# SONIFEX

AoIP – Dante<sup>®</sup>/AES67 Audio over IP Products

# Catalogue





# AoIP – Dante®/AES67 Audio over IP Products

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## **AVN-CU2-DANTE Configurable Dante Commentary Unit for 2 Commentators**



















**Product Function:** Commentator unit for two commentary positions with 16 x 16 Dante matrix mixers.

Typical Applications: Used in sports grounds and arenas by commentators to produce commentary & hear talkback, 16 x 16 matrix mixer with configurable controls for live events.

#### Features:

- · Web GUI for configuration.
- · Full web GUI remote control.
- · Lock-out of all individual controls and/or button row/encoder pages.
- PoE using 2 x Neutrik EtherCON<sup>®</sup> RJ45s.
- Dual redundant network ports on RJ45 and SFP.
- 2 x locking mic/line inputs with +48V phantom power indication.
- 2 x headphone outputs on locking 6.35mm jack sockets.
- 1 x stereo analogue/digital input and 1 x stereo analogue/digital output on RJ45s.

- 4 x talkback buses.
- · On-air buttons with lazy talkback ability.
- Supports up to 16 x input and output streams and 16 input and output multicast flows/streams.
- · Configurable headphone source selection.
- 6 x rotary encoders to control volume, or signal panning, and colour coded with the screen.
- 6 x detailed colour displays showing channel name, level metering & amp; limit indication
- Fully configurable buttons and rotary encoders, for inputs and outputs.
- 4 x pages of 6 x encoder settings, e.g. one for talkback inputs, one for outputs, one to monitor other sources.
- Metering per input/output, with output configured as pre or post level adjustment.
- Built-in line-up tone generator.
- · Limiter available on every output.
- · Output metering section shows limiter indication, the name of the output & level metering.
- 10 x configurable GPIO with 1 x switched changeover output.
- GPIO & VGPIO matrix programmable with events.

The AVN-CU2-DANTE takes a new approach to provide a multi-purpose configurable tool for commentary teams. Its power lies in the impressive mix engine which overlays the usual Dante&reg: Controller settings. Once Dante&reg: flows have been made, inputs and outputs can be mixed freely to AoIP or physical inputs and outputs, controlled using the programmable buttons and rotary encoders, which control the gain and pan of inputs. outputs or cross-points, allowing total flexibility for different situations.

It can be used in any number of different commentary situations, controlled manually or remotely and controls can be 'locked down' so that they can't be tampered with or altered, to guarantee reliable operation. Housed in a rugged and intuitive, userfriendly package, it's a truly different way of looking at how commentary units should operate.

Traditional commentary units have fixed analogue and digital I/O and fixed controls in fixed positions on the unit. Their inputs and outputs are defined at hardware design and are thus limited by that initial design, including limited routing, mixing and DSP of the audio pathways.

We've taken a different approach with the AVN-CU2-DANTE. It was designed from the ground up to be totally flexible in operation and the use of Dante AoIP means that inputs and outputs can be added as required (up to a max of 16 per unit). Because any physical analogue or digital input can be mixed and routed with any



Dante AoIP input to any physical output or Dante output, you can define your own audio pathways. Additionally you can choose which of those pathways need to be controlled (volume and pan) by the use of rotary encoders.

Using a built-in web GUI, up to 4 nameable pages of 6 rotary encoders (24 in total) can be placed on the mix matrix at inputs, outputs or cross-points. Encoders can control the volume and pan of headphones and volume of inputs, outputs or cross-points. Each rotary encoder has a separate colour-coded meter section showing the channel name, detailed level metering and left/right panning on a bright daylight reading display. Colours can be programmed per encoder to quickly identify particular source groups, so headphone source selection becomes intuitive.

The unit can be fully remotely controlled from the web interface, which mimics the front panel of the unit and includes live metering. All the controls can be operated as if at the unit and there are front-panel lock-out options for every button and encoder, with 'quick lock-out' for the rows of buttons and pages of encoders. This allows you to configure the unit for simple operation by the talent but with more complex setups, say, for mixing the talkback or feeds from an OB truck.

The AVN-CU2-DANTE provides two mic/line inputs with a wide, adjustable gain range and has two stereo headphone outputs with lockable jack sockets, suitable for operation by two commentators.

It's powered using Power over Ethernet (PoE), using Neutrik EtherCON® connectors, with primary and secondary ports for power and data redundancy. There's an additional 4 pin XLR 12V DC input. The unit supports up to 16 input and

output AoIP channels and up to 16 simultaneous input and output AoIP streams.

The 6 push-button rotary encoders control input and output levels and panning. The 12 key-cap buttons are fully configurable for any button function and can have different colours for the on and off functions.

Metering is available per input/output, with output metering configurable as pre or post level adjustment. The top of the display shows output metering, a limiter indication and the name of the output. An adjustable limiter is available on every output and is applied automatically to prevent signal clipping.

The unit has 2 x locking mic/line inputs with +48V phantom power indication and 2 x headphone outputs on locking 6.35mm jack sockets.

Fourwire I/O on rear panel RJ45 connectors provide an AES3 or analogue input and output that can be assigned as mic outputs (line level), talkback outputs, programme inputs or talkback inputs as desired. In addition, the AES/analogue connections can be used as an insert or exit point into/out from the AoIP network.

The unit has dual redundant network ports on both RJ45 (PoE using 2 x Neutrik EtherCON® connectors) and SFP cages for long fibre runs.

There are 10 x configurable GPIO on a 15 way D-type connector with 1 x switched changeover output. GPIO & VGPIO (virtual

GPIO, i.e. network commands) can be configured on a matrix to visually show actions, combined with button presses and event triggers, such as input/output muting, which enables some automation. Button presses can control buttons and functions on other network connected AVN-CU units using VGPIO.

The complete configuration of the unit, including all encoder settings, button setups and Dante mix matrix audio routes, can be saved off to a text editable .json file. Up to 8 of these files can be stored on the unit and loaded from the main Menu, meaning that the unit can be pre-configured for any venue, or event and with these setups instantly recalled onsite.

"All of the buttons have key-cap text and can be configured for any button in any position. There are some standard operations available:

- 2 x On-Air buttons can be used to connect mic audio to the main output, either over AoIP or via the AES digital audio connection. The On-Air buttons can be locked if required.
- A Menu button can be used to access limited setup options on the TFT display.
- 2 x Page buttons change the display and encoders to monitor an additional set of sources, mix points or outputs. Up to 4 pages can be pre-programmed, e.g. one for talkback inputs, one for outputs, one to monitor other sources.



- 2 x Cough buttons take the commentator off-air while pressed.
- A User button can be programmed to perform any function, using the web server.
- 4 x T/B (talkback) buttons can be configured to initiate talkback over AoIP or AES digital audio connection, using 4 x talkback buses. The talkback buttons operate with lazy talkback, taking the commentator off-air when invoked.
- 6 x rotary encoders.
- 4 x pages of 6 rotary encoder positions (24 in total).

The illuminated 'Sonifex' logo acts as a power indication and there are illuminated LEDs to indicate network clock status, AoIP Primary and AoIP Secondary link status, PoE Primary, PoE Secondary and DC power active.

#### **Technical Specification For AVN-CU2-DANTE**

| Network and AoIP                      |   |
|---------------------------------------|---|
| AoIP Standard:                        | Dante   |
| Number of Channels:                   | 16 receive, 16 transmit   |
| Number of Streams:                    | 16 receive, 16 transmit   |
| Sample Rate:                          | 48 kHz  |
| Format:                               | Linear PCM, 16 or 24 bit  |
| AES67 Support:                        | Yes   |
| Connectivity:                         | 2 x RJ45 and 2 x SFP  |
| Speed:                                | 1Gbps and 100Mbps   |
| Network Modes:                        | Switched or redundant   |
| Dante Domain                          |   |
| Manager Ready:                        | Yes   |
| Mic/Line Inputs                       |   |
| Input Impedance                       |   |
| (Mic Mode):                           | 2.5kΩ electronically balanced   |
| Input Impedance                       | > 10kO alastraniaslių balance -                                       |
| (Line Mode):                          | >10kΩ electronically balanced   |
| Preamp Gain<br>(Mic Mode):            | User selectable 16dB to 76dB in 3dB                                   |
| (IVIIC IVIOUC).                       | steps (Ref. +18dBu = 0dBFS)   |
| OdBFS Line-up                         |   |
| (Mic Mode):                           | -58dBu @ Max preamp gain, +2dBu                                       |
|                                       | @ Min preamp gain   |
| OdBFS Line-up                         |   |
| (Line Mode):                          | User selectable +15dBu / +18dBu /<br>+20dBu / +22dBu / +24dBu = 0dBFS |
| Frequency Response                    | 120dBd / 122dBd / 124dBd = 0dBl 3                                     |
| (Mic Mode):                           | +0/-0.2dB 20Hz to 20kHz Ref 40dB                                      |
|                                       | gain @ 1kHz   |
| Frequency Response                    |   |
| (Line Mode):                          | +0/-0.2dB 20Hz to 20kHz Ref 0dBu<br>@ 1kHz                            |
| THD+N (Mic Mode):                     | <-90dBFS, -64dBFS, 20Hz – 20kHz,                                      |
|                                       | 40dB gain, 20kHz BW   |
| THD+N (Line Mode):                    | <-98dBFS, -30dBFS, 20Hz – 20kHz,                                      |
| Facility I and I am a 40 °C           | 20kHz BW  |
| Equivalent Input Noise<br>(Mic Mode): | 125dB Ref. 76dB preamp gain, Rs =                                     |
| (IVIIC IVIOUC).                       | 200Ω  |
| Noise (Line Mode):                    | <-100dBFS, 20kHz BW, Rs = 200Ω  |
| C.M.R.R.                              | >60dB @ 1kHz  |
| Phantom Power                         |   |
| (Mic Mode):                           | +48V  |
| High Pass Filter:                     | 12dB/octave, user selectable  |
|                                       | frequency   |
| Stereo Analogue/Digi                  | tal Input   |
| Input Impedance                       | •   |
| (Analogue Mode):                      | >20kΩ electronically balanced   |
| Input Impedance                       | -   |
| (Digital Mode):                       | 110Ω  |
| Supported Input Rates                 | 375H-7 44 15H-7 485H-7 88 75H-7                                       |
|                                       |   |

32kHz, 44.1kHz, 48kHz, 88.2kHz,

96kHz, 176.4kHz and 192kHz

| NTE                                       |   |
|---|---|
| OdBFS Line-up<br>(Analogue Mode):         | User selectable +15dBu / +18dBu /<br>+20dBu / +22dBu / +24dBu =<br>0dBFS  |
| Frequency Response (Analogue Mode):       | +0/-0.2dB 20Hz to 20kHz Ref 0dBu<br>@ 1kHz                                |
| THD+N<br>(Analogue Mode):                 | <-107dBFS, -30dBFS, 20Hz – 20kHz, 20kHz BW                                |
| Noise<br>(Analogue Mode):                 | <-108dBFS, 20kHz BW, Rs = 200Ω  |
| C.M.R.R<br>(Analogue Mode):               | >60dB @ 1kHz  |
| Stereo Analogue/Digi                      | tal Output  |
| Output Impedance (Analogue Mode):         | <50Ω  |
| Output Impedance<br>(Digital Mode):       | 110Ω  |
| Supported Output<br>Rates (Digital Mode): | 48kHz   |
| OdBFS Line-up<br>(Analogue Mode):         | User selectable 0dBFS = +15dBu /<br>+18dBu / +20dBu / +22dBu /<br>+24dBu  |
| Frequency Response<br>(Analogue Mode):    | +0/-0.5dB 20Hz to 20kHz Ref 0dBu<br>@ 1kHz                                |
| THD+N<br>(Analogue Mode):                 | <-107dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW                             |
| Noise<br>(Analogue Mode):                 | <-107dBFS, 20kHz BW   |
| <b>Headphone Outputs</b> Drive Capability | Drives 150mW into $32\Omega$ to $600\Omega$ headphones                    |
| OdBFS Line-up:                            | Fixed OdBFS = +18dBu  |
| Frequency Response:                       | +0/-0.2dB 20Hz to 20kHz Ref 0dBu<br>@ 1kHz                                |
| THD+N:                                    | <-107dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW                             |
| Noise:                                    | <-110dBFS, 20kHz BW   |
| Main Panel Operation<br>Power LED         | nal Controls & Indicators  Sonifex logo illuminates when power is present |
| D la la                                   | 42 - Illinoite and broken a short and                                     |

12 x Illuminated buttons that can be assigned to any of the following functions, and configured: On-Air / Talkback / Page # / Page Cycle / GPO / Cough / Menu / Brightness

480 x 107 pixel colour TFT display

metering, control type, source/ mix/destination names and main

showing pan status, level,

output metering

Pushbuttons:

Display:

| Rotary Encoders:                    | PSU status LEDs to monitor power<br>and connectivity. Can be disabled<br>if required  |  |  |
|-------------------------------------|---|--|--|
| Rotary Encoders:                    | 6 x robust soft touch rotary<br>encoders with adjacent<br>configurable light bars for group/<br>source/commentator grouping and<br>identification |  |  |
| Front Panel Connection              | ons   |  |  |
| Headphone Outputs:                  | 2 x Neutrik ¼" (6.35mm) locking jack sockets  |  |  |
| Microphone/Line<br>Inputs:          | 2 x Neutrik 3 pin latching XLR sockets  |  |  |
| Rear Panel Connectio                | ns  |  |  |
| Stereo Analogue/<br>Digital Input:  | RJ45 socket (electronically balanced)   |  |  |
| Stereo Analogue/<br>Digital Output: | RJ45 socket (electronically balanced)   |  |  |
| GPIO Port:                          | 15-way 'D'-type socket  |  |  |
| DC Input:                           | Neutrik 4 pin XLR plug, 12VDC,<br>Maximum 12W   |  |  |
| Network:                            | 2 x 1Gbps Neutrik EtherCON<br>receptacles, with PoE, maximum<br>12W 2 x SFP slots for alternative<br>interfaces                                   |  |  |
| PoE Power                           |   |  |  |
| Standard:                           | 802.3af   |  |  |
| Class:                              | 0   |  |  |
| PD Power Range:                     | 0.44 W to 12.94 W   |  |  |
|                                     |   |  |  |

Clock / AoIP status / PoE status and

Status LEDs:



AVN-CU2-DANTE Front View.

| Typical PSE Power                          |  |
|--|--|
| Usage:                                     | 13W  |
| Max PSE Power Usage                        | : 15.4W  |
| Equipment Type                             |  |
| AVN-CU2-DANTE                              | Dante® Commentator Unit,<br>2 Commentators                                       |
| Physical Specification                     |  |
| Dimensions (Raw):                          | 13.5cm (W) x 23cm (D) x 9.0cm (H - rear<br>6.6cm (H - Front) 9.3cm (H - Maximum) |
|  | 5.3" (W) x 9" (D) x 3.5" (H - front) x 2.6" (H - rear) 3.7" (H - Maximum)        |
| Dimensions (Boxed):                        | 30.5cm (W) x 21.5cm (D) x 16.3cm (H)   |
|  | 12" (W) x 8.5" (D) x 6.4" (H)  |
| Weight:                                    | Nett: 1.8kg Gross: 2.1kg<br>Nett: 3.96lbs Gross: 4.62lbs                         |
|  |  |
| Accessories                                | BOD 6 16 NWB 4:  |
| AVN-DCX60                                  | DC Power Supply for AVN Range, 4 pin<br>XLR socket, 60W                          |
| AVN-TC3 (Case Only)                        | Transport Case, AVN-CU2-DANTE &<br>Power Supply Cut-out                          |
| AVN-TC3 (Case Only)                        | 455 (14) 25 (2) 47.5 (11)  |
| Dimensions (Raw):                          | 46.5cm (W) x 36cm (D) x 17.6cm (H)   |
| AVN-TC3 (Case Only)<br>Dimensions (Raw):   | 18.3" (W) x 14.2" (D) x 6.9" (H)   |
| AVN-TC3 (Case Only)<br>Dimensions (Boxed): | 52cm (W) x 21.5cm (D) x 43cm (H)   |
| AVN-TC3 (Case Only)<br>Dimensions (Boxed): | 20.5" (W) x 8.5" (D) x 16.9" (H)   |
| AVN-TC3 (Case Only)<br>Weight:             | Net: 3.0kg Gross: 3.5kg  |
| AVN-TC3 (Case Only)<br>Weight:             | Net:6.6lbs Gross: 7.7lbs   |
| CM-GM2:                                    | Professional Gooseneck Condenser   |



Microphone

AVN-CU2-DANTE Rear View.

(Digital Mode):





# **AVN-CU4-DANTE Configurable Dante Commentary Unit for 4 Commentators**















Category: Dante Commentary.

**Product Function:** Commentator unit for four commentary positions with dual 16 x 16 Dante matrix mixers.

Typical Applications: Used in sports grounds and arenas by commentators to produce commentary & hear talkback, 16 x 16 matrix mixer with configurable controls for live events.

#### Features:

- · Web GUI for configuration.
- Full web GUI remote control.
- Lock-out of all individual controls and/or button row/encoder pages.
- PoE using 2 x Neutrik EtherCON<sup>®</sup> RJ45s.
- Dual redundant network ports on RJ45 and SFP.
- 4 x locking mic/line inputs with +48V phantom power indication.
- 4 x headphone outputs on locking 6.35mm jack sockets.
- 4 x analogue inputs and 6 analogue outputs on XLR.
- 1 x stereo AES/EBU input and output on RJ45s.

- 4 x talkback buses.
- On-air buttons with lazy talkback ability.
- Supports up to 16 x input and output streams and 16 input and output multicast flows/streams.
- Configurable headphone source selection.
- 12 x rotary encoders to control volume, or signal panning, and colour coded with the screen.
- 12 x detailed colour displays showing channel name, level metering & Dimit indication.
- Fully configurable buttons and rotary encoders, for inputs and outputs.
- 4 x pages of encoder settings, e.g. one for talkback inputs, one for outputs, one to monitor other sources.
- Metering per input/output, with output configured as pre or post level adjustment.
- Built-in line-up tone generator.
- Limiter available on every output.
- Output metering section shows limiter indication, the name of the output & level metering.
- 10 x configurable GPIO with 1 x switched changeover output.
- GPIO & VGPIO matrix programmable with events.

The AVN-CU4-DANTE is a configurable portable commentator unit using Dante® AoIP. It is a dual version of the AVN-CU2-DANTE providing four mic/line inputs with a wide, adjustable gain range and four stereo headphone outputs with lockable jack sockets, suitable for operation by three or four commentators.

Fully featured, this unit allows you to handle virtually any commentary situation with both Dante AoIP and 4 wire connections,

dual redundant Dante networking and multiple AC/DC/PoE power options. Up to 48 rotary encoders can be used on inputs,



outputs or cross-points, allowing talkback feeds, commentary and audio mixing to be handled in one unit

It's been designed from the ground up to be configurable for different situations with config files that can be saved and recalled for quick setup. Additionally, buttons and rotary encoders can be locked out from use, if the unit is being operated by less-technical commentators.

It has an abundance of 4 wire connections on the rear panel: 4 x analogue line inputs on XLR sockets with latching locks, 6 x analogue line outputs on XLR plugs and an RJ45 AES3 stereo input & output. These connections can act as a simultaneous analogue backup to the Dante AoIP connections.

#### Comparison with AVN-CU2-DANTE

The featureset is as per the AVN-CU2-DANTE, with the following differences. There are two displays with a doubling of operational controls:

- 4 x On-air buttons.
- 4 x Page buttons, 2 for each half of the display.
- 4 x Cough buttons.
- 8 x Talkback buttons, up to 4 for each user.
- 3 x User buttons.
- 12 x rotary encoders.
- 8 x pages of 6 rotary encoder positions (48 in total).

Similar to the AVN-CU2-DANTE, the illuminated 'Sonifex' logo acts as a power indication and illuminated LEDs indicate network clock status, AoIP Primary and AoIP Secondary link status, PoE Primary, PoE Secondary and AC power active.

The front panel houses 4 x locking mic/line inputs with +48V phantom power indication and 4 x headphone outputs on locking 6.35mm jack sockets.

The unit has dual redundant network ports on both RJ45 (PoE+ using 2 x Neutrik EtherCON® connectors) and SFP cages.

To power the unit, as well as the dual PoE ports and 12V 4 pin XLR DC input, there is an AC mains input on an IEC inlet, with a universal supply.

#### **Full Description**

The unit supports up to 16 input and output AoIP channels and up to 16 simultaneous input and output AoIP streams. Stream setup to and from the unit is initially via Dante&reg: Controller with more detailed configuration performed using the built-in web GUI. The power of this unit lies in the impressive mix engine which overlays the usual Dante® Controller settings. Once Dante® flows have been made, inputs and outputs can be mixed freely to AoIP or physical inputs and outputs, controlled using the programmable buttons and rotary encoders, which control the gain and pan of inputs, outputs or cross-points, allowing total flexibility for different situations.

It can be used in any number of different commentary situations, controlled manually or remotely and controls can be 'locked down' so that they can't be tampered with or altered, to guarantee reliable operation. Housed in a rugged and intuitive, user-friendly package, it's a truly different way of looking at how commentary units should operate.

Traditional commentary units have fixed analogue and digital I/O and fixed controls in fixed positions on the unit. Their inputs and outputs are defined at hardware design and are thus limited by that initial design, including limited routing, mixing and DSP of the audio pathways.

We've taken a different approach with the AVN-CU4-DANTE. It was designed from the ground up to be totally flexible in operation and the use of Dante AoIP means that inputs and outputs can be added as required (up to a max of 16 per unit). Because any physical analogue or digital input can be mixed and routed with any





AVN-CU4-DANTE Front View.

AVN-CU4-DANTE Rear View.

Dante AoIP input to any physical output or Dante output, you can define your own audio pathways. Additionally you can choose which of those pathways need to be controlled (volume and pan) by the use of rotary encoders.

Using a built-in web GUI, up to 8 nameable pages of 6 rotary encoders (48 in total) can be placed on the mix matrix at inputs, outputs or cross-points. Encoders can control the volume and pan of headphones and volume of inputs, outputs or cross-points. Each rotary encoder has a separate colour-coded meter section showing the channel name, detailed level metering and left/right panning on a bright daylight reading display. Colours can be programmed per encoder to quickly identify particular source groups, so headphone source selection becomes intuitive.

The unit can be fully remotely controlled from the web interface, which mimics the front panel of the unit and includes live metering. All the controls can be operated as if at the unit and there are front-panel lock-out options for every button and encoder, with 'quick lock-out' for the rows of buttons and pages of encoders. This allows you to configure the unit for simple operation by the talent but with more complex setups, say, for mixing the talkback or feeds from an OB truck.

The AVN-CU4-DANTE provides four locking mic/line inputs, each with +48V phantom power indication and a wide, adjustable gain range. It has four stereo headphone outputs with locking 6.35mm jack sockets, suitable for operation by four commentators.

It's powered using Power over Ethernet (PoE), using Neutrik EtherCON® connectors, with primary and secondary ports for power and data redundancy. There's an additional 4 pin XLR 12V DC input and an AC mains input on an IEC inlet, with a universal supply. The unit supports up to 16 input and output AoIP channels and up to 16 simultaneous input and output AoIP streams.

The 12 x push-button rotary encoders and 24 x key-cap buttons are fully configurable, to control input & output levels and panning. The 12 key-cap buttons can have different colours for the on and off functions

Metering is available per input/output, with output metering configurable as pre or post level adjustment. The top of the display shows output metering, a limiter indication and the name of the output. An adjustable limiter is available on every output and is applied automatically to prevent signal clipping.

The unit has 4 x locking mic/line inputs with +48V phantom power indication and 4 x headphone outputs on locking 6.35mm jack sockets.

Four wire connections on the rear panel provide a number of ways of linking to the outside world. There are:

- 4 x analogue line inputs on XLR sockets with latching locks
- 6 x analogue line outputs on XLR plugs
- 1 x RJ45 AES3 stereo input

• 1 x RJ45 AES3 stereo output

The I/O can be assigned as programme inputs, talkback inputs, mic outputs (line level), PA outputs, zone outputs or talkback outputs, as desired. In addition, the AES/ analogue connections can be used as an insert or exit point into/out from the AoIP network.

The unit has dual redundant network ports on both RJ45 (PoE using 2 x Neutrik EtherCON® connectors) and SFP cages for long fibre runs.

There are 10 x configurable GPIO on a 15 way D-type connector with 1 x switched changeover output. GPIO & VGPIO (virtual GPIO, i.e. network commands) can be configured on a matrix to visually show actions, combined with button presses and event triggers, such as input/output muting, which enables some automation. Button presses can control buttons and functions on other network connected AVN-CU units using VGPIO.

The complete configuration of the unit, including all encoder settings, button setups and Dante mix matrix audio routes, can be saved off to a text editable .json file. Up to 8 of these files can be stored on the unit and loaded from the main Menu, meaning that the unit can be pre-configured for any venue, or event and with these setups instantly recalled onsite.

All of the buttons have key-cap text and can be configured for any button in any position. There are some standard operations available:

- 4 x On-Air buttons can be used to connect mic audio to the main output, either over AoIP or via the AES digital audio connection. The On-Air buttons can be locked if required.
- A Menu button can be used to access limited setup options on the TFT display.
- Page buttons change the display and encoders to monitor an additional set of sources, mix points or outputs. Up to 8 pages can be pre-programmed, e.g. one for talkback inputs, one for outputs, one to monitor other sources.
- 4 x Cough buttons take the commentator off-air while pressed.
- 3 x User buttons can be programmed to perform any function, using the web server.
- 8 x T/B (talkback) buttons can be configured to initiate talkback over AoIP, analogue or AES digital audio connections, using 4 x talkback buses.
   The talkback buttons operate with lazy talkback, taking the commentator off-air when invoked.
- 12 x rotary encoders.
- 8 x pages of 6 rotary encoder positions (48 in total).

The illuminated 'Sonifex' logo acts as a power indication and there are illuminated LEDs to indicate network clock status, AoIP Primary and AoIP Secondary link status, PoE Primary, PoE Secondary and AC/DC power active.

#### **Technical Specification For AVN-CU4-DANTE**

| Network and AoIP                      |   |
|---------------------------------------|---|
| AoIP Standard:                        | Dante   |
| Number of Channels:                   | 16 receive, 16 transmit   |
| Number of Streams:                    | 16 receive, 16 transmit   |
| Sample Rate:                          | 48 kHz  |
| Format:                               | Linear PCM, 16 or 24 bit  |
| AES67 Support:                        | Yes   |
| Connectivity:                         | 2 x RJ45 and 2 x SFP  |
| Speed:                                | 1Gbps and 100Mbps   |
| Network Modes:                        | Switched or redundant   |
| Dante Domain<br>Manager Ready:        | Yes   |
| Mic/Line Inputs                       |   |
| Input Impedance<br>(Mic Mode):        | 2.5kΩ electronically balanced   |
| Input Impedance<br>(Line Mode):       | >10kΩ electronically balanced   |
| Preamp Gain<br>(Mic Mode):            | User selectable 0dB to 60dB in 3dB steps                              |
| OdBFS Line-up<br>(Mic Mode):          | -58dBu @ Max preamp gain, +2dBu<br>@ Min preamp gain                  |
| OdBFS Line-up<br>(Line Mode):         | User selectable +15dBu / +18dBu /<br>+20dBu / +22dBu / +24dBu = 0dBFS |
| Frequency Response<br>(Mic Mode):     | +0/-0.2dB 20Hz to 22kHz   |
| Frequency Response (Mic Mode):        | Ref 40dB gain @ 1kHz  |
| Frequency Response (Line Mode):       | +0/-0.2dB 20Hz to 22kHz   |
| Frequency Response (Line Mode):       | Ref OdBu @ 1kHz   |
| THD+N (Mic Mode):                     | <-90dBFS, -64dBFS, 20Hz – 20kHz,<br>40dB gain, 20kHz BW               |
| THD+N (Line Mode):                    | <-98dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW                          |
| Equivalent Input<br>Noise (Mic Mode): | 127dB Ref. 60dB preamp gain, Rs = 200Ω                                |
| Noise (Line Mode):                    | <-110dBFS, 20kHz BW, Rs = 200Ω  |
| C.M.R.R.                              | >60dB   |
| Stereo Digital Input                  |   |
| Input Impedance:                      | 110Ω  |
|                                       | 32kHz, 44.1kHz, 48kHz, 88.2kHz,<br>96kHz, 176.4kHz and 192kHz         |
| Output Impedance:                     | 110Ω  |
| Supported output rates:               | 48kHz   |
| Analogue Line Inputs                  |   |
| Input Impedance:                      | >20kΩ electronically balanced   |
| OdBFS Line-up:                        | User selectable +15dBu / +18dBu /<br>+20dBu / +22dBu / +24dBu = 0dBFS |
| Frequency Response:                   | +0/-0.2dB 20Hz to 22kHz   |
| Frequency Response:                   | Ref OdBu @ 1kHz   |

| NTE<br>THD+N:          | * 1104DEC 204DEC 2011= 2011=   |
|------------------------|--|
|                        | <-110dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW  |
| Noise:                 | <-110dBFS, 20kHz BW, Rs = 200Ω   |
| C.M.R.R.               | >60dB  |
| Analogue Line Output   | ts   |
| Output Impedance:      | <50Ω   |
| OdBFS Line-up:         | User selectable 0dBFS = +15dBu /<br>+18dBu / +20dBu / +22dBu /<br>+24dBu   |
| Frequency Response:    | +0/-0.5dB 20Hz to 22kHz  |
| Frequency Response:    | Ref OdBu @ 1kHz  |
| THD+N:                 | <-110dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW  |
| Noise:                 | <-110dBFS, 20kHz BW  |
| Headphone Outputs      |  |
| Drive Capability       | Drives 150mW into $32\Omega$ to $600\Omega$ Headphones   |
| OdBFS Line-up:         | Fixed OdBFS = +18dBu   |
| Frequency Response:    | +0/-0.2dB 20Hz to 22kHz  |
| Frequency Response:    | Ref OdBu @ 1kHz  |
| THD+N:                 | <-108dBFS, -30dBFS, 20Hz – 20kHz,<br>20kHz BW  |
| Noise:                 | <-110dBFS, 20kHz BW  |
| Main Panel Operation   | al Controls & Indicators   |
| Power LED              | Sonifex logo illuminates when  |
|                        | power is present   |
| Pushbuttons:           | 24 x Illuminated buttons that can be assigned to any of the following functions, and configured:   |
| Pushbuttons:           | On-Air / Talkback / Page # / Page<br>Cycle / GPO / Cough / Menu /<br>Brightness  |
| Display:               | 2 x 480 x 107 pixel colour TFT   |
|                        | display showing pan status, level,<br>metering, control type, source/mix/<br>destination names and main output<br>metering                         |
| Status LEDs:           | Clock / AoIP status / PoE status and<br>PSU status LEDs to monitor power<br>and connectivity. Can be disabled if<br>required                       |
| Rotary Encoders:       | 12 x robust soft touch rotary<br>encoders with adjacent configurable<br>light bars for group/source/<br>commentator grouping and<br>identification |
| Front Panel Connection | nne  |
| Headphone Outputs:     | 4 x Neutrik ¼" (6.35mm) locking jack sockets   |
| Microphone/Line        | SOCKELS  |
| Inputs:                | 4 x Neutrik 3 pin latching XLR sockets   |
| Rear Panel Connection  | ns   |
| Stereo Digital Input:  | RJ45 socket (electronically balanced   |
|                        |  |

Stereo Digital Output: RJ45 socket (electronically balanced)

| Analogue Line Inputs:     | 4 x Neutrik locking XLR sockets                                     |
|---------------------------|---|
| Analogue Line<br>Outputs: | 6 x Neutrik locking XLR plugs                                       |
| GPIO Port:                | 15-way 'D'-type socket  |
| Network:                  | 2 x 1Gbps Neutrik EtherCON<br>receptacles, with PoE, Maximum<br>12W |
| Network:                  | 2 x SFP slots for alternative interfaces                            |
| DC Input:                 | Neutrik 4 pin XLR plug, 12VDC,<br>Maximum 12W                       |
| Mains Input:              | Filtered IEC, 85 – 264VAC, 47-63 Hz, 60W max                        |
| Fuse Rating:              | Anti-surge fuse 2A 20 x 5mm   |
| PoE Power                 |   |
| Standard:                 | 802.3at Type 2  |
| Class:                    | 4   |
| PD Power Range:           | 12.95 W to 25.50 W  |
| Typical PSE Power         |   |
| Usage:                    | 20W   |
| Max PSE Power Usage       | : 30W   |
| Equipment Type            |   |
| AVN-CU4-DANTE             | AVN Dante commentator unit.   |

4 commentators

| 25.7cm (W) x 22.9cm (D) x 9.2cm<br>(H - max), 6.5cm (H - front), 8.9cm<br>(H - rear) |
|--|
| 10.1" (W) x 9" (D) x 3.6" (H - max),<br>2.6" (H - front), 3.5" (H - rear)            |
| 37cm (W) x 35.5cm (D) x 19cm (H)   |
| 14.6" (W) x 14" (D) x 7.5" (H)   |
|  |
| DC Power Supply for AVN Range,<br>4 pin XLR socket, 60W                              |
| Transport Case, AVN-CU2-DANTE & Power Supply Cut-out                                 |
| 46.5cm (W) x 36cm (D) x 17.6cm (H)   |
| 18.3" (W) x 14.2" (D) x 6.9" (H)   |
| 52cm (W) x 21.5cm (D) x 43cm (H)   |
| 20.5" (W) x 8.5" (D) x 16.9" (H)   |
| Net: 3.0kg Gross: 3.5kg  |
| Net:6.6lbs Gross: 7.7lbs   |
| Professional Gooseneck Condenser<br>Microphone                                       |
|  |

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# AVN-AIO4 4 Input, 4 Output Dante® Interface, PoE



Category: Dante Audio Interfaces.

**Product Function:** Converts up to four analogue inputs and four analogue outputs to and from the Dante Audioover-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 4 x balanced analogue inputs on XLR.
- 4 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 1 x RJ45 Dante connector (100Mb Ethernet Port).
- PoE, Link, and Clock LED status indicators.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE.
- 1U 19" rack-mount form factor.



The AVN-AIO4 audio converter and interface converts up to four analogue inputs and four analogue outputs to and from the Dante Audio-over-IP networking standard. This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue inputs and outputs are on highquality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE, Link, and Clock, Global OdBFS line-up can be set to +12dBu, +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 4 x balanced analogue inputs on XLR.
- 4 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).

- 1 x RJ45 Dante connector (100Mb Ethernet Port).
- · PoE, Link, and Clock LED status indicators.
- Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

#### **Technical Specification For AVN-AIO4** Network and AoIP

Dante

AoIP Standard:

| Number of Channels:   | 4 receive, 4 transmit             |  |
|---|-----------------------------------|--|
| Number of Streams:  | 4 receive, 4 transmit             |  |
| Sample Rates:   | 44.1kHz, 48 kHz, 88.2kHz or 96kHz |  |
| (Note: 2 channels I/O only when >48kHz. Either 4 inputs or 4 outputs at 96kHz, can be configured on request with a separate firmware download). |                                   |  |
| Format:   | Linear PCM, 16, 24 or 32 bit      |  |
| AES67 Support:  | Yes                               |  |
| Connectivity:   | 1 x RJ45                          |  |
| Speed:  | 100Mbps                           |  |
| Dante Domain<br>Manager Ready:  | Yes                               |  |

| <b>Audio Input Performa</b> | nce Specifications                                |
|-----------------------------|---|
| Parameter                   | Line Input  |
| Input Impedance:            | >20kΩ balanced                                    |
| OdBFS line-up:              | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB                          |
| THD+N:                      | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |
| Noise:                      | -110dBFS, 20kHz BW, Rs=200Ω                       |
| Crosstalk:                  | <-100 dB  |
| Common Mode                 |   |
| Rejection:                  | >60dB @ 1kHz                                      |

#### Multi-Channel Dante® Audio Interfaces

| Audio Output Performance Specifications | ŝ |
|---|---|
|---|---|

| Addio Odtput i cironnance Specinications |   |
|--|---|
| Parameter                                | Line Output                                       |
| Output Impedance:                        | <50Ω balanced                                     |
| OdBFS line-up:                           | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |
| Frequency Response:                      | 20Hz to 20kHz, +0/-0.5dB                          |
| THD+N:                                   | <-107dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |
| Noise (Mic=EIN):                         | -107dBFS, 20kHz BW                                |
| Crosstalk:                               | <100dB  |
|  |   |

#### Balanced Line Input XLR Pinout

| Pin | Func | Function                 |  |
|-----|------|--------------------------|--|
|     | 1    | Chassis Ground/Screen    |  |
|     | 2    | Input Phase/Positive     |  |
|     | 3    | Input Non-Phase/Negative |  |
|     |      |                          |  |

#### **Balanced Line Output XLR Pinout**

| Pin | Function                  |
|-----|---------------------------|
| 1   | Chassis Ground/Screen     |
| 2   | Output Phase/Positive     |
| 3   | Output Non-Phase/Negative |
|     |                           |

| PoE Power                     |                   |  |
|-------------------------------|-------------------|--|
| Standard                      | 802.3af           |  |
| Class                         | 0                 |  |
| PD Power Range                | 0.44 W to 12.94 W |  |
| Typical PSE Power Usage 4.5 W |                   |  |
| Max PSE Power Usage 15.4 W    |                   |  |
|                               |                   |  |

#### Equipment Type

| Edarbinent . Abc |  |
|------------------|--|
| AVN-AIO4:        | 4 Input, 4 Output Dante® Interface,<br>PoE |
|                  |  |

#### Physical Specification

| Dimensions (Raw):<br>(H)(1U) | 48cm (W) x 11cm (D) x 4.3cm   |
|------------------------------|-------------------------------|
| (11)(10)                     | 19" (W) x 4.3" (D) x 1.7" (H) |
| Weight:                      | Nett: 0.9kg Gross: 1.4kg      |
|                              | Nett: 1.9lbs Gross: 3.1lbs    |



AVN-AIO4 Front View.



AVN-AIO4 Rear View.











# AVN-AIO8 8 Input, 8 Output Dante® Interface, PoE

















**Product Function:** Converts up to eight analogue inputs and eight analogue outputs to and from the Dante Audioover-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 8 x balanced analogue inputs on XLR.
- 8 x balanced analogue outputs on XLR.
- 1 x RJ45 Dante connector (1Gb/s Ethernet Port).
- . POE, Link, and Clock LED status indicators.
- Configuration using Dante Controller.
- AES67 & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

The AVN-AIO8 audio converter and interface converts up to eight analogue inputs and eight analogue outputs to and from the Dante Audio-over-IP networking standard. This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue inputs and outputs are on highquality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE. Link, and Clock, Global OdBFS line-up can be set to +12dBu. +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 8 x balanced analogue inputs on XLR.
- 8 x balanced analogue outputs on XLR.
- 1 x RJ45 Dante connector (1Gb/s Ethernet Port).

- . POE, Link, and Clock LED status indicators.
- · Configuration using Dante Controller.
- AES67 & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

(The AVN-AIO8R is available with dual redundant Ethernet ports).



#### **Technical Specification For AVN-AIO8**

| Dante  |
|--|
| 8 receive, 8 transmit                                  |
| 8 receive, 8 transmit                                  |
| 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |
| Linear PCM, 16, 24 or 32 bit                           |
| Yes  |
| 1 x RJ45   |
| 1Gbps or 100Mbps                                       |
| Yes  |
|  |

| Manager Ready:      | Yes   |
|---------------------|---|
| Audio Input Perforn | nance Specifications                              |
| Parameter           | Line Input  |
| Input Impedance:    | >20kΩ balanced                                    |
| OdBFS line-up:      | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |
|                     |   |

| Frequency Response:                     | 20Hz to 20kHz, +0/-0.2dB                          |  |
|---|---|--|
| THD+N:                                  | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |  |
| Noise:                                  | -110dBFS, 20kHz BW, Rs=200Ω                       |  |
| Crosstalk:                              | <-100 dB  |  |
| Common Mode                             |   |  |
| Rejection:                              | >60dB @ 1kHz                                      |  |
|   |   |  |
| Audio Output Performance Specifications |   |  |
| Parameter                               | Line Output                                       |  |
| Output Impedance:                       | <50Ω balanced                                     |  |
| OdBFS line-up:                          | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |  |
| Frequency Response:                     | 20Hz to 20kHz, +0/-0.5dB                          |  |
| THD+N:                                  | <-107dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |  |
| Noise (Mic=EIN):                        | -107dBFS, 20kHz BW                                |  |
| Crosstalk:                              | <100dB  |  |
|   |   |  |

| Pin                             | Function                  |
|---------------------------------|---------------------------|
| 1                               | Chassis Ground/Screen     |
| 2                               | Input Phase/Positive      |
| 3                               | Input Non-Phase/Negative  |
|                                 |                           |
| Balanced Line Output XLR Pinout |                           |
| Pin                             | Function                  |
| 1                               | Chassis Ground/Screen     |
| 2                               | Output Phase/Positive     |
| 3                               | Output Non-Phase/Negative |
|                                 |                           |
| PoE Power                       |                           |
| Standard                        | 802.3af                   |
| Class                           | 0                         |
| PD Power Range                  | 0.44 W to 12.94 W         |
| Typical PSE Power Usage 6.5 W   |                           |
| Typical PSE Power U:            | sage 6.5 W                |

| Equipment Type         |  |  |
|------------------------|--|--|
| AVN-AIO8:              | 8 Input, 8 Input, 8 Output Dante®<br>Interface, PoE                  |  |
|                        |  |  |
| Physical Specification | n  |  |
| Dimensions (Raw):      | 48cm (W) x 11cm (D) x 4.3cm (H)(1U)<br>19" (W) x 4.3" (D) x 1.7" (H) |  |
| Weight:                | Nett: 1.1kg Gross: 1.6kg<br>Nett: 2.4lbs Gross: 3.5lbs               |  |



AVN-AIO8 Front View.



AVN-AIO8 Rear View.









# RAVENNA



# AVN-AIO8R 8 Input, 8 Output, Dual Dante® Interface, PoE





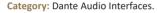












**Product Function:** Converts up to eight analogue inputs and eight analogue outputs to and from the Dante Audioover-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 8 x balanced analogue inputs on XLR.
- 8 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.
- · PoE and Link LED status indicators for each Ethernet port.
- · Clock LED status indicator.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.



The AVN-AIO8R audio converter and interface converts up to eight analogue inputs and eight analogue outputs to and from the Dante Audio-over-IP networking standard. Dual Ethernet ports allow the unit to operate in redundant mode. ensuring audio routing is maintained in the event of loss of link on either of the network connections.

This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue inputs and outputs are on highquality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE, Link, and Clock. Global OdBFS line-up can be set to +12dBu, +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 8 x balanced analogue inputs on XLR.
- 8 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.

- · PoE and Link LED status indicators for each Ethernet port.
- Clock LFD status indicator.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.

(The AVN-AIO8 is available with a single Ethernet port).

#### **Technical Specification For AVN-AIO8R**

| Network and AoIP               |  |
|--------------------------------|--|
| AoIP Standard:                 | Dante  |
| Number of Channels:            | 8 receive, 8 transmit                                  |
| Number of Streams:             | 8 receive, 8 transmit                                  |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           |
| AES67 Support:                 | Yes  |
| Connectivity:                  | 2 x RJ45   |
| Speed:                         | 1Gbps or 100Mbps                                       |
| Network Modes:                 | Switched or redundant                                  |
| Dante Domain<br>Manager Ready: | Yes  |

| Audio Input Performance Specifications |   |
|--|---|
| Parameter                              | Line Input  |
| Input Impedance:                       | >20kΩ balanced                                    |
| OdBFS line-up:                         | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |

| Frequency Response:                   | 20Hz to 20kHz, +0/-0.2dB  |
|---------------------------------------|---|
| THD+N:                                | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW                                       |
| Noise:                                | -110dBFS, 20kHz BW, Rs=200Ω   |
| Crosstalk:                            | <-100 dB  |
| Common Mode<br>Rejection:             | >60dB @ 1kHz  |
| Audio Output Perforn                  | nance Specifications  |
| Parameter .                           | Line Output   |
| Output Impedance:                     | <50Ω balanced   |
|                                       |   |
| OdBFS line-up:                        | User Selectable +12dBu/+18dBu/<br>+24dBu balanced                                 |
| OdBFS line-up:<br>Frequency Response: |   |
| ·                                     | +24dBu balanced   |
| Frequency Response:                   | +24dBu balanced<br>20Hz to 20kHz, +0/-0.5dB<br><-107dBFS, -30dBFS, 20Hz to 20kHz, |

| Balanced Line Input XLR Pinout |                           |
|--------------------------------|---------------------------|
| Pin                            | Function                  |
| 1                              | Chassis Ground/Screen     |
| 2                              | Input Phase/Positive      |
| 3                              | Input Non-Phase/Negative  |
|                                |                           |
| Balanced Line Outp             | ut XLR Pinout             |
| Pin                            | Function                  |
| 1                              | Chassis Ground/Screen     |
| 2                              | Output Phase/Positive     |
| 3                              | Output Non-Phase/Negative |
|                                |                           |
| PoE Power                      |                           |
| Standard                       | 802.3af                   |
| Class                          | 0                         |
| PD Power Range                 | 0.44 W to 12.94 W         |
| Typical PSE Power              |                           |
| Usage                          | 7.5 W                     |

Max PSE Power Usage 15.4 W

| Equipment Type                |  |
|-------------------------------|--|
| AVN-AIO8R:                    | 8 Input, 8 Output, Dual Dante®<br>Interface, PoE |
|                               |  |
| <b>Physical Specification</b> | n  |
| Dimensions (Raw):             | 48cm (W) x 11cm (D) x 4.3cm (H)(1U)              |

Weight:

Nett: 1.1kg Gross: 1.6kg Nett: 2.4lbs Gross: 3.5lbs



AVN-AIO8R Front View.



AVN-AIO8R Rear View.











# AVN-AI16 16 Input Dante® Interface, PoE















Category: Dante Audio Interfaces.

**Product Function:** Converts sixteen analogue inputs to the Dante Audio-over-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 16 x balanced analogue inputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 1 x RJ45 Dante connector (1 Gb/s Ethernet Port).
- · PoE. Link, and Clock LED status indicators.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

All analogue inputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE, Link, and Clock, Global OdBFS line-up can be set to +12dBu, +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 16 x balanced analogue inputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 1 x RJ45 Dante connector (1 Gb/s Ethernet Port).

- PoE, Link, and Clock LED status indicators.
- Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

(The AVN-AI16R is available with dual redundant Ethernet ports).

The AVN-AI16 audio converter and interface converts up to sixteen analogue inputs to the Dante Audio-over-IP networking standard. This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).



#### **Technical Specification For AVN-AI16**

| rediffical ope                 | omoundment of Aviv Airo                                |           |
|--------------------------------|--|-----------|
| Network and AoIP               |  | Audio Ir  |
| AoIP Standard:                 | Dante  | Paramet   |
| Number of Channels:            | 16 transmit  | Input Im  |
| Number of Streams:             | 16 transmit  | OdBFS lii |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz | Frequen   |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           | THD+N:    |
| AES67 Support:                 | Yes  |           |
| Connectivity:                  | 1 x RJ45   | Noise:    |
| Speed:                         | 1Gbps or 100Mbps                                       | Crosstal  |
| Dante Domain<br>Manager Ready: | Yes  | Commo     |

| Parameter                 | Line Input  |
|---------------------------|---|
| Input Impedance:          | >20kΩ balanced                                    |
| OdBFS line-up:            | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                          |
| THD+N:                    | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |
| Noise:                    | -110dBFS, 20kHz BW, Rs=200Ω                       |
| Crosstalk:                | <-100 dB  |
| Common Mode<br>Rejection: | >60dB @ 1kHz                                      |

| Balanced Line Input XLR Pinout |                          |  |
|--------------------------------|--------------------------|--|
| Pin                            | Function                 |  |
| 1                              | Chassis Ground/Screen    |  |
| 2                              | Input Phase/Positive     |  |
| 3                              | Input Non-Phase/Negative |  |
| PoE Power                      |                          |  |
| Standard                       | 802.3af                  |  |
| Class                          | 0                        |  |
| PD Power Range                 | 0.44 W to 12.94 W        |  |
| Typical PSE Power<br>Usage     | 6 W                      |  |
| Max PSE Power Usage            | 15.4 W                   |  |

| Equipment Type<br>AVN-AI16: | 16 Input Dante® Interface, PoE                          |
|-----------------------------|---|
| Physical Specification      | n   |
| Dimensions (Raw):           | 48cm (W) x 11cm (D) x 4.3cm (H)(1U)                     |
| Weight:                     | 19" (W) x 4.3" (D) x 1.7" (H)  Nett: 1.1kg Gross: 1.6kg |
| wcigiit.                    | Nett: 2.4lbs Gross: 3.5lbs                              |



AVN-AI16 Front View.



AVN-AI16 Rear View.











# AVN-AI16R 16 Input Dual Dante® Interface, PoE





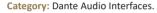












**Product Function:** Converts sixteen analogue inputs to the Dante Audio-over-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 16 x balanced analogue inputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/ +24dBu).
- 2 x RJ45 Dante connector (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.
- · PoE and Link LED status indicators for each Ethernet port.
- · Clock LED status indicator.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.



The AVN-AI16R audio converter and interface converts up to sixteen analogue inputs to the Dante Audio-over-IP networking standard. Dual Ethernet ports allow the unit to operate in redundant mode, ensuring audio routing is maintained in the event of loss of link on either of the network connections.

This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue inputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE, Link, and Clock. Global OdBFS line-up can be set

to +12dBu, +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 16 x balanced analogue inputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 2 x RJ45 Dante connector (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.

- PoE and Link LED status indicators for each Ethernet port.
- Clock LFD status indicator.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.

(The AVN-AI16 is available with a single Ethernet port).

#### **Technical Specification For AVN-AI16R**

#### Network and AoIP

Parameter

Input Impedance:

| receiver and Aon               |  |
|--------------------------------|--|
| AoIP Standard:                 | Dante  |
| Number of Channels:            | 16 transmit  |
| Number of Streams:             | 16 transmit  |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           |
| AES67 Support:                 | Yes  |
| Connectivity:                  | 2 x RJ45   |
| Speed:                         | 1Gbps and 100Mbps                                      |
| Network Modes:                 | Switched or redundant                                  |
| Dante Domain<br>Manager Ready: | Yes  |
| Audia Innut Danfarma           | una Cuasifications                                     |
| Audio Input Performa           | ince Specifications                                    |

Line Input >20kΩ balanced

| OdBFS line-up:            | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |
|---------------------------|---|
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                          |
| THD+N:                    | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW       |
| Noise:                    | -110dBFS, 20kHz BW, Rs=200Ω                       |
| Crosstalk:                | <-100 dB  |
| Common Mode<br>Rejection: | >60dB @ 1kHz                                      |
|                           | >60dB @ 1kHz                                      |

Function Chassis Ground/Screen Input Phase/Positive Input Non-Phase/Negative

**Balanced Line Input XLR Pinout** 

| er Selectable +12dBu/+18dBu/<br>IdBu balanced |
|---|
| Hz to 20kHz, +0/-0.2dB                        |
| 10dBFS, -30dBFS, 20Hz to 20kHz,<br>kHz BW     |
| OdBFS, 20kHz BW, Rs=200Ω                      |
| 00 dB   |
| OdB @ 1kHz                                    |

|          | Class       |
|----------|-------------|
| o 20kHz, | PD P        |
| 0Ω       | Typic       |
|          | Max         |
|          | Equi<br>AVN |
|          |             |

| PoE Power           |                   |
|---------------------|-------------------|
| Standard            | 802.3af           |
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 7 W               |
| Max PSE Power Usage | 15.4 W            |
|                     |                   |
| Fauinment Tyne      |                   |

| _ | Equipment  |
|---|------------|
| _ | AVN-AI16R: |

16 Input Dual Dante® Interface, PoE

| Physical Specification |
|------------------------|
| D: ' (D )              |

| Dimensions (Raw): | 48cm (W) x 11cm (D) x 4.3cm (H)(1U)<br>19" (W) x 4.3" (D) x 1.7" (H) |
|-------------------|--|
| Weight:           | Nett: 1.1kg Gross: 1.6kg<br>Nett: 2.4lbs Gross: 3.5lbs               |



AVN-AI16R Front View.



AVN-AI16R Rear View.











# AVN-AO16 16 Output Dante® Interface, PoE















Category: Dante Audio Interfaces.

**Product Function:** Converts sixteen analogue outputs to the Dante Audioover-IP networking standard.

**Typical Applications:** System integration to Dante.

#### Features:

- 16 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 1 x RJ45 Dante connector (1 Gb/s Ethernet Port).
- . PoE. Link, and Clock LED status indicators.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

The AVN-AO16 audio converter and interface converts up to sixteen analogue outputs from the Dante Audio-over-IP networking standard. This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue outputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE, Link. and Clock, Global OdBFS line-up can be set

to +12dBu, +18dBu or +24dBu to meet your specific requirement via the front panel recessed toggle switch.

- 16 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 1 x RJ45 Dante connector (1 Gb/s Ethernet Port).
- · PoE, Link, and Clock LED status indicators.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

(The AVN-AO16R is available with dual redundant Ethernet ports).



#### **Technical Specification For AVN-AO16**

| Network and AoIP    |   |  |
|---------------------|---|--|
| AoIP Standard:      |   |  |
| Number of Channels: | , |  |

| AoIP Standard:                 | Dante  |
|--------------------------------|--|
| Number of Channels:            | 16 receive   |
| Number of Streams:             | 16 receive   |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           |
| AES67 Support:                 | Yes  |
| Connectivity:                  | 1 x RJ45   |
| Speed:                         | 1Gbps or 100Mbps                                       |
| Dante Domain<br>Manager Ready: | Yes  |

| Audio Output Performance Specifications |   |  |
|---|---|--|
| Parameter                               | Line Output                                       |  |
| Output Impedance:                       | <50Ω balanced                                     |  |
| OdBFS line-up:                          | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |  |
| Frequency Response:                     | 20Hz to 20kHz, +0/-0.5dB                          |  |
| THD+N:                                  | <-107dBFS, -30dBFS, 20Hz to 20kHz<br>20kHz BW     |  |
| Noise (Mic=EIN):                        | -107dBFS, 20kHz BW                                |  |
| Crosstalk:                              | <100dB  |  |
|   |   |  |

| Balanced Line Output XLR Pinout |                           | E |
|---------------------------------|---------------------------|---|
| Pin                             | Function                  | Д |
| 1                               | Chassis Ground/Screen     |   |
| 2                               | Output Phase/Positive     | P |
| 3                               | Output Non-Phase/Negative |   |
|                                 |                           | ( |
| PoE Power                       |                           |   |
| Standard                        | 802.3af                   | V |
| Class                           | 0                         |   |
| PD Power Range                  | 0.44 W to 12.94 W         |   |
| Typical PSE Power               |                           |   |
| Usage                           | 7 W                       |   |

Max PSE Power Usage 15.4 W

#### Equipment Type AVN-AO16:

16 Output Dante® Interface, PoE

#### Physical Specification

Dimensions (Raw): 48cm (W) x 11cm (D) x 4.3cm (H)(1U)

### 19" (W) x 4.3" (D) x 1.7" (H)

Nett: 1.1kg Gross: 1.6kg Nett: 2.4lbs Gross: 3.5lbs Weight:



AVN-AO16 Front View.



AVN-AO16 Rear View.











# AVN-AO16R 16 Output Dual Dante Interface, PoE















**Product Function:** Converts sixteen analogue outputs to the Dante Audioover-IP networking standard.

Typical Applications: System integration to Dante.

#### Features:

- 16 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.
- · PoE and Link LED status indicators for each Ethernet port.
- · Clock LED status indicator.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.



The AVN-AO16R audio converter and interface converts up to sixteen analogue outputs from the Dante Audio-over-IP networking standard. Dual Ethernet ports allow the unit to operate in redundant mode, ensuring audio routing is maintained in the event of loss of link on either of the network connections.

This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up, except line-up levels, done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All analogue outputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE. Link. and Clock. Global OdBFS line-up can be set to +12dBu, +18dBu or +24dBu to meet your

specific requirement via the front panel recessed toggle switch.

- 16 x balanced analogue outputs on XLR.
- Adjustable global OdBFS line up selection (+12dBu/+18dBu/+24dBu).
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.
- PoE and Link LED status indicators for each Ethernet port.

- · Clock LED status indicator.
- Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both)
   Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.

(The AVN-AO16 is available with a single Ethernet port).

#### **Technical Specification For AVN-AO16R**

| Network and AoIP               |  |
|--------------------------------|--|
| AoIP Standard:                 | Dante  |
| Number of Channels:            | 16 receive   |
| Number of Streams:             | 16 receive   |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           |
| AES67 Support:                 | Yes  |
| Connectivity:                  | 2 x RJ45   |
| Speed:                         | 1Gbps or 100Mbps                                       |
| Network Modes:                 | Switched or redundant                                  |
| Dante Domain<br>Manager Ready: | Yes  |
|                                |  |

| Audio Output Performance Specifications |   |
|---|---|
| Parameter                               | Line Output                                       |
| Output Impedance:                       | <50Ω balanced                                     |
| OdBFS line-up:                          | User Selectable +12dBu/+18dBu/<br>+24dBu balanced |

| Frequency Response:  | 20Hz to 20kHz, +0/-0.5dB                    |
|----------------------|---|
| THD+N:               | <-107dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW |
| Noise (Mic=EIN):     | -107dBFS, 20kHz BW                          |
| Crosstalk:           | <100dB                                      |
| Balanced Line Output | XIR Pinout                                  |
| Pin                  | Function                                    |
| 1                    | Chassis Ground/Screen                       |
| 2                    | Output Phase/Positive                       |
| 3                    | Output Non-Phase/Negative                   |
| PoE Power            |   |
| Standard             | 802.3af                                     |
|                      |   |

0.44 W to 12.94 W

|   | Equipment Type         |  |
|---|------------------------|--|
|   | AVN-AO16R:             | 16 Output Dual Dante®Interface, PoE                                  |
| _ |                        |  |
| _ | Physical Specification |  |
| - | Dimensions (Raw):      | 48cm (W) x 11cm (D) x 4.3cm (H)(1U)<br>19" (W) x 4.3" (D) x 1.7" (H) |
|   | Weight:                | Nett: 1.1kg Gross: 1.6kg<br>Nett: 2.4lbs Gross: 3.5lbs               |



Class PD Power Range

Typical PSE Power

Max PSE Power Usage 15.4 W

AVN-AO16R Front View.



AVN-AO16R Rear View.



# Dante AES67 now! RAVENNA EMBER







# AVN-AESIO8 8 AES3 Input, 8 AES3 Output Dante® Interface, PoE

















Product Function: Converts up to eight digital stereo AES3 inputs and eight digital stereo AES3 outputs to and from the Dante Audio-over-IP networking standard.

Typical Applications: System integration to Dante.

#### Features:

- 8 x balanced digital stereo AES3 inputs on XLR, supporting input rates of 32kHz - 192kHz.
- Sample rate conversion on physical inputs to Dante system sample rate.
- 8 x balanced digital stereo AES3 outputs on XLR, output rate matches Dante system sample rate.
- 1 x RJ45 Dante connector (1Gb/s Ethernet Port).
- · PoE, Link, and Clock LED status indicators.
- AES3 Lock LED status indicators for each input.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- · Powered by PoE.
- 1U 19" rack-mount form factor.

The AVN-AESIO8 audio converter and interface converts up to eight digital stereo AES3 inputs and eight digital stereo AES3 outputs to and from the Dante Audio-over-IP networking standard. Each input can accept sample rates from 32kHz to 192kHz. which will be sample rate converted to the Dante system sample rate. A valid input signal is confirmed by front panel AES3 lock LEDs for each input. All outputs follow the Dante system sample rate.

This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All digital AES3 inputs and outputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE. Link and Clock.

- 8 x balanced digital stereo AES3 inputs on XLR, supporting input rates of 32kHz -192kHz.
- Sample rate conversion on physical inputs to Dante system sample rate.
- 8 x balanced digital stereo AES3 outputs on XLR, output rate matches Dante system sample rate.
- 1 x RJ45 Dante connector (1Gb/s Ethernet Port).

- . PoE. Link, and Clock LED status indicators.
- · AES3 Lock LED status indicators for each input.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE.
- •1U 19" rack-mount form factor.

(The AVN-AESIO8R is available with dual redundant Ethernet ports).



#### **Technical Specification For AVN-AESIO8**

| Network and AoIP               |  |  |
|--------------------------------|--|--|
| AoIP Standard:                 | Dante  |  |
| Number of Channels:            | 16 receive, 16 transmit                                |  |
| Number of Streams:             | 16 receive, 16 transmit                                |  |
| Sample Rates:                  | 44.1kHz, 48 kHz, 88.2kHz, 96kHz,<br>176.4kHz or 192kHz |  |
| Format:                        | Linear PCM, 16, 24 or 32 bit                           |  |
| AES67 Support:                 | Yes  |  |
| Connectivity:                  | 1 x RJ45   |  |
| Speed:                         | 1Gbps or 100Mbps                                       |  |
| Dante Domain<br>Manager Ready: | Yes  |  |
| AES3 Input Specifications      |  |  |

| Support Input Rates:           | 32kHz, 44.1kHz, 48kHz, 88.2kHz,                      |  |
|--------------------------------|--|--|
|                                | 96kHz, 176.4kHz and 192kHz                           |  |
|                                | (sample rate converted to Dante system sample rate). |  |
| Signal Level:                  | AES3-2009 compliant                                  |  |
| Bit Depth:                     | Up to 24bits   |  |
| Output Impedance:              | 110Ω balanced  |  |
| Support Output Rates:          | 44.1kHz, 48kHz, 88.2kHz, 96kHz,                      |  |
|                                | 176.4kHz and 192kHz (set as per                      |  |
|                                | Dante system sample rate).                           |  |
| Signal Level:                  | AES3-2009 compliant                                  |  |
| Bit Depth:                     | 24 bits  |  |
|                                |  |  |
| Balanced Line Input XLR Pinout |  |  |
| Pin                            | Function   |  |
| 1                              | Chassis Ground                                       |  |
| 2                              | Input Positive                                       |  |
| 3                              | Input Negative                                       |  |

| Pin                           | Function                           |
|-------------------------------|------------------------------------|
| 1                             | Chassis Ground                     |
| 2                             | Output Positive                    |
| 3                             | Output Negative                    |
| PoE Power                     |                                    |
| Standard                      | 802.3af                            |
| Class                         | 0                                  |
| PD Power Range                | 0.44 W to 12.94 W                  |
| Typical PSE Power<br>Usage    | 6W                                 |
| Max PSE Power Usage           | 15.4 W                             |
| Equipment Type                |                                    |
| AVN-AESIO8:<br>Interface, PoE | 8 AES3 Input, 8 AES3 Output Dante® |

| <b>Physical Specification</b> |   |
|-------------------------------|---|
| Dimensions (Raw):             | 48cm (W) x 11cm (D) x 4.3cm (H)(1U) 19" (W) x 4.3" (D) x 1.7" (H) |
| Weight:                       | Nett: 1.0kg Gross: 1.5kg<br>Nett: 2.2lbs Gross: 3.3lbs            |



AVN-AESIO8 Front View.

Input Impedance: 110Ω balanced



AVN-AESIO8 Rear View.











# AVN-AESIO8R 8 AES3 Input, 8 AES3 Output Dual Dante® Interface, PoE















Product Function: Converts up to eight digital stereo AES3 inputs and eight digital stereo AES3 outputs to and from the Dante Audio-over-IP networking standard.

Typical Applications: System integration to Dante.

#### Features:

- 8 x balanced digital stereo AES3 inputs on XLR, supporting input rates of 32kHz - 192kHz.
- Sample rate conversion on physical inputs to Dante system sample rate.
- 8 x balanced digital stereo AES3 outputs on XLR, output rate matches Dante system sample rate.
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.

- PoF and Link LED status indicators for each Ethernet port.
- · Clock LED status indicator.
- · AES3 Lock LED status indicators for each input.
- · Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both) Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.

The AVN-AFSIO8R audio converter and interface converts up to eight digital stereo AES3 inputs and eight digital stereo AES3 outputs to and from the Dante Audio-over-IP networking standard. Each input can accept sample rates from 32kHz to 192kHz. which will be sample rate converted to the Dante system sample rate. A valid input signal is confirmed by front panel AES3 lock LEDs for each input. All outputs follow the Dante system sample rate.

Dual Ethernet ports allow the unit to operate in redundant mode, ensuring audio routing is maintained in the event of loss of link on either of the network connections. This cost effective 1U rack-mount unit offers an easy solution for AV professionals and system integrators. It is simple to configure and operate, with all set-up done via the standard Dante Controller software and power via PoE (Power Over Ethernet).

All digital AES3 inputs and outputs are on high-quality Neutrik XLR connectors and there are front panel status/confidence LEDs for PoE. Link and Clock.

- 8 x balanced digital stereo AES3 inputs on XLR, supporting input rates of 32kHz -192kHz.
- Sample rate conversion on physical inputs to Dante system sample rate.

#### Multi-Channel Dante® Audio Interfaces

- 8 x balanced digital stereo AES3 outputs on XLR, output rate matches Dante system sample rate.
- 2 x RJ45 Dante connectors (1Gb/s Ethernet Port) allowing the unit to operate in redundant or switched modes.
- PoE and Link LED status indicators for each Ethernet port.
- Clock LED status indicator.
- AES3 Lock LED status indicators for each input.

- Configuration using Dante Controller.
- AES67 operation & Dante Domain Manager compliant.
- Powered by PoE on either (or both)
   Ethernet ports, offering power supply redundancy.
- 1U 19" rack-mount form factor.

(The AVN-AESIO8 is available with a single Ethernet port).

#### **Technical Specification For AVN-AESIO8R**

| Network and AoIP               |   |
|--------------------------------|---|
| AoIP Standard:                 | Dante   |
| Number of Channels:            | 16 receive, 16 transmit   |
| Number of Streams:             | 16 receive, 16 transmit   |
| Sample Rate:                   | 44.1kHz, 48 kHz, 88.2kHz, 96kHz<br>176.4kHz or 192kHz   |
| Format:                        | Linear PCM, 16, 24 or 32 bit  |
| AES67 Support:                 | Yes   |
| Connectivity:                  | 2 x RJ45  |
| Speed:                         | 1Gbps or 100Mbps  |
| Network Modes:                 | Switched or redundant   |
| Dante Domain<br>Manager Ready: | Yes   |
| AES3 Input Specification       | ons   |
| Input Impedance:               | 110Ω balanced   |
| Support Input Rates:           | 32kHz, 44.1kHz, 48kHz, 88.2kHz, 96kHz, 176.4kHz and 192kHz (sample rate converted to Dante system sample rate). |
| Signal Level:                  | AES3-2009 compliant   |
| Bit Depth:                     | Up to 24bits  |
| Output Impedance:              | 110Ω balanced   |
| Support Output Rates:          | 44.1kHz, 48kHz, 88.2kHz, 96kHz, 176.4kHz and 192kHz (set as per Dante system sample rate).                      |
|                                | ,   |

AES3-2009 compliant

24 bits

| Balanced Line Input XLR Pinout |                |  |
|--------------------------------|----------------|--|
| Pin                            | Function       |  |
| 1                              | Chassis Ground |  |
| 2                              | Input Positive |  |
| 3                              | Input Negative |  |

| Balanced Line Output XLR Pinout |                 |  |
|---------------------------------|-----------------|--|
| Pin                             | Function        |  |
| 1                               | Chassis Ground  |  |
| 2                               | Output Positive |  |
| 3                               | Output Negative |  |
|                                 |                 |  |

| PoE Power                  |                   |
|----------------------------|-------------------|
| Standard                   | 802.3af           |
| Class                      | 0                 |
| PD Power Range             | 0.44 W to 12.94 W |
| Typical PSE Power<br>Usage | 7W                |
| Max PSE Power Usage        | 15.4 W            |
|                            |                   |
| Equipment Type             |                   |

| AVN-AESIO8R:           | 8 AES3 Input, 8 AES3 Output Dual<br>Dante® Interface, PoE |
|------------------------|---|
| Physical Specification |   |

| Dimensions (Raw): | 48cm (W) x 11cm (D) x 4.3cm (H)(<br>19" (W) x 4.3" (D) x 1.7" (H) |
|-------------------|---|
| Weight:           | Nett: 1.0kg Gross: 1.5kg<br>Nett: 2.2lbs Gross: 3.3lbs            |



Signal Level:

Bit Depth:

AVN-AESIO8R Front View.



AVN-AESIO8R Rear View.











# **AVN-DIO01 Dante to Analogue XLR Stereo Output**

The AVN-DIO01 is a Dante AoIP network to analogue XLR stereo output converter in the Sonifex DIO range of Dante input/output devices. It features two balanced analogue XLR outputs and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.















Category: Dante Audio Interfaces.

**Product Function:** Dante to Analogue XLR Stereo Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 2 x balanced XLR analogue outputs.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- Ultra-high quality, wide dynamic range D/A conversion, >120dB.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

The superior A/D and D/A circuitry used in this product is designed for optimal audio performance and offers 120dB of dynamic range - ten times better than similar competing products. All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

- 2 x balanced XLR analogue outputs.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.



- AES67 compatible.
- Dante Domain Manager compliant.
- · Ultra-high quality, wide dynamic range D/A conversion. >120dB.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

#### **Technical Specification For AVN-DIO01**

#### Analogue XLR Output Pin-out:

| Pin | Function         |
|-----|------------------|
| 1   | Chassis Ground   |
| 2   | Output Phase     |
| 3   | Output Non Phase |
|     |                  |

| Line Output - XLR/Terminal: |  |  |
|-----------------------------|--|--|
| Parameter:                  | Description                                    |  |
| Output Impedance:           | <200Ω balanced                                 |  |
| OdBFS Line-Up:              | +18dBu   |  |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.5dB (600Ω load, ref 1kHz) |  |
| THD+N:                      | <-100dBu, -30dBFS, 20Hz to 20kHz, 20kHz BW     |  |
| Dynamic Range:              | 120dB, 20kHz BW                                |  |
| Crosstalk:                  | <-110dB  |  |
|                             |  |  |

#### Dante:

| Parameter     | Description                   |
|---------------|-------------------------------|
| Sample Rates: | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| ncoding:      | PCM 16, PCM 24, PCM 32        |
|               |                               |

#### PoE Power:

| Standard            | 802.3af           |
|---------------------|-------------------|
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 6 W               |
| Max PSE Power Usage | 15.4 W            |
|                     |                   |

#### Equipment Type:

#### Physical Specification:

| Dimensions: (Raw)   | 10.6cm (W) x 7.3cm (D) x 4.3cm (H)<br>4.2" (W) x 2.9" (D) x 1.7" (H) |
|---------------------|--|
| Dimensions (Boxed): | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H)    |
| Weight:             | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs             |



AVN-DIO01 Front View.



AVN-DIO01 Rear View.











# AVN-DIO02 Analogue XLR Stereo Input to Dante®

The AVN-DIO02 is an analogue XLR stereo input to Dante AoIP network converter in the Sonifex DIO range of Dante input/output devices. It features two balanced analogue XLR inputs and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.









**Product Function:** Analogue XLR Stereo Input to Dante®.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 2 x balanced XLR analogue inputs.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- · Dante Domain Manager compliant.
- · Ultra-high quality, wide dynamic range A/D conversion, >120dB.
- Powered via PoE (Power over Ethernet).
- · Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

The superior A/D and D/A circuitry used in this product is designed for optimal audio performance and offers 120dB of dynamic range - ten times better than similar competing products. All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

- 2 x balanced XLR analogue inputs.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.



- Dante Domain Manager compliant.
- Ultra-high quality, wide dynamic range A/D conversion. >120dB.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

#### **Technical Specification For AVN-DIO02**

#### Analogue XLR Input Pin-out:

| Pin                        | Function                                       |
|----------------------------|--|
| 1                          | Chassis Ground                                 |
| 2                          | Input Phase                                    |
| 3                          | Input Non Phase                                |
| Line Input - XLR/Terminal: |  |
| Parameter                  | Description                                    |
| Input Impedance            | >10kΩ balanced                                 |
| OdBFS Line-Up              | +18dBu   |
| Frequency Response         | 20Hz to 20kHz, +0/-0.2dB (600Ω load, ref 1kHz) |
| THD+N                      | <-118dBFS, -12dBu, 20Hz to 20kHz,<br>20kHz BW  |
| Dynamic Range              | 120dB, 20kHz BW, Rs=200Ω                       |
| Crosstalk                  | <-110dB  |
| Common Mode                | s code o divis                                 |
| Rejection                  | >60dB @ 1kHz                                   |

#### Dante:

| Parameter           | Description                   |
|---------------------|-------------------------------|
| Sample Rates:       | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| Encoding:           | PCM 16, PCM 24, PCM 32        |
|                     |                               |
| PoE Power:          |                               |
| Standard            | 802.3af                       |
| Class               | 0                             |
| PD Power Range      | 0.44 W to 12.94 W             |
| Typical PSE Power   |                               |
| Usage               | 3 W                           |
| Max PSE Power Usage | 15.4 W                        |
| Equipment Type:     |                               |
| AVN-DIO02:          | Analogue XLR Stereo Input to  |
| AVIV-DIOUZ.         | Dante®                        |

| Dimensions: (Raw)   | 10.9cm (W) x 7.3cm (D) x 4.3cm (H) 4.3" (W) x 2.9" (D) x 1.7" (H) |
|---------------------|---|
| Dimensions (Boxed): | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H) |
| Weight:             | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs          |



AVN-DIO02 Front View.



AVN-DIO02 Rear View.











**AVN-DIO03 Dante to Headphone Outputs (1/4" & 3.5mm Jacks)** with Volume Control & Limiter















**Product Function:** Dante to Headphone Outputs (1/4" & 3.5mm Jacks) With Volume Control & Limiter.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

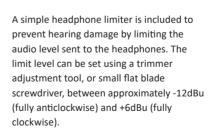
- 1/4-inch and 3.5mm jack analogue headphone outputs.
- · Headphones volume control.
- Limiter on/off, threshold control and LED indicator.
- Neutrik FtherCon® Fthernet connection.
- · Fully Dante compliant device.
- AES 67 compatible.
- Ultra-high quality, wide dynamic range D/A conversion.
- · Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

The AVN-DIO03 is Dante AoIP network-to-headphones convertor in the Sonifex DIO range of Dante input/output devices. It features a single stereo headphone output available on two connections for 1/2" and 3.5mm jacks and a volume control for headphones level. A limiter with threshold setting and LED indication can be switched in to prevent overshooting and protect hearing.

The AVN-DIO03 is Dante AoIP network-toheadphones convertor in the Sonifex DIO range of Dante input/output devices. It features a single stereo headphone output available on two connections for 1/2" and 3.5mm jacks and a volume control for headphones level. A limiter with threshold setting and LED indication can be switched

in to prevent over-shooting and protect hearing.

The front panel potentiometer adjusts headphone volume from mute (fully anticlockwise) to +6dB of gain when fully clockwise. This is useful if the Dante stream level is low and requires boosting.



When limiting, the blue limit LED illuminates. This should prompt you to turn the headphone volume down until the blue LED extinguishes, as audio quality will be

reduced whilst the limiter is active.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

The superior A/D and D/A circuitry used in this product is designed for optimal audio performance and offers 120dB of dynamic range - ten times better than similar competing products. All DIO products use



#### **AVN-DIO Audiophile Dante® Audio Interface**

Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

\* Note: Because the headphone output of this device is unbalanced, performance is 6dB lower than for balanced AVN-DIO products.

- 1/4-inch and 3.5mm jack analogue headphone outputs.
- Headphones volume control.
- Limiter on/off, threshold control and LED indicator.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES 67 compatible.
- Ultra-high quality, wide dynamic range D/A conversion.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

#### **Technical Specification For AVN-DIO03**

#### Headphone Jack Output Pin-out:

Dynamic Range:

Audio Limiter Range:

Crosstalk:

| Pin                 | Function   |
|---------------------|--|
| Tip                 | Left   |
| Ring                | Right  |
| Screen              | 0V Common  |
|                     |  |
| Headphone Output    | Jack:  |
| Parameter           | Description  |
| Output Impedance:   | Capable of driving 150mW into $32\Omega$ to $600\Omega$ headphones |
| Maximum Output      |  |
| Level:              | +16dBu   |
| D to A Line-Up:     | -6dBFS = +16dBu (max volume)                                       |
| Gain:               | Mute (min Volume) to +6dB  |
|                     | (max volume)   |
| Frequency Response: | 20Hz to 20kHz, +0/-0.5dB (ref 1kHz)                                |
| THD+N:              | <0.005%, -14dBFS input, max<br>volume, 20Hz to 20kHz, 20kHz BW     |

<100dB

-12dBu to +6dBu

114dBu, 20kHz BW, 0dB gain

#### Dante:

Dimensions (Boxed):

Weight:

| Danic.                     |   |
|----------------------------|---|
| Parameter                  | Description   |
| Sample Rates:              | 4.1kHz, 48kHz, 88.2kHz, 96kHz                                     |
| Encoding:                  | PCM 16, PCM 24, PCM 32  |
| PoE Power:                 |   |
| Standard                   | 802.3af   |
| Class                      | 0   |
| PD Power Range             | 0.44 W to 12.94 W   |
| Typical PSE Power<br>Usage | 4 W   |
| Max PSE Power Usage        | 15.4 W  |
| Equipment Type:            |   |
| AVN-DIO03:                 | Dante® to Headphone Output  |
| Physical Specification:    |   |
| Dimensions: (Raw)          | 10.6cm (W) x 7.3cm (D) x 4.3cm (H) 4.2" (W) x 2.9" (D) x 1.7" (H) |
|                            |   |

17.4cm (W) x 9.5cm (D) x 5.6cm (H)

6.9" (W) x 3.7" (D) x 2.2" (H)

Nett: 0.2kg Gross: 0.3kg Nett: 0.44lbs Gross: 0.66lbs







AVN-DIO03 Rear View.











# AVN-DIO04 Dante<sup>®</sup> to Analogue Phono Stereo Input & Output

The AVN-DIO04 is a Dante to analogue phono stereo input & output convertor in the Sonifex DIO range of Dante input/output devices. It features stereo analogue input and output phono connections and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.



Category: Dante Audio Interfaces.

Product Function: Dante® to Analogue Phono Stereo Input & Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 2 x analogue phono-type inputs.
- 2 x analogue phono-type outputs.
- Neutrik FtherCon® Fthernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- · Ultra-high quality, wide dynamic range D/A and A/D conversion.
- Powered via PoE (Power over Ethernet).

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

The superior A/D and D/A circuitry used in this product is designed for optimal audio performance and offers 114dB \* of dynamic range - ten times better than similar competing products. All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

\* Note: Because the inputs & outputs of this device are unbalanced, performance is 6dB lower than for balanced AVN-DIO products.



- 2 x analogue phono-type inputs.
- 2 x analogue phono-type outputs.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.

- Dante Domain Manager compliant.
- Ultra-high quality, wide dynamic range D/A and A/D conversion.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

#### **Technical Specification For AVN-DIO04**

#### Analogue Phono Input & Output Pin-out:

| Pin                   | Function                                      |
|-----------------------|---|
| Inner                 | Signal (White - Left, Red - Right)            |
| Outer                 | Chassis Ground                                |
|                       |   |
| Line Input - RCA Phor | 10:   |
| Parameter             | Description                                   |
| Input Impedance       | >5kΩ unbalanced                               |
| 0dBFS Line-Up         | +12dBu  |
| Frequency Response    | 20Hz to 20kHz, +0/-0.5dB<br>(ref 1kHz)        |
| THD+N                 | <-114dBFS, -18dBu, 20Hz to<br>20kHz, 20kHz BW |
| Dynamic Range         | >114dB A-weighted, 20kHz BW,<br>Rs=200Ω       |
| Crosstalk             | <-100dB                                       |
|                       |   |

#### Line Output - RCA Phono:

| Ellic Output Real lic      | 110.                                       |
|----------------------------|--|
| Output Impedance           | <200Ω balanced                             |
| OdBFS Line-Up              | +12dBu                                     |
| Frequency Response         | 20Hz to 20kHz, +0/-0.5dB (ref 1kHz)        |
| THD+N                      | <-114dBu, -30dBFS, 20Hz to 20kHz, 20kHz BW |
| Dynamic Range              | >114dB A-weighted, 20kHz BW                |
| Crosstalk                  | <-100dB                                    |
| Dante:                     |  |
| Parameter                  | Description                                |
| Sample Rates:              | 4.1kHz, 48kHz, 88.2kHz, 96kHz              |
| Encoding:                  | PCM 16, PCM 24, PCM 32                     |
| PoE Power:                 |  |
| Standard                   | 802.3af                                    |
| Class                      | 0  |
| PD Power Range             | 0.44 W to 12.94 W                          |
| Typical PSE Power<br>Usage | 5 W  |
| Max PSE Power Usage        | 15.4 W                                     |
|                            |  |

#### Equipment Type:

AVN-DIO04: Dante® to Analogue Phono Stereo Input & Output

#### Physical Specification:

 $\begin{array}{lll} \mbox{Dimensions: (Raw)} & 11.1 cm \ (W) \times 7.3 cm \ (D) \times 4.3 cm \ (H) \\ & 4.4'' \ (W) \times 2.9'' \ (D) \times 1.7'' \ (H) \\ \mbox{Dimensions (Boxed):} & 17.4 cm \ (W) \times 9.5 cm \ (D) \times 5.6 cm \ (H) \\ & 6.9'' \ (W) \times 3.7'' \ (D) \times 2.2'' \ (H) \\ \mbox{Weight:} & Nett: 0.2 kg \ Gross: 0.3 kg \\ & Nett: 0.44 lbs \ Gross: 0.66 lbs \\ \end{array}$ 



AVN-DIO04 Front View.



AVN-DIO04 Rear View.











## AVN-DIO05 Dante® to Analogue Terminal Block Stereo Input & Output

The AVN-DIO05 is a Dante to analogue terminal block input and output convertor in the Sonifex DIO range of Dante input/output devices. It features balanced stereo analogue inputs and outputs on a terminal block connector and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.





Product Function: Dante® to Analogue Terminal Block Stereo Input & Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 12 x terminal block connections (balanced stereo inputs and outputs).
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES 67 compatible.
- Dante Domain Manager compliant.
- · Ultra-high quality, wide dynamic range D/A and A/D conversion.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

The superior A/D and D/A circuitry used in this product is designed for optimal audio performance and offers 120dB of dynamic range - ten times better than similar competing products. All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack, All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultrareliable connectivity.

- 12 x terminal block connections (balanced) stereo inputs and outputs).
- Neutrik EtherCon® Ethernet connection.
- · Fully Dante compliant device.
- AES 67 compatible.



- · Dante Domain Manager compliant.
- Ultra-high quality, wide dynamic range D/A and A/D conversion.
- · Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

## **Technical Specification For AVN-DIO05**

| Analogue | Terminal | Stereo | Input & | Output | Pin-out: |
|----------|----------|--------|---------|--------|----------|
|----------|----------|--------|---------|--------|----------|

| Anaiogue Terminai Ste | reo input & Output Pin-out: |
|-----------------------|-----------------------------|
| Pin                   | Function                    |
| 1                     | Chassis Ground              |
| 2                     | Left Input Phase            |
| 3                     | Left Input Non Phase        |
| 4                     | Chassis Ground              |
| 5                     | Right Input Phase           |
| 6                     | Right Input Non Phase       |
| 7                     | Chassis Ground              |
| 8                     | Left Output Phase           |
| 9                     | Left Output Non Phase       |
| 10                    | Chassis Ground              |
| 11                    | Right Output Phase          |
| 12                    | Right Output Non Phase      |
|                       |                             |

| Line Input – Terminal Block: |  |  |
|------------------------------|--|--|
| Parameter                    | Description  |  |
| Input Impedance              | >10kΩ balanced   |  |
| OdBFS Line-Up                | +18dBu   |  |
| Frequency Response           | 20Hz to 20kHz, +0/-0.2dB (600 $\Omega$ load, ref 1kHz) |  |
|                              |  |  |

| <-118dBFS, -12dBu, 20Hz to 20kHz,<br>20kHz BW |
|---|
| 120dB, 20kHz BW, Rs=200Ω                      |
| <-110dB                                       |
| >60dB @ 1kHz                                  |
|   |

## Line Output – Terminal Block:

| Output Impedance                   | <200Ω balanced                    |
|------------------------------------|-----------------------------------|
| 0dBFS Line-Up                      | +18dBu                            |
| Frequency Response load, ref 1kHz) | 20Hz to 20kHz, +0/-0.5dB (600Ω    |
| THD+N<br>20kHz BW                  | <-100dBu, -30dBFS, 20Hz to 20kHz, |
| Dynamic Range                      | 120dB, 20kHz BW                   |
| Crosstalk                          | <-110dB                           |
| Dante:                             |                                   |
| Parameter                          | Description                       |
| Sample Rates:                      | 4.1kHz, 48kHz, 88.2kHz, 96kHz     |
| Encoding:                          | PCM 16, PCM 24, PCM 32            |

#### PoE Power:

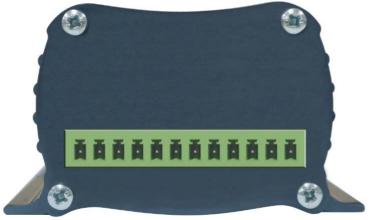
| Standard            | 802.3af           |
|---------------------|-------------------|
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 5 W               |
| Max PSE Power Usage | 15.4 W            |
|                     |                   |

## Equipment Type:

| darbinent ilber |  |
|-----------------|--|
| VN-DIO05:       | Dante® to Analogue Terminal Block<br>Stereo Input & Output |
|                 |  |

## Physical Specification:

| Dimensions: (Raw)   | 10.6cm (W) x 7.3cm (D) x 4.3cm (H)<br>4.2" (W) x 2.9" (D) x 1.7" (H) |
|---------------------|--|
| Dimensions (Boxed): | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H)    |
| Weight:             | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs             |







AVN-DIO05 Rear View.











## AVN-DIO06 Dante® to AES3 XLR Stereo Input & Output

The AVN-DIO06 is a Dante to AES3 digital input and output audio convertor in the Sonifex DIO range of Dante input/output devices. It features stereo AES3 digital audio inputs and outputs on Neutrik XLR connectors, and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.



Category: Dante Audio Interfaces.

Product Function: Dante® to AES3 XLR Stereo Input & Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

## Features:

- 1 x stereo AES3 XLR input.
- 1 x stereo AES3 XLR output.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

- 1 x stereo AES3 XLR input.
- 1 x stereo AES3 XLR output.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.



- · Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

## **Technical Specification For AVN-DIO06**

| VEC 3 | VID | Storoo | Innut | Din out  |
|-------|-----|--------|-------|----------|
| AE53  | XLK | Stereo | Input | Pin-out: |

| Pin | Function        |  |
|-----|-----------------|--|
| 1   | Chassis Ground  |  |
| 2   | Input Phase     |  |
| 3   | Input Non Phase |  |

## AES3 XLR Stereo Output Pin-out:

| Pin | Function         |  |
|-----|------------------|--|
| 1   | Chassis Ground   |  |
| 2   | Output Phase     |  |
| 3   | Output Non Phase |  |

## Power Supply - Class 0 802.3af PoE:

| Device    | Power Consumption (Watts) |
|-----------|---------------------------|
| DIO06:    | < 3 W                     |
| Parameter | Description               |
|           |                           |

## AES3 Input - XLR:

| Output Impedance | 110Ω balanced                  |
|------------------|--------------------------------|
| Output Format    | AES3                           |
| Supported Sample |                                |
| Rates            | 44.1kHz, 48kHz, 88.2kHz, 96kHz |

#### Dante:

| Parameter     | Description                   |
|---------------|-------------------------------|
| Sample Rates: | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| Encoding:     | PCM 16, PCM 24, PCM 32        |

## PoE Power:

| Standard            | 802.3af           |
|---------------------|-------------------|
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 2 W               |
| Max PSE Power Usage | 15.4 W            |

## Equipment Type:

| AVN-DIO06: | Dante® to AES3 XLR Stereo Input |
|------------|---------------------------------|
|            | & Output                        |

## Physical Specification:

| r nysicai specification | •   |
|-------------------------|---|
| Dimensions: (Raw)       | 10.6cm (W) x 7.3cm (D) x 4.3cm (H)                                |
|                         | 4.2" (W) x 2.9" (D) x 1.7" (H)                                    |
| Dimensions (Boxed):     | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H) |
| Weight:                 | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs          |



AVN-DIO06 Front View.



AVN-DIO06 Rear View.











## AVN-DIO07 Dante® to AES-3id BNC Stereo Input & Output

The AVN-DIO07 is a Dante to AES-3id digital input and output audio convertor in the Sonifex DIO range of Dante input/output devices. It features stereo AES-3id digital audio inputs and outputs on BNC connectors, and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.















Product Function: Dante® to AES-3id BNC Stereo Input & Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 1 x stereo AES-3id BNC input.
- 1 x stereo AES-3id BNC output.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- · Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- · Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

- 1 x stereo AES-3id BNC input.
- 1 x stereo AES-3id BNC output.
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.



- Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

## **Technical Specification For AVN-DIO07**

## AES-3id BNC Input & Output Pin-out:

| Pin   | Function       |  |
|-------|----------------|--|
| Inner | Signal         |  |
| Outer | Chassis Ground |  |
|       |                |  |

## AES-3id Input - BNC:

| Parameter        | Description                    |
|------------------|--------------------------------|
| Output Impedance | 75Ω unbalanced                 |
| Output Format    | AES-3id                        |
| Supported Sample |                                |
| Rates            | 44.1kHz, 48kHz, 88.2kHz, 96kHz |

## Dante:

| Parameter     | Description                   |
|---------------|-------------------------------|
| Sample Rates: | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| Encoding:     | PCM 16, PCM 24, PCM 32        |

#### PoE Power:

| Standard            | 802.3af           |
|---------------------|-------------------|
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 2 W               |
| Max PSE Power Usage | 15.4 W            |

## Equipment Type:

| AVN-DIO07: | Dante® to AES-3id BNC Stereo Input |
|------------|------------------------------------|
|            | & Output                           |

## Physical Specification:

| rilysical Specification | •   |
|-------------------------|---|
| Dimensions: (Raw)       | 11.6cm (W) x 7.3cm (D) x 4.3cm (H) 4.6" (W) x 2.9" (D) x 1.7" (H) |
| Dimensions (Boxed):     | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H) |
| Weight:                 | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs          |



AVN-DIO07 Front View.



AVN-DIO07 Rear View.











## AVN-DIO08 Dante® to AES3 Terminal Block Stereo Input & Output

The AVN-DIO08 is a Dante to AES3 digital input and output audio convertor in the Sonifex DIO range of Dante input/output devices. It features stereo AES3 digital audio inputs and outputs on terminal block connectors, and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.















Category: Dante Audio Interfaces.

Product Function: Dante® to AES3 Terminal Block Stereo Input & Output.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

- 6 x terminal block connections (balanced stereo inputs and outputs).
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- · Dante Domain Manager compliant.
- · Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

- 6 x terminal block connections (balanced stereo inputs and outputs).
- Neutrik EtherCon® Ethernet connection.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.



- · Ultra-high quality digital audio.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

## **Technical Specification ForAVN-DIO08**

## AES3 Terminal Block Stereo Input & Output Pin-out:

| 1 | Chassis Ground   |
|---|------------------|
| 2 | Input Phase      |
| 3 | Input Non Phase  |
| 4 | Chassis Ground   |
| 5 | Output Phase     |
| 6 | Output Non Phase |
|   |                  |

| AES3 Input – Terminal | Block:      |
|-----------------------|-------------|
| Darameter             | Description |

| Input Impedance  | 110Ω balanced                   |
|------------------|---------------------------------|
| Input Format     | AES3                            |
| Supported Sample |                                 |
| Rates            | 32kHz, 44.1kHz, 48kHz, 88.2kHz, |
|                  | 96kHz, 176,4kHz, 192kHz         |

## AES3 Output - Terminal Block:

| Parameter        | Description                    |
|------------------|--------------------------------|
| Output Impedance | 110Ω balanced                  |
| Output Format    | AES3                           |
| Supported Sample |                                |
| Rates            | 44.1kHz, 48kHz, 88.2kHz, 96kHz |
|                  |                                |

#### ante.

| Parameter     | Description                   |
|---------------|-------------------------------|
| Sample Rates: | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| Encoding:     | PCM 16, PCM 24, PCM 32        |

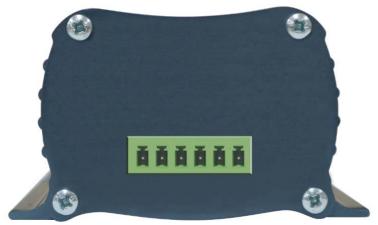
| PoE Power:          |                   |
|---------------------|-------------------|
| Standard            | 802.3af           |
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 2 W               |
| Max PSE Power Usage | 15.4 W            |
|                     |                   |

## Equipment Type:

| AVN-DIO08: | Dante® to AES3 Terminal Block |
|------------|-------------------------------|
|            | Storeo Innut & Outnut         |

## Physical Specification:

| Dimensions: (Raw)   | 10.6cm (W) x 7.3cm (D) x 4.3cm (H) 4.2" (W) x 2.9" (D) x 1.7" (H) |
|---------------------|---|
| Dimensions (Boxed): | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H) |
| Weight:             | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs          |



AVN-DIO08 Front View.



AVN-DIO08 Rear View.











## **AVN-DIO09 Microphone Input to Dante®**





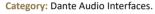












**Product Function: Microphone Input to** Dante®.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

#### Features:

• 1 x balanced microphone input on XLR socket with latch lock.

- Neutrik EtherCon® Ethernet connection.
- Single turn pot setting fine mic gain (0dB - 36dB).
- Coarse mic gain switch (+20db/+50dB).
- High pass filter on/off switch.
- Phantom power on/off switch.
- · Phantom power LED indicator.
- · Level LED indicator.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- Ultra-high quality E.I.N. of >129dB.
- Powered via PoE (Power over Ethernet).
- · Five units fit into an AVN-DIORK 1U rack.

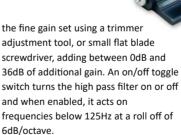
The AVN-DIO09 is a Microphone input to Dante converter in the Sonifex DIO range of Dante input/output devices with A/D circuitry offering a world-class E.I.N. of 129dB. It features a single high quality mic preamp with balanced XLR input, coarse and fine gain controls, high pass filter, phantom power, tri-colour level LED and one Neutrik EtherCon® connector for direct connection to a Dante AoIP network.

All Sonifex DIO interfaces provide a simple, convenient, and elegant plug and play method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

All DIO products use Dante Controller for configuration, are AES67 and Dante Domain Manager compliant and are powered by PoE (Power over Ethernet).

They use rugged aluminium boxes with side slots for screw-mounting and five units can be fitted into the optional AVN-DIORK 1U rack. All feature rugged Neutrik EtherCon® connectors and Neutrik lockable audio connectors for ultra-reliable connectivity.

The AVN-DIO09 has coarse and fine mic gain with the coarse gain set using a toggle switch, providing 20dB/50dB of gain, and



Phantom power is enabled/disabled via a toggle switch on the front panel and when enabled, a 48V DC supply is provided to power an appropriate microphone. A red LED illuminates to show when phantom is enabled.

A front panel audio level LED helps to set the gain and shows the audio level being sent to the Dante network.

**Note:** If using a phantom powered microphone, it may be necessary to earth the unit using the rear panel earth tag to eliminate mains hum, depending on the nature of your PoE supply.

- 1 x balanced microphone input on XLR socket with latch lock.
- Neutrik EtherCon® Ethernet connection.
- · Single turn pot setting fine mic gain (0dB - 36dB).



## AVN-DIO Audiophile Dante® Audio Interface

- Coarse mic gain switch (+20db/+50dB).
- High pass filter on/off switch.
- Phantom power on/off switch.
- Phantom power LED indicator.
- Level LED indicator.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- Ultra-high quality E.I.N. of >129dB.
- Powered via PoE (Power over Ethernet).
- Five units fit into an AVN-DIORK 1U rack.

## **Technical Specification For AVN-DIO09**

## Microphone XLR Input Pin-out :

| Pin | Function        |
|-----|-----------------|
| 1   | Chassis Ground  |
| 2   | Input Phase     |
| 3   | Input Non Phase |
|     |                 |

## Earthing Point:

Single earth tag.

| evel LED Thresholds: |                                 |
|----------------------|---------------------------------|
| off                  | Level is under -38dBFS (-20dBu) |
| ireen                | Level is between -38dBFS (-20dB |

| Green | Level is between -38dBFS (-20dBu) and -18dBFS (0dBu)   |
|-------|--|
| Amber | Level is between -18dBFS (0dBu)<br>and -10dBFS (+8dBu) |
| Red   | Level is above -10dBFS (+8dBu).                        |

## Microphone Input - XLR:

| Parameter           | Description                |
|---------------------|----------------------------|
| Input Impedance     | 2.2kΩ balanced             |
| Maximum Input Level | -68dBu (max gain) to -8dBu |
|                     | (min gain)                 |

| Gain               | Adjustable 20dB to 86dB                                     |
|--------------------|---|
| Frequency Response | 20Hz to 20kHz, +0/-0.5dB (ref 1kHz)                         |
| High Pass Filter   |   |
| Response           | Fc = 125Hz @ 6dB per octave                                 |
| THD+N              | <0.01%, -32dBu input, 40dB gain,<br>20Hz to 20kHz, 20kHz BW |
| E.I.N.             | 129dBu, 20kHz BW, max gain,<br>Rs=200Ω                      |
| Common Mode        |   |

>60dB @ 1kHz

## Rejection Phantom Power

| Dante:        |                               |
|---------------|-------------------------------|
| Parameter     | Description                   |
| Sample Rates: | 4.1kHz, 48kHz, 88.2kHz, 96kHz |
| Encoding:     | PCM 16. PCM 24. PCM 32        |

+48V ± 4V

## PoE Power:

| Standard            | 802.3af           |
|---------------------|-------------------|
| Class               | 0                 |
| PD Power Range      | 0.44 W to 12.94 W |
| Typical PSE Power   |                   |
| Usage               | 4 W               |
| Max PSE Power Usage | 15.4 W            |

## **Equipment Type:**

| AVN-DIO09:                    | Microphone Input to Dante®  |
|-------------------------------|---|
|                               |   |
| <b>Physical Specification</b> | :   |
| Dimensions: (Raw)             | 10.6cm (W) x 7.3cm (D) x 4.3cm (H) 4.2" (W) x 2.9" (D) x 1.7" (H) |
| Dimensions (Boxed):           | 17.4cm (W) x 9.5cm (D) x 5.6cm (H) 6.9" (W) x 3.7" (D) x 2.2" (H) |
| Weight:                       | Nett: 0.2kg Gross: 0.3kg<br>Nett: 0.44lbs Gross: 0.66lbs          |







AVN-DIO09 Rear View.











## AVN-DIO10 Dante® to 3G/HD/SD-SDI Embedder/De-Embedder















Category: Dante Audio Interfaces.

Product Function: Dante® to 3G/HD/SD-SDI Embedder/De-Embedder.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital

audio equipment to the Dante AoIP audio network.

## Features:

- 1 x 3G/HD/SD-SDI input.
- 1 x reclocked 3G/HD/SD-SDI output.
- Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCon® Ethernet connectors.
- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.

- AES67 compatible.
- Dante Domain Manager compliant.
- · Web interface for configuration.
- · Clock, SDI Lock, PoE and Sync LEDs.
- DIPSwitch selection of embed channel pairs.
- · Overwrite or insert into existing SDI audio groups.
- · Test tones available on embedded outputs.
- SDI audio sample rate support at 48kHz.

- All available Dante sample rates supported.
- Sample rate conversion of audio between Dante and SDI.
- Dante clock domain can be optionally synchronised from the SDI clock.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and underdesk mounting plate.
- 3 x units rackmount in the AVN-DIORK.



The easiest way to connect legacy SDI equipment to the Dante® network, the AVN-DIO10 can be used for simultaneous embedding and de-embedding. This simple plug and play audio/video interface provides a convenient and elegant method of connecting legacy 3G/ HD/SD-SDI equipment to the Dante® AoIP audio network.

The AVN-DIO10 takes an SDI feed, deembeds the 16 audio channels and places them on channels 1-16 of the Dante network, mapped using Dante Controller. It simultaneously takes the 16 input channels mapped to the device on Dante Controller and re-embeds them onto the SDI output.

Switches on the unit allow embedding of Dante channels onto the SDI output per channel pair and there are two modes of operation: Insert Mode enabled allows embedding to overwrite existing SDI audio selectively per channel pair. Insert Mode disabled clears any incoming audio channels on the SDI output and then allows selective embedding onto the SDI output per channel pair.

A Test Tone Mode allows 1kHz, 2kHz, 3kHz and 4kHz signals to be output on channels 1 to 4 respectively, for any group where embedding is enabled. This is so that downstream SDI audio outputs can be tested without the need of Dante sources.

It's powered using Power over Ethernet (PoE), using Neutrik EtherCON connectors, with primary and secondary ports for power and data redundancy. The AVN-DIO10 uses the latest Audinate Dante ® chipsets so is AES67 and Dante Domain Manager® compliant.

## **AVN-DIO Audiophile Dante® Audio Interface**

There are front panel LEDs to indicate network clock status, SDI lock status, AoIP Primary and AoIP Secondary link status, PoE Primary power and PoE Secondary power active.

A web interface is available for firmware updates, status information and network settings.

The AVN-DIO10 is supplied with a belt-clip, shoulder strap hooks (but not the strap) and an underdesk mounting plate. Up to 3 of the AVN-DIO10 units can be rackmount in the 1U AVN-DIORK.

- 1 x 3G/HD/SD-SDI input.
- 1 x reclocked 3G/HD/SD-SDI output.
- Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCon® Ethernet connectors.
- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.
- AES67 compatible.

- Dante Domain Manager compliant.
- · Web interface for configuration.
- Clock, SDI Lock, PoE and Sync LEDs.
- DIPSwitch selection of embed channel pairs.
- Overwrite or insert into existing SDI audio groups.
- Test tones available on embedded outputs.
- SDI audio sample rate support at 48kHz.
- All available Dante sample rates supported.
- Sample rate conversion of audio between Dante and SDI.
- Dante clock domain can be optionally synchronised from the SDI clock.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and underdesk mounting plate.
- 3 x units rackmount in the AVN-DIORK.

## **Technical Specification For AVN-DIO10**

| Parameter SDI Input:        |   |
|-----------------------------|---|
| Input Impedance:            | 75Ω Unbalanced  |
| SDI Supported<br>Standards: | 270Mbps SMPTE-259M-C (SD)                                 |
| SDI Supported Standards:    | 1.485 or 1.4835Gbps<br>SMPTE-292M (HD)                    |
| SDI Supported Standards:    | 2.97 or 2.967Gbps<br>SMPTE-424M (3G)                      |
| Supported Video Formats:    | 525/59.94 (SMPTE-125M)                                    |
| Supported Video Formats:    | 625/50 (ITU-R BT.656)                                     |
| Supported Video Formats:    | 720p/23.98, 24, 25, 29.97, 30, 50, 59.94, 60 (SMPTE-296M) |
| Supported Video Formats:    | 1035i/59.94, 60 (SMPTE-260M)                              |
| Supported Video Formats:    | 1080i/50, 59.94, 60<br>(SMPTE-274M)                       |
| Supported Video Formats:    | 1080p/23.98, 24, 25, 50, 59.94, 60 (SMPTE-274M)           |
| Supported Video Formats:    | 1080pSF/23.98, 24, 25, 29.97, 30 (RP-211)                 |
| Supported Video Formats:    | 1080i/50 (SMPTE-295M)                                     |
| Supported Video Formats:    | 1080p/50 (SMPTE-295M)                                     |
| Embedded Audio:             | 48kHz, synchronous  |
| Embedded Audio:             | SMPTE-272M-ABC  |
| Embedded Audio:             | SMPTE-299M  |
| Supported Image Mapping     | : SMPTE-425M-AB   |

| Parameter SDI Output:    |                      |
|--------------------------|----------------------|
| Output Impedance:        | 75Ω Unbalanced       |
| Alignment Jitter:        | <0.2UI               |
| Output Level:            | 800mV ±10%           |
| Return Loss:             | <15dB @ 1.5GHz       |
| SDI Supported Standards: | Output follows input |
| Supported Video Formats: | Output follows input |
| Embedded Audio:          | 48kHz, synchronous   |
| Embedded Audio:          | SMPTE-272M-C         |
| Embedded Audio:          | SMPTE-299M           |
|                          |                      |

| Network and AoIP               |  |
|--------------------------------|--|
| AoIP Standard:                 | Dante  |
| Channels:                      | 16 receive, 16 transmit  |
| Flows:                         | 16 receive, 16 transmit  |
| Sample Rates:                  | 44.1kHz, 48kHz, 88.2kHz, 96kHz,<br>176.4kHz, 192kHz                                  |
| Encoding:                      | PCM 16, PCM 24, PCM 32   |
| AES67 Support                  | Yes  |
| Connectivity:                  | 2 x etherCON (RJ45 compatible)   |
| Speed:                         | 1Gbps or 100Mbps   |
| Network Modes:                 | Switched or redundant  |
| Dante Domain Manager<br>Ready: | Yes  |
| Clock Source:                  | Internal (PTP Leader), Network<br>PTP Leader or from SDI input<br>(Sync to External) |
|                                |  |

| Standard                | 802.3af           |
|-------------------------|-------------------|
| Redundancy              | Yes               |
| Class                   | 0                 |
| PD Power Range          | 0.44 W to 12.94 W |
| Typical PSE Power Usage | 6 W               |
| Max PSE Usage           | 15.4 W            |
|                         |                   |
| Equipment Type:         |                   |

PoF Power

| AVN-DIO10:              | Dante to 3G/HD/SD-SDI<br>Embedder/De-Embedder                            |
|-------------------------|--|
|                         |  |
| Physical Specification: |  |
| Dimensions (Raw):       | 14.0cm (W) x 13.6cm (D) x<br>4.2cm (H)<br>5.5" (W) x 5.4" (D) x 1.7" (H) |
| Weight:                 | Nett: 0.5kg Gross: 0.3kg<br>Nett: 1.1lbs Gross: 0.7lbs                   |

| Accessories: |  |
|--------------|--|
| AVN-DIORK    | AVN-DIO 1U 19" Rack Kit (5 x Small DIO or 3 x Large DIO) |





AVN-DIO10 Front View.

AVN-DIO10 Rear View.











## AVN-DIO19 Dante® to AES3 16 Channel I/O Converter



Category: Dante Audio Interfaces.

Product Function: Dante® to AES3 16 Channel I/O Converter.

Typical Applications: These simple plug and play audio interfaces provide a

convenient and elegant method of connecting legacy analogue and digital audio equipment to the Dante AoIP audio network.

## Features:

- 8 x balanced digital stereo AES3 inputs and outputs on 2 x 25-way D-types.
- Sample rate conversion of physical inputs to Dante system sample rate.
- Physical output sample rate matches Dante system sample rate.

- · Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCON® Ethernet connectors.
- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.

- · Web interface for configuration.
- · Clock, PoE, Network link and AES3 input lock LEDs.
- All available Dante sample rates supported.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and under desk mounting plate.
- 3 x units rackmount in the AVN-DIORK.



The AVN-DIO19 audio converter and interface converts up to eight digital stereo AES3 inputs and eight digital stereo AES3 outputs to and from the Dante Audio-over-IP networking standard. Each input can accept sample rates from 32kHz to 192kHz, which will be sample rate converted to the Dante system sample rate. All outputs follow the Dante system sample rate.

It's powered using Power over Ethernet (PoE), using Neutrik EtherCON® connectors, with primary and secondary ports for power and data redundancy. The AVN-DIO19 uses the latest Audinate Dante® chipsets so is AES67 and Dante Domain Manager® compliant.

There are front panel LEDs to indicate network clock status, AoIP Primary and AoIP Secondary link status, PoE Primary power and PoE Secondary power active. In addition to these there are 8 AES3 input lock status LEDs.

A web interface is available for firmware updates, status information, network and device settings.

The AVN-DIO19 is supplied with a belt-clip, shoulder strap hooks (but not the strap) and an under desk mounting plate. Up to 3 of the AVN-DIO19 units can be rackmount in the 1U AVN-DIORK.

- 8 x balanced digital stereo AES3 inputs and outputs on 2 x 25-way D-types.
- Sample rate conversion of physical inputs to Dante system sample rate.

## AVN-DIO Audiophile Dante® Audio Interface

- Physical output sample rate matches Dante system sample rate.
- Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCON® Ethernet connectors.
- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.

- Web interface for configuration.
- Clock, PoE, Network link and AES3 input lock LEDs.
- All available Dante sample rates supported.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and under desk mounting plate.
- 3 x units rackmount in the AVN-DIORK.

## **Technical Specification For AVN-DIO19**

| AES3 I/O                       |   |
|--------------------------------|---|
| Input & Output                 |   |
| Impedance:                     | 110Ω Balanced                                       |
| Sample Rates:                  | 32 – 192kHz   |
|                                |   |
| Network and AoIP               |   |
| AoIP Standard:                 | Dante   |
| Channels:                      | 16 receive, 16 transmit                             |
| Flows:                         | 16 receive, 16 transmit                             |
| Sample Rates:                  | 44.1kHz, 48kHz, 88.2kHz, 96kHz,<br>176.4kHz, 192kHz |
| Encoding:                      | PCM 16, PCM 24, PCM 32                              |
| Connectivity:                  | 2 x etherCON (RJ45 compatible)                      |
| Speed:                         | 1Gbps or 100Mbps                                    |
| Network Modes:                 | Switched or redundant                               |
| AES67 Support:                 | Yes   |
| Dante Domain<br>Manager Ready: | Yes   |
| Clock Source:                  | Internal (PTP Leader) or from<br>Network PTP Leader |
|                                |   |
| PoE Power                      |   |
| Standard                       | 802.3af   |
| Redundancy                     | Yes   |
| Class                          | 0   |
| PD Power Range                 | 0.44 W to 12.94 W                                   |
| Typical PSE Power<br>Usage     | 4.6W  |
|                                |   |

15.4 W

| Equipment Type         |   |
|------------------------|---|
| AVN-DIO19:             | Dante to AES3 16 channel I/O converter                              |
| Physical Specification |   |
| Dimensions (Raw):      | 14.0cm (W) x 13.6cm (D) x 4.2cm (H) 5.5" (W) x 5.4" (D) x 1.7" (H)  |
| Dimensions (Boxed):    | 17.8cm (W) x 17cm (D) x 5.6cm (H)<br>7.0" (W) x 6.7" (D) x 2.2" (H) |
| Weight:                | TBD   |
| Accessories            |   |
| AVN-DIORK:             | AVN-DIO 1U 19" rack kit (5 x small<br>DIO or 3 x large DIO)         |
| CBL-D25-4XI4XO:        | AES3 balanced cable, DB25 to 4 x<br>XLR3M and 4 x XLR3F, 3m         |
| CBL-D25-D25:           | AES3 balanced cable, DB25 to<br>DB25, 3m                            |



AVN-DIO19 Front View.



AVN-DIO19 Rear View.

Max PSE Usage









## RAVENNA



## AVN-DIO20 Dante® to MADI AFS3 64 Channel I/O Converter













Category: Dante Audio Interfaces.

Product Function: Dante® to MADI AES3 64 Channel I/O Converter.

Typical Applications: These simple plug and play audio interfaces provide a convenient and elegant method of connecting legacy analogue and digital

audio equipment to the Dante AoIP audio network.

## Features:

- 1 x AES10 MADI input and output, on either SFP or coaxial BNC (up to 64 channels of MADI I/O).
- 8 x stereo AES3 inputs and outputs on 2 x 25-way D-types, using AES59 digital pinout.
- Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCON® Ethernet connectors.

- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.
- AES67 compatible.
- Dante Domain Manager compliant.
- · Web interface for configuration.
- Clock, PoE, Network link, AES3 input lock and MADI active LEDs.
- MADI 64, 56, 28 and 16 channel support and audio sample rate support up to 192kHz.

- MADLSEP to BNC automatic failurer.
- All available Dante sample rates supported (44.1kHz to 192kHz).
- Sample rate conversion of audio between Dante and MADI/AES3.
- · Dante clock domain can be optionally synchronised from the MADI source.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and under desk mounting plate.
- 3 x units rackmount in the AVN-DIORK.

The AVN-DIO20 is a MADI and AES3 to Dante bridging device allowing MADI to connect bidirectionally with AoIP, together with 8 stereo channels of AES3. This simple plug and play audio interface provides a quick and convenient method of connecting legacy MADI equipment to the Dante AoIP audio network.



The AVN-DIO20 takes a MADI feed, sample rate converts all 64 channels and places them on the Dante network, mapped using Dante Controller. It simultaneously takes the 64 channels mapped to the device on Dante Controller, optionally sample rate converts them, and transmits them on the MADI output. The unit accepts 8 stereo AES3 inputs and place them on the Dante network, replacing 16 of the selected MADI input channels. The unit also sends 8 stereo AES3 outputs which can be sourced from the Dante channels, in blocks of 16.

The unit accepts MADI signals from either a compatible SFP module or the coaxial BNC input. The input can be selected via the built-in web GUI or, by default, it will use an SFP module if it is inserted into the unit. The coaxial BNC output is disabled while

the SFP is in use. There is also an additional automatic failover mode which, when selected, allows for automatic switching to the BNC input when the SFP signal is lost. providing glitch free audio redundancy.

The MADI output audio can be clocked from either the Dante network, in which case the output SRC is bypassed, or the recovered clock obtained from the MADI input.

It's powered using Power over Ethernet (PoE), using Neutrik EtherCON® connectors, with primary and secondary ports for power and data redundancy. The AVN-DIO20 uses the latest Audinate Dante™ chipsets so is AES67 and Dante Domain Manager™ compliant.

There are front panel LEDs to indicate network clock status, AoIP Primary and AoIP

## **AVN-DIO Audiophile Dante® Audio Interface**

Secondary link status, PoE Primary power and PoE Secondary power active. In addition there are 8 AES3 input lock status LEDs and an active MADI input indication.

A web interface is available for network and device settings, status information and firmware updates.

The AVN-DIO20 is supplied with a belt-clip, shoulder strap hooks (but not the strap) and an under desk mounting plate. Up to 3 of the AVN-DIO20 units can be rackmount in the 1U AVN-DIORK.

- 1 x AES10 MADI input and output, on either SFP or coaxial BNC (up to 64 channels of MADI I/O).
- 8 x stereo AES3 inputs and outputs on 2 x 25-way D-types, using AES59 digital pinout.
- Dual redundant Primary and Secondary Dante network ports using Neutrik EtherCON® Ethernet connectors.
- Powered via PoE (Power over Ethernet) with PoE dual redundancy.
- Fully Dante compliant device.

- AES67 compatible.
- · Dante Domain Manager compliant.
- Web interface for configuration.
- Clock, PoE, Network link, AES3 input lock and MADI active LEDs.
- MADI 64, 56, 28 and 16 channel support and audio sample rate support up to 192kHz.
- · MADI SFP to BNC automatic failover.
- All available Dante sample rates supported (44.1kHz to 192kHz).
- Sample rate conversion of audio between Dante and MADI/AES3.
- Dante clock domain can be optionally synchronised from the MADI source.
- Supplied with belt-clip, shoulder strap hooks (but not the strap) and under desk mounting plate.
- 3 x units rackmount in the AVN-DIORK.

## **Technical Specification For AVN-DIO20**

| MADI I/O           |                                    |
|--------------------|------------------------------------|
| BNC Input & Output |                                    |
| Impedance:         | 75Ω Unbalanced                     |
| MADI Input Sample  |                                    |
| Rate/Channels:     | 48kHz – 56/64 channels             |
|                    | 96kHz – 28/32 channels             |
|                    | 192kHz – 16 channels               |
| MADI Output Sample |                                    |
| Rate/Channels:     | Follow input, Dante sample rate    |
| Connections:       | Coaxial BNC input and output       |
|                    | SFP - LVDS SFP 100MB/s             |
|                    |                                    |
| AES3 I/O           |                                    |
| Input& Output      |                                    |
| Impedance:         | 110Ω Balanced                      |
| Sample Rates:      | 32 – 192kHz                        |
| Connections:       | 8 x stereo AES3 inputs and outputs |
|                    | on 2 x 25-way D-types using AES59  |
|                    | pinout                             |
|                    |                                    |
| Network and AoIP   |                                    |
| AoIP Standard:     | Dante                              |
| Channels:          | 64 receive, 64 transmit            |
| Flows:             | 32 receive, 32 transmit            |
| Sample Rates:      | 44.1kHz, 48kHz, 88.2kHz, 96kHz,    |
|                    | 176 AVH- 102VH-                    |

| Network and AoIP               |  |
|--------------------------------|--|
| AoIP Standard:                 | Dante  |
| Channels:                      | 64 receive, 64 transmit  |
| Flows:                         | 32 receive, 32 transmit  |
| Sample Rates:                  | 44.1kHz, 48kHz, 88.2kHz, 96kHz,<br>176.4kHz, 192kHz              |
| Encoding:                      | PCM 16, PCM 24, PCM 32   |
| AES67 Support:                 | Yes  |
| Connectivity:                  | 2 x etherCON (RJ45 compatible)                                   |
| Speed:                         | 1Gbps or 100Mbps   |
| Network Modes:                 | Switched or redundant  |
| Dante Domain<br>Manager Ready: | Yes  |
| Clock Source:                  | Internal (PTP Leader), Network PTP<br>Leader or from MADI input) |

| PoE Power                  |  |
|----------------------------|--|
| Standard                   | 802.3af  |
| Redundancy                 | Yes  |
| Class                      | 0  |
| PD Power Range             | 0.44 W to 12.94 W  |
| Typical PSE<br>Power Usage | 6.1W   |
| Max PSE Usage              | 15.4 W   |
| Equipment Type             |  |
| AVN-DIO20:                 | Dante to MADI/AES3 64 channel I/O converter                        |
| Physical Specification     |  |
| Dimensions (Raw):          | 14.0cm (W) x 13.6cm (D) x 4.2cm (H) 5.5" (W) x 5.4" (D) x 1.7" (H) |
| Dimensions (Boxed):        | 17.8cm (W) x 17cm (D) x 5.6cm (H) 7.0" (W) x 6.7" (D) x 2.2" (H)   |
| Weight:                    | Nett: 0.5kg Gross: 0.9kg<br>Nett: 1.1lbs Gross: 2.0lbs             |
| Accessories                |  |
| AVN-DIORK:                 | AVN-DIO 1U 19" rack kit (5 x small<br>DIO or 3 x large DIO)        |
| CBL-D25-4XI4XO:            | AES3 balanced cable, DB25 to 4 x                                   |

XLR3M and 4 x XLR3F, 3m

DB25, 3m

AES3 balanced cable, DB25 to

| SONIFEX Audiophile Dante* Interface  DIO20 |            |     | MADI  |        |
|--|------------|-----|-------|--------|
| Clock Pri PoE Sec Dante to MADI Converter  | 1-2-3-4 BN | SFP | Input | Output |



CBL-D25-D25:

AVN-DIO20 Rear View.



## **AVN-DIORK 19" AVN-DIO Mounting Rack**















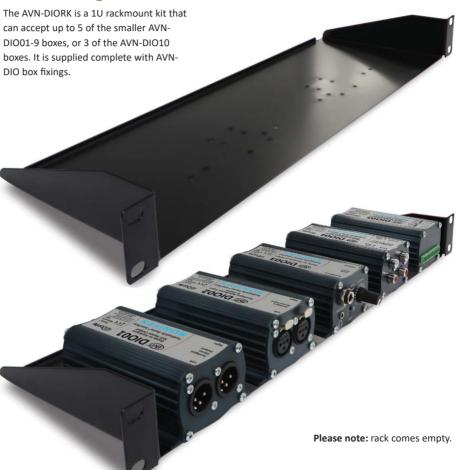


**Product Function:** A 1U rackmount kit for AVN-DIO Audiophile boxes.

Typical Applications: A 1U rackmount kit, providing a tray to hold AVN-DIO Audiophile boxes.

## Features:

- 1U rackmount tray.
- Can take up to 5 x AVN-DIO01-09 units.
- Alternatively, can take up to 3 x AVN-DIO10 sized boxes.
- Supplied with 5 sets of AVN-DIO mounting screws and nuts..











## **AVN-GMCS IEEE1588 PTP Grandmaster Clock with GPS Receiver**



Category: AES67 PTP Grandmaster Clocks.

**Product Function: IEEE1588 PTP** Grandmaster Clock with GPS Receiver.

Typical Applications: A PTPv2 grandmaster clock for use with AoIP applications.

#### Features:

- AES67/RAVENNA/AES-R16 compatible.
- · GPS satellites received indicator.
- · Master and slave sync modes.
- 8ns PTPv2 time stamping resolution.
- · Holdover drift < 1ppm, with an option for < 0.01ppm.
- AES-3id, wordclock & amp; variable PPS outputs.
- · Analogue master input.
- Dual power supply inputs.
- · Front panel display.
- GPS sync and power supply present LED indicators.
- · Real time clock for accurate date/time.
- · Responsive design Ethernet webserver and front panel control & amp; configuration.



The AVN-GMCS is a PTPv2 grandmaster clock for use with AoIP applications. IEEE1588-2008 PTPv2 (precision time protocol) is used to synchronise all the nodes within a network. To achieve this one of the nodes must become the master clock and distribute time packets to the others. The AVN-GMCS is designed to perform this role simply and accurately, enabling sub micro second synchronisation between all nodes.

RAVENNA (of which AES67 is a subset) allows for the distribution of audio across a network. For this to be possible, each of the nodes needs to be time synchronised with one another. RAVENNA uses PTP time stamping to achieve this, which distributes the network time but also works out the latency involved in the delivery and adjusts the time at each node accordingly.

Unit configuration is achieved easily either with the front panel controls or the webserver, including the setup of the PTP profiles.

The AVN-GMCS supports the Default (RAVENNA), Media (AES67) and AES-R16-2016 (SMPTE-ST 2059-2 & AES67 compatible) profiles and has a 'Custom' profile page for you to define your own.

In normal operation, the unit has PTPv2 time stamping resolution to 8nsec. It uses a combination of a GPS receiver, a PLL (phase lock loop) and a specialist onboard clock device to create the precise, low jitter clock signals required to drive the physical transceiver's time stamping circuitry, also providing holdover if the GPS signal is lost.

The specialist on board clock is available in two different types: TCXO & OXCO, which vary in both price and accuracy:

AVN-GMCS - TCXO Temperature

Compensated Oscillator accurate to 1 part per million (worst case 1 sec gain/loss every 11.5 days). \*

AVN-GMCOS - OCXO Oven Controlled Oscillator accurate to 0.1 parts per million (worst case 1 sec gain/loss every 116 days). \*

GPS presence and the number of satellites received is shown on the front panel, together with status information on output sample rates, sync type and profile type. The unit also has a screen-saver option which shows the current time.

Although designed as a grandmaster clock, a separate clock input can act as an alternative



AVN-GMCS Front View.



AVN-GMCS Rear View.

reference source to GPS which the unit can 'slave' to. Clock outputs, driven from the physical transceiver, can be used to provide media clocks for external equipment local to the AVN-GMCS when it is in both 'master' and 'slave' states. The clock outputs are available as a single AES-3id output and two outputs which can be selected as either word clock or variable PPS. The wordclock can operate at 32, 44.1, 48, 96, 176.4 and 192kHz. When set as a variable PPS output, the unit can act as a clock master to distribute a reference frequency to test and measurement equipment.

The unit shows UTC as standard, but can be set to show 'local time' on the front panel, by adding a time offset. Daylight saving time changes can be accommodated by entering Spring Forward and Fall Back dates. It has a real time clock so that accurate date and time is available even after the unit is repowered without GPS access.

The built-in webserver, or front panel OLED display, can be used to configure the unit. The webserver is a responsive design meaning that it can be used with small screens on smartphones and tablets.

Front panel LEDs show the synchronisation status, GPS lock and the status of the AC and DC power supply inputs.

The brightness of the OLED display and LED indicators can be adjusted for low or high lighting conditions 4 general purpose outputs indicate critical states for the unit using a 9 way D-type connector mounted on the rear panel. Pull down when active pins are supplied for GPS lock status, external sync present, AC power present and DC power present.

The unit has a front panel power button and dual power connectors - an IEC mains input and a 12V DC input, which allows the AVN-GMCS to be used for both studio and mobile installations. Moreover this allows for a secondary power source to reduce the effect of power down events. In any case, the unit monitors the status of both power sources and displays this on the front panel.

The unit can be put into a low-power sleep mode when not in use, with an instant start when power is re-applied. In power off situations, a super capacitor is used to keep the GPS receiver powered in a low power mode for more than 20 hours, enabling the receiver to regain lock immediately rather than having to 'cold' start.

- AES67/RAVENNA/AES-R16 compatible.
- GPS satellites received indicator.
- Master and slave sync modes.
- 8ns PTPv2 time stamping resolution.
- Holdover drift <1ppm, with an option for &lt;0.01ppm.
- AES-3id, wordclock & amp; variable PPS outputs.
- Analogue master input.
- Dual power supply inputs.
- Front panel display.
- GPS sync and power supply present LED indicators.
- Real time clock for accurate date/time.
- Responsive design Ethernet webserver and front panel control & amp; configuration.

## **Technical Specification For AVN-GMCS**

| Timing Specification:            |  |
|----------------------------------|--|
| Profile Support:                 | Default (RAVENNA), Media (AES67),<br>AES-R16-2016 (SMPTE-ST 2059-2 &<br>AES67 compatible), Custom profile      |
| Timing Protocol:                 | PTPv2, IEEE1588-2008   |
| Timing Accuracy:                 | PTP time stamping resolution 8ns   |
| Holdover Drift: TCXO:            | <90ms  |
| Holdover Drift:                  | These figures are over 24 hours at constant temperature  |
| GPS Performance:                 | 50 channel GPS receiver  |
| GPS Frequency:                   | 1575.42MHz, L1 band  |
|                                  |  |
| Clock Specification:             |  |
| Input Impedance:                 | 75Ω  |
| Output Impedance:                | <75Ω   |
| Antenna Impedance:               | 50Ω  |
|                                  |  |
| Connections:                     |  |
| Clocking Input:                  | BNC female   |
| Clocking Outputs:                | 3 x BNC female Wordclock, AES-3id<br>@ 32, 44.1, 48, 96, 176.4 & 192kHz<br>Variable PPS (1, 10, 100, 1000) TTL |
| GPS Input:                       | SMA socket   |
| GPIO:                            | D-Type female 9 way  |
| Ethernet Port:                   | RJ45 socket, 100 BASE-T  |
| Mains AC Input:                  | Universal filtered IEC socket,<br>continuously rated 85-264 VAC<br>@47-63Hz, max 10W                           |
| DC Input:                        | 1 x 12V, KPJX-45 socket, positive pins 1 and 3   |
| Maximum Operating<br>Range (DC): | 10.3V to 13.2V DC  |
|                                  |  |

| Equipment Type:        |   |
|------------------------|---|
| AVN-GMCS               | Grandmaster clock for PTP systems,<br>GPS, IP, TCXO, 1ppm, rackmount          |
| Physical Specification | :   |
| Dimensions (Raw):      | 4.4cm (H) x 48.3cm (W) x 17.8cm<br>(D(1U)<br>1.8" (H) x 19" (W) x 7" (D) (1U) |
| Dimensions (Boxed):    | 6.8cm (H) x 58.8cm (W) x 27cm (D)<br>2.7" (H) x 23" (W) x 10.6" (D)           |
| Weight:                | Nett: 1.5kg Gross: 2.2kg<br>Nett: 3.2lbs Gross: 4.8lbs                        |
| AVN-DC150              | 150W DC power supply with<br>KPJX-4S plug                                     |









## AVN-GMCOS IEEE1588 PTP Grandmaster Clock with GPS Receiver



Category: AES67 PTP Grandmaster Clocks.

**Product Function: IEEE1588 PTP** Grandmaster Clock with GPS Receiver.

Typical Applications: A PTPv2 grandmaster clock for use with AoIP applications.

#### Features:

- AES67/RAVENNA/AES-R16 compatible.
- · GPS satellites received indicator.
- Master and slave sync modes.
- 8ns PTPv2 time stamping resolution.
- Holdover drift < 1ppm, with an option for < 0.01ppm.
- AES-3id, wordclock & amp; variable PPS outputs.<
- · Analogue master input.
- Dual power supply inputs.
- Front panel display.
- GPS sync and power supply present LED indicators.
- · Real time clock for accurate date/time.
- Responsive design Ethernet webserver and front panel control & amp; configuration.



The AVN-GMCOS is a PTPv2 grandmaster clock for use with AoIP applications which uses a temperature controlled crystal for GPS holdover accuracy. IEEE1588-2008 PTPv2 (precision time protocol) is used to synchronise all the nodes within a network. To achieve this one of the nodes must become the master clock and distribute time packets to the others. The AVN-GMCOS is designed to perform this role simply and accurately, enabling sub micro second synchronisation between all nodes.

RAVENNA (of which AES67 is a subset) allows for the distribution of audio across a network. For this to be possible, each of the nodes needs to be time synchronised with one another. RAVENNA uses PTP time stamping to achieve this, which distributes the network time but also works out the latency involved in the delivery and adjusts the time at each node accordingly.

Unit configuration is achieved easily either with the front panel controls or the webserver, including the setup of the PTP profiles.

The AVN-GMCOS supports the Default (RAVENNA), Media (AES67) and AES-R16-2016 (SMPTE-ST 2059-2 & AES67 compatible) profiles and has a 'Custom' profile page for you to define your own.

In normal operation, the unit has PTPv2 time stamping resolution to 8nsec. It uses a combination of a GPS receiver, a PLL (phase lock loop) and a specialist on-board clock device to create the precise, low jitter clock signals required to drive the physical transceiver's time stamping circuitry, also providing holdover if the GPS signal is lost.

The specialist on board clock is available in two different types: TCXO & OXCO, which vary in both price and accuracy:

**AVN-GMCS - TCXO** Temperature Compensated Oscillator accurate to 1 part per million (worst case 1 sec gain/loss every 11.5 days). \*

AVN-GMCOS - OCXO Oven Controlled Oscillator accurate to 0.1 parts per million (worst case 1 sec gain/loss every 116 days).

GPS presence and the number of satellites received is shown on the front panel. together with status information on output sample rates, sync type and profile type. The unit also has a screen-saver option which shows the current time.

Although designed as a grandmaster clock, a separate clock input can act as an alternative



AVN-GMCOS Front View.



AVN-GMCOS Rear View.

reference source to GPS which the unit can 'slave' to. Clock outputs, driven from the physical transceiver, can be used to provide media clocks for external equipment local to the AVN-GMCOS when it is in both 'master' and 'slave' states. The clock outputs are available as a single AES-3id output and two outputs which can be selected as either word clock or variable PPS. The wordclock can operate at 32, 44.1, 48, 96, 176.4 and 192kHz. When set as a variable PPS output, the unit can act as a clock master to distribute a reference frequency to test and measurement equipment.

The unit shows UTC as standard, but can be set to show 'local time' on the front panel, by adding a time offset. Daylight saving time

changes can be accommodated by entering Spring Forward and Fall Back dates. It has a real time clock so that accurate date and time is available even after the unit is repowered without GPS access.

The built-in webserver, or front panel OLED display, can be used to configure the unit. The webserver is a responsive design meaning that it can be used with small screens on smartphones and tablets.

Front panel LEDs show the synchronisation status, GPS lock and the status of the AC and DC power supply inputs.

The brightness of the OLED display and LED indicators can be adjusted for low or high lighting conditions 4 general purpose

outputs indicate critical states for the unit using a 9 way D-type connector mounted on the rear panel. Pull down when active pins are supplied for GPS lock status, external sync present, AC power present and DC power present.

The unit has a front panel power button and dual power connectors - an IEC mains input and a 12V DC input, which allows the AVN-GMCOS to be used for both studio and mobile installations. Moreover this allows for a secondary power source to reduce the effect of power down events. In any case, the unit monitors the status of both power sources and displays this on the front panel.

The unit can be put into a low-power sleep mode when not in use, with an instant start

when power is re-applied. In power off situations, a super capacitor is used to keep the GPS receiver powered in a low power mode for more than 20 hours, enabling the receiver to regain lock immediately rather than having to 'cold' start.

- AES67/RAVENNA/AES-R16 compatible.
- · GPS satellites received indicator.
- Master and slave sync modes.
- 8ns PTPv2 time stamping resolution.
- Holdover drift < 1ppm, with an option for < 0.01ppm.
- AES-3id, wordclock & amp; variable PPS outputs.

- Analogue master input.
- Dual power supply inputs.
- Front panel display.
- GPS sync and power supply present LED indicators.
- Real time clock for accurate date/time.
- Responsive design Ethernet webserver and front panel control & amp; configuration.

## **Technical Specification For AVN-GMCOS**

Timing Specification:

| riming specification:    |  |
|--------------------------|--|
| Profile Support:         | Default (RAVENNA), Media (AES67),<br>AES-R16-2016 (SMPTE-ST 2059-2 &<br>AES67 compatible), Custom profile      |
| Timing Protocol:         | PTPv2, IEEE1588-2008   |
| Timing Accuracy:         | PTP time stamping resolution 8ns   |
| Holdover Drift:<br>OCXO: | <900μs   |
| Holdover Drift:          | These figures are over 24 hours at constant temperature  |
| GPS Performance:         | 50 channel GPS receiver  |
| GPS Frequency:           | 1575.42MHz, L1 band  |
|                          |  |
| Clock Specification:     |  |
| Input Impedance:         | 75Ω  |
| Output Impedance:        | <75Ω   |
| Antenna Impedance:       | 50Ω  |
|                          |  |
| Connections:             |  |
| Clocking Input:          | BNC female   |
| Clocking Outputs:        | 3 x BNC female Wordclock, AES-3id<br>@ 32, 44.1, 48, 96, 176.4 & 192kHz<br>Variable PPS (1, 10, 100, 1000) TTL |
| GPS Input:               | SMA socket   |

| GPIO:                         | D-Type female 9 way  |
|-------------------------------|--|
| Ethernet Port:                | RJ45 socket, 100 BASE-T  |
| Mains AC Input:               | Universal filtered IEC socket,<br>continuously rated 85-264 VAC<br>@47-63Hz, max 10W |
| DC Input:                     | 1 x 12V, KPJX-45 socket, positive pins 1 and 3                                       |
| Maximum Operating Range (DC): | 10.3V to 13.2V DC  |
| Equipment Type:               |  |
| AVN-GMCOS                     | Grandmaster clock for PTP systems, GPS, IP, OCXO, 0.01ppm, rackmount                 |
| Physical Specification        | <u> </u>   |
| Dimensions (Raw):             | 4.4cm (H) x 48.3cm (W) x 17.8cm (D)(1U) 1.8" (H) x 19" (W) x 7" (D) (1U)             |
| Dimensions (Boxed):           | 6.8cm (H) x 58.8cm (W) x 27cm (D)<br>2.7" (H) x 23" (W) x 10.6" (D)                  |
| Weight:                       | Nett: 1.5kg Gross: 2.2kg<br>Nett: 3.2lbs Gross: 4.8lbs                               |
| AVN-DC150                     | 150W DC power supply with<br>KPJX-4S plug  |









## AVN-TB10AR 10 Button Advanced Talkback Intercom. AoIP Portal



AVN-TB10AR Front View.



Category: AES67 Talkback, IFB, Intercom & Commentary.

Product Function: An advanced talkback/ listening/paging intercom unit to enable voice/audio communication between different areas in a facility or building complex.

Typical Applications: OB truck comms. theatre comms. inter-studio comms in a TV or radio station.

#### Features:

- 10 illuminated key-cap Talk buttons plus Listen & amp; Page buttons.
- Mic & amp; headset inputs, headphone &amp: speaker outputs with volume control.

- Sources from AoIP, balanced, 2 x unbalanced or S/PDIF digital inputs.
- Destinations to AoIP or rear panel balanced & amp: unbalanced outputs.
- Advanced echo cancellation & amp: mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & amp: 1Gb SFP fibre port.
- 10 user assignable GPIO ports.
- Dual AC & amp: DC power supply inputs.
- Front panel display providing source &amp: destination information.
- Ethernet webserver and front panel control & amp: configuration.
- Speaker & amp; microphone mute buttons.
- Callback button with source display.

The AVN-TB products are IP audio based talkback intercom units with an advanced feature set, allowing them to be used in multiple applications. With both 4-wire analogue inputs and outputs, as well as AoIP network audio connectivity, the AVN-TB units can be used with existing legacy 4-wire systems and with new AES67 AoIP networked audio infrastructure. Additionally, all audio is at 48kHz sample rate, meaning that it's broadcast quality audio as standard.

RAVENNA (of which AES67 is a subset) allows for the distribution of audio across a network. The AVN range use RAVENNA as the communication method providing compatibility with other AES67 systems.

The AVN-TB10AR is a 10 button intercom meaning that 10 other 'stations' can be defined, one per button, for communication. Comms can be made as a Talk action, a Listen action or a duplex Talk/Listen action to/from each station. Coloured LEDs in the buttons help to show which action is being used and there is also a Callback button for when you're unavailable to receive a call.

The stations can be from anywhere on the AoIP network and the use of Bonjour Device Discovery means that other stations can be found quickly and sometimes automatically.

The Page button is used to speak to all stations (or a defined list of stations) and Group Talk functions can be enabled to page particular groups of stations.

Two monitor buttons allow for routing audio directly to the speaker e.g. to take an IFB feed or an off-air transmission signal. Signals can be ducked or mixed when a talkback input is received to the speakers or headphones.

Three user defined buttons can be programmed for different functions, such as for Group Talk.

The speaker mutes automatically when headphones are inserted and the volume level of headphones, speaker and incoming sources can all be controlled with one front



AVN-TB10AR Rear View.

panel rotary encoder volume control knob, which shows the level using RGB LEDs around the outside of the knob.

Advanced acoustic echo cancellation & built-in microphone AGC (automatic gain control) ensure that there's no acoustic feedback between microphone and speaker.

Buttons are available for microphone mute (cough) and speaker mute actions and these can be controlled remotely by GPI or network commands.

Each unit has a built-in webserver which is where the majority of settings and configurations are made. The front panel OLED display can also be used to configure the unit, although more functionality is available by using the webserver. The webserver is a responsive design meaning that it can be used with small screens on smartphones and tablets.

The unit can act as a PTP masterclock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

Front panel LEDs show the AoIP network status, synchronisation status, whether AGC is being used and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be adjusted for low or high lighting conditions.

The unit has a front panel power button and dual power connectors - an IEC mains input and a 12V DC input, which allows the AVN-TB10AR to be used for both studio and mobile installations. Also, a secondary power source reduces the effect of power down events. In any case, the unit monitors the status of both power sources and displays this on the front panel.

10 GPIOs (general purpose inputs/outputs) and a programmable relay output can be configured to indicate critical states for the unit, via the 15 way D-type connector, for example, to show loss of DC power, or to show a button press action.

## **Webserver Software**

Each AVN-TB talkback unit has a built-in

webserver for setup and configuration. The webserver is responsive, and resizes depending on the size of your screen, meaning that it can be used on large monitors or small handheld devices such as smart-phones. Help information is shown on the right hand side of the screen so it's a good place to go to find out how the unit operates.

- 10 illuminated key-cap Talk buttons plus Listen & Damp; Page buttons.
- Mic & Description
   Mic & Description<
- Sources from AoIP, balanced, 2 x unbalanced or S/PDIF digital inputs.
- Destinations to AoIP or rear panel balanced & mp; unbalanced outputs.
- Advanced echo cancellation & mp; mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & Dual 1Gb SFP fibre port.

- 10 user assignable GPIO ports.
- Dual AC & amp; DC power supply inputs.
- Front panel display providing source & amp; destination information.
- Ethernet webserver and front panel control & amp: configuration.
- Speaker & microphone mute buttons.
- · Callback button with source display.

## **Technical Specification For AVN-TB10AR**

## Audio-Over-IP Specification

| Addio Over ii Speciii | Julion                      |
|-----------------------|-----------------------------|
| Open Standards:       | RAVENNA, AES67              |
| Device Discovery:     | Bonjour (mDNS / DNS-SD)     |
| Audio Delivery:       | RTP/UDP over IPv4 multicast |
| QoS:                  | DiffServ                    |
| Stream Management:    | RTSP/SDP                    |
| Control:              | Ember+/webserver            |
| Format:               | Linear PCM 24-bit (L24)     |
| Channels Per Stream:  | 2                           |
| Frames Per Packet:    | 48                          |
| Maximum Streams:      | RX 6, TX 5 (fixed)          |
| Sample Rate:          | 48 kHz                      |
| ,                     |                             |
|                       |                             |

## **Timing Synchronisation**

| Profile Support: | Default, media & custom profiles |
|------------------|----------------------------------|
| Timing Protocol: | PTPv2, IEEE1588-2008             |

## Technical Specification

## Microphone and Headset Input

| Input Impedance:    | >2.5kΩ balanced                                 |
|---------------------|---|
| Gain Range:         | 0dB to +60dB                                    |
| OdBFS Line-Up:      | Adjustable in steps of 3dB from -58dBu to +2dBu |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                        |
| Noise:              | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain    |

## Unbalanced Line Inputs

| Input Impedance:    | >2UKΩ  |
|---------------------|--|
| OdBFS Line-Up:      | +12dBu   |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                               |
| THD+N:              | <-97dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:              | -100dBFS, 20kHz BW, Rs=200Ω                            |
| Crosstalk:          | <-97dB   |
|                     |  |

## Balanced Line Inputs

| Input | Impedance: | >20kΩ balanced |
|-------|------------|----------------|
|       |            |                |

| OdBFS Line-Up:            | Adjustable +15/+18/+20/+22/<br>+24dBu                   |
|---------------------------|---|
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                                |
| THD+N:                    | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW             |
| Noise:                    | -110dBFS, 20kHz BW, Rs=200Ω                             |
| Crosstalk:                | <-100dB   |
| Common Mode<br>Rejection: | >70dB @ 1kHz  |
| Headphone Output          |   |
| Output Impedance:         | Drives 150mW into $32\Omega$ to $600\Omega$ headphones  |
| OdBFS Line-Up:            | +20dBu  |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                                |
| THD+N:                    | <-108dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:                    | -110dBFS, 20kHz BW                                      |
| Unbalanced Line Out       | outs  |
| Output Impedance:         | <50Ω  |
| OdBFS Line-Up:            | +12dBu  |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                                |
| THD+N:                    | <-95dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW  |
| Noise:                    | -100dBFS, 20kHz BW                                      |
| Balanced Line Output      | s   |
| Output Impedance:         | <50Ω balanced   |
| OdBFS Line-Up:            | Adjustable +15/+18/+20/+22/<br>+24dBu                   |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                                |
| TUDAN                     | <-110dBFS, -30dBFS, 20Hz to 20kHz                       |
| THD+N:                    | 20kHz BW  |

4W

Mute to full volume via front panel

Power Output:

Volume:

## Connections

| Microphone:         | XLR-3 pin female (electronically balanced)                         |
|---------------------|--|
| Headphone:          | ¼ inch (6.35mm) stereo jack socket                                 |
| Headset:            | 2 x XLR-5 pin female (front/rear electronically balanced input)    |
| Audio Inputs:       | 2 x unbalanced stereo, RCA phono                                   |
| Audio Inputs:       | 1 x S/PDIF, RCA phono  |
| Audio Outputs:      | 1 x balanced stereo, RJ45  |
| Audio Outputs:      | 1 x unbalanced stereo, 2 x RCA phono                               |
| Audio Outputs:      | 1 x loudspeaker output   |
| Audio Input/Output: | 1 x balanced stereo input or mono input/output, RJ45               |
| GPIO:               | 15-way 'D'-type socket   |
| Network:            | 2 x gigabit Ethernet, RJ45   |
| Network:            | 1 x SFP fibre  |
| Mains AC Input:     | Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W |
| DC Input:           | 4 pin 7.5A power jack socket,<br>9.5-14VDC                         |
| Fuse Rating:        | Anti-surge fuse 2A 20mm x 5mm                                      |
|                     |  |
|                     |  |

## Equipment Type AVN-TB10AR

310AR 10 channel rackmount talkback intercom control unit with RAVENNA AOIP

## Physical Specification

| Dimensions: (Raw)   | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U) |
|---------------------|--|
| Dimensions (Boxed): | 59cm (W) x 28cm (D) x 11cm (H)<br>23" (W) x 11" (D) x 4.3" (H)             |
| Weight:             | Nett: 2.4kg Gross: 3.1kg<br>Nett: 5.3lbs Gross: 6.8lbs                     |

## Accessories

| AVN-DC150 | KPJX-4S plug                     |
|-----------|----------------------------------|
| CM-GM2:   | Professional Gooseneck Condenser |







## RAVENNA



## AVN-TB20AR 20 Button Advanced Talkback Intercom. AoIP Portal



AVN-TB20AR Front View.





Product Function: An advanced talkback/ listening/paging intercom unit to enable voice/audio communication between different areas in a facility or building complex.

Typical Applications: OB truck comms, theatre comms, inter-studio comms in a TV or radio station.

## Features:

- 20 illuminated key-cap Talk buttons plus Listen & Page buttons.
- · Phone button for remote dialling and control of an external telephone hybrid.

- Page button and Group Talk facilities.
- · Callback button with callback source display.
- Three user definable buttons.
- Speaker & microphone mute buttons.
- Mic & headset inputs (front & rear panel headset connection), headphone & speaker outputs.
- Front panel volume control which operates on speaker/headphone outputs and incoming source levels.
- +48V phantom power for the mic inputs.
- Ethernet webserver and front panel control & configuration.
- Front panel display providing source & destination information.
- Sources from AoIP, 1 x balanced, 2 x unbalanced or S/PDIF digital inputs.

- Destinations to AoIP or rear panel balanced & unbalanced outputs.
- · Advanced echo cancellation & mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & 1Gb SFP fibre port.
- 10 user assignable GPIO ports.
- GPI/O button for triggering external events, via physical GPIO or network commands.
- Front panel LEDs for network audio presence, Talk activity, AGC activity, clock sync and power supply activity.
- Two front panel monitor buttons for routing audio directly to the speaker e.g. to take an IFB feed or off-air transmission signal.
- Ducking or mixing of inputs to speaker/ headphones.
- Dual AC & DC power supply inputs.

The AVN-TB20AR is a 2U rackmount 20 button version of the AVN-TB10AR with the same specification, but more station buttons allowing greater communication for larger facilities. It also has 'GPI/O' and 'Phone' buttons for remote control of external equipment and Sonifex DHY-04 telephone hybrids, useful in an OB truck or production gallery.

"The AVN-TB20AD is a desktop version of the rackmount AVN-TB20AR intercom with a smaller form factor and an elegant sloped front. It has the same feature set and connectivity \*.

- \* Except for an additional headset connector on the rear of the AVN-TB20AR.
- 20 illuminated key-cap Talk buttons plus Listen & Page buttons.



AVN-TB20AR Front View.

- Phone button for remote dialling and control of an external telephone hybrid.
- · Page button and Group Talk facilities.
- Callback button with callback source display.
- · Three user definable buttons.
- · Speaker & microphone mute buttons.
- Mic & headset inputs (front & rear panel headset connection), headphone & speaker outputs.
- Front panel volume control which operates on speaker/headphone outputs and incoming source levels.
- +48V phantom power for the mic inputs.

- Ethernet webserver and front panel control & configuration.
- Front panel display providing source & destination information.
- Sources from AoIP, 1 x balanced, 2 x unbalanced or S/PDIF digital inputs.
- Destinations to AoIP or rear panel balanced & unbalanced outputs.
- Advanced echo cancellation & mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & 1Gb SFP fibre port.
- 10 user assignable GPIO ports.
- GPI/O button for triggering external events, via physical GPIO or network commands.

- Front panel LEDs for network audio presence, Talk activity, AGC activity, clock sync and power supply activity.
- Two front panel monitor buttons for routing audio directly to the speaker e.g. to take an IFB feed or off-air transmission signal.
- Ducking or mixing of inputs to speaker/ headphones.
- Dual AC & DC power supply inputs.

## **Technical Specification For AVN-TB20AR**

## Audio-Over-IP Specification

| Open Standards:      | RAVENNA, AES67              |
|----------------------|-----------------------------|
| Device Discovery:    | Bonjour (mDNS / DNS-SD)     |
| Audio Delivery:      | RTP/UDP over IPv4 multicast |
| QoS:                 | DiffServ                    |
| Stream Management:   | RTSP/SDP                    |
| Control:             | Ember+/webserver            |
| Format:              | Linear PCM 24-bit (L24)     |
| Channels Per Stream: | 2                           |
| Frames Per Packet:   | 48                          |
| Maximum Streams:     | RX 6, TX 5 (fixed)          |
| Sample Rate:         | 48 kHz                      |
|                      |                             |

## **Timing Synchronisation**

| Profile Support: | Default, media & custom profiles |
|------------------|----------------------------------|
| Timing Protocol: | PTPv2, IEEE1588-2008             |

## Technical Specification

## Microphone and Headset Input

| Input Impedance:    | >2.5kΩ balanced                                 |
|---------------------|---|
| Gain Range:         | 0dB to +60dB                                    |
| OdBFS Line-Up:      | Adjustable in steps of 3dB from -58dBu to +2dBu |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                        |
| Noise:              | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain    |

## **Unbalanced Line Inputs**

| Input Impedance:    | >20kΩ  |
|---------------------|--|
| OdBFS Line-Up:      | +12dBu   |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                               |
| THD+N:              | <-97dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:              | -100dBFS, 20kHz BW, Rs=200Ω                            |
| Crosstalk:          | <-97dB   |

## **Balanced Line Inputs**

| Input Impedance:          | >20kΩ balanced                              |
|---------------------------|---|
| OdBFS Line-Up:            | Adjustable +15/+18/+20/+22/<br>+24dBu       |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                    |
| THD+N:                    | <-110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW |
| Noise:                    | -110dBFS, 20kHz BW, Rs=200Ω                 |
| Crosstalk:                | <-100dB                                     |
| Common Mode<br>Rejection: | >70dB @ 1kHz                                |

## Headphone Output

| Output Impedance:<br>headphones | Drives 150mW into 32 $\Omega$ to 600 $\Omega$           |
|---------------------------------|---|
| OdBFS Line-Up:                  | +20dBu  |
| Frequency Response:             | 20Hz to 20kHz, +0/-0.2dB                                |
| THD+N:                          | <-108dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:                          | -110dBFS, 20kHz BW                                      |

## Unbalanced Line Outputs

| Output Impedance:   | <50Ω   |
|---------------------|--|
| OdBFS Line-Up:      | +12dBu   |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                               |
| THD+N:              | <-95dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:              | -100dBFS, 20kHz BW                                     |

## **Balanced Line Outputs**

| Output Impedance:   | <50Ω balanced                                  |
|---------------------|--|
| OdBFS Line-Up:      | Adjustable +15/+18/+20/+22/<br>+24dBu          |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                       |
| THD+N:              | <-110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW |

| Noise.        | -1100BF3, 20KH2 BW, KS=20012                |
|---------------|---|
| Loudspeaker   |   |
| Power Output: | 4W  |
| Volume:       | Mute to full volume via front panel control |
|               |   |

balanced)

440-IDEC 201-11- DW/ D- 2000

XLR-3 pin female (electronically

2 x XLR-5 pin female (front/rear

15-way 'D'-type socket

2 x gigabit Ethernet, RJ45 1 x SFP fibre

¼ inch (6.35mm) stereo jack socket

## Connections Microphone:

Headphone: Headset:

GPIO:

Network:

Network:

|     |                     | electronically balanced input)                       |
|-----|---------------------|--|
|     | Audio Inputs:       | 2 x unbalanced stereo, RCA phono                     |
|     | Audio Inputs:       | 1 x S/PDIF, RCA phono                                |
|     | Audio Outputs:      | 1 x balanced stereo, RJ45                            |
| _   | Audio Outputs:      | 1 x unbalanced stereo, 2 x RCA phono                 |
| lz, | Audio Outputs:      | 1 x loudspeaker output                               |
| _   | Audio Input/Output: | 1 x balanced stereo input or mono input/output, RJ45 |
|     |                     |  |

#### Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W DC Input: 4 pin 7.5A power jack socket, 9.5-14VDC Fuse Rating: Anti-surge fuse 2A 20mm x 5mm

## **Equipment Type**

| AVN-TB2UAR | 20 channel rackmount talkbac |
|------------|------------------------------|
|            | intercom control unit with   |
|            | RAVENNA AoIP                 |
|            |                              |

## **Physical Specification**

| Dimensions: (Raw)        | 48.3cm (W) x 17.5cm (D) x 8.8cm (H)(2U)<br>19" (W) x 6.9" (D) x 3.6" (H) (2U) |
|--------------------------|---|
| Dimensions (Boxed):      | 59cm (W) x 28cm (D) x 15cm (H)<br>23" (W) x 11" (D) x 5.9" (H)                |
| Weight:                  | Nett: 3.7kg Gross: 4.4kg<br>Nett: 8.1lbs Gross: 9.7lbs                        |
| Accessories<br>AVN-DC150 | 150W DC power supply with   |

## KPJX-4S plug Professional Gooseneck Condenser Microphone CM-GM2:



# AES67 now!





## AVN-TB20AD 20 Button Advanced Talkback Intercom, AoIP Desktop Portal

















Product Function: An advanced talkback/ listening/paging intercom unit to enable voice/audio communication between different areas in a facility or building complex.

Typical Applications: OB truck comms. theatre comms, inter-studio comms in a TV or radio station.

## Features:

- 20 illuminated key-cap Talk buttons plus Listen & amp: Page buttons.
- Phone button for remote dialling and control of an external telephone hybrid.
- Page button and Group Talk facilities.
- Callback button with callback source display.
- Three user definable buttons.
- Speaker & amp; microphone mute buttons.
- Mic & amp; headset inputs, headphone & speaker outputs.
- Front panel volume control which operates on speaker/headphone outputs and incoming source levels.

- +48V phantom power for the mic inputs.
- Ethernet webserver and front panel control & amp; configuration.
- Front panel display providing source & amp; destination information.
- Sources from AoIP, 1 x balanced, 2 x unbalanced or S/PDIF digital inputs.
- Destinations to AoIP or rear panel balanced & amp; unbalanced outputs.
- Advanced echo cancellation & amp; mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & amp; 1Gb SFP fibre port.
- 10 user assignable GPIO ports.<
- · GPIO button for triggering external events, via physical GPIO or network commands.
- Front panel LEDs for network audio presence, Talk activity, AGC activity, clock sync and power supply activity.
- Two front panel monitor buttons for routing audio directly to the speaker e.g. to take an IFB feed or off-air transmission signal.
- Ducking or mixing of inputs to speaker/ headphones.
- Dual AC & amp; DC power supply inputs.



AVN-TR20AD Front View

The AVN-TB20AD is a desktop version of the rackmount AVN-TB20AR intercom with a smaller form factor and an elegant sloped front. It has the same feature set and connectivity \*.

- \* Except for an additional headset connector on the rear of the AVN-TB20AR.
- 20 illuminated key-cap Talk buttons plus Listen & amp; Page buttons.
- Phone button for remote dialling and control of an external telephone hybrid.
- Page button and Group Talk facilities.
- Callback button with callback source display.

- · Three user definable buttons.
- Speaker & amp; microphone mute buttons.
- Mic & amp; headset inputs, headphone & speaker outputs.
- · Front panel volume control which operates on speaker/headphone outputs and incoming source levels.



## AVN-TB20AD Rear View.

- +48V phantom power for the mic inputs.
- Ethernet webserver and front panel control & Damp; configuration.
- Front panel display providing source & Description
   & Section 1
   & Section 2
   & Section 3
   & Section 3
- Sources from AoIP, 1 x balanced, 2 x unbalanced or S/PDIF digital inputs.
- Destinations to AoIP or rear panel balanced & Destination balanced outputs.
- Advanced echo cancellation & amp; mic AGC to prevent acoustic feedback.
- Dual 1Gb lan ports & Dual 1Gb SFP fibre port.

- 10 user assignable GPIO ports.<
- GPIO button for triggering external events, via physical GPIO or network commands.
- Front panel LEDs for network audio presence, Talk activity, AGC activity, clock sync and power supply activity.
- Two front panel monitor buttons for routing audio directly to the speaker e.g. to take an IFB feed or off-air transmission signal.
- Ducking or mixing of inputs to speaker/ headphones.
- Dual AC & amp; DC power supply inputs.

## **Technical Specification For AVN-TB20AD**

| Open Standards:      | RAVENNA, AES67              |
|----------------------|-----------------------------|
| Device Discovery:    | Bonjour (mDNS / DNS-SD)     |
| Audio Delivery:      | RTP/UDP over IPv4 multicast |
| QoS:                 | DiffServ                    |
| Stream Management:   | RTSP/SDP                    |
| Control:             | Ember+/webserver            |
| Format:              | Linear PCM 24-bit (L24)     |
| Channels Per Stream: | 2                           |
| Frames Per Packet:   | 48                          |
| Maximum Streams:     | RX 6, TX 5 (fixed)          |
| Sample Rate:         | 48 kHz                      |

| tion                             |
|----------------------------------|
| Default, media & custom profiles |
| PTPv2, IEEE1588-2008             |
|                                  |
|                                  |

| adset Input  |
|--|
| >2.5kΩ balanced                                    |
| 0dB to +60dB                                       |
| Adjustable in steps of 3dB from<br>-58dBu to +2dBu |
|  |

| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB                               |
|-----------------------------|--|
| Noise:                      | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain           |
| Input Impedance:            | >20kΩ  |
|                             |  |
| Unbalanced Line Inpu        | its  |
| OdBFS Line-Up:              | +12dBu   |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB                               |
| THD+N:                      | <-97dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |
| Noise:                      | -100dBFS, 20kHz BW, Rs=200Ω                            |
| Crosstalk:                  | <-97dB   |
|                             |  |
| <b>Balanced Line Inputs</b> |  |
| Input Impedance:            | >20kΩ balanced   |
| OdBFS Line-Up:              | Adjustable +15/+18/+20/+22/<br>+24dBu                  |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB                               |
| THD+N:                      | <-110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW         |
| Noise:                      | -110dBFS, 20kHz BW, Rs=200Ω                            |
| Crosstalk:                  | <-100dB  |
| Common Mode<br>Rejection:   | >70dB @ 1kHz   |
|                             |  |

anni Dasnansa. 2011-ta 2011- 10/0 2dD

## Technical Specification For AVN-TB20AD continued...

Mute to full volume via front panel

| Headphone Output     |  |
|----------------------|--|
| Output Impedance:    | Drives 150mW into 32 $\Omega$ to 600 $\Omega$ headphones   |
| OdBFS Line-Up:       | +20dBu   |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                                   |
| THD+N:               | <-108dBFS, -30dBFS, 20Hz to<br>20kHz, unity gain, 20kHz BW |
| Noise:               | -110dBFS, 20kHz BW   |
|                      |  |
| Unbalanced Line Outp | outs   |
| Output Impedance:    | <50Ω   |
| OdBFS Line-Up:       | +12dBu   |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                                   |
| THD+N:               | <-95dBFS, -30dBFS, 20Hz to 20kHz<br>unity gain, 20kHz BW   |
| Noise:               | -100dBFS, 20kHz BW   |
|                      |  |
| Balanced Line Output |  |
| Output Impedance:    | <50Ω balanced  |
| OdBFS Line-Up:       | Adjustable +15/+18/+20/+22/<br>+24dBu                      |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                                   |
| THD+N:               | <-110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW             |
|                      | -110dBFS, 20kHz BW, Rs=200Ω                                |

| Connections            |   |
|------------------------|---|
| Microphone:            | XLR-3 pin female (electronically balanced)                                |
| Headphone:             | ¼ inch (6.35mm) stereo jack socket  |
| Headset:               | 1 x XLR-5 pin female (front electronically balanced input)                |
| Audio Inputs:          | 2 x unbalanced stereo, RCA phono  |
| Audio Inputs:          | 1 x S/PDIF, RCA phono   |
| Audio Outputs:         | 1 x balanced stereo, RJ45   |
| Audio Outputs:         | 1 x unbalanced stereo, 2 x RCA phono                                      |
| Audio Outputs:         | 1 x loudspeaker output  |
| Audio Input/Output:    | 1 x balanced stereo input or mono input/output, RJ45                      |
| GPIO:                  | 15-way 'D'-type socket  |
| Network:               | 2 x gigabit Ethernet, RJ45  |
| Network:               | 1 x SFP fibre   |
| Mains AC Input:        | Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W        |
| DC Input:              | 4 pin 7.5A power jack socket,<br>9.5-14VDC                                |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm   |
| Equipment Type         |   |
| AVN-TB20AD             | 20 channel desktop talkback<br>intercom control unit with<br>RAVENNA AOIP |
| Physical Specification |   |
| Dimensions: (Raw)      | 29.4cm (W) x 16.5cm (D) x 8.5cm (H) 11.6" (W) x 6.5" (D) x 3.3" (H)       |
|                        |   |

| Weight:     | 16" (W) x 11" (D) x 5.9" (H)                           |
|-------------|--|
|             | Nett: 2.5kg Gross: 3.2kg<br>Nett: 5.5lbs Gross: 7.0lbs |
| Accessories |  |
| AVN-DC150   | 150W DC power supply with<br>KPJX-4S plug              |
| CM-GM2:     | Professional Gooseneck Condenser<br>Microphone         |

Power Output: Volume:









## **AVN-TB6 6 Button Talkback Intercom**



Category: AES67 Talkback, IFB, Intercom & Commentary,

**Product Function:** Provides broadcast quality audio communication using RAVENNA/AES67 as the transport mechanism, allowing simple CAT 5 cabling and expansion.

Typical Applications: Ideal for broadcast quality audio communication between studios, offices and different areas in a facility or building complex.

#### Features:

- 6 illuminated key-cap Talk buttons plus Listen & amp; Page buttons.
- Dual 1Gb Ethernet & Dual 1Gb SFP port.
- Mic & amp; headset inputs, headphone & speaker outputs with volume control.
- Loudspeaker & Direction States
- Dual AC & Dower supply inputs.
- Advanced echo cancellation & amp; mic AGC to prevent acoustic feedback.
- 10 user assignable GPIO ports.
- Responsive design Ethernet webserver.
- AVN-TB6RK 19&rdguo; rack kit available.



AVN-TB6 Front View.

The AVN-TB6 is a 6 channel talkback intercom control unit from the Sonifex AVN range of IP based products. The AVN-TB6 is a freestanding version which can be rackmounted with the AVN-TB6RK 1U rack kit. There is also a desktop version, the AVN-TB6D which can be fitted flush into a work surface if required. Both models offers similar functionality.

This unit provides broadcast quality audio communication between studios, offices and different areas in a facility or building complex, using RAVENNA/AES67 as the transport mechanism, allowing simple CAT 5 cabling and expansion. RAVENNA (of which AES67 is a subset) allows for the distribution of audio across a network. The AVN range use RAVENNA as the communication method providing compatibility with other AES67 systems.

Each of the 6 channels on the AVN-TB6 can be configured to provide communications

with other remote networked units, and an independently configurable 'page' function can contact selected units with priority over standard intercom calls if required.

There is a monitor channel that can route the audio from an AoIP source to the headphones and speaker. This could be used to take an IFB feed or an off-air transmission signal or simply to listen to any audio source.

A user configurable GPIO system, with 10 physical ports and 10 virtual ports, can be used to control operational functions on local or networked units, or drive outputs as selected states change, and a voltage free relay contact can be used to operate external equipment.

A built-in web server provides complete configuration control of the units and also allows for firmware updates and configuration backup. An Ember+ interface also gives access to the configuration options as well as

providing remote control and monitoring of the GPIO and virtual GPIO ports.

## Webserver Software

The AVN-TB6 has a built-in webserver for setup and configuration. The webserver is responsive, and resizes depending on the size of your screen, meaning that it can be used on large monitors or small handheld devices such as smart-phones. Help information is shown on the right hand side of the screen so it's a good place to go to find out how the unit operates."

- 6 illuminated key-cap Talk buttons plus Listen & amp; Page buttons.
- Dual 1Gb Ethernet & Dual 1Gb SFP port.
- Mic & amp: headset inputs, headphone & speaker outputs with volume control.
- Loudspeaker & amp; Mic Mute buttons.
- Dual AC & Dower supply inputs.

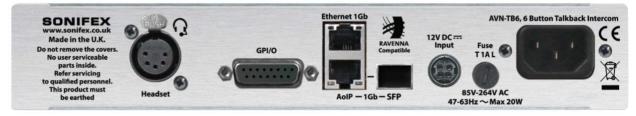
- Advanced echo cancellation & mp; mic AGC to prevent acoustic feedback.
- 10 user assignable GPIO ports.
- Responsive design Ethernet webserver.
- AVN-TB6RK 19" rack kit available.

## **Technical Specification For AVN-TB6**

| rechnical Specification For AVIV-1B6 |   |  |
|--------------------------------------|---|--|
| Audio-Over-IP Specific               | cation  |  |
| Open Standards:                      | RAVENNA, AES67                                    |  |
| Device Discovery:                    | Bonjour (mDNS / DNS-SD)                           |  |
| Audio Delivery:                      | RTP/UDP over IPv4 multicast                       |  |
| QoS:                                 | DiffServ  |  |
| Stream Management:                   | RTSP/SDP  |  |
| Control:                             | Ember+/Webserver                                  |  |
| Format:                              | Linear PCM 24-bit (L24)                           |  |
| Channels Per Stream:                 | 2   |  |
| Frames Per Packet:                   | 48  |  |
| Transit Streams:                     | 1 (fixed)   |  |
| Sample Rate:                         | 48 kHz  |  |
|                                      |   |  |
| Ember+ Interface Con                 | nection   |  |
| Interface Type:                      | Provider  |  |
| Network Interface:                   | Ethernet port and AoIP port                       |  |
| Port:                                | 9000  |  |
|                                      |   |  |
| Timing Synchronisatio                | n   |  |
| Profile Support:                     | Default, AES67 Media & Custom profiles            |  |
| Timing Protocol:                     | PTPv2, IEEE1588-2008                              |  |
|                                      |   |  |
| Microphone and Head                  | lset Input  |  |
| Input Impedance:                     | > 2.5kΩ balanced                                  |  |
| Gain Range:                          | 0dB to +60dB                                      |  |
| OdBFS Line-Up:                       | Adjustable in steps of 3dB from -58dBu to +2dBu   |  |
| Frequency Response:                  | 20Hz to 20kHz, +0/-0.2dB                          |  |
| Noise:                               | -127dBu, 20kHz BW, Rs=200 $\Omega$ ref. 76dB gain |  |
|                                      |   |  |

| Output Impedance:   | Drives 150mW into 320 to 6000                                     |
|---------------------|---|
| output impedance.   | headphones  |
| OdBFS Line-Up:      | +20dBu  |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N:<br>20kHz,    | < -108dBFS, -30dBFS, 20Hz to<br>unity gain, 20kHz BW              |
| Noise:              | -110dBFS, 20kHz BW  |
| Loudspeaker         |   |
| Power Output:       | 4W  |
| Volume:             | Mute to full volume via front pane control                        |
|                     |   |
| Connections         |   |
| Microphone:         | XLR-3 pin female (electronically balanced)                        |
| Headphone:          | 1/4 inch (6.35mm) stereo jack socke                               |
| Headset:            | 1 x XLR-5 pin female (electronicall balanced input)               |
| GPIO:               | 15-way 'D'-type socket  |
| Network:            | 2 x gigabit Ethernet, RJ45 1 x SFP                                |
| Mains AC Input:     | Universal filtered IEC, continuousl rated 85-264VAC, 47-63Hz, 20W |
| DC Input:           | 4 pin 7.5A power jack socket,<br>10-14VDC                         |
| Fuse Rating:        | Anti-surge fuse 2A 20mm x 5mm                                     |
| Equipment Type      |   |
| AVN-TB6:            | 6 channel freestanding talkback                                   |
|                     | intercom control unit with RAVENNA AOIP                           |

| Physical Specification |   |
|------------------------|---|
| Dimensions: (Raw)      | 25.5cm (W) x 19.4cm (D) x 4.4cm<br>(H) (1U)<br>10.1" (W) x 7.7" (D) x 1.8" (H) (1U) |
| Dimensions (Boxed):    | 33cm (W) x 26cm (D) x 17cm (H)<br>13" (W) x 10.3" (D) x 6.7" (H)                    |
| Weight:                | Nett: 2kg Gross: 3kg<br>Nett: 4.4lbs Gross: 6.6lbs                                  |
| Accessories            |   |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug                                 |
| AVN-TB6RK              | 1U Rack Kit   |
| CM-GM2:                | Professional Gooseneck Condenser<br>Microphone                                      |



AVN-TB6 Rear View.









## **AVN-TB6D 6 Button Talkback Intercom**



Category: AES67 Talkback, IFB, Intercom & Commentary,

**Product Function:** Provides broadcast quality audio communication using RAVENNA/AES67 as the transport mechanism, allowing simple CAT 5 cabling and expansion.

Typical Applications: Ideal for broadcast quality audio communication between studios, offices and different areas in a facility or building complex.

#### Features:

- 6 illuminated key-cap Talk buttons plus Listen & amp; Page buttons.
- Dual 1Gb Ethernet & Dual 1Gb SFP port.
- Mic & amp; headset inputs, headphone & speaker outputs with volume control.
- Loudspeaker & amp; Mic Mute buttons.
- Dual AC & Dower supply inputs.
- Advanced echo cancellation & amp; mic AGC to prevent acoustic feedback.
- 10 user assignable GPIO ports.
- Responsive design Ethernet webserver.



AVN-TB6D Front View.

The AVN-TB6D is a 6 channel talkback intercom control unit from the Sonifex AVN range of IP based products. The AVN-TB6D is a desktop version which can be fitted flush into a work surface if required. There is also a freestanding version which can be rackmounted with the AVN-TB6RK 1U rack kit. Both models offers similar functionality.

This unit provides broadcast quality audio communication between studios, offices and different areas in a facility or building complex, using RAVENNA/AES67 as the

transport mechanism, allowing simple CAT 5 cabling and expansion. RAVENNA (of which AES67 is a subset) allows for the distribution of audio across a network. The AVN range

use RAVENNA as the communication method providing compatibility with other AES67 systems.

Each of the 6 channels on the AVN-TB6D can be configured to provide communications with other remote networked units, and an independently configurable 'page' function can contact selected units with priority over standard intercom calls if required.

There is a monitor channel that can route the audio from an AoIP source to the headphones and speaker. This could be used to take an IFB feed or an off-air transmission signal or simply to listen to any audio source.

A user configurable GPIO system, with 10 physical ports and 10 virtual ports, can be used to control operational functions on local or networked units, or drive outputs as selected states change, and a voltage free relay contact can be used to operate external equipment.

A built-in web server provides complete configuration control of the units and also allows for firmware updates and configuration backup. An Ember+ interface also gives access to the configuration options as well as providing remote control and monitoring of the GPIO and virtual GPIO ports.

## Webserver Software

The AVN-TB6D has a built-in webserver for setup and configuration. The webserver is responsive, and resizes depending on the size of your screen, meaning that it can be used on large monitors or small handheld devices such as smart-phones. Help information is shown on the right hand side of the screen so it's a good place to go to find out how the unit operates.

- 6 illuminated key-cap Talk buttons plus Listen & Damp; Page buttons.
- Dual 1Gb Ethernet & Dual 1Gb SFP port.
- Mic & Description
   Mic & Description<
- Loudspeaker & amp; Mic Mute buttons.
- Dual AC & amp; DC power supply inputs.
- Advanced echo cancellation & mic AGC to prevent acoustic feedback.
- 10 user assignable GPIO ports.
- · Responsive design Ethernet webserver.

## **Technical Specification For AVN-TB6D**

| Audio-Over-IP Specification |                             |
|-----------------------------|-----------------------------|
| Open Standards:             | RAVENNA, AES67              |
| Device Discovery:           | Bonjour (mDNS / DNS-SD)     |
| Audio Delivery:             | RTP/UDP over IPv4 multicast |
| QoS:                        | DiffServ                    |
| Stream Management:          | RTSP/SDP                    |
| Control:                    | Ember+/Webserver            |
| Format:                     | Linear PCM 24-bit (L24)     |
| Channels Per Stream:        | 2                           |
| Frames Per Packet:          | 48                          |
| Transit Streams:            | 1 (fixed)                   |
| Sample Rate:                | 48 kHz                      |
|                             |                             |

| Ember+ Interface Connection |                             |  |
|-----------------------------|-----------------------------|--|
| Interface Type:             | Provider                    |  |
| Network Interface:          | Ethernet port and AoIP port |  |
| Port:                       | 9000                        |  |
|                             |                             |  |

Default, AES67 Media & Custom

Timing Synchronisation
Profile Support:

Noise:

|                     | profiles  |
|---------------------|---|
| Timing Protocol:    | PTPv2, IEEE1588-2008  |
|                     |   |
| Microphone and Head | dset Input  |
| Input Impedance:    | > 2.5kΩ balanced  |
| Gain Range:         | 0dB to +60dB  |
| OdBFS Line-Up:      | Adjustable in steps of 3dB from -58dBu to +2dBu             |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                                    |
| Noise:              | -127dBu, 20kHz BW, Rs=200 $\Omega$ ref. 76dB gain           |
| Headphone Output    |   |
| Output Impedance:   | Drives 150mW into 32 $\Omega$ to 600 $\Omega$ headphones    |
| OdBFS Line-Up:      | +20dBu  |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                                    |
| THD+N:              | < -108dBFS, -30dBFS, 20Hz to<br>20kHz, unity gain, 20kHz BW |

-110dBFS, 20kHz BW

| zoudopeane.     |  |
|-----------------|--|
| Power Output:   | 4W   |
| Volume:         | Mute to full volume via front panel control                        |
| Connections     |  |
| Microphone:     | XLR-3 pin female (electronically balanced)                         |
| Headphone:      | ¼ inch (6.35mm) stereo jack socket                                 |
| Headset:        | 1 x XLR-5 pin female (electronically balanced input)               |
| GPIO:           | 15-way 'D'-type socket   |
| Network:        | 2 x gigabit Ethernet, RJ45 1 x SFP                                 |
| Mains AC Input: | Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W |
| DC Input:       | 4 pin 7.5A power jack socket,<br>10-14VDC                          |
| Fuse Rating:    | Anti-surge fuse 2A 20mm x 5mm                                      |

Loudspeaker

| 6 channel desktop talkback<br>intercom control unit with<br>RAVENNA AOIP   |
|--|
|  |
| 22.2cm (W) x 17cm (D) x 11.5cm (H) (1U) 8.7" (W) x 6.7"(D) x 4.5" (H) (1U) |
| 41cm (W) x 31cm D) x 23cm (H)<br>16" (W) x 12" (D) x 9" (H)                |
| Nett: 1.8kg Gross:2.6kg<br>Nett: 4.0llbs Gross:5.7llbs                     |
|  |
| 60W DC power supply for AVN range with KPJX-4S plug                        |
|  |



AVN-TB6D Rear View.







## **RAVENNA**



### **AVN-PXH12 12 x 2 Channel Mix Monitor, AoIP Portal**



AVN-PXH12 Front View.

The Sonifex AVN-PXH12 is a monitor-mixer primarily designed for monitoring Audio Over IP audio channels in a compact 1U format rack. Any AES67 audio channels on your network can be assigned to input channels on the unit and mixed down to analogue outputs, a headphone output and built-in speaker.

It is a stereo monitoring device that allows you to monitor up to 12 audio sources, from an input total of 24, at any one time. The 24 audio sources can be selected from 4 discrete stereo analogue audio inputs (1 x front panel 3.5mm jack socket, 2 x rear panel 3.5mm jack sockets and 1 x rear panel stereo XLR input pair) or from any

RAVENNA, AES67 or AES67-enabled Dante® AoIP connected streams.

These stereo signals are routed to the 12 x control channels on the front panel, each of which have a 'Normal' and an 'Alternate' input selection. Each channel has three buttons: one for input selection, another to Mute the channel and the third to select

whether the channel input is routed to the left, right or stereo output legs.

The knob for each channel controls the level of the input routed to the output and the knob also illuminates either green, amber or red to show input level. Pressing the knob 'Solos' the channel input to the output.



Category: AES67 Mix Monitor

Product Function: A mix monitor for monitoring 2 x 12 AES67 Audio Over IP audio channels in a compact 1U format rack to analogue outputs, a headphone output and built-in speaker.

Typical Applications: Production gallery or control room channel monitoring positions, OB truck monitoring, confidence channel monitoring of any AoIP installation.

### Features:

- AES67 as a format has been established. providing compatibility with most other AoIP products - the unit uses RAVENNA audio to ensure AES67 compatibility.
- A built-in web server is used for all configuration. Sources for all channels are simply assigned on one webpage and can be freely selected and altered at will.
- · SAP can be used as a discovery mechanism to discover Dante devices and monitor Dante® AES67 streams. Dante® is a trademark of Audinate Ptv Ltd.

- Confidence monitoring on the translucent volume knob for each channel so vou've got at-a-glance' monitoring available.
- The front panel Mute button and the Solo feature on the control knob allow a single channel, or a handful of, channels to be auditioned quickly.
- · For each channel. 'Normal' and 'Alternate' inputs can be switched quickly (with <1msec accuracy) for direct comparison.
- Each channel can be directed to headphone left ear, right ear, or a stereo mix, providing you with the ability to set up a familiar headphone mix.

- 6.35mm (1/4") & 3.5mm headphone outputs and a speaker output with separate LS & HP volume controls.
- Sources from AoIP, balanced or 3 x unbalanced inputs.
- Destinations to AoIP or rear panel balanced outputs.
- Dual 1Gb lan ports & 1Gb SFP fibre port.
- 10 user assignable GPIO ports.
- Dual AC & DC power supply inputs.
- Ethernet webserver control & configuration.
- Speaker mute button.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs.

The front panel has 3 outputs: paralleled stereo headphones on 6.35mm (%") jack and 3.5mm jack sockets, each with their own individual attenuation settings, and a mono-mix speaker output. There are discrete volume controls for the headphones and the speaker, and the latter also has a mute button.

The rear panel has an additional 3 line level XLR-3 audio outputs, which can be designated as mono mix or left or right channel outputs of the mixed audio content (similar to the speaker and headphone outputs respectively), or any of the physical inputs or AoIP input sources.

The unit also sends to the network, as AoIP AES67 streams, the 8 channels of the 4 physical stereo inputs, together with a stereo mix of the speaker output.

The unit can act as a PTP masterclock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

The rear panel contains IEC mains and secondary DC power inputs which provide

power redundancy to the product. There are two Ethernet RJ45 connections (control and AoIP) and there is an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software controlled functionality. A voltage free relay contact can be used to operate external equipment.

A built-in web server provides complete configuration control of the unit including source assignment to each channel and also allows for firmware updates and configuration backup. The unit can be controlled by suitable Ember+ commands.

### Features:

- AES67 as a format has been established, providing compatibility with most other AoIP products – the unit uses RAVENNA audio to ensure AES67 compatibility.
- A built-in web server is used for all configuration. Sources for all channels are simply assigned on one webpage and can be freely selected and altered at will.
- SAP can be used as a discovery mechanism to discover Dante devices

- and monitor Dante® AES67 streams.

  Dante® is a trademark of Audinate Pty

  Itd.
- Confidence monitoring on the translucent volume knob for each channel so you've got 'at-a-glance' monitoring available.
- The front panel Mute button and the Solo feature on the control knob allow a single channel, or a handful of, channels to be auditioned guickly.
- For each channel, 'Normal' and 'Alternate' inputs can be switched quickly (with <1msec accuracy) for direct comparison.
- Each channel can be directed to headphone left ear, right ear, or a stereo mix, providing you with the ability to set up a familiar headphone mix.
- 6.35mm (1/4") & 3.5mm headphone outputs and a speaker output with separate LS & HP volume controls.
- Sources from AoIP, balanced or 3 x unbalanced inputs.
- Destinations to AoIP or rear panel balanced outputs.
- Dual 1Gb lan ports & 1Gb SFP fibre port.

- 10 user assignable GPIO ports.
- Dual AC & DC power supply inputs.
- Ethernet webserver control & configuration.
- · Speaker mute button.

### Webserver Software

The mix monitor has a built-in webserver for setup and configuration. The webserver is responsive, and resizes depending on the size of your screen, meaning that it can be used on large monitors or small handheld devices such as smart-phones. Help information is shown on the right hand side of the screen so it's a good place to go to find out how the unit operates.



### **Technical Specification For AVN-PXH12**

### Audio-Over-IP Specification

| Open Standards:      | RAVENNA, AES67                |
|----------------------|-------------------------------|
|                      |                               |
| Device Discovery:    | Bonjour (mDNS / DNS-SD) & SAP |
| Audio Delivery:      | RTP/UDP over IPv4 multicast   |
| QoS:                 | DiffServ                      |
| Stream Management:   | RTSP/SDP                      |
| Control:             | Ember+/webserver              |
| Format:              | Linear PCM 24-bit (L24)       |
| Channels Per Stream: | Maximum of 8                  |
| Frames Per Packet:   | 48                            |
| Maximum Streams:     | RX 15, TX 2                   |
| Sample Rate:         | 48 kHz                        |

### Timing Synchronisation Profile Support: Default, media & custom profiles Timing Protocol: PTPv2, IEEE1588-2008

| Technical Specification |  |  |
|-------------------------|--|--|
| Unbalanced Line Inpu    | ts   |  |
| Input Impedance:        | >20kΩ  |  |
| OdBFS Line-Up:          | +12dBu   |  |
| Frequency Response:     | 20Hz to 20kHz, +0/-0.2dB                               |  |
| THD+N:                  | <-97dBFS, -30dBFS, 20Hz to 20kHz, unity gain, 20kHz BW |  |

| Noise:               | -100dBFS, 20kHz BW, Rs=200Ω                               |  |
|----------------------|---|--|
| Crosstalk:           | <-97dB  |  |
|                      |   |  |
| Balanced Line Inputs |   |  |
| Input Impedance:     | >20kΩ balanced  |  |
| OdBFS Line-Up:       | Adjustable +15/+18/+20/+22/<br>+24dBu                     |  |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                                  |  |
| THD+N:               | <-110dBFS, -30dBFS, 20Hz to 20kHz<br>20kHz BW             |  |
| Noise:               | -110dBFS, 20kHz BW, Rs=200Ω                               |  |
| Crosstalk:           | <-100dB   |  |
| Common Mode          |   |  |
| Rejection:           | >70dB @ 1kHz  |  |
| Headphone Output     |   |  |
| Output Impedance:    | Drives 150mW into $32\Omega$ to $600\Omega$ headphones    |  |
| OdBFS Line-Up:       | +20dBu  |  |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                                  |  |
| THD+N:               | <-108dBFS, -30dBFS, 20Hz to 20kHz<br>unity gain, 20kHz BW |  |
| Noise:               | -110dBFS, 20kHz BW  |  |

### Balanced Line Outputs

Network:

Network:

| balanceu Line Output | .3  |  |  |
|----------------------|---|--|--|
| Output Impedance:    | <50Ω balanced                                 |  |  |
| OdBFS Line-Up:       | Adjustable +15/+18/+20/+22/<br>+24dBu         |  |  |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                      |  |  |
| THD+N:               | <-110dBFS, -30dBFS, 20Hz to 20kHz<br>20kHz BW |  |  |
| Noise:               | -110dBFS, 20kHz BW, Rs=200Ω                   |  |  |
| Loudspeaker          |   |  |  |
| Power Output:        | 4W  |  |  |
| Volume:              | Mute to full volume via front panel control   |  |  |
| Connections          |   |  |  |
| Headphone:           | ¼ inch (6.35mm) stereo jack socket            |  |  |
| Headphone:           | 3.5mm stereo jack socket                      |  |  |
| Audio Inputs:        | 3 x unbalanced 3.5mm stereo jack socket       |  |  |
| Audio Inputs:        | 2 x balanced XLR-3 female socket              |  |  |
| Audio Outputs:       | 3 x balanced XLR-3 male plug                  |  |  |
| Audio Outputs:       | 1 x loudspeaker output                        |  |  |
| GPIO:                | 15-way 'D'-type socket                        |  |  |
|                      |   |  |  |

2 x gigabit Ethernet, RJ45

1 x SFP fibre

### **Equipment Type**

| AVN-PXH12: | 2 x 12 Channel mix monitor, AoIP portal |
|------------|---|
|            |   |

| Physical Specification |   |
|------------------------|---|
| Dimensions: (Raw)      | 48.3cm (W) x 20.12cm (D) x 4.4cm (H)(1U)<br>19" (W) x 7.92" (D) x 1.8" (H) (1U) |
| Dimensions (Boxed):    | 55.2cm (W) x 29.5cm (D) x 16.5cm (H) 21.7" (W) x 11.6" (D) x 6.5" (H)           |
| Weight:                | Nett: 2.9kg Gross: 3.5kg<br>Nett: 6.38lbs Gross: 7.7lbs                         |

### Accessories

AVN-DC150: 150W DC power supply with KPJX-4S









## AVN-PA8, 8 Stereo Analogue Line Inputs & 8 Stereo Analogue Line Outputs, AES67 Portal



AVN-PA8 Front View.

The AVN-PA8 is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight stereo analogue line inputs, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small formfactor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67 stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or

receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PA8 becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required.

With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & Controls

The AVN-PA8 unit is supplied with the standard OLED front panel display, which provides detailed status information on device name, network addresses, PTP clocking info. power status/voltages, version information, and basic metering (The 'D' version adds twin TFT metering with









Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer. 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier. IFB distribution to analogue belt-packs & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

 Eight stereo analogue inputs and eight stereo analague outputs on D-type

- sockets with AES59 pinout, paralleled with eight RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.<
- · Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition
- Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PA8 Rear View.

configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PA8 uses D-type sockets with AES59 analogue

pinout, paralleled with eight RJ45 connectors using the StudioHub® pinout.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage AVN-HA1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HA1 can be used as another GPI for muting the output.

The AVN-PA8 is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo analogue inputs and eight stereo analague outputs on D-type sockets with AES59 pinout, paralleled with eight RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.

- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition
- Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PA8**

| Audio-Over-IP Specific | ration                                 |
|------------------------|--|
| Open Standards:        | RAVENNA. AES67                         |
| Device Discovery:      | Bonjour (mDNS / DNS-SD) or SAP         |
| Audio Delivery:        | RTP/UDP over IPv4 multicast            |
| QoS:                   | DiffServ                               |
| Stream Management:     | RTSP/SDP                               |
| Control:               | Web server/Ember+                      |
| Format:                | Linear PCM 24-bit (L24)                |
| Channels Per Stream:   | Up to 8                                |
| Frames Per Packet:     | 48                                     |
| Transmit Streams:      | Up to 8                                |
| Sample Rate:           | 48 kHz                                 |
| Ember+ Interface Con   | nection                                |
| Interface Type:        | Provider                               |
| Network Interface:     | Ethernet port and AoIP port            |
| Port:                  | 9000                                   |
| Timing Synchronisation | on                                     |
| Profile Support:       | Default, AES67 Media & Custom profiles |
| Timing Protocol:       | PTPv2, IEEE1588-2008                   |
| Balanced Line Inputs   |  |
| Input Impedance:       | > 20kΩ balanced                        |
| OdBFS Line-Up:         | Adjustable +15/+18/+20/+22/<br>+24dBu  |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB               |
| THD+N:                 | <-110dBFS, -30dBFS, 20Hz to 20kHz      |

-110dBFS, 20kHz BW, Rs=200Ω

| ,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,         |                      |  |
|---|----------------------|--|
| ication   | Crosstalk:           | < -100dB   |
| RAVENNA, AES67                                  | Common Mode          |  |
| Bonjour (mDNS / DNS-SD) or SAP                  | Rejection:           | > 70dB @ 1kHz  |
| RTP/UDP over IPv4 multicast                     |                      |  |
| DiffServ  | Balanced Line Output | is .   |
| : RTSP/SDP                                      | Output Impedance:    | < 50Ω balanced   |
| Web server/Ember+                               | OdBFS Line-Up:       | Adjustable +15/+18/+20/+22/  |
| Linear PCM 24-bit (L24)                         |                      | +24dBu   |
| : Up to 8                                       |                      | 20Hz to 20kHz, +0/-0.2dB   |
| 48  | THD+N:               | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW                    |
| Up to 8   | Noise:               | -110dBFS, 20kHz BW, Rs=200Ω  |
| 48 kHz  |                      | ,                            |
|   | Connections          |  |
| nnection  | Inputs:              | 2 x D-Sub (DB-25) connections                                      |
| Provider  |                      | (TASCAM AES-59 analogue pinout)                                    |
| Ethernet port and AoIP port                     |                      | paralleled with 8 RJ45 connection                                  |
| 9000  |                      | (StudioHub+ pinout).   |
|   | Outputs:             | 2 x D-Sub (DB-25) connections                                      |
| on  |                      | (TASCAM AES-59 analogue pinout) paralleled with 8 RJ45 connections |
| Default, AES67 Media & Custom                   |                      | (StudioHub+ pinout).   |
| profiles  | GPIO:                | 1 x D-Sub (DA-15) connections.                                     |
| PTPv2, IEEE1588-2008                            | Network:             | 2 x Gigabit Ethernet, RJ45's.                                      |
|   | Network:             | 1 x SFP fibre.   |
| i   | Power:               | Mains AC Input: Universal filtered                                 |
| > 20kΩ balanced                                 |                      | IEC, continuously rated 85-264VAC,                                 |
| Adjustable +15/+18/+20/+22/                     |                      | 47-63Hz, 20W.  |
| +24dBu  | DC Input:            | 4-pin 7.5A power jack socket, 10                                   |
| 20Hz to 20kHz, +0/-0.2dB                        |                      | 14VDC.   |
| < -110dBFS, -30dBFS, 20Hz to 20kHz,<br>20kHz BW | Fuse Rating:         | Anti-surge fuse 2A 20mm x 5mm.                                     |
|   |                      |  |

| Equipment Type<br>AVN-PA8: | 8 Stereo analogue line inputs & 8 stereo analogue line outputs, AES67 portal |
|----------------------------|--|
| Physical Specification     | ALSO PORTA   |
| Dimensions (Raw):          | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)   |
| Dimensions (Boxed):        | 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)                |
| Weight:                    | Nett: 3.12 kg Gross:4.16 kg<br>Nett: 6.86 lbs Gross:9.15 lbs                 |
| Accessories                |  |
| AVN-DC060:                 | 60W DC power supply for AVN range with KPJX-4S plug                          |
| AVN-HA1:                   | Analogue Headphone Amplifier.  |

Noise:









## AVN-PA8T, 8 Stereo Analogue Line Inputs & 8 Stereo Analogue Line Outputs on Terminal **Blocks, AES67 Portal**



AVN-PA8T Front View.

The AVN-PA8T is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight stereo analogue line inputs, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding. Physical audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67 stream output, with gain adjustment at the input,

the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/

modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PA8T becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support. multiple stream outputs can be provided,













Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue belt-packs & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

· Eight stereo analogue inputs and eight stereo analague outputs on terminal block connections.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).

- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).

combined with remote handling via GPIO, VGPIO. SNMP and Ember+ support.

# Front Panel Displays, Metering & Controls

The AVN-PA8T unit is supplied with the standard OLED front panel display, which provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, version information, and basic metering (The 'D' version adds twin TFT metering with configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is

disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections and GPIO the AVN-PA8 uses terminal block connections.

There are two Ethernet RJ45 connections - one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

The rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

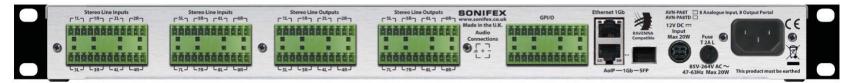
For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

The AVN-PA8T is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three

additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo analogue inputs and eight stereo analague outputs on terminal block connections.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition.

- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).



### **Technical Specification For AVN-PA8T**

| Aud | io-Ov | er-IF | Spec | ification | ١ |
|-----|-------|-------|------|-----------|---|
|     |       |       |      |           |   |

| Open Standards:      | RAVENNA, AES67                 |
|----------------------|--------------------------------|
| Device Discovery:    | Bonjour (mDNS / DNS-SD) or SAP |
| Audio Delivery:      | RTP/UDP over IPv4 multicast    |
| QoS:                 | DiffServ                       |
| Stream Management:   | RTSP/SDP                       |
| Control:             | Web server/Ember+              |
| Format:              | Linear PCM 24-bit (L24)        |
| Channels Per Stream: | Up to 8                        |
| Frames Per Packet:   | 48                             |
| Transmit Streams:    | Up to 8                        |
| Sample Rate:         | 48 kHz                         |
|                      |                                |

### Ember+ Interface Connection

| Interface Type:    | Provider                    |
|--------------------|-----------------------------|
| Network Interface: | Ethernet port and AoIP port |
| Port:              | 9000                        |
|                    |                             |

### Timing Synchronisation

| Profile Support: | Default, AES67 Media & Custom profiles |
|------------------|--|
| Timing Protocol: | PTPv2, IEEE1588-2008                   |

| Balanced Line Inputs      |   |
|---------------------------|---|
| Input Impedance:          | > 20kΩ balanced                                 |
| OdBFS Line-Up:            | Adjustable +15/+18/+20/+22/<br>+24dBu           |
| Frequency Response:       | 20Hz to 20kHz, +0/-0.2dB                        |
| THD+N:                    | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW |
| Noise:                    | -110dBFS, 20kHz BW, Rs=200Ω                     |
| Crosstalk:                | < -100dB  |
| Common Mode<br>Rejection: | > 70dB @ 1kHz                                   |
| Output Impedance:         | < 50Ω balanced                                  |
| OdBFS Line-Up:            | Adjustable +15/+18/+20/+22/<br>+24dBu           |
|                           |   |

| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB  |
|------------------------|---|
| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW   |
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω   |
| Connections            |   |
| Inputs:                | 2 x 24-Pin Phoenix style terminal blocks (Analogue pinout).                                     |
| Outputs:               | 2 x 24-Pin Phoenix style terminal blocks. (Analogue pinout).                                    |
| GPIO:                  | 1 x 24-Pin Phoenix style terminal blocks.   |
| Network:               | 2 x Gigabit Ethernet, RJ45's.   |
| Network:               | 1 x SFP fibre.  |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.             |
| DC Input:<br>10-14VDC. | 4-pin 7.5A power jack socket,   |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.  |
| Equipment Type         |   |
| AVN-PA8T:              | 8 Stereo analogue line inputs & 8 stereo analogue line outputs on terminal blocks, AES67 portal |
| Physical Specification |   |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)                      |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)                                |
| Weight:                | Nett: 2.96 kg Gross: 4.00kg<br>Nett: 6.51lbs Gross: 8.80lbs                                     |
| Accessories            |   |
| AVN-DC060:             | 60W DC power supply for AVN range<br>with KPJX-4S plug  |
| AVALUAT.               | Analasia Hardalasa Asaalifiaa   |

Analogue Headphone Amplifier.

AVN-HA1:

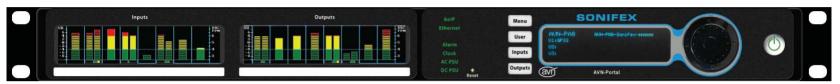








## AVN-PA8D, 8 Stereo Analogue Line Inputs & 8 Stereo Analogue Line Outputs, AES67 Portal, with Detailed Meter Displays



AVN-PA8D Front View.

The AVN-PA8D is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight stereo analogue line inputs, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or

AES67 stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created. supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PA8D becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & **Controls**

The AVN-PARD has the standard OLFD front panel display, and the additional twin TFT



Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue belt-packs & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

 Eight stereo analogue on D-type sockets with AES59 pinout, paralleled with eight RJ45 connectors using StudioHub® pinout.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- · Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PA8D Rear View.

meter displays. The OLED display provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view."

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU. Metering can be set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be setup to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PA8D uses D-type sockets with AES59 analogue pinout, paralleled with eight RJ45 connectors using the StudioHub® pinout."

There are two Ethernet RJ45 connections - one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

### 8-Way Analogue Headphone Distribution System

The AVN-PA8D StudioHub® outputs can be combined with multiple AVN-HA1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HA1 can be used as another GPI for muting the output."

The AVN-PA8D is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright

TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo analogue on D-type sockets with AES59 pinout, paralleled with eight RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PA8D**

| Audio-Over-IP Specification |                                |  |
|-----------------------------|--------------------------------|--|
| Open Standards:             | RAVENNA, AES67                 |  |
| Device Discovery:           | Bonjour (mDNS / DNS-SD) or SAP |  |
| Audio Delivery:             | RTP/UDP over IPv4 multicast    |  |
| QoS:                        | DiffServ                       |  |
| Stream Management:          | RTSP/SDP                       |  |
| Control:                    | Web server/Ember+              |  |
| Format:                     | Linear PCM 24-bit (L24)        |  |
| Channels Per Stream:        | Up to 8                        |  |
| Frames Per Packet:          | 48                             |  |
| Transmit Streams:           | Up to 8                        |  |
| Sample Rate:                | 48kHz                          |  |

| Ember+ Interface Connection |  |  |
|-----------------------------|--|--|
| Provider                    |  |  |
| Ethernet port and AoIP port |  |  |
| 9000                        |  |  |
|                             |  |  |

| Timing Synchronisation |   |  |
|------------------------|---|--|
| Profile Support:       | Default, AES67 Media & Custom profiles          |  |
| Timing Protocol:       | PTPv2, IEEE1588-2008                            |  |
|                        |   |  |
| Balanced Line Inputs   |   |  |
| Input Impedance:       | > 20kΩ balanced                                 |  |
| 0dBFS Line-Up:         | Adjustable +15/+18/+20/+22/<br>+24dBu           |  |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB                        |  |
| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW |  |
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω                     |  |
| Crosstalk:             | < -100dB  |  |
|                        |   |  |

| <b>Balanced Line Output</b> | s   |
|-----------------------------|---|
| Output Impedance:           | < 50Ω balanced                                  |
| OdBFS Line-Up:              | Adjustable +15/+18/+20/+22/<br>+24dBu           |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB                        |
| THD+N:                      | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW |
| Noise:                      | -110dBFS, 20kHz BW, Rs=200Ω                     |

> 70dB @ 1kHz

Common Mode Rejection:

| Connections |  |
|-------------|--|
| Inputs:     | 2 x D-Sub (DB-25) connections<br>(TASCAM AES-59 analogue pinout)<br>paralleled with 8 RJ45 connections<br>(StudioHub+ pinout). |

| Outputs:                                | 2 x D-Sub (DB-25) connections  |
|---|--|
|   | (TASCAM AES-59 analogue pinout)  |
|   | paralleled with 8 RJ45 connections   |
|   | (StudioHub+ pinout).   |
| GPIO:                                   | 1 x D-Sub (DA-15) connections.   |
| Network:                                | 2 x Gigabit Ethernet, RJ45's.  |
| Network:                                | 1 x SFP fibre.   |
| Power:                                  | Mains AC Input: Universal filtered   |
|   | IEC, continuously rated 85-264VAC,<br>47-63Hz, 20W.                        |
| DC Input:                               | 4-pin 7.5A power jack socket, 10   |
|   | 14VDC.   |
| Fuse Rating:                            | Anti-surge fuse 2A 20mm x 5mm.   |
| Equipment Type                          |  |
| AVN-PA8D:                               | 8 Stereo analogue line inputs & 8  |
| ,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,,, | stereo analogue line outputs, AES67  |
|   | portal, with detailed meter displays                                       |
| Physical Specification                  |  |
| Dimensions (Raw):                       | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U) |
| Dimensions (Boxed):                     | 55.5cm (W) x 29cm (D) x 18cm (H)   |

| Dimensions (Boxed): | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H) |
|---------------------|--|
| Weight:             | Nett: 2.98 kg Gross: 4.02kg<br>Nett: 6.56 lbs Gross:8.84 lbs     |
|                     |  |
| Accessories         |  |
| AVN-DC060:          | 60W DC power supply for AVN range wit<br>KPJX-4S plug            |
| AVN-HA1:            | Analogue Headphone Amplifier.                                    |

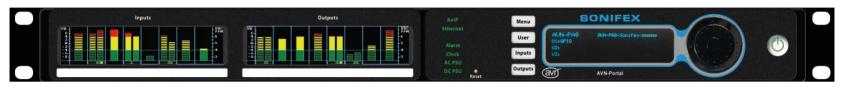








## **AVN-PA8TD 8 Stereo Analogue Line Inputs & 8 Stereo Analogue Line Outputs on Terminal** Blocks, AES67 Portal, with Detailed Meter Displays



AVN-PA8TD Front View.

The AVN-PA8TD is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight stereo analogue line inputs. eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding. Physical audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or

routed to any physical audio output or AES67 stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain

and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master

clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PA8TD becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO. VGPIO. SNMP and Ember+ support.

Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue belt-packs & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

· Eight stereo analogue inputs and eight stereo analague outputs on terminal block connections

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

# Front Panel Displays, Metering & Controls

The AVN-PA8TD has the standard OLED front panel display and the additional twin TFT meter displays. The OLED display provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view.

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU. Metering can be set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to

networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections and GPIO the AVN-PA8TD uses terminal block connections.

There are two Ethernet RJ45 connections - one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as

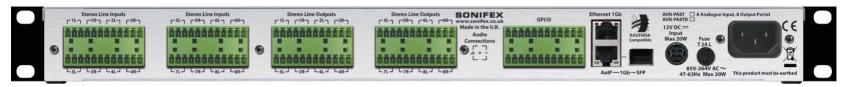
inputs or outputs and provide softwarecontrolled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

The AVN-PA8TD is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo analogue inputs and eight stereo analague outputs on terminal block connections
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.

- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



### **Technical Specification For AVN-PA8TD**

| rediffical ope                 | omoundment of Aviv I Ad                        |
|--------------------------------|--|
| Audio-Over-IP Specific         | cation   |
| Open Standards:                | RAVENNA, AES67                                 |
| Device Discovery:              | Bonjour (mDNS / DNS-SD) or SAP                 |
| Audio Delivery:                | RTP/UDP over IPv4 multicast                    |
| QoS:                           | DiffServ                                       |
| Stream Management:             | RTSP/SDP                                       |
| Control:                       | Web server/Ember+                              |
| Format:                        | Linear PCM 24-bit (L24)                        |
| Channels Per Stream:           | Up to 8  |
| Frames Per Packet:             | 48   |
| Transmit Streams:              | Up to 8  |
| Sample Rate:                   | 48kHz  |
|                                |  |
| <b>Ember+ Interface Con</b>    | nection  |
| Interface Type:                | Provider                                       |
| Network Interface:             | Ethernet port and AoIP port                    |
| Port:                          | 9000   |
|                                |  |
| Timing Synchronisation         |  |
| Profile Support:               | Default, AES67 Media & Custom<br>profiles      |
| Timing Protocol:               | PTPv2, IEEE1588-2008                           |
|                                |  |
| Balanced Line Inputs           |  |
| Input Impedance:               | > 20kΩ balanced                                |
| OdBFS Line-Up:                 | Adjustable +15/+18/+20/+22/<br>+24dBu          |
| Frequency Response:            | 20Hz to 20kHz, +0/-0.2dB                       |
| THD+N:                         | < -110dBFS, -30dBFS, 20Hz to 20kHz<br>20kHz BW |
| Noise:                         | -110dBFS, 20kHz BW, Rs=200Ω                    |
| Crosstalk:                     | < -100dB                                       |
| Common Mode                    |  |
| Rejection:                     | > 70dB @ 1kHz                                  |
| Outros de lancia a de la casa. | . FOO h-l                                      |

Output Impedance: < 50Ω balanced

Frequency Response: 20Hz to 20kHz, +0/-0.2dB

Adjustable +15/+18/+20/+22/ +24dBu

OdBFS Line-Up:

| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to 20kHz,<br>20kHz BW   |
|------------------------|---|
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω   |
| Connections            |   |
| Inputs:                | 2 x 24-Pin Phoenix style terminal blocks (Analogue pinout).   |
| Outputs:               | 2 x 24-Pin Phoenix style terminal blocks. (Analogue pinout).  |
| GPIO:                  | 1 x 24-Pin Phoenix style terminal blocks.   |
| Network:               | 2 x Gigabit Ethernet, RJ45's.   |
| Network:               | 1 x SFP fibre.  |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.   |
| DC Input:              | 4-pin 7.5A power jack socket, 10-14VDC.   |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.  |
| Equipment Type         |   |
| AVN-PA8TD:             | 8 Stereo analogue line inputs & 8 stereo analogue line outputs on terminal blocks, AES67 portal, with detailed meter displays |
| Physical Specification |   |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)  |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)  |
| Weight:                | Nett: 2.90kg Gross: 3.94 kg<br>Nett: 6.38lbs Gross: 8.67lbs   |
| Accessories            |   |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug   |
| AVN-HA1:               | Analogue Headphone Amplifier.   |
|                        |   |









## AVN-PD8, 8 Stereo AES3 Digital Inputs & 8 Stereo AES3 Digital Outputs, AES67 Portal



AVN-PD8 Front View.

The AVN-PD8 is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight stereo AES3 digital inputs, eight stereo AES3 digital outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67 stream output, with gain

adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created,

supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PD8 becomes an advanced problem-solving box for any

applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & Controls

The AVN-PD8 unit is supplied with the standard OLED front panel display, which provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, version



Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

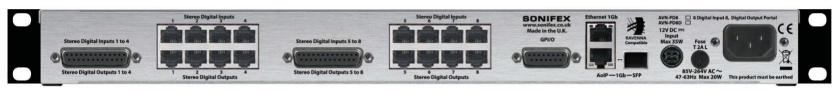
Typical Applications: A powerful AES3 & AES67 mix engine which allows for multiple applications: 8 output AES3 zone mixer, 8 channel digital mixer, 64 channel AES67 stream distribution amplifier from digital sources, 8 channel headphone distribution system (with AVN-HD1 units).

### Features:

· Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on D-type sockets with AES59 pinout, paralleled with 8 x RJ45 connectors using StudioHub® pinout.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PD8 Rear View.

information, and basic metering (The 'D' version adds twin TFT metering with configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PD8 uses D-type sockets with AES59 digital pinout, paralleled with eight RJ45 connectors using the StudioHub® pinout.

There are two Ethernet RJ45 connections - one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

### 8 Way Digital Headphone Distribution System

The AVN-PD8 StudioHub® outputs can be combined with multiple AVN-HD1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HD1 can be used as another GPI for muting the output.

The AVN-PD8 is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block

connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on D-type sockets with AES59 pinout, paralleled with 8 x RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition.

### Multi-Channel Audio Mix Engine Interfaces

- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PD8**

### Audio-Over-IP Specification

| Open Standards:      | RAVENNA, AES67                 |
|----------------------|--------------------------------|
| Device Discovery:    | Bonjour (mDNS / DNS-SD) or SAP |
| Audio Delivery:      | RTP/UDP over IPv4 multicast    |
| QoS:                 | DiffServ                       |
| Stream Management:   | RTSP/SDP                       |
| Control:             | Web server/Ember+              |
| Format:              | Linear PCM 24-bit (L24)        |
| Channels Per Stream: | Up to 8                        |
| Frames Per Packet:   | 48                             |
| Transmit Streams:    | Up to 8                        |
| Sample Rate:         | 48kHz                          |
|                      |                                |

| Ember+ Interface Connection |                             |  |
|-----------------------------|-----------------------------|--|
| Interface Type:             | Provider                    |  |
| Network Interface:          | Ethernet port and AoIP port |  |
| Port:                       | 9000                        |  |

### Timing Synchronisation

| Profile Support: | Default, AES67 Media & Custom profiles |
|------------------|--|
| Timing Protocol: | PTPv2, IEEE1588-2008                   |

### Stereo Digital Input

Network:

Power:

| Stereo Digital Ilipat      |  |
|----------------------------|--|
| Input Impedance:           | 110Ω balanced  |
| Supported Input Rate       | s:32kHz, 44.1kHz, 48kHz, 88.2kHz,<br>96kHz, 176.4kHz and 192kHz<br>(sample rate converted to 48kHz)                          |
| Output Impedance:          | 110Ω balanced  |
| Supported Output<br>Rates: | 48kHz  |
|                            |  |
| Connections                |  |
| Inputs:                    | 1 x D-Sub (DB-25) connection<br>(TASCAM AES-5 digital pinout)<br>paralleled with 8 RJ45 connections<br>(StudioHub+ pinout).  |
| Outputs:                   | 1 x D-Sub (DB-25) connection<br>(TASCAM AES-59 digital pinout)<br>paralleled with 8 RJ45 connections<br>(StudioHub+ pinout). |
| GPIO:                      | 1 x D-Sub (DA-15) connections.   |
| Network:                   | 2 x Gigabit Ethernet, RJ45's.  |
|                            |  |

1 x SFP fibre.

Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.

| DC Input:              | 4-pin 7.5A power jack socket, 10 14VDC.  |
|------------------------|--|
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.   |
| Equipment Type         |  |
| AVN-PD8:               | 8 Stereo AES3 digital inputs & 8<br>stereo AES3 digital outputs, AES67<br>portal |
| Physical Specification |  |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)       |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)                 |
| Weight:                | Nett: 2.88 kg Gross:3.92 kg<br>Nett: 6.34 lbs Gross:8.62 lbs                     |
| Accessories            |  |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug                              |
| AVN-HD1:               | Digital Headphone Amplifier.   |
|                        |  |









### AVN-PD8T 8 Stereo AES3 Digital Inputs & 8 Stereo AES3 Digital Outputs on Terminal **Blocks, AES67 Portal**



AVN-PD8T Front View.

The AVN-PD8T is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight stereo AES3 digital inputs, eight stereo AES3 digital outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding. Physical digital audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67

stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering. can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams, or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/ modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PD8T becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & Controls

The AVN-PD8T unit is supplied with the standard OLED front panel display, which provides detailed status information on







Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

Product Function: Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

Typical Applications: A powerful AES3 & AES67 mix engine which allows for multiple applications: 8 output AES3 zone mixer, 8 channel digital mixer, 64 channel AES67 stream distribution amplifier from digital sources, 8 channel headphone distribution system (with AVN-HD1 units).

### Features:

 Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on on terminal block connections.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).

- Audio file playback.
- DTMF tone generation/recognition.
- · Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).

### Multi-Channel Audio Mix Engine Interfaces

device name, network addresses, PTP clocking info, power status/voltages, version information, and basic metering (The 'D' version adds twin TFT metering with configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### Physical Inputs & Outputs

For physical digital audio connections and GPIO the AVN-PD8T uses terminal block connections.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands. The AVN-PD8T is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on on terminal block connections.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.

- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).



AVN-PD8T Rear View.

### **Technical Specification For AVN-PD8T**

| Audio-Over-IP Specific                  |   |
|---|---|
| Open Standards:                         | RAVENNA, AES67  |
| Device Discovery:                       | Bonjour (mDNS / DNS-SD) or SAP  |
| Audio Delivery:                         | RTP/UDP over IPv4 multicast   |
| QoS:                                    | DiffServ  |
| Stream Management:                      | RTSP/SDP  |
| Control:                                | Web server/Ember+   |
| Format:                                 | Linear PCM 24-bit (L24)   |
| Channels Per Stream:                    | Up to 8   |
| Frames Per Packet:                      | 48  |
| Transmit Streams:                       | Up to 8   |
| Sample Rate:                            | 48kHz   |
| Ember+ Interface Con<br>Interface Type: | <b>nection</b><br>Provider  |
| Network Interface:                      | Ethernet port and AoIP port   |
| Port:                                   | 9000  |
| Timing Synchronisation                  | n   |
| Profile Support:                        | Default, AES67 Media & Custom profiles  |
| Timing Protocol:                        | PTPv2, IEEE1588-2008  |
| Stereo Digital Input                    |   |
| Input Impedance:                        | 110Ω balanced   |
| Supported Input Rates                   | ::32kHz, 44.1kHz, 48kHz, 88.2kHz,<br>96kHz, 176.4kHz and 192kHz<br>(sample rate converted to 48kHz) |
| Output Impedance:                       | 110Ω balanced   |
|   |   |

48kHz

Supported Output Rates:

| blocks (Digital pinout).   | Inputs:                | 2 x 24-Pin Phoenix style terminal  |         |
|--|------------------------|------------------------------------|---------|
| Outputs:  2 x 24-Pin Phoenix style terminal blocks. (Digital pinout).  GPIO:  1x 24-Pin Phoenix style terminal blocks.  Network:  2 x Gigabit Ethernet, RJ45's.  Network:  1 x SFP fibre.  Power:  Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.  DC Input:  4-pin 7.5A power jack socket, 10-14VDC.  Fuse Rating:  Anti-surge fuse 2A 20mm x 5mm.  Equipment Type  AVN-PD8T:  8 Stereo AES3 digital inputs & 8 stereo AES3 digital outputs on terminal blocks, AES67 portal  Physical Specification  Dimensions (Raw):  48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed):  55.5cm (W) x 20cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight:  Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060:  60W DC power supply for AVN range with KPIX-45 plug | inputs.                |                                    |         |
| blocks   | Outputs:               | 2 x 24-Pin Phoenix style terminal  |         |
| Network:   | GPIO:                  |                                    |         |
| Power: Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20U.  DC Input: 4-pin 7.5A power jack socket, 10-14VDC.  Fuse Rating: Anti-surge fuse 2A 20mm x 5mm.  Equipment Type  AVN-PD8T: 8 Stereo AES3 digital inputs & 8 stereo AES3 digital outputs on terminal blocks, AES67 portal  Physical Specification  Dimensions (Raw): 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed): 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPIX-45 plug  | Network:               | 2 x Gigabit Ethernet, RJ45's.      |         |
| IEC, continuously rated 85-264VAC, 47-63Hz, 20W.   | Network:               | 1 x SFP fibre.                     |         |
| 10-14VDC.  Fuse Rating: Anti-surge fuse 2A 20mm x 5mm.  Equipment Type  AVN-PD8T: 8 Stereo AES3 digital inputs & 8 stereo AES3 digital outputs on terminal blocks, AES67 portal  Physical Specification  Dimensions (Raw): 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed): 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPIX-4S plug  | Power:                 | IEC, continuously rated 85-264VAC, |         |
| Stereo AES3 digital inputs & 8   stereo AES3 digital inputs & 8   stereo AES3 digital outputs on terminal blocks, AES67 portal   | DC Input:              |                                    |         |
| AVN-PD8T: 8 Stereo AES3 digital inputs & 8 stereo AES3 digital outputs on terminal blocks, AES67 portal  Physical Specification  Dimensions (Raw): 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed): 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPIX-4S plug  | Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.     |         |
| AVN-PD8T: 8 Stereo AES3 digital inputs & 8 stereo AES3 digital outputs on terminal blocks, AES67 portal  Physical Specification  Dimensions (Raw): 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed): 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPIX-4S plug  | Equipment Type         |                                    |         |
| Stereo AES3 digital outputs on terminal blocks, AES67 portal   | AVN-PD8T:              | 8 Stereo AES3 digital inputs & 8   |         |
| terminal blocks, AES67 portal  |                        |                                    |         |
| Dimensions (Raw):  |                        |                                    |         |
| 19" (W) x 6.9" (D) x 1.8" (H) (1U)  Dimensions (Boxed): 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPJX-4S plug  | Physical Specification |                                    |         |
| 22" (W) x 11.4" (D) x 7" (H)  Weight: Nett: 2.86 kg Gross: 3.90kg Nett: 6.29 lbs Gross: 8.58lbs  Accessories  AVN-DC060: 60W DC power supply for AVN range with KPIX-4S plug   | Dimensions (Raw):      |                                    | (H)(1U) |
| Nett: 6.29 lbs Gross: 8.58lbs  Accessories AVN-DC060: 60W DC power supply for AVN range with KPJX-4S plug  | Dimensions (Boxed):    |                                    |         |
| AVN-DC060: 60W DC power supply for AVN range with KPJX-4S plug   | Weight:                |                                    |         |
| KPJX-4S plug   | Accessories            |                                    |         |
| AVN-HD1: Digital Headphone Amplifier.  | AVN-DC060:             |                                    | e with  |
|  | AVALUD1.               | Digital Headphone Amplifier        |         |









## AVN-PD8D, 8 Stereo AES3 Digital Inputs & 8 Stereo AES3 Digital Outputs, AES67 Portal, with **Detailed Meter Displays**



AVN-PD8D Front View.

The AVN-PD8D is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight stereo AES3 digital inputs, eight stereo AES3 digital outputs, plus a dedicated AES67 Audio over IP (AoIP) portal. available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or

routed to any physical audio output or AES67 stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and

edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PD8D becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote



Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

mix engine which allows for multiple applications: 8 output AES3 zone mixer, 8 channel digital mixer, 64 channel AES67 stream distribution amplifier from digital sources, 8 channel headphone distribution system (with AVN-HD1 units).

### Features:

 Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on D-type sockets with AES59 pinout, paralleled with 8 x RJ45 connectors using StudioHub® pinout.

- Typical Applications: A powerful AES3 & AES67 AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
  - · Gain and DSP functions available at inputs, outputs, and crosspoints.
  - Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
  - Responsive webserver software controlled router/mixer.
  - Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition
- Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PD8D Front View.

handling via GPIO, VGPIO, SNMP and Ember+support.

# Front Panel Displays, Metering & Controls

The AVN-PD8D has the standard OLED front panel display, and the additional twin TFT meter displays. The OLED display provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view.

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU. Metering can be

set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PD8D uses D-type sockets with AES59 digital pinout, paralleled with eight RJ45 connectors using the StudioHub® pinout.

There are two Ethernet RJ45 connections - one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external

equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

# 8 Way Digital Headphone Distribution System

The AVN-PD8 StudioHub® outputs can be combined with multiple AVN-HD1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HD1 can be used as another GPI for muting the output.

The AVN-PD8D is part of the Sonifex AVN-Portal series, which includes the AVN-PA8

### **Multi-Channel Audio Mix Engine Interfaces**

(analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on D-type sockets with AES59 pinout, paralleled with 8 x RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.

- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition
- Optional multi-channel Opus codec encoding and decoding
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PD8D**

| Technical Spe              | cification For AVN-PD8L                                       |
|----------------------------|---|
| Audio-Over-IP Specific     |   |
| Open Standards:            | RAVENNA, AES67  |
| Device Discovery:          | Bonjour (mDNS / DNS-SD) or SAP                                |
| Audio Delivery:            | RTP/UDP over IPv4 multicast                                   |
| QoS:                       | DiffServ  |
| Stream Management:         | RTSP/SDP  |
| Control:                   | Web server/Ember+   |
| Format:                    | Linear PCM 24-bit (L24)                                       |
| Channels Per Stream:       | Up to 8   |
| Frames Per Packet:         | 48  |
| Transmit Streams:          | Up to 8   |
| Sample Rate:               | 48kHz   |
|                            |   |
| Ember+ Interface Con       | nection   |
| Interface Type:            | Provider  |
| Network Interface:         | Ethernet port and AoIP port                                   |
| Port:                      | 9000  |
| Timing Synchronisation     | on  |
| Profile Support:           | Default, AES67 Media & Custom profiles                        |
| Timing Protocol:           | PTPv2, IEEE1588-2008  |
| Stereo Digital Input       |   |
| Input Impedance:           | 110Ω balanced   |
|                            | ::32kHz, 44.1kHz, 48kHz, 88.2kHz,                             |
| supported input nates      | 96kHz, 176.4kHz and 192kHz                                    |
|                            | (sample rate converted to 48kHz)                              |
| Output Impedance:          | 110Ω balanced   |
| Supported Output<br>Rates: | 48kHz   |
|                            |   |
| Connections                |   |
| Inputs:                    | 1 x D-Sub (DB-25) connection<br>(TASCAM AES-5 digital pinout) |
|                            | paralleled with 8 RJ45 connections                            |

(StudioHub+ pinout).

| Outputs:               | 1 x D-Sub (DB-25) connection<br>(TASCAM AES-59 digital pinout)<br>paralleled with 8 RJ45 connections<br>(StudioHub+ pinout). |
|------------------------|--|
| GPIO:                  | 1 x D-Sub (DA-15) connections.   |
| Network:               | 2 x Gigabit Ethernet, RJ45's.  |
| Network:               | 1 x SFP fibre.   |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.  |
| DC Input:              | 4-pin 7.5A power jack socket, 10 14VDC.  |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.   |
| Equipment Type         |  |
| AVN-PD8D:              | 8 Stereo AES3 digital inputs & 8<br>stereo AES3 digital outputs, AES67<br>portal, with detailed meter displays               |
|                        |  |
| Physical Specification |  |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)  |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  |
| Weight:                | Nett: 2.82 kg Gross: 3.86kg<br>Nett: 6.20 lbs Gross:8.49lbs  |
|                        |  |
| Accessories            |  |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug  |
| AVN-HD1:               | Digital Headphone Amplifier.   |
|                        |  |

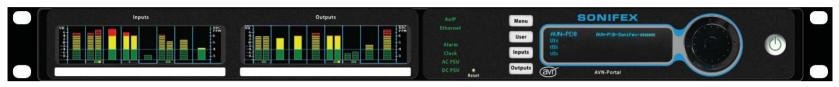








## AVN-PD8TD 8 Stereo AES3 Digital Inputs & 8 Stereo AES3 Digital Outputs on Terminal Blocks, AES67 Portal, with Detailed Meter Displays



AVN-PD8TD Front View.

The AVN-PD8TD is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight stereo AES3 digital inputs, eight stereo AES3 digital outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding. Physical audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67

stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering. can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created. supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams, or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/ modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PD8TD becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & Controls

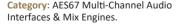
The AVN-PD8TD has the standard OLED front panel display, and the additional twin TFT meter displays. The OLED display provides











Product Function: Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

Typical Applications: A powerful AES3 & AES67 mix engine which allows for multiple applications: 8 output AES3 zone mixer, 8 channel digital mixer, 64 channel AES67 stream distribution amplifier from digital sources, 8 channel headphone distribution system (with AVN-HD1 units).

### Features:

· Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on terminal black connections.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### Multi-Channel Audio Mix Engine Interfaces

detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view.

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU. Metering can be set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical AES3 digital audio connections and GPIO, the AVN-PD8TD uses terminal block connections.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events

over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

The AVN-PD8TD is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight stereo digital AES3 inputs and eight stereo digital AES3 outputs on terminal black connections
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.

- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



### **Technical Specification For AVN-PD8TD**

| Audio-Over-IP Specific       | cation   |
|------------------------------|--|
| Open Standards:              | RAVENNA, AES67   |
| Device Discovery:            | Bonjour (mDNS / DNS-SD) or SAP                                 |
| Audio Delivery:              | RTP/UDP over IPv4 multicast                                    |
| QoS:                         | DiffServ   |
| Stream Management:           | RTSP/SDP   |
| Control:                     | Web server/Ember+  |
| Format:                      | Linear PCM 24-bit (L24)  |
| Channels Per Stream:         | Up to 8  |
| Frames Per Packet:           | 48   |
| Transmit Streams:            | Up to 8  |
| Sample Rate:                 | 48kHz  |
|                              |  |
| Ember+ Interface Con         | nection  |
| Interface Type:              | Provider   |
| Network Interface:           | Ethernet port and AoIP port                                    |
| Port:                        | 9000   |
|                              |  |
| <b>Timing Synchronisatio</b> | n  |
| Profile Support:             | Default, AES67 Media & Custom profiles                         |
| Timing Protocol:             | PTPv2, IEEE1588-2008   |
|                              |  |
| Stereo Digital Input         |  |
| Input Impedance:             | 110Ω balanced  |
| Supported Input              | ·  |
| Rates:                       | 32kHz, 44.1kHz, 48kHz, 88.2kHz,                                |
|                              | 96kHz, 176.4kHz and 192kHz<br>(sample rate converted to 48kHz) |
| Outnut Impodonce             | 110Ω balanced  |
| Output Impedance:            | TTOM DAIGHTER  |
| Supported Output<br>Rates:   | 48kHz  |
|                              |  |

| Connections            |  |
|------------------------|--|
| Inputs:                | 2 x 24-Pin Phoenix style terminal blocks (Digital pinout).                                       |
| Outputs:               | 2 x 24-Pin Phoenix style terminal blocks. (Digital pinout).                                      |
| GPIO:                  | 1x 24-Pin Phoenix style terminal blocks.   |
| Network:               | 2 x Gigabit Ethernet, RJ45's.  |
| Network:               | 1 x SFP fibre.   |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.              |
| DC Input:              | 4-pin 7.5A power jack socket, 10-14VDC.  |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.   |
|                        | stereo AES3 digital outputs on<br>terminal blocks, AES67 portal, with<br>detailed meter displays |
| Physical Specification |  |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)                       |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)                                 |
| Weight:                | Nett: 2.80kg Gross: 3.84kg<br>Nett: 6.16 lbs Gross: 8.45lbs                                      |
| Accessories            |  |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug  |
| AVN-HD1:               | Digital Headphone Amplifier.   |
|                        | •  |







## **RAVENNA**



## **AVN-PM8 8 Mic/Line Inputs, 8 Stereo Analogue Line Outputs, AES67 Portal**



AVN-PM8 Front View.

The AVN-PM8 is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight mic/line inputs with switchable phantom power, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port, Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67 stream

output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created. supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PM8 becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO, SNMP and Ember+ support.

### Front Panel Displays, Metering & **Controls**

The AVN-PM8 unit is supplied with the standard OLED front panel display, which provides detailed status information on











Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue beltpacks & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

 Eight balanced XLR mic/line inputs and eight stereo line outputs on RJ45 connectors using StudioHub® pinout.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- · Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PM8 Rear View.

device name, network addresses, PTP clocking info, power status/voltages, version information, and basic metering (The 'D' version adds twin TFT metering with configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PM8 uses eight mic/line XLR sockets for the inputs and eight RJ45 connectors using StudioHub® pinout for the stereo analogue line outputs. +48V phantom power is available for each microphone input with a red LED presence indication.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

# 8-Way Digital Headphone Distribution System

The AVN-PM8 StudioHub® outputs can be combined with multiple AVN-HA1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HA1 can be used as another GPI for muting the output.

The AVN-PM8D is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs

and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight balanced XLR mic/line inputs and eight stereo line outputs on RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).

- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PM8**

| echnical Specification For AVN-PING |   |  |
|-------------------------------------|---|--|
| udio-Over-IP Specific               | cation  |  |
| pen Standards:                      | RAVENNA, AES67                                      |  |
| evice Discovery:                    | Bonjour (mDNS / DNS-SD) or SAP                      |  |
| udio Delivery:                      | RTP/UDP over IPv4 multicast                         |  |
| loS:                                | DiffServ  |  |
| tream Management:                   | RTSP/SDP  |  |
| Control:                            | Web server/Ember+                                   |  |
| ormat:                              | Linear PCM 24-bit (L24)                             |  |
| hannels Per Stream:                 | Up to 8   |  |
| rames Per Packet:                   | 48  |  |
| ransmit Streams:                    | Up to 8   |  |
| ample Rate:                         | 48kHz   |  |
|                                     |   |  |
| mber+ Interface Con                 | nection   |  |
| nterface Type:                      | Provider  |  |
| letwork Interface:                  | Ethernet port and AoIP port                         |  |
| ort:                                | 9000  |  |
|                                     |   |  |
| iming Synchronisatio                | n   |  |
| rofile Support:                     | Default, AES67 Media & Custom profiles              |  |
| iming Protocol:                     | PTPv2, IEEE1588-2008                                |  |
|                                     |   |  |
| Aicrophone Inputs                   |   |  |
| nput Impedance:                     | > 2.5kΩ balanced                                    |  |
| iain Range:                         | 0dB to +60dB  |  |
| dBFS Line-Up:                       | Adjustable in steps of 3dB from<br>-58dBu to +2dBu  |  |
| requency Response:                  | 20Hz to 20kHz, +0/-0.2dB                            |  |
| loise:                              | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain        |  |
|                                     |   |  |
| alanced Line Inputs                 |   |  |
| nput Impedance:                     | > 20kΩ balanced                                     |  |
|                                     | 0dBFS Line-Up: Adjustable<br>+15/+18/+20/+22/+24dBu |  |
| requency Response:                  | 20Hz to 20kHz, +0/-0.2dB                            |  |
| HD+N:                               | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW     |  |
| loise:                              | -110dBFS, 20kHz BW, Rs=200Ω                         |  |
| rosstalk:                           | < -100dB  |  |
|                                     |   |  |
| ommon Mode                          |   |  |
| ejection:                           | > 70dB @ 1kHz                                       |  |
|                                     |   |  |

| <b>Balanced Line Output</b> | s   |
|-----------------------------|---|
| Output Impedance:           | < 50Ω balanced  |
| OdBFS Line-Up:              | Adjustable +15/+18/+20/+22/<br>+24dBu   |
| Frequency Response:         | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N:                      | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW   |
| Noise:                      | -110dBFS, 20kHz BW, Rs=200Ω   |
| Connections                 |   |
| Inputs:                     | XLR connectors with phantom power (toggle).   |
| Outputs:                    | RJ45 connections (StudioHub+ pinout).   |
| GPIO:                       | 1 D-Sub (DA-15) connections.  |
| Network:                    | 2 x Gigabit Ethernet, RJ45's.   |
| Network:                    | 1 x SFP fibre.  |
| Power:                      | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.                                       |
| DC Input:                   | 4-pin 7.5A power jack socket, 10-14VDC.   |
| Fuse Rating:                | Anti-surge fuse 2A 20mm x 5mm.  |
| Equipment Type              |   |
| AVN-PM8:                    | Advanced audio routing, metering and equalisation unit with analogue mic/line inputs, analogue outputs, and RAVENNA AOIP. |
|                             |   |
| Physical Specification      |   |
| Dimensions (Raw):           | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)   |
| Dimensions (Boxed):         | 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)   |
| Weight:                     | Nett: 2.98 kg Gross:4.02 kg<br>Nett: 6.56 lbs Gross:8.84 lbs  |
|                             |   |
| Accessories                 |   |
| AVN-DC060:                  | 60W DC power supply for AVN<br>range with KPJX-4S plug  |
| AVN-HA1:                    | Analogue Headphone Amplifier.   |









## AVN-PM8T Advanced Audio Routing, Metering and Equalisation Unit with Terminal Type Analogue Mic/Line Inputs, Terminal Type Analogue Outputs, and RAVENNA AoIP



AVN-PM8T Front View

The AVN-PM8T is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, and web-enabled control and configuration. It features eight mic/line inputs with switchable phantom power, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small form-factor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/ recognition, and optional Opus codec encoding & decoding. Physical digital audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or routed to any physical audio output or AES67 stream output, with gain adjustment at the input,

the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PM8T becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with













Interfaces & Mix Engines.



**Product Function:** Mix and route analogue & AES67 stream inputs to analogue & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue beltpacks & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

### Features:

 Eight balanced mic/line inputs and eight stereo line outputs on on terminal block connections.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).

- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).

remote handling via GPIO, VGPIO, SNMP and Ember+ support.

# Front Panel Displays, Metering & Controls

The AVN-PM8T unit is supplied with the standard OLED front panel display, which provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, version information, and basic metering (The 'D' version adds twin TFT metering with configurable ballistics). The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PM8T uses terminal block connections for the eight mic/line XLR inputs, the eight stereo analogue line outputs, and the GPIO. +48V phantom power is available for each microphone input with a red LED presence indication.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

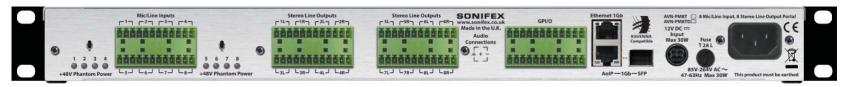
For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

The AVN-PM8T is part of the Sonifex AVN-Portal series, which includes the AVN-PA8

(analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight balanced mic/line inputs and eight stereo line outputs on on terminal block connections.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.

- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports (terminal block connectors).



### **Technical Specification For AVN-PM8T**

| Technical Specification For AVN-PM8T |   |  |  |  |
|--------------------------------------|---|--|--|--|
| Audio-Over-IP Specific               | cation  |  |  |  |
| Open Standards:                      | RAVENNA, AES67                                  |  |  |  |
| Device Discovery:                    | Bonjour (mDNS / DNS-SD) or SAP                  |  |  |  |
| Audio Delivery:                      | RTP/UDP over IPv4 multicast                     |  |  |  |
| QoS:                                 | DiffServ  |  |  |  |
| Stream Management:                   | RTSP/SDP  |  |  |  |
| Control:                             | Web server/Ember+                               |  |  |  |
| Format:                              | Linear PCM 24-bit (L24)                         |  |  |  |
| Channels Per Stream:                 | Up to 8   |  |  |  |
| Frames Per Packet:                   | 48  |  |  |  |
| Transmit Streams:                    | Up to 8   |  |  |  |
| Sample Rate:                         | 48kHz   |  |  |  |
|                                      |   |  |  |  |
| Ember+ Interface Con                 | nection   |  |  |  |
| Interface Type:                      | Provider  |  |  |  |
| Network Interface:                   | Ethernet port and AoIP port                     |  |  |  |
| Port:                                | 9000  |  |  |  |
|                                      |   |  |  |  |
| Timing Synchronisation               | on  |  |  |  |
| Profile Support:                     | Default, AES67 Media & Custom profiles          |  |  |  |
| Timing Protocol:                     | PTPv2, IEEE1588-2008                            |  |  |  |
|                                      |   |  |  |  |
| Microphone Inputs                    |   |  |  |  |
| Input Impedance:                     | > 2.5kΩ balanced                                |  |  |  |
| Gain Range:                          | 0dB to +60dB                                    |  |  |  |
| 0dBFS Line-Up:                       | Adjustable in steps of 3dB from -58dBu to +2dBu |  |  |  |
| Frequency Response:                  | 20Hz to 20kHz, +0/-0.2dB                        |  |  |  |
| Noise:                               | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain    |  |  |  |
| Balanced Line Inputs                 |   |  |  |  |
| Input Impedance:                     | > 20kΩ balanced                                 |  |  |  |
| OdBFS Line-Up:                       | Adjustable +15/+18/+20/+22/<br>+24dBu           |  |  |  |
| Frequency Response:                  | 20Hz to 20kHz, +0/-0.2dB                        |  |  |  |
| THD+N:                               | < -110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW    |  |  |  |
| Noise:                               | -110dBFS, 20kHz BW, Rs=200Ω                     |  |  |  |
| Crosstalk:                           | <-100dB   |  |  |  |
|                                      |   |  |  |  |

> 70dB @ 1kHz

Common Mode Rejection:

| Balanced Line Output   | s   |
|------------------------|---|
| Output Impedance:      | < 50Ω balanced  |
| OdBFS Line-Up:         | Adjustable +15/+18/+20/+22/<br>+24dBu   |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to 20kHz, 20kHz BW  |
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω   |
| Connections            |   |
| Inputs:                | 1 24-Pin Phoenix style terminal blocks. (Analogue pinout)   |
| Outputs:               | 2 24-Pin Phoenix style terminal blocks. (Analogue pinout)   |
| GPIO:                  | 1 24-Pin Phoenix style terminal blocks  |
| Network:               | 2 x Gigabit Ethernet, RJ45's  |
| Network:               | 1 x SFP fibre   |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.   |
| DC Input:              | 4-pin 7.5A power jack socket, 10-14VDC.   |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.  |
| Equipment Type         |   |
| AVN-PM8T:              | Advanced audio routing, metering and equalisation unit with terminal type analogue mic/line inputs, terminal type analogue outputs, and RAVENNA AoIP. |
| Physical Specification |   |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)  |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)  |
| Weight:                | Nett: 2.96 kg Gross: 4.00kg<br>Nett: 6.51 lbs Gross: 8.80lbs  |
| Accessories            |   |
| AVN-DC060:             | 60W DC power supply for AVN range with KPJX-4S plug   |
| AVN-HA1:               | Analogue Headphone Amplifier.   |

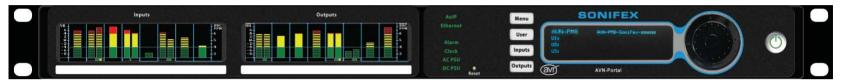








## AVN-PM8D Advanced Audio Routing, Metering and Equalisation Unit with Analogue Mic/Line Inputs, Analogue Outputs, RAVENNA AoIP, and a Detailed Customisable Display



AVN-PM8D Front View.

The AVN-PM8D is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight mic/line inputs with switchable phantom power, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small formfactor pluggable) port. Recent firmware additions include tone generation, audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and AES67 stream input can be mixed or

routed to any physical audio output or AES67 stream output, with gain adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams. or AES67-enabled Dante® flows (discovered using SAP). Input and output AES67 streams can be individually added/modified and the SDP of each stream can be checked and

edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PM8Dbecomes an advanced problemsolving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support, multiple stream outputs can be provided, combined with remote

Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

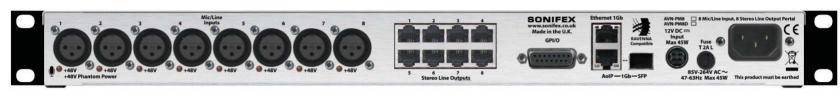
Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue belt-packs & AES67 streams, 8 channel • Twin TFT Meter displays with selectable headphone distribution system (with AVN-HA1 units).

### Features:

 Eight balanced XLR mic/line inputs and eight stereo line outputs on RJ45 connectors using StudioHub® pinout.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



AVN-PM8D Rear View.

handling via GPIO, VGPIO, SNMP and Ember+ support.

# Front Panel Displays, Metering & Controls

The AVN-PM8D has the standard OLED front panel display, and the additional twin TFT meter displays. The OLED display provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view.

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU.

Metering can be set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PM8D uses eight mic/line XLR sockets for the inputs and eight RJ45 connectors using StudioHub® pinout for the stereo analogue line outputs. +48V phantom power is available for each microphone input with a red LED presence indication.

There are two Ethernet RJ45 connections one for control and one for the AoIP network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as

inputs or outputs and provide softwarecontrolled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

# 8-Way Digital Headphone Distribution System

The AVN-PM8D StudioHub® outputs can be combined with multiple AVN-HA1 headphone amplifiers to provide a headphone distribution system. Output connections can supply analogue power to the satellite headphone amplifiers. Each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs. The switches on the front panel of the AVN-HA1 can be used as another GPI for muting the output.

### Multi-Channel Audio Mix Engine Interfaces

The AVN-PM8D is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight balanced XLR mic/line inputs and eight stereo line outputs on RJ45 connectors using StudioHub® pinout.
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.

- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

### **Technical Specification For AVN-PM8D**

| Open Standards:        | RAVENNA, AES67                                  |
|------------------------|---|
| Device Discovery:      | Bonjour (mDNS / DNS-SD) or SAP                  |
| Audio Delivery:        | RTP/UDP over IPv4 multicast                     |
| QoS:                   | DiffServ  |
| Stream Management:     | RTSP/SDP  |
| Control:               | Web server/Ember+                               |
| Format:                | Linear PCM 24-bit (L24)                         |
| Channels Per Stream:   | Up to 8   |
| Frames Per Packet:     | 48  |
| Transmit Streams:      | Up to 8   |
| Sample Rate:           | 48kHz   |
| Ember+ Interface Con   | nection   |
| Interface Type:        | Provider  |
| Network Interface:     | Ethernet port and AoIP port                     |
| Port:                  | 9000  |
| Timing Synchronisation | on .  |
| Profile Support:       | Default, AES67 Media & Custom profiles          |
| Timing Protocol:       | PTPv2, IEEE1588-2008                            |
| Microphone Inputs      |   |
| Input Impedance:       | > 2.5kΩ balanced                                |
| Gain Range:            | 0dB to +60dB                                    |
| OdBFS Line-Up:         | Adjustable in steps of 3dB from -58dBu to +2dBu |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB                        |
| Noise:                 | -127dBu, 20kHz BW, Rs=200Ω ref<br>76dB gain     |
| Balanced Line Inputs   |   |
| Input Impedance:       | > 20kΩ balanced                                 |
| OdBFS Line-Up:         | Adjustable +15/+18/+20/+22/<br>+24dBu           |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB                        |
| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW |
|                        |   |
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω                     |

> 70dB @ 1kHz

Common Mode Rejection:

| <b>Balanced Line Output</b>                     | s  |
|---|--|
| Output Impedance:                               | < 50Ω balanced   |
| OdBFS Line-Up:                                  | Adjustable +15/+18/+20/+22/<br>+24dBu  |
| Frequency Response:                             | 20Hz to 20kHz, +0/-0.2dB   |
| THD+N:  | < -110dBFS, -30dBFS, 20Hz to<br>20kHz, 20kHz BW  |
| Noise:  | -110dBFS, 20kHz BW, Rs=200Ω  |
| Connections                                     |  |
| AVN-PM8/D                                       |  |
| Inputs:   | XLR connectors with phantom power (toggle).  |
| Outputs:  | RJ45 connections (StudioHub+ pinout).  |
| GPIO:   | 1 D-Sub (DA-15) connections.   |
| Network:  | 2 x Gigabit Ethernet, RJ45's.  |
| Network:  | 1 x SFP fibre.   |
| Power:  | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.  |
| DC Input:                                       | 4-pin 7.5A power jack socket,<br>10-14VDC.   |
| Fuse Rating:                                    | Anti-surge fuse 2A 20mm x 5mm.   |
| Facilities and Tour                             |  |
| Equipment Type<br>AVN-PM8D:                     | Advanced audio routing, metering and equalisation unit with analogue mic/line inputs, analogue outputs, RAVENNA AoIP, and a detailed customisable display. |
|   |  |
| Physical Specification                          |  |
| Physical Specification<br>Dimensions (Raw):     | 48.3cm (W) x 17.5cm (D) x 4.4cm (H)(1U) 19" (W) x 6.9" (D) x 1.8" (H) (1U)   |
|   |  |
| Dimensions (Raw):                               | (H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)<br>55.5cm (W) x 29cm (D) x 18cm (H)  |
| Dimensions (Raw):  Dimensions (Boxed):  Weight: | (H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)<br>55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)<br>Nett: 2.92 kg Gross: 3.96kg           |
| Dimensions (Raw):  Dimensions (Boxed):          | (H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)<br>55.5cm (W) x 29cm (D) x 18cm (H)<br>22" (W) x 11.4" (D) x 7" (H)<br>Nett: 2.92 kg Gross: 3.96kg           |

. . . . . . . .

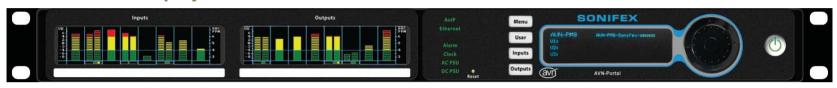








**AVN-PM8TD Advanced Audio Routing, Metering and Equalisation Unit with Terminal Type** Analogue Mic/Line Inputs, Terminal Type Analogue Outputs, RAVENNA AoIP, and a Detailed **Customisable Display** 



AVN-PM8TD Front View

The AVN-PM8TD is a powerful audio mix engine and AES67 interface with built-in mixer matrix, assignable DSP functions, GPIO, audio playback, web-enabled control and configuration, and twin TFT meter displays. It features eight mic/line inputs with switchable phantom power, eight stereo analogue line outputs, plus a dedicated AES67 Audio over IP (AoIP) portal, available via RJ45 Ethernet connection or an SFP (small formfactor pluggable) port. Recent firmware additions include tone generation. audio file playback, DTMF generation/recognition, and optional Opus codec encoding & decoding. Physical audio inputs and outputs, and GPIO provided on terminal block connectors.

At the heart of this unit is a powerful mix engine with web-enabled configuration and control. Any physical audio input and

AES67 stream input can be mixed or routed to any physical audio output or AES67 stream output, with gain

adjustment at the input, the mix point or the output. DSP functions, such as gain and filtering, can be added at inputs, outputs and cross-points.

Up to 16 x AES67 input channels and 64 x AES67 output channels can be created, supporting the full range of AES67 packet times and channel counts. It can send or receive RAVENNA and AES67 AoIP streams, or AES67-enabled Dante® flows (discovered using SAP). Input and output

AES67 streams can be individually added/ modified and the SDP of each stream can be checked and edited. The unit can act as a PTP master clock or slave clock and supports IEEE1588-2008 PTPv2 media and default profiles.

With this flexibility, the AVN-PM8TD becomes an advanced problem-solving box for any applications where monitoring of inputs and outputs and mixing of signals is required. With native AES67 support,



Category: AES67 Multi-Channel Audio Interfaces & Mix Engines.

**Product Function:** Mix and route AES3 & AES67 stream inputs to AES3 & AES67 stream outputs.

Typical Applications: A powerful analogue & AES67 mix engine which allows for multiple applications: 8 output analogue zone mixer, 8 channel clean-feed generator, 64 channel AES67 stream distribution amplifier, IFB distribution to analogue belt-packs & AES67 streams, 8 channel headphone distribution system (with AVN-HA1 units).

#### Features:

 Eight balanced mic/line inputs and eight stereo line outputs on on terminal block connections.

- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- · Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.

- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.

multiple stream outputs can be provided, combined with remote handling via GPIO, VGPIO. SNMP and Ember+ support.

# Front Panel Displays, Metering & Controls

The AVN-PM8TD has the standard OLED front panel display, and the additional twin TFT meter displays. The OLED display provides detailed status information on device name, network addresses, PTP clocking info, power status/voltages, and version information. The TFT screens provide live display physical input and output levels. A rotary navigation control can be used to select a single input or output and view its metering data in a more detailed horizontal view."

The metering scale used is user configurable to one of nine different metering scales, with relevant ballistics. The metering scales available are: Dual PPM + Standard VU, EBU PPM, BBC PPM, Nordic PPM, AES Digital PPM, DIN PPM, German PPM, SMPTE RP.0155, Standard VU & Extended VU. Metering can be set to be either 'Discrete' or 'Continuous', which changes the appearance of the meter bar. Phase metering can be displayed per stereo channel, and channel idents can be shown either above or below the metering to identify each input/output.

The display and navigation controls allow editing of certain functions, limited to networking (IP addresses, friendly name, etc) and display (brightness and contrast). The front panel controls also include user configurable buttons which can be set-up to perform actions such as activating a GPIO or as a shortcut button to jump to a specified menu on the OLED display.

Front panel LEDs show the AoIP network status, synchronisation status and the status of the AC and DC power supply inputs. The brightness of the OLED display and LED indicators can be continuously adjusted for low or high lighting conditions.

A front panel power button is available to turn the unit on and off. The power button is disabled by default but can be enabled through the 'Display Settings' web page.

#### **Physical Inputs & Outputs**

For physical audio connections, the AVN-PM8TD uses terminal block connections for the eight mic/line XLR inputs, the eight stereo analogue line outputs, and the GPIO. +48V phantom power is available for each microphone input with a red LED presence indication.

There are two Ethernet RJ45 connections - one for control and one for the AoIP

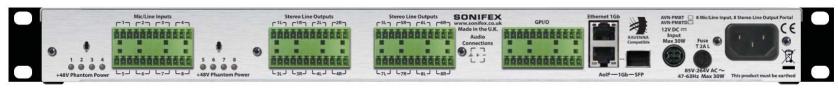
network. There is also an Ethernet SFP module that, when used, replaces the AoIP RJ45 connection, e.g. for a 1Gbit/s copper or optical SFP transceiver. When an SFP is used, this replaces the AoIP RJ45 connection.

A rear panel GPIO connector provides 10 local ports which can be user configured as inputs or outputs and provide software-controlled functionality. A voltage free relay contact can be used to operate external equipment. There are virtual GPIO ports which can be used to trigger events over the network between devices.

For remote operation and monitoring, SNMP V2 is supported and the units can be controlled using Ember+ commands.

The AVN-PM8TD is part of the Sonifex AVN-Portal series, which includes the AVN-PA8 (analogue physical I/O), the AVN-PD8 (digital physical I/O), and the AVN-PM8 (mic inputs, line outputs). Each of those has three additional variants indicated by letter suffixes: The AVN-PA8D features two bright TFT meter displays which provide a live display of the levels of the physical inputs and outputs respectively; the AVN-PA8T features rear panel terminal block connectors for all physical inputs and outputs; the AVN-PA8TD features both the TFT displays and terminal block connections.

- Eight balanced mic/line inputs and eight stereo line outputs on on terminal block connections
- AES67 portal on 1Gb Ethernet (RJ45) and 1Gb SFP ports.
- Gain and DSP functions available at inputs, outputs, and crosspoints.
- Twin TFT Meter displays with selectable ballistics, plus phase and channel ident display.
- Responsive webserver software controlled router/mixer.
- Up to eight AoIP input streams with a maximum of 16 channels to be routed.
- Up to eight AoIP output streams with a maximum of eight channels each (64 channels).
- · Audio file playback.
- DTMF tone generation/recognition.
- Optional multi-channel Opus codec encoding and decoding.
- Dual AC and DC power supply inputs.
- 10 user assignable GPIO ports.



### **Technical Specification For AVN-PM8TD**

#### Audio-Over-IP Specification

| Open Standards:      | RAVENNA, AES67                 |
|----------------------|--------------------------------|
| Device Discovery:    | Bonjour (mDNS / DNS-SD) or SAP |
| Audio Delivery:      | RTP/UDP over IPv4 multicast    |
| QoS:                 | DiffServ                       |
| Stream Management:   | RTSP/SDP                       |
| Control:             | Web server/Ember+              |
| Format:              | Linear PCM 24-bit (L24)        |
| Channels Per Stream: | Up to 8                        |
| Frames Per Packet:   | 48                             |
| Transmit Streams:    | Up to 8                        |
| Sample Rate:         | 48kHz                          |

# Ember+ Interface Connection Interface Type: Provider Network Interface: Ethernet port and AoIP port Port: 9000

#### **Timing Synchronisation**

| Default, AES67 Media & Custom<br>profiles |
|---|
| PTPv2, IEEE1588-2008                      |
|   |

| Microphone Inputs   |   |
|---------------------|---|
| Input Impedance:    | > 2.5kΩ balanced                                |
| Gain Range:         | 0dB to +60dB                                    |
| OdBFS Line-Up:      | Adjustable in steps of 3dB from -58dBu to +2dBu |
| Frequency Response: | 20Hz to 20kHz, +0/-0.2dB                        |
| Noise:              | -127dBu, 20kHz BW, Rs=200Ω ref.<br>76dB gain    |

| Balanced Line Inputs |  |
|----------------------|--|
| Input Impedance:     | > 20kΩ balanced                                |
| OdBFS Line-Up:       | Adjustable +15/+18/+20/+22/<br>+24dBu          |
| Frequency Response:  | 20Hz to 20kHz, +0/-0.2dB                       |
| THD+N:               | < -110dBFS, -30dBFS, 20Hz to 20kHz<br>20kHz BW |
| Noise:               | -110dBFS, 20kHz BW, Rs=200Ω                    |
| Crosstalk:           | < -100dB                                       |

| Common Mode |               |
|-------------|---------------|
| Rejection:  | > 70dB @ 1kHz |

#### **Balanced Line Outputs**

AVN-DC060:

| Output Impedance:      | < 50Ω balanced   |
|------------------------|--|
| 0dBFS Line-Up:         | Adjustable +15/+18/+20/+22/<br>+24dBu  |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB   |
| THD+N:                 | < -110dBFS, -30dBFS, 20Hz to 20kHz,<br>20kHz BW  |
| Noise:                 | -110dBFS, 20kHz BW, Rs=200Ω  |
| Connections            |  |
|                        | 1 24-Pin Phoenix style terminal  |
| Inputs:                | blocks. (Analogue pinout)  |
| Outputs:               | 2 24-Pin Phoenix style terminal blocks. (Analogue pinout)  |
| GPIO:                  | 1 24-Pin Phoenix style terminal blocks   |
| Network:               | 2 x Gigabit Ethernet, RJ45's   |
| Network:               | 1 x SFP fibre  |
| Power:                 | Mains AC Input: Universal filtered IEC, continuously rated 85-264VAC, 47-63Hz, 20W.  |
| DC Input:              | 4-pin 7.5A power jack socket,<br>10-14VDC.   |
| Fuse Rating:           | Anti-surge fuse 2A 20mm x 5mm.   |
| Equipment Type         |  |
| AVN-PM8TD:             | Advanced audio routing, metering and equalisation unit with terminal type analogue mic/line inputs, terminal type analogue outputs, RAVENNA AoIP, and a detailed customisable display. |
|                        |  |
| Physical Specification |  |
| Dimensions (Raw):      | 48.3cm (W) x 17.5cm (D) x 4.4cm<br>(H)(1U)<br>19" (W) x 6.9" (D) x 1.8" (H) (1U)   |
| Dimensions (Boxed):    | 55.5cm (W) x 29cm (D) x 18cm (H) 22" (W) x 11.4" (D) x 7" (H)  |
| Weight:                | Nett: 2.90kg Gross: 3.94kg<br>Nett: 6.38 lbs Gross: 8.67lbs  |
|                        |  |
| Accessories            |  |

60W DC power supply for AVN range with KPJX-4S plug

Analogue Headphone Amplifier.



# AVN-HA1 Analogue Headphone Amp for AVN-PA8/D & AVN-PM8/D Portals















**Category:** Headphone Distribution Amplifiers.

Product Function: Simple headphone amplifier connected to the output of the AVN-PA8/PM8 portals, with volume control and button tally. Up to 8 can be connected to each AVN portal.

Typical Applications: Headphone amplifiers in a radio studio, a talks studio of a radio station/TV station, a voice-over booth or news booth.

- Front panel 6.35mm (1/4")
  headphone socket and volume
  control knob, with mute/GPO push
  button.
- Locking DC power connector if a portal is not being used to supply the unit with power.
- Analogue audio input on RJ45 (the connector provides power to the unit and a GPO back to the portal).
- Loop through audio output on RJ45 (power and GPO signal are not connected).

The AVN-PA8/D and the AVN-PM8/D Portals can be combined with multiple **AVN-HA1** headphone amplifiers to provide a headphone distribution system – the portal output connections can supply analogue power to satellite headphone amplifiers and the unit can be mounted using the CM-MNT1 desk mount panel.

The AVN-PA8/D and AVN-PM8/D can be combined with multiple Sonifex **AVN-HA1** headphone amplifiers to provide 8 separate headphone signals where each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs.

- Front panel 6.35mm (1/4") headphone socket and volume control knob.
- Front panel Mute/GPO push button.

- Analogue audio input on RJ45 (the connector provides power to the unit and a GPO back to the portal).
- Loop through audio output on RJ45 (power and GPO signal are not connected).
- Locking DC power connector (if a portal is not being used to supply the unit with power).

**Note:** The AVN-HA1 is an analogue input product taking an analogue audio feed from the AVN portals. It can be used independently of the portals by using the separate DC input for power and a separate analogue input.

## **Technical Specification For AVN-HA1**

| Parameter Output Impedance: headphones. | Drives 150mW into 32 $\Omega$ to 600 $\Omega$   |
|---|---|
| Max Input Level:                        | +24dBu  |
| Max Output Level:                       | +18dBu  |
| Frequency Response:                     | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N:                                  | <-90dBu, -12dBu, 20Hz to 20kHz,<br>unity gain, 20kHz BW   |
| Noise:                                  | -92dBu, A weighted  |
| Crosstalk:                              | -90dB, 1kHz, +4dBu  |
| Common Mode:                            | -40dB, 1kHz   |
| Cable:                                  | Cat5/Cat5e/Cat6, Maximum length<br>of 20m between Portal and AVN-<br>HA1 if no external PSU is used |
|   |   |

|                        | PA8/D & AVN-PM8/D Portals   |
|------------------------|---|
|                        |   |
| Physical Specification | ı   |
| Dimensions: (Raw)      | 7.7cm (W) x 8.3cm (D) x 4.2cm (H)<br>3.0" (W) x 3.3" (D) x 1.7" (H) |
| Dimensions (Boxed):    | 12.7cm (W) x 22.9cm (D) x 7.6cm (H) 5" (W) x 9" (D) x 3" (H)        |
| Weight:                | Nett: 0.22kg Gross: 0.33kg<br>Nett: 0.49lbs Gross: 0.73lbs          |
|                        |   |
| Accessories            |   |
| CM-MNTI                | Desk Mount Pane   |

Analogue Headphone Amp for AVN-

Equipment Type AVN-HA1:





# **AVN-HD1** Digital Headphone Amp for AVN-PD8/D Portal



**Category:** Headphone Distribution Amplifiers.

Product Function: Simple AES3 digital headphone amplifier connected to the output of the AVN-PD8/D portals, with volume control and button tally. Up to 8 can be connected to each AVN portal.

Typical Applications: Headphone amplifiers in a radio studio, a talks studio of a radio station/TV station, a voice-over booth or news booth.

- Front panel 6.35mm (1/4";) headphone socket and volume control knob, with mute/GPO push button.
- Locking DC power connector if a portal is not being used to supply the unit with power.
- AES3 digital input on RJ45 (the connector provides power to the unit and a GPO back to portal).
- AES digital output on RJ45 (power and GPO signal are not connected).

The AVN-PD8/DPortal can be combined with multiple **AVN-HD1** headphone amplifiers to provide a headphone distribution system – the portal output connections can supply analogue power to satellite headphone amplifiers and the unit can be mounted using the CM-MNT1 desk mount panel.

The AVN-PD8/D can be combined with multiple Sonifex AVN-HD1 headphone amplifiers to provide 8 separate headphone signals where each headphone amplifier can be sent a separate feed, mixed from any physical or stream inputs.

 Front panel 6.35mm ( ¼") headphone socket and volume control knob.

- Front panel Mute/GPO push button.
- AES3 digital input on RJ45 (the connector provides power to the unit and a GPO back to the portal).
- AES3 digital output on RJ45 (power and GPO signal are not connected).
- Locking DC power connector if a portal is not being used to supply the unit with power.

Note: The AVN-HD1 is an AES3 digital input product taking an AES3 audio feed from the AVN portals. It can be used independently of the portals by using the separate DC input for power and a separate AES3 audio input.

#### **Technical Specification For AVN-HD1**

| Parameter              |  |
|------------------------|--|
| Output Impedance:      | Drives 150mW into $32\Omega$ to $600\Omega$ headphones   |
| D to A Line-up:        | -6dBFS = +18dBu  |
| Frequency Response:    | 20Hz to 20kHz, +0/-0.2dB   |
| THD+N:                 | <-90dBu, -30dBFS, 20Hz to 20kHz,<br>unity gain, 20kHz BW   |
| Noise:                 | -92dBu, A weighted   |
| Crosstalk:             | -90dB, 1kHz, +14dBFS   |
| Cable:                 | Cat5/Cat5e/Cat6, Maximum length of<br>20m between Portal and AVN-HD1 if<br>no external PSU is used |
| Equipment Type         |  |
| AVN-HD1:               | Digital Headphone Amp for AVN-<br>PD8/D Portal   |
| Physical Specification |  |
|                        |  |
| Dimensions: (Raw)      | 7.7cm (W) x 8.3cm (D) x 4.2cm (H)<br>3.0" (W) x 3.3" (D) x 1.7" (H)                                |
| Dimensions (Boxed):    | 12.7cm (W) x 22.9cm (D) x 7.6cm (H) 5" (W) x 9" (D) x 3" (H)                                       |
| Weight:                | Nett: 0.22kg Gross: 0.33kg<br>Nett: 0.49lbs Gross: 0.73lbs   |
|                        |  |

Desk Mount Panel

Accessories

CM-MNTI











# AVN-MPPR 4 Channel Presenter In-Ear Monitoring Remote Controller, AES67



Category: AES67 Presenter In Ear Monitoring System.

Product Function: The AVN-MPPR presenter remote provides the mixed audio to the presenter's earpiece via front panel 14" & 3.5mm headphone outputs, together with a rear panel XLR output.

**Typical Applications: Production** galleries, control rooms and studios.

- 4 rotary encoders control the 4 channels of a selected virtual mixer.
- 1. XLR. plus ¼":& 3.5mm headphone outputs.
- Software-configurable output level control (one for each headphone output) to set the attenuation of the signal.
- Virtual GPO.
- IP address reset.
- · Can be mounted on a standard microphone screw thread or under a desktop using optional accessories.

The presenter mixer system allows on-air presenters to adjust the audio levels fed to their radio or wired IEM systems, whilst also allowing Sound Control Room operators to adjust the levels remotely. The system consists of 3 types of unit all connected on a common Ravenna Audio over IP (AoIP) network with added audio sources derived from Ravenna based AoIP mixers or input portals. Setup and control can be provided via webpages and/or 3rd party applications. The system uses mono audio throughout.

The 3 units are:

AVN-PX8x4C - a 2U rack mounted Mix Engine Rack.

AVN-MPPR - a small desk, or microphone stand, mounted Presenters' Remote Unit. AVN-MPTR - a 1U rack mounted Technicians' Remote Unit.

The AVN-MPPR presenter remote provides the mixed audio to the presenter's earpiece via front panel ¼" & 3.5mm headphone outputs, together with a rear panel XLR output.

The four rotary encoders control the 4 channels of a selected virtual mixer, and give both visual feedback for mix level of the source and actual input level.

Multiple units can simultaneously control any of the 8 virtual mixers in the main mix engine with the VM number displayed on the front panel.

Audio is transported from the mix engine to the remote using AES67 AoIP with a single PoE Neutrik ™ Ethercon network connection.

Units can be mounted on a standard microphone screw thread or under a desktop using optional accessories.

By default, the audio on the XLR output on the rear of the unit is not muted when headphones are connected to one of the sockets on the front of the unit. However, there is a "Headphone XLR Mute" configuration setting in the "Misc Settings" section of the webpages to enable the muting of the XLR output on headphone connection if required.

The headphones will have an additional software-configurable output level control (one for each headphone output) to set the attenuation of the signal to the headphone driver for comfortable levels at normal signal level from the output.

GPO Button - The button between the STATUS LEDs and the rotary controls provides a virtual GPO from the unit that







can be used on the Mix Engine to trigger a GPO on its remote port. This button will illuminate red when it has been selected by this MPPR and will be yellow when initiated by a paralleled MPPR or MPTR.

IP Address Reset - As well as being able to set the IP address of the unit via the web pages, it is also possible to reset the IP address of the unit from the front panel to a factory default.

At the rear of the AVN-MPPR there are three connectors and two controls:

**Network Connector** - This is a Neutrik etherCON RJ45 CAT5e connector with latching lock with PoE plus for unit power.

Mix Output Mono Analogue Output - The mix output is a XLR 3 pin XLR male output which is electronically balanced.

Standard GPIO Connector - This is a 15-way female 'D'-type connector with the same pin connects as for the AVN-PX8x4C Mix Engine.

- 4 rotary encoders control the 4 channels of a selected virtual mixer.
- 1, XLR, plus ¼" & 3.5mm headphone outputs.
- Software-configurable output level control (one for each headphone output) to set the attenuation of the signal.
- Virtual GPO.
- IP address reset.
- Can be mounted on a standard microphone screw thread or under a desktop using optional accessories.

#### **Technical Specification For AVN-MPPR**

| Parameter                               |   |
|---|---|
| Output Impedance<br>Line Input:         | <50Ω balanced   |
| Output Impedance<br>Headphone Output:   | Drives 150mW into $8\Omega$ to $600\Omega$ headphones                   |
| OdBFS Line-up Line<br>Input:            | +15/+18/+20/+22/+24dBu<br>Balanced*                                     |
| OdBFS Line-up<br>Headphone Output:      | +20dBu Maximum when no attenuation factor applied.                      |
| Frequency Response<br>Line Input:       | 20Hz to 20kHz, +0/-0.5dB  |
| Frequency Response<br>Headphone Output: | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N Line Input:                       | <-110dBFS, -30dBFS, 20Hz to 20kHz,<br>All input gain settings, 20kHz BW |
| THD+N Headphone<br>Output:              | <-108dBFS, -30dBFS, 20Hz to 20kHz,<br>unity gain, 20kHz BW              |
| Noise Line Input:                       | -110dBFS, 20kHz BW, Rs=200Ω   |
| Noise Headphone<br>Output:              | -110dBFS, 20kHz BW  |
|   |   |
| PoE Power                               |   |
| Standard:                               | 802.3af   |
| Class:                                  | 3   |
| PD Power Range:                         | 6.49W to 12.95W   |

| Typical PSE Power Usage:      | 14W   |
|-------------------------------|---|
| Max PSE Power Usage           | e: 15.4W  |
|                               |   |
| Equipment Type                |   |
| AVN-MPPR                      | 4 Channel Presenter In-Ear<br>Monitoring Remote Controller,<br>AES67  |
|                               |   |
| <b>Physical Specification</b> | n   |
| Dimensions: (Raw)             | 20cm (W) x 15cm (D) x 4.2 (H) (1U) 7.9" (W) x 5.9" (D) x 1.7" (H) (1U)  |
| Dimensions (Boxed):           | 28cm (W) x 19cm(D) x 9cm(H)<br>11" (W) x 7.5(D) x 3.5(H)  |
| Weight:                       | Nett: 0.6kg Gross: 1.13kg<br>Nett: 1.3lbs Gross: 2.45lbs  |
| Dimensions (Boxed):           | 7.9" (W) x 5.9" (D) x 1.7" (H) (1)<br>28cm (W) x 19cm(D) x 9cm(H)<br>11" (W) x 7.5(D) x 3.5(H)<br>Nett: 0.6kg Gross: 1.13kg |



AVN-MPPR Rear View.









## **AVN-MPTR Technician Remote Controller**



Category: AES67/Dante AoIP Products.

Product Function: Allows selection of each of 8 VMs and remote adjustment of volume controls.

**Typical Applications: Production** galleries, control rooms and studios.

- Allows selection of each of 8 VMs and remote adjustment of volume controls.
- · 8 virtual mixer select buttons.
- 4 x rotary encoders showing input level metering and output volume control.
- · GPO enable and activation buttons.
- 1Gb PoE Ethernet and power using Neutrik Ethercon.
- · Status LEDs and OLED display.
- 10 x GPIO ports.
- · Configured via webserver.

The presenter mixer system allows on-air presenters to adjust the audio levels fed to their radio or wired IEM systems, whilst also allowing Sound Control Room operators to adjust the levels remotely. The system consists of 3 types of unit all connected on a common Ravenna Audio over IP (AoIP) network with added audio sources derived from Ravenna based AoIP mixers or input portals. Setup and control can be provided via webpages and/or 3rd party applications. The system uses mono audio throughout.

#### The 3 units are:

AVN-PX8x4C - a 2U rack mounted Mix Engine Rack.

AVN-MPPR - a small desk, or microphone stand, mounted Presenters' Remote Unit. AVN-MPTR - a 1U rack mounted Technicians' Remote Unit.

The AVN-MPTR Technician Remote Unit allows a technician to have control of a Virtual Mixer, paralleling the Presenter Remote Unit controls.

The MPTRs are assigned to a gallery and hence are normally tied to a Mix Engine. Note that no audio is present on the

Technician's Remote Unit - the monitor audio for this is provided by an output on the Mix Engine routed to the technician's area, but this product has similar connectivity to the MPPR and we can supply audio enabled versions of the MPTR if required.





AVN-MPTR Front View.

The Virtual Mixer (VM) to be controlled is selectable using the Virtual Mixer buttons labelled Pres 1 – Pres 8 on the front panel. The mixer selection buttons are only active when the AVN-MPTR is grabbed by a mix engine (as indicated by a green Auth LED). When a VM is selected, the corresponding button will be illuminated green (or blue if MPPR lock out is set for that VM). If no VM is selected, then all the mixer select button LEDs will be off. Once a VM has been selected, the rotary encoders act in the same way as on the Presenter Unit.

The AVN-MPTR is able stop MPPRs that are assigned to a particular VM from controlling the source levels.

- Allows selection of each of 8 VMs and remote adjustment of volume controls.
- 8 virtual mixer select buttons.
- 4 x rotary encoders showing input level metering and output volume control.
- · GPO enable and activation buttons.
- 1Gb PoE Ethernet and power using Neutrik Ethercon.
- Status LEDs and OLED display.
- 10 x GPIO ports.
- · Configured via webserver.

#### **Technical Specification For AVN-MPTR**

| PoE Power                |   |
|--------------------------|---|
| Standard:                | 802.3af   |
| Class:                   | 3   |
| PD Power Range:          | 0.44W to 12.94W   |
| Typical PSE Power Usage: | 7W  |
| Max PSE Power Usage:     | 15.4W   |
| Equipment Type           |   |
| AVN-MPTR                 | Technician Remote Controller  |
| Physical Specification   |   |
| Dimensions: (Raw)        | 48cm (W) x 15.0cm(D) x4.2cm (H) (1U) 19"(W) x 5.9" (D) x1.7" (H) (1U) |
| Dimensions (Boxed):      | 56cm (W) x 29.5cm (D) x 16.5cm (H)<br>22" (W) x 11.6" (D) x 6.5" (H)  |
| Weight:                  | Nett: 1.9kg Gross: 2.8kg<br>Nett: 4.2lbs Gross: 6.2llbs               |
|                          |   |



AVN-MPTR Rear View.









# AVN-PX8X4C 8 x 4 Channel Mix Engine, 24 Inputs, 16 Outputs, AoIP AES67



Category: AES67/Dante AoIP Products.

Product Function: The mix engine houses the hardware where the complex routing. mixing and DSP functions are performed.

Typical Applications: Production galleries. control rooms and studios.

 8 virtual mixers, each with 4 mono channel inputs to one mono output.

- · Virtual mixers controlled by connected technician and presenter remote controls.
- · Sophisticated configuration via webserver.
- 24 analogue inputs and 16 analogue outputs on D-type connectors.
- 32 logical inputs & amp; 10 logical outputs using AoIP.
- Dual hot-swappable AC power supplies.
- 2 x 1Gb Ethernet ports & amp; 1 x SFP port.
- · Status LEDs and OLED display.
- 20 x configurable GPIO ports.

The presenter mixer system allows on-air presenters to adjust the audio levels fed to their radio or wired IEM systems, whilst also allowing Sound Control Room operators to adjust the levels remotely. The system consists of 3 types of unit all connected on a common Ravenna Audio over IP (AoIP) network with added audio sources derived from Ravenna based AoIP mixers or input portals. Setup and control can be provided via webpages and/or 3rd party applications. The system uses mono audio throughout.

#### The 3 units are:

AVN-PX8x4C - a 2U rack mounted Mix Engine Rack.

AVN-MPPR- a small desk, or microphone stand, mounted Presenters' Remote Unit. AVN-MPTR - a 1U rack mounted Technicians' Remote Unit.

The AVN-PX8X4C Mix Engine, as well as using AES67 audio streams, accepts 24 analogue mono input channels on 3 x DB25 connectors and output 16 analogue mono channels on 2 x DB25 connectors. Each mixer unit is able to create and control 8 virtual mixers, each capable of mixing any 4 sources (analogue or AoIP) to a single mixed output that is available locally at an output and via an AoIP stream (to the Presenter

Unit and other destinations). The analogue audio line-up level will default to +18dBU = 0dBFS.

A built-in webserver is used for configuration settings, input/output routing and for defining which Presenter and Technician Remote Units are assigned to the particular Mix Engine and Virtual Mixers. Up to 4 MPPRs can be assigned to each Virtual





AVN-PX8X4C Front View

Mixer on a first-come first-served basis. It's also possible that both MPTRs could also select the same VM so a maximum of 6 devices can control the mix levels of a VM.

The Mix Engine will allow a maximum of 32 audio over IP input streams (each stream can be up to 8 channels). There are 8 virtual mixer outputs available. Each Virtual Mixer has 4 sources and 1 mixed output. There are also 2 more monitor outputs which play out the audio of the VM selection made on each of the 2 connected Technician Remote Units.

The Mix Engine has dual universal AC power supplies which are hot-swappable in the event of failure and have a green LED to indicate within spec operation. The front panel has LED indications and GPOs to warn of power supply failure. There are also front panel LEDs for network and sync/clock failure.

- 8 virtual mixers, each with 4 mono channel inputs to one mono output.
- Virtual mixers controlled by connected technician and presenter remote controls.
- Sophisticated configuration via webserver.
- 24 analogue inputs and 16 analogue outputs on D-type connectors.
- 32 logical inputs & amp; 10 logical outputs using AoIP.
- Dual hot-swappable AC power supplies.
- 2 x 1Gb Ethernet ports & Dort.
- · Status LEDs and OLED display.
- 20 x configurable GPIO ports.

#### Technical Specification For AVN-PX8x4C

AVN-PX8x4C Audio Inputs – Gain setting 0dBFS = +18dBu unless otherwise stated

| Parameter                             | Line Input  |
|---------------------------------------|---|
| Input Impedance:                      | >20kΩ balanced  |
| OdBFS line-up:<br>Balanced*           | +15/+18/+20/+22/+24dBu  |
| Frequency Response:                   | 20Hz to 20kHz, +0/-0.2dB  |
| THD+N:                                | <-110dBFS, -30dBFS, 20Hz to 20kHz,<br>All input gain settings, 20kHz BW |
| Noise:                                | -110dBFS, 20kHz BW, Rs=200Ω   |
| Crosstalk:                            | <-100dB   |
| Common Mode<br>Rejection:<br>settings | >70dB @ 1kHz, All input gain  |
|                                       |   |

AVN-PX8x4C Audio Outputs – Gain setting 0dBFS = +18dBu unless otherwise stated

| Parameter                                | Line Output   |
|--|---|
| Output Impedance:                        | <50Ω balanced   |
| OdBFS line-up:<br>Balanced*              | +15/+18/+20/+22/+24dBu  |
| Frequency Response:                      | 20Hz to 20kHz, +0/-0.5dB  |
| THD+N:                                   | <-110dBFS, -30dBFS, 20Hz to 20kHz,<br>All input gain settings, 20kHz BW |
| Noise:                                   | -110dBFS, 20kHz BW, Rs=200Ω   |
| *+18dB, +22dB and +2<br>and +20dB in DSP | 24dB are handled in hardware, +15dB                                     |

AVN-PX8x4C Power – Dual redundant IEC 85 - 264VAC 47 – 63Hz Max 60W 2Amp Fused type T 2A L

Equipment Type

AVN-PX8X4C 8 x 4 Channel Mix Engine, 24 Inputs, 16 Outputs, AoIP AES67

Physical Specification

Dimensions: (Raw) 48cm (W) x 27cm (D) x 8.8cm (H) (1U) 19" (W) x 10.6" (D) x 3.5" (H) (1U)

Dimensions (Boxed): 59cm (W) x 49cm (D) x 38cm (H) 23" (W) x 19" (D) x 15" (H)

Weight: 12.51b c gross: 7.2kg

Nett: 12.51b c gross: 7.2kg

Nett: 12.51b c gross: 15.8llbs



AVN-PX8X4C Rear View.

# SONIFEX

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