

Robust Location Two Channel Dante® / AES67 Network Audio Interface





AoIP22 MKII

Highlights

Powered by PoE

Two Balanced
Output Channels

Dante® / AES67 Network Audio

Up to 96k/ 24 bit

Two Balanced Input Channels

Robust Aluminium Construction

Overview

Packaged in a small & rugged format and designed for outside broadcast, theatre and location applications.

The AoIP22 MKII is designed to easily and quickly interface existing analogue equipment to a Dante® / AES67 network audio system. Being powered by PoE means that only one cable needs to be connected to the network to carry both audio and power, providing flexibility and saving time on installation.

Robust proven construction techniques, simple reliable interface and excellent specification will help make your technician's life hassle free. Whilst the low cost and long asset life will keep the accountant satisfied.







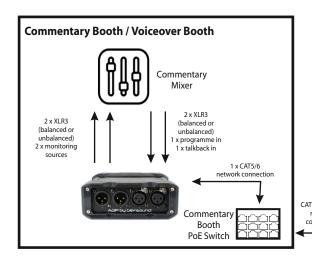


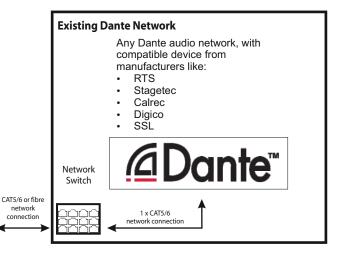


EXAMPLE APPLICATION

Interface To Existing Dante Network

Commentary Unit Connection To An Existing Dante Network





There are over 150 manufacturers that design Dante compatible equipment. In this example, the broadcaster has an established Dante® network. This is most likely to be an audio console, or an intercom system.

The broadcaster needs to expand the network by adding an existing commentary booth / voice over position. Using the AoIP22MKII, the commentary position can be added to the existing Dante® network very cost effectively.

The AoIP22 MKII is located in the commentary booth, and connected into the broadcast network via a single CAT5/6 cable.

The Glensound GS-CU008A is a popular and widely used commentary unit by broadcasters. It provides a programme output, and has a talkback outputs These connect to the inputs of the AoIP22 MKII via XLR connections.

The AoIP22 MKII outputs connect to the headphone monitoring inputs on the GS-CU008A via XLR. This allows the commentator to hear 2 audio channels. These are typically the programme audio, and talkbackreturn.

The audio routing can be configured via the Dante® Controller software. Simply run Dante® Controller on any PC connected to the network. All available sources and destinations will be shown on the AoIP22 MKII, and all other Dante® compatible devices on the network. Click the check boxes to route the audio in the desired directions, and the system is now ready to be used.

Audio routing only needs to be configured by Dante® Controller on the first time the AoIP22 MKII is connected to the network as all routes are stored locally on the device.





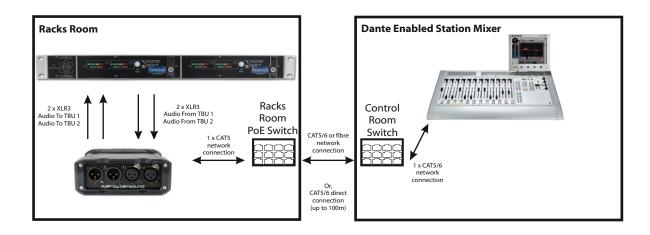




EXAMPLE APPLICATION

Interface To Existing Codec or TBU in Racks Room

Two channel audio interface to existing equipment



The AoIP22 MKII is the perfect choice for installing in the back of a rack for interfacing legacy analogue audio equipment to your Dante network.

In this example the AoIP22 MKII is used to interface to a two channel Telephone Hybrid but this could also easily be a two channel ISDN or Ip Codec. The AoIP22 MKII is installed in the back of the rack using cable ties through the fastening points.

It is then connected to the network switch in the station's racks room, this switch provides power to the AoIP22 MKII via PoE. If the switch doesn't supply PoE then a mid span PoE injector can be used.

The two audio inputs to the AoIP22 MKII are connected to the two output audio circuits of the telephone hybrid, and they are converted to network audio streams sent across the network to stations desk where they appear as two input channels.

Two return clean feed output circuits are routed across the network from the desk to the AoIP22 MKIIs outputs. These are then connected on XLR cables to the two audio inputs on the telephone hybrids.

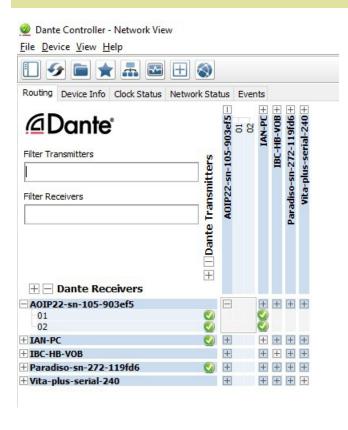
Using the AoIP22 MKII in this way will allow you to continue using your existing analogue based equipment on a modern networked audio infrastructure.







EASY TO USE



Dante® Controller by Audinate is a free application that controls all your Dante® enabled devices on your network.

It will automatically find the AoIP22 MKII and allow you just by the click of a mouse to route audio circuits to/ from it.

Dante® network audio has become very popular because it just works and it's so easy to use. Yet underneath the simple trouble free setup is a very sophisticated Audio over IP (AoIP) protocol working in real time across standard network switches with clever digital clock synchronisation to keep all equipment working in perfect harmony.

PRACTICAL FFATURES

CABLE TIE HOLES

All four front corners of the AoIP22 feature large extremely strong holes that are ideal for threading cable ties through. This makes temporarily installing the unit in out of the way locations very easy.











SPECIFICATION

AUDIO

Frequency Response

20Hz to 22k ≤ \pm 0.25dB

Maximum Input Before Clip

+18dBu

Maximum Output Level

+18dBu

Input Impedance

>20 k Ω

Output Impedance

50 Ω

Distortion (Analogue in to Dante Out)

0.0013% THD+N @ 1kHz Reference to +18dBu output

Distortion (Dante in to Analogue Out)

0.0026% THD+N @ 1kHz Reference to +18dBu output

Noise (Analogue In & Out @ 0dBu)

-92dBu (Residual)

Dynamic Range

108dBs

Crosstalk (0dBu input to output 1k tone)

-135dBu

Output Type

Electronically balanced (can be wired unbalanced) on Neutrik 3 pin XLR plug

Input Type

Electronically balanced (can be wired Unbalanced) on Neutrik 3 pin XLR socket

Digital Full Scale

+18dBu = 0dBFs

PHYSICAL

Size

125 x 122 x 48 mm (WxDxH)

Weight

0.45Kg

Mechanics

All aluminium construction, anodized and laser etched.

Front and rear hard rubber bumpers.

POWER

PoE

2.6 Watts

Source

PoE Enabled Switch or Mid-Span PoE Injector **DC: USB**

NETWORK AUDIO

Compatible Audio Networks

Dante® uncompressed, low latency audio. AES67

Network Connection

Neutrik RJ45 EtherCON

Dante Network Sample Rate

44.1k, 48k, 88.2k, 96k

AES67 Network Sample Rate

48k

INCLUDED ITEMS

Handbook

Download available

Rj45 Network Cable

2 metre Cat5 Rj45 plug/ Rj45 plug cable





E & OE