



AUDIO PROCESSOR

Nevion Virtuoso

AUD-PROC-MADI-IP, AUD-AES3, AUD-RPRO

Nevion Virtuoso's AUD-PROC-MADI-IP offers an attractive set of audio adaptation, processing and mixing functions for use in live audio production applications.

The AUD-PROC-MADI-IP media function provides MADI and SMPTE 2110 and AES67 IP audio interfacing, monitoring, routing and processing of audio signals. The AUD-AES3 and RPRO cards and media functions provide additional audio interfacing capability.

Four audio processor engines are available for flexible routing/mono shuffling with per-channel control of polarity, gain and delay. Each of the processing engines can also be configured as an audio summing matrix mixer with up to 512 cross-points.

The AUD-PROC-MADI-IP media functions runs on the Virtuoso HBR card and supports two 1/10 GigE ports for IP audio, with up to 128 input and 128 output streams, fully compliant to AES67 and ST 2110-30/31. ST2022-7 is supported for all inputs and outputs.

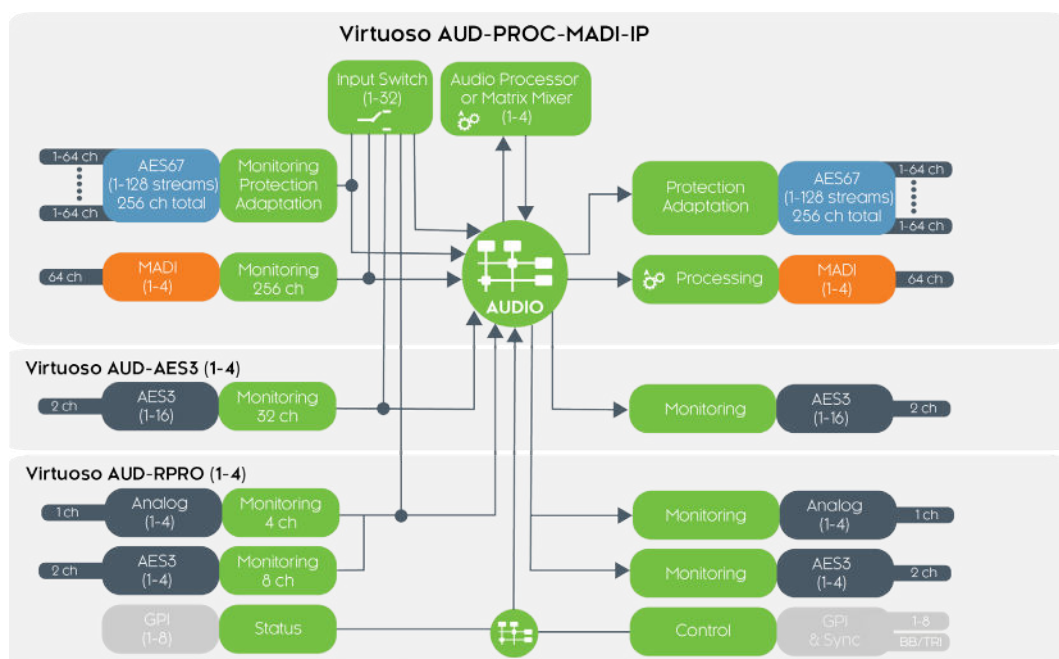
The Audio Processor media function is optimized for high-speed processing to ensure that the end-to-end latency is kept at a minimum, making it well suited for any live production application.

Applications

- IP in the facilities infrastructure
- Outside broadcast
- Remote and distributed production
- In-house/campus media networks
- Audio over IP contribution and distribution

Key features

- Multi-standard audio connectivity
 - MADI/AES10 (optical and electrical)
 - AES3 inputs and outputs (balanced and unbalanced)
 - Analog input and outputs
 - IP Audio compliant with AES67 & ST 2110
 - Any-to-any routing and conversion
- Audio processing
 - Audio routing/shuffling, delay and gain
 - Audio summing matrix mixer
- RTP/IP flow protection
 - SMPTE 2022-7 hitless/seamless protection
 - Alarm-based input switching
- Powerful control, monitoring and alarms
 - HTML5 web user interface and REST API
 - In-depth service monitoring with audio presence, peak level and silence detection
 - Configurable GPIO interfacing for alarm status



Powerful audio processing

Nevion Virtuoso's Audio Processor media function has flexible audio interfacing with support for MADI/AES10 and IP audio compliant to AES67 and SMPTE ST 2110-30/31. In addition, AES3 and analog interfacing is available via adapter cards.

The four internal 64-channel audio processors enable fully flexible audio routing (mono shuffling), per channel delay of up to 10 seconds, as well as audio level / gain control and polarity inversion. Similar processing is available on MADI outputs.

Audio matrix mixer

An audio processor can be configured as an audio summing matrix mixer with up to 512 crosspoints. With support for gain/delay pre-processing of inputs as well as gain/polarity/mute control for every crosspoint, the audio matrix mixer provides a powerful toolbox for a variety of audio summing and mix-minus audio mixer requirements in live production, remote production and MCR applications.

High density and flexibility

The Audio Processor Media Function running on an HBR accelerator handles up to 8 MADI streams, 4 inputs and 4 outputs with Nevision breakout cables or 2 inputs and 2 outputs with optical/electrical MADI SFPs.

For IP audio, up to 128 input and 128 output AES67 / ST 2110-30/31 streams are supported, with ST 2022-7.

For pure ST 2110 IP audio routing applications, each instance of the media function is equivalent to a 256x256 IP audio router.

Seamless IP Protection Switching

Transmitting the same RTP/IP stream across dual, fully diverse network links enables receivers/ decoders to utilize SMPTE ST 2022-7 Seamless IP Protection Switching (SIPS), which gives error-free transport even in case of severe packet loss or link outages as long as a packet arrives on either of the two network links. Support for ST 2022-7 receivers requires the protection license.

Timing and sync

Precision Timing Protocol (PTP) provides accurate synchronization for IP audio, using IEEE 1588v2 profiles or SMPTE 2059-2. Locking to PTP/TAI ensures fully synchronous operation across any network and any distance. For IP audio in an AES67/SMPTE 2110 environment, playout can be receiver buffer-based on link-offset based. For wide-area network environments where PTP is not available, adaptive clock recovery is supported to enable point-to-point transport of MADI audio over IP. Analogue video, BB and Tri level (RPRO) and audio word clock locked to SMPTE 2059.

Monitoring

The software supports in-depth audio service monitoring for up to 256 audio channels, with audio bars in the user interface as well as advanced template based monitoring and alarms. Template monitoring is configurable per audio channel with presence, peak level threshold and silence detection (requires the monitoring license).

Audio formats

Digital Audio - MADI	MADI / AES10 optical SFP (1 in + 1 out) MADI / AES10 electrical SFP (2 in, 2 out) MADI electrical breakout cable (4 in, 4 out) Up to 64 audio mono channels per MADI.
Audio format	48 kHz, 24 or 32-bit
Input SRC	Asynchronous Sample Rate Conversion (ASRC) is a licensed option per MADI input
ASRC ports	Available on all input ports
Audio over IP	AES67 up to 64 channels/stream SMPTE ST 2110-30 Level A, B, C SMPTE ST 2110-31 Level A, B, C
Max audio streams	128 input + 128 output (1-8 channels/stream) 4 input + 4 output (64 channels/stream)

Networking and Protection

Ethernet ports	2 x 10GBase-R (10 Gigabit Ethernet) or 2 x 1GBase-X (1 Gigabit Ethernet)
Protocols	RTP, UDP, IP, ICMP, ARP, IGMPv2/v3, Diffserv/ TOS, 802.1Q (VLAN tag), 802.1P (VLAN priority)
Max VLANs	Up to 128 VLANs per network interface
Link redundancy	SMPTE ST 2022-7 support for all IP audio input streams (optional licensed feature).
Max IP flows	2 IP input / 2 IP output per audio stream. Multicast IP destination address can only be used once per card for input/output.
Input switching	32x Audio input switches supporting IP audio inputs (AES67, ST2110-30/31) Manual or automatic switching based on alarm status and severity

Audio processing

Processors	Audio Processor elements (1-4) MADI outputs (1-4)
Audio matrix	64 output channels per audio processor
Fade out/in	Configurable per matrix (slow/medium/fast)
Audio delay	Up to 10 seconds per channel, configurable in samples or milliseconds (0,1 ms steps)
Audio level/gain	Individual gain control per channel -80 dB to + 36 dB (0,1 dB steps)
Audio mute	Individual control per channel
Polarity invert	Individual control per channel

Timing and synchronization

Sync input format	PTP (IEEE 1588v2:2008)
PTP profile support	PTP default and media profile SMPTE ST 2059-2 PTP profile
PTP redundancy	Internal PTP failover in Virtuoso FA/MI
Media timing	SMPTE ST 2059-1, SMPTE ST 2110-10 AES67 Link Offset mode IP Receiver buffer with PTP fixed rate mode IP Receiver buffer with adaptive clock recovery for MADI over IP in WANs.

Monitoring

Audio bars for MADI, Audio Processor and IP audio inputs
Advanced audio monitoring license option, providing:
- Audio silence detection and alarm
- Audio peak level threshold detection and alarm
- Audio level stuck detection and alarm

Media Server Appliance support

Please refer to Nevion Virtuoso Platform datasheet for details.	
Virtuoso MI	Supported in version 1.2 or higher
Virtuoso RE	Supported in version 1.0.6 or higher

Accelerator requirement

Accelerator	HBR Media Accelerator
Description	Multi-channel high bitrate Media Accelerator (HW module). 4x SFP+ ports that can accommodate a combination of 1GE/10GE SFP+ and MADI SFPs.
Product codes	VIRTUOSO-HW-HBR-SFP4 (24204)
SFP configuration	Port 1: MADI SFP Port 2: MADI SFP Port 3: 1GE / 10GE Port 4: 1GE / 10GE
Optical MADI SFPs	Single mode 1310nm (SFP-TR1-13T-LR) Multimode 850nm (SFP-TR1-850-SR) For more options please contact Nevion.
Electrical MADI SFPs	SFP-MADI-TRX-1-DIN (24686) SFP-MADI-TRX-1-HDBNC (24687) SFP-MADI-RX-2-HDBNC (24688) SFP-MADI-TX-2-HDBNC (24689)
Sync input format	PTP (IEEE 1588v2:2008, SMPTE ST2059)
Power consumption	Maximum 45W

Audio formats

Digital Audio - AES3	AES3-1992 110 ohms Up to 16 channels per AES3 card. Input/output direction is configurable on a port-by-port basis. Two audio mono channels per AES3.
Audio format	48 kHz, 24-bit linear PCM audio 32-bit transparent AES3 Synchronized to clock master (PTP locked)
Input SRC	Asynchronous Sample Rate Conversion (ASRC) is a licensed option per AES3 input
ASRC ports	Available on all input ports.
ASRC modes	On, Auto, Off
ASRC spec	Output frequency: 48kHz THD+N > 120dB Frequency range: 24kHz - 96 kHz Passband ripple < +/- 0.02dB Passband response edge: 21.7 Hz Stopband edge: 26.2 Hz Stopband response > 145 dB

Audio processing

Processors	Audio Processor elements (1-2)
Audio matrix	32 output channels per audio processor
Audio mute	Individual control per channel

Protection

Input switching	32x Audio input switches supporting IP audio inputs Manual or automatic switching based on alarm status and severity
-----------------	---

Timing and synchronization

Sync input format	PTP (IEEE 1588v2:2008) via HBR10 in Virtuoso MI
PTP profile support	PTP default and media profile SMPTE ST 2059-2 PTP profile
PTP redundancy	Internal PTP failover in Virtuoso MI
Media timing	SMPTE ST 2059-1

Monitoring

Audio bars for AES3 input, AES3 output and Audio Processors

Advanced audio monitoring license option, providing:

- Audio silence detection and alarm
- Audio peak level threshold detection and alarm
- Audio level stuck detection and alarm

GPIO

4 GPIO (direction configurable on a port by port basis)	
Inputs:	Internal pull-up to 3.3V (asserted below 1 volt)
Outputs:	Open collector transistor to common ground Max. current 190mA Max voltage 24VDC

Media Server Appliance support

Virtuoso MI	Supported in version 1.2 or higher
Virtuoso RE	Supported in version 1.0.6 or higher

Accelerator requirement

Description	Virtuoso AES3 adapter module with 16 AES3 inputs or outputs (direction configurable on a port by port basis). 4 GPIO (direction configurable on a port-by-port basis). DB-37 female connector (DB-37 cable not included, supplied with Breakout Panel). An HBR accelerator with AUD-PROC-MADI-IP software and licenses required for IP audio.
Product codes	VIRTUOSO-HW-AUD-AES3 (24772)
Number of ports	16 XLR - configurable as input or output 4 pin Phoenix GPIO ports (direction configurable) 1 DB-37 female (Audio, GPIO)
Breakout panels	VIRTUOSO-HW-AUD-BRK-BNC16 (24774) 1RU passive breakout panel for AES3 digital audio adapter with 16 unbalanced female BNC connectors (75 Ohm). DB-37 female connector (DB-37 cable included). VIRTUOSO-HW-AUD-BRK-XLR16 (24775) 1RU passive breakout panel for AES3 digital audio adapter with 8 male and 8 female XLR connectors. DB-37 female connector (DB-37 cable included).
Power consumption	Maximum 20W

Audio formats

Digital Audio - AES3	AES3-1992 110 ohm Up to 4 channels per RPRO card. Input/output direction is configurable on a port-by-port basis. Two audio mono channels per AES3.
Audio format	48 kHz, 24-bit linear PCM audio 32-bit transparent AES3 Synchronized to clock master (PTP locked)
Input SRC	Asynchronous Sample Rate Conversion (ASRC) is a licensed option per AES3 input
ASRC ports	Available on all input ports.
ASRC modes	On, Auto, Off
ASRC spec	Output frequency: 48kHz THD+N > 120dB Frequency range: 24kHz - 96 kHz Passband ripple < +/- 0.02dB Passband response edge: 21.7 Hz Stopband edge: 26.2 Hz Stopband response > 145 dB
Analog Audio	Up to 4 bidirectional balanced audio channels per RPRO card (4 x 4-wire). High impedance inputs, low impedance with high common mode impedance to earth Ref level: 0 dBFS @+12 to +24 dBu Freq. response: +/-0.1 20Hz to 20kHz Crosstalk: >-100dB 20Hz to 20kHz Out of band filtering: >-100dB > 246kHz (inputs) >-76dB > 26kHz (outputs) Dynamic range: >100dB dB(A) THD+N:>-86dB/0.0005%, 20Hz - 20kHz at -9 dBFS level SMPTE IMD: >-86dB/0.0005% at -12dBu

Audio processing

Processors	Audio Processor elements (1-2)
Audio matrix	32 output channels per audio processor
Audio mute	Individual control per channel

Timing and synchronization

Sync input format	PTP (IEEE 1588v2:2008) via HBR10 in Virtuoso MI
PTP profile support	PTP default and media profile SMPTE ST 2059-2 PTP profile
PTP redundancy	Internal PTP failover in Virtuoso MI
Media timing	SMPTE ST 2059-1

Monitoring

Audio bars for Mono, AES3 input, AES3 output and Audio Processors

Advanced audio monitoring license option, providing:

- Audio silence detection and alarm
- Audio peak level threshold detection and alarm
- Audio level stuck detection and alarm

GPIO

8 GPIO (direction configurable on a port by port basis)	
Inputs:	Internal pull-up to 3.3V (asserted below 1 volt)
Outputs:	Open collector transistor to common ground Max. current 190mA Max voltage 24VDC

Sync outputs

2 Analog independant sync outputs (75 ohm). Analogue video black (without colour burst) or Tri-level
Synchronized to clock master (PTP locked)
Word Clock sync output

Media Server Appliance support

Virtuoso MI	Supported in version 1.2 or higher
Virtuoso RE	Supported in version 1.0.6 or higher

Accelerator requirement

Description	Virtuoso RPRO adapter card with 4 balanced analog inputs and 4 balanced analog outputs, 4 balanced AES3 inputs/outputs (direction configurable on a port by port basis), 8 GPIO (direction configurable on a port-by-port basis) and 2 video sync outputs. DB-37 female connector (DB-37 cable not included, supplied with Breakout Panel). An HBR accelerator with AUD-PROC-MADI-IP software and licenses required for IP audio.
Product codes	VIRTUOSO-HW-AUD-RPRO (24773)
Number of ports	4 XLR balanced analog inputs 4 XLR balanced analog outputs 4 XLR AES3 ports (direction configurable) 8 pin Phoenix GPIO ports (direction configurable) 2 BNC analogue video sync output ports 1 DB-37 female (Audio, GPIO, Sync)
Breakout panels	VIRTUOSO-HW-AUD-BRK-RPRO (24776) 1RU passive breakout panel for RPRO audio interface module (IRU) with balanced XLR connectors for analog/digital audio, 8x GPIO ports and 2 BNCs for analog video sync output. DB-37 female connector (DB-37 cable included).
Power consumption	Maximum 20W



Nevion near you!

Nevion has a presence in all the major regions, and an extensive network of partners to reach customers anywhere in the world.

Visit our website for your nearest sales contact

[nevion.com](https://www.nevion.com)

Copyright © Nevion, 2024, all rights reserved.

No part of this documentation may be reproduced in any form or by any means or be used to make any derivative work (including translation, transformation or adaptation) without explicit written consent of Nevion.

Nevion reserves the right to make changes without notice to equipment specification or design. The information provided in this document is for guidance purposes only and shall not form part of any contract.