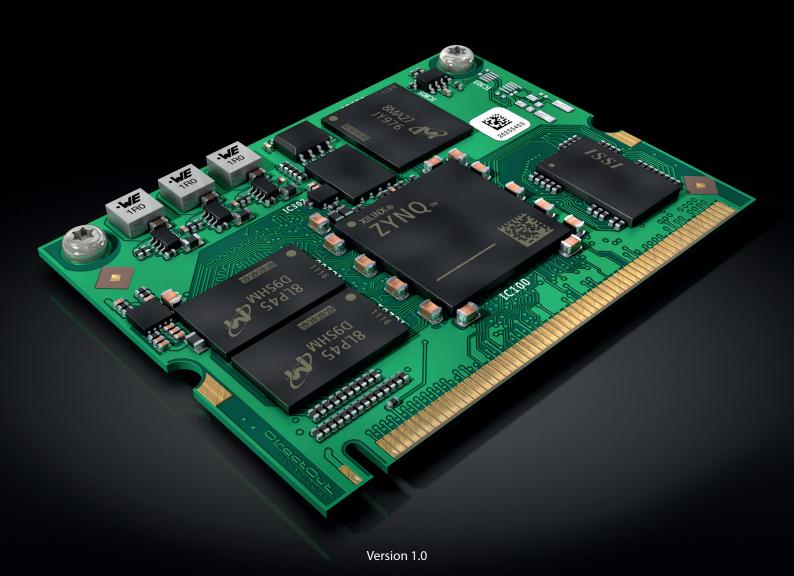


AVT RAVENNA Module

Software Manual



A publication of AVT Audio Video Technologies GmbH

Nordostpark 91 90411 Nürnberg GERMANY

Phone +49-911-5271-0 Telefax +49-911-5271-100

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Version: 1.1

Release date: April 2023

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AVT System Software

Introduction

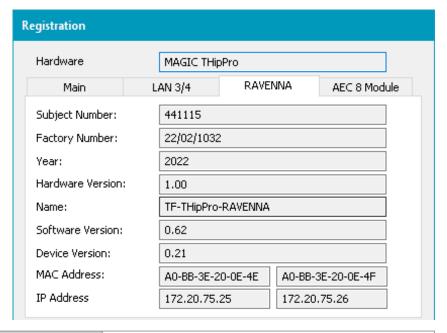
If equipped, the MAGIC RAVENNA Module shows in the System configuration of the MAGIC Telephone Hybrids and MAGIC DAB Encoders. The next 10 pages specify the system specific configuration within the AVT MAGIC System user interface. The configuration of the Ravenna module is performed via a HTML5 webbrowser. The configuration PC must have access to the network the Ravenna interface is connected to. This guide adresses the configuration of the Ravenna Interface itself from page 14.

Telephone Hybrids

Registration

Find information about the installed Ravenna module in the THipPro PC Software under *Menu » Administration » Registration*.

Select the RAVENNA tab at the top.



SUBJECT NUMBER	The AVT ID of the Ravenna module
FACTORY NUMBER	The AVT serial number of the Ravenna module
YEAR	Manufacturing Year
HARDWARE VERSION	Version of the complete Ravenna module
SOFTWARE VERSION	Version of the DirectOut Ravenna software
DEVICE VERSION	Version of the DirectOut Ravenna module
MAC ADDRESS	Hardware addresses of LAN 1 and LAN 2 of the Ravenna module
IP ADDRESS	IP addresses of the LAN interface 1 and 2 of the Ravenna module

Mode & Audio Line

Audio signals of the THipPro are assigned to Ravenna channels in the RAVENNA tab of the MODE & AUDIO LINE configuration page of the THipPro.

The RAVENNA tab is only displayed if the THipPro is equipped with a Ravenna module and the built-in AES67 feature is disabled.

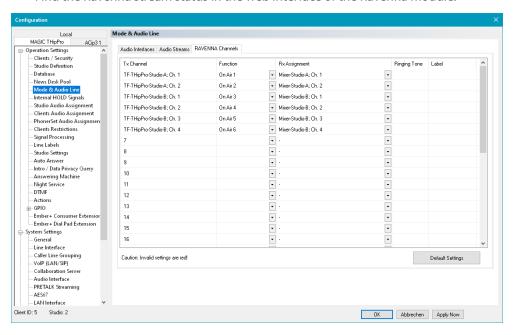
The THipPro can utilize 32 channels of the Ravenna module for both directions (input and output).

It is recommended to configure the input and output streams in the Ravenna web interface first. The channels and stream names are then available in the THipPro configuration.

Remark: It can take up to a minute until the streams are available to the configuration dialog. Try to switch to another configuration page and back to Mode & Audio Line to update the table.

NOTE

- The IP address of the Ravenna module can be found under *Menu* » *Administration* » *Registration*. Then, select *RAVENNA* at the top.
- An overview of Ravenna streams and channels used by the THipPro can be found in the System Monitor under Menu » System Monitor » Ravenna State » Show Streams.
- Find the Ravenna stream status in the web interface of the Ravenna module.



TX CHANNEL	Displays the Ravenna stream names and channel numbers if the output
FUNCTION	Select an audio line
RX ASSIGNMENT	Select a channel from a stream which is currently received by the Ravenna module to assign it to the audio line
	Note: Configure the Input streams in the Ravenna web interface first. Only then are they available in the configuration



Ravenna Input Streams

In the Ravenna web interface, up to 32 input streams can be configured.

The Ravenna module can send up to 32 mono audio channels to the ThipPro base board. For each stream a START CHANNEL needs to be set in the Ravenna web interface. This channel number refers to the internal channel number of the ThipPro.

The Ravenna module can only receive a whole stream. There is no way of selecting a single channel or a part of a stream. This means if a stream with 8 channels is received and START CHANNEL is set to 1, Ravenna audio channels 1-8 of the THipPro are occupied. For the next stream the START CHANNEL must be set to 9 or higher to avoid overlapping audio signals.

Ravenna Output Streams

In the Ravenna web interface, up to 32 output streams can be configured. For each stream a START CHANNEL and NUMBER OF CHANNELS must be set in the Ravenna web interface. These channel numbers refer to the Ravenna Tx channels of the THipPro. Example: THipPro Ravenna Tx channels 1-6 should be transmitted via one Ravenna stream, the START CHANNEL should be set to 1 and NUMBER OF CHANNELS should be set to 6.

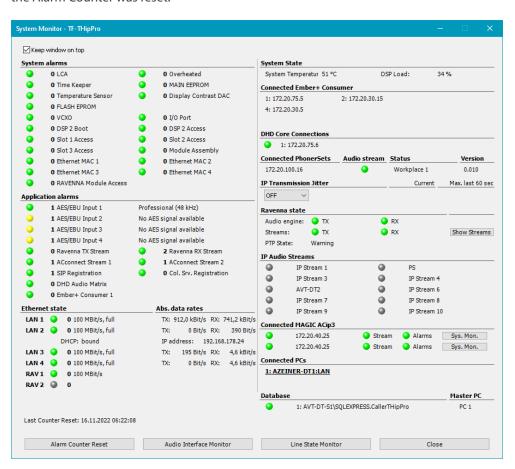
System Monitor

Open *Menu* » *System Monitor* to get an overview of the system status.

The status is displayed as text as well as LEDs:

- : Alarm is active. There is an error.
- : Alarm is active but not relevant for the current configuration.
- : Status is OK.

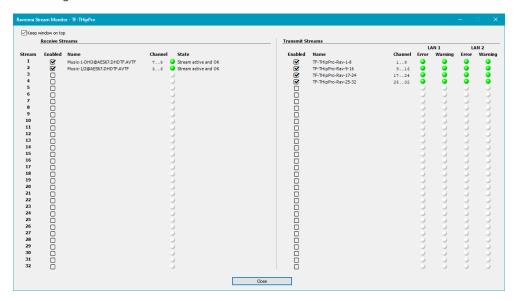
Find alarm counters next to the LEDs which indicate how often the alarm occurred since the Alarm Counter was reset.



KEEP WINDOW ON TOP	Activate to keep the windows always on top of the screen
RAVENNA STATE	Displays the state of the optional Ravenna module
AUDIO ENGINE	Internal state of the Ravenna module for RX and TX. Refer to the Ravenna web interface for more information
STREAMS	State of the configured Rx and Tx streams. Refer to the Ravenna web interface or press SHOW STREAMS for more information
PTP STATE	State of the audio clocks of the Ravenna module. Refer to the Ravenna web interface for more information
SHOW STREAMS	Shows an overview of the configured streams and their status

Ravenna Streams

Open *Menu* » *System Monitor* » *Ravenna State* » *Show Streams* to get an overview of the configured Ravenna streams.

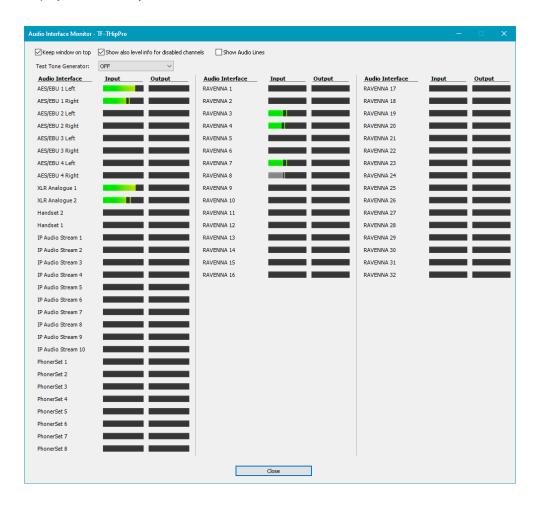


KEEP WINDOW ON TOP	Activate to keep the windows always on top of the screen
STREAM	The stream number as displayed in the Ravenna web interface
RECEIVE STREAMS	Shows the configured Ravenna input streams
	ENABLED:
	Shows whether the input stream is activated in the Ravenna web
	interface.
	NAME:
	The name of the Ravenna stream.
	CHANNEL:
	Shows which audio channels the Ravenna module uses to transmit the
	audio signal to the THipPro base board. STATE:
	Shows whether the stream is currently being received.
TD ANGLAIT CTD ANAC	· · ·
TRANSMIT STREAMS	Shows the configured Ravenna output streams
	ENABLED:
	Shows whether the output stream is activated in the Ravenna web
	interface.
	NAME:
	The name of the Ravenna stream.
	CHANNEL:
	Shows which Ravenna Tx channels of the THipPro are included in that
	stream.
	LAN 1 /2, ERROR, WARNING:
	Shows if there are any problems on any of the LAN interfaces of the
	Ravenna module. Get more information in the Ravenna web interface.

Audio Interface Monitor

Open *Menu* » *System Monitor* » *Audio Interface Monitor* to get an overview of the audio interfaces.

For each audio channel of the device, **AUDIO INTERFACE**, INPUT and OUTPUT are displayed individually.

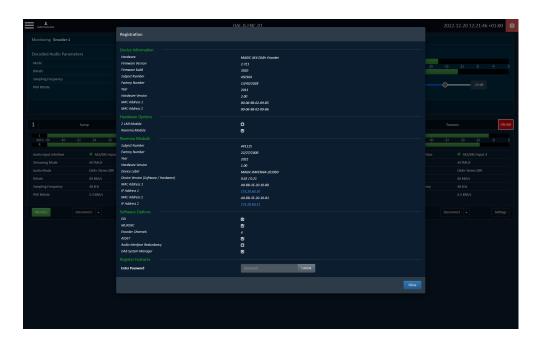


KEEP WINDOW ON TOP	Activate to keep the windows always on top of the screen
TEST TONE GENERATOR	Select an audio interface from the drop-down box
	to generate a sine test signal on the respective audio
	output
ALIDIO INTEREA CE	
AUDIO INTERFACE	Name of the audio channel
	AES/EBU:
	Digital audio interfaces at the back of the device.
	XLR ANALOGUE:
	Analogue audio interfaces at the back of the device.
	HANDSET:
	Handset audio interfaces at the front of the device.
	IP AUDIO STREAM:
	Pretalk streams over IP.
	PHONERSET:
	PhonerSet audio streams over IP.
	AES67:
	AES67 audio channels of the AES67 software module.
	DANTE:
	Audio channels of the Dante hardware module.
	RAVENNA:
	Audio channels of the Ravenna hardware module.
INPUT	Audio level of the signal fed to the THipPro
OUTPUT	Audio level of the signal Audio level of the signal emitted
	by the THipPro

DAB

Registration

Find information about the installed RAVENNA module in the MAGIC AE4 Web Interface under *Menu* » *Administration* » *Registration* » *Module Slot 2*.



SUBJECT NUMBER	The AVT ID of the RAVENNA module
FACTORY NUMBER	The AVT serial number of the RAVENNA module
YEAR	Manufacturing Year
HARDWARE VERSION	Version of the complete RAVENNA module
SOFTWARE VERSION	Version of the DirectOut RAVENNA software
DEVICE VERSION	Version of the DirectOut RAVENNA module
MAC ADDRESS	Hardware addresses of LAN 1 and LAN 2 of the RAVENNA module
IP ADDRESS	IP addresses of the LAN interface 1 and 2 of the RAVENNA module

Audio

To enable RAVENNA functionality the corresponding option has to be set in *Menu* » *Settings* » *Audio*.

The RAVENNA option is only visible if the module is equipped and enabled by software license.

Only one of RAVENNA and AES67 functionality can be enabled simultaneously. Make sure to disable the other option first.

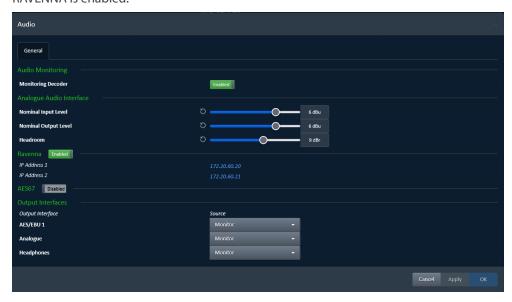
The MAGIC AE4 can utilize the first 4 stereo channels of the RAVENNA module for audio input.

It is recommended to configure the input and output streams in the Ravenna web interface first. The channels and stream names are then available in the MAGIC AE4 configuration.



NOTE

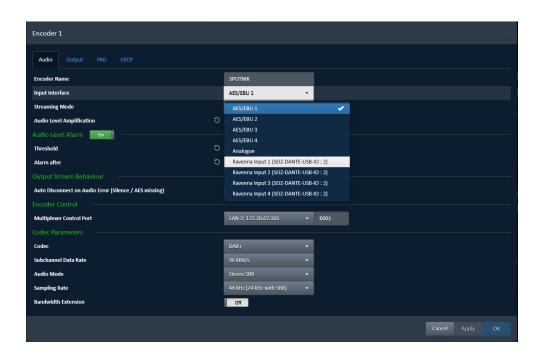
The IP addresses of the Ravenna module can be found under *Menu* » *Administration* » *Registration* and in the Audio settings page under *Menu* » *Settings* » *Audio* once RAVENNA is enabled.



Encoder

To allocate RAVENNA channels the corresponding option has to be set in *Menu* » *Settings* » *Encoder*.

The Input Interface dropdown displays all available RAVENNA inputs.

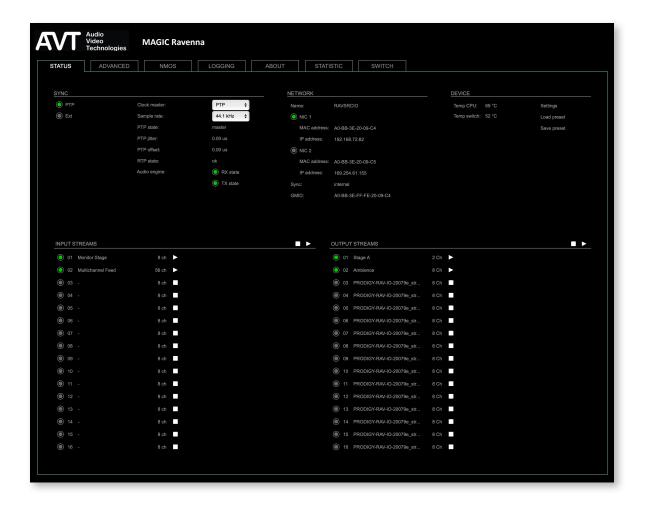


RAVENNA Module Software

Introduction

The AVT RAVENNA Module is an audio network module for RAVENNA / AES67 based on the RAV2 mudule.

All functions of the device are accessible through a browser based interface (hmtl5 / javascript). The size of the window and the zoom level can be varied. The page is organized in tabs, pulldown menus or hyperlinks offer access to the values of a parameter. Some values use an input field (e.g. IP address).



Connecting Audio Network

To access the control page:

- · connect the network with one port
- enter http://<IP Address> (default IP @ PORT 1: 192.168.96.111) in the navigation bar of your browser

Two independent network interfaces (NICs) can be configured in the switch ,configuration. Port 1 is fixed assigned to NIC 1.

Status - Overview

The tab 'STATUS' is divided into several sections:

SYNC - monitoring sync state, clock selection, links to I/O

settings

NETWORK - display network info, link to network settings

DEVICE - monitoring device info, link to device settings,

phones level control

• INPUT STREAMS - monitoring and control input streams,

link to input stream settings

OUTPUT STREAMS - monitoring and control output streams,

link to output stream settings

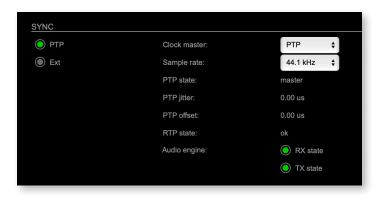
Hyperlinks open a popup window to adjust related settings. Most settings are updated immediately without further notification. To exit a popup window click the button in the top right corner.

Mouse overs are used to display further information (e.g. connection speed of network link).

NOTE

The web user interface updates itself when changes are applied by other instances (other browsers, external control commands).

Status - Sync



PTP, Ext	Displays clock source and state for the main frame: (OFF) = not locked (ON) = locked and in sync with clock master (blinking) = locked but not in sync with clock master
Clock master	Pulldown menu to select clock source of the main frame (PTP, extern)
Sample rate	Pulldown menu to adjust sample rate of the main frame (44.1 / 48 / 88.2 / 96 / 176.4 / 192 kHz).
PTP state	State of PTP (Master / Slave).
PTP jitter	PTP-clock jitter per second
PTP offset	Offet relative to PTP-clock master
RTP state	Status of packet processing (OK, Error*)
Audio engine RX state	State of module's audio engine - receiving (ON) = ok, receiving data (blinking) = not all received packets can be processed
Audio engine TX state	State of module's audio engine - sending (ON) = ok, sending data (blinking) = not all packets can be sent to the network

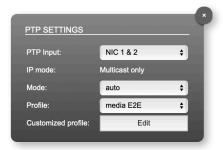
* Error: packet time stamps are out of bounds.

Possible reasons: stream offset may be too small or transmitter or receiver are not synced properly to the Grandmaster.

Hyperlinks:

• PTP / PTP state (p 17)

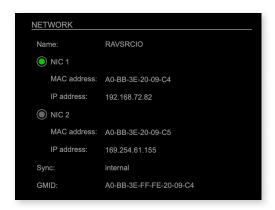
PTP Settings



PTP Input	NIC selection for PTP clock input. 'NIC 1 & 2' means input redundancy.
IP Mode	PTP via multicast, unicast or in hybrid mode. *
Mode	PTP-clock master / slave configuration is auto negotiated between devices in the network. Module's master / slave state may change automatically.
Profile	PTP profile selection (default E2E, default P2P, media E2E, media P2P, customized)
Customized profile	Edit opens the tab 'ADVANCED' to adjust the custom profile.

See "Advanced - PTP Clock Setting" on page 43 for more details.

Status - Network



Name	Module's name in the network. Used e.g. for mDNS service. The name needs to be unique throughout the network.
NIC 1 / NIC 2	Monitoring state of network interface controller ○ (OFF) = not connected ○ (ON) = connected with the network
MAC address	Hardware identification of network interface controller.
IP address	IP address of device
Sync	Selected NIC for PTP sync
GMID	Grand Master ID (PTP)

Hyperlinks

• Name / IP address (p 19)

Mouse over:

- LED NIC 1 indicating link state and connection speed
- LED NIC 2 indicating link state and connection speed



NOTE

If NIC 1 and NIC 2 are connected to the same switch, they must be configured to different subnets - see "Network Settings" on page 19.

Network Settings

The two network interface controllers (NIC 1 / NIC 2) are configured individually.



Device name	Input field - Module's name in the network. Used e.g. for mDNS service. The name needs to be unique throughout the network.
Dynamic IP address (IPv4)	Switch to enable the device's DHCP client. IP address is assigned by DHCP server. If no DHCP is available the IP address is determined via Zeroconf.
Static IP address (IPv4)	Switch to disable the device's DHCP client. Manual configuration of network parameters.
IP address (IPv4)	Module´s IP Address
Subnet mask (IPv4)	Module's subnet mask
Gateway (IPv4)	IP address of gateway
DNS server (IPv4)	IP address of DNS server
Apply	Button to confirm changes. Another popup window will appear to confirm a reboot of the module.
Direct routing	IP addresses of devices outside the subnet, to enable multicast traffic; e.g. Grandmaster or IGMP querier. Mark checkbox to activate.

Status - Device

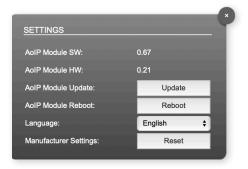


Temp CPU	Display temperature of CPU core in degree Celsius. It may reach 110 °C without effecting the performance of the device.
Temp switch	Display temperature of network switch in degree Celsius
Settings	Opens a popup window to configure the device.
Load preset	Opens a dialog to store the device settings to a file. Filetype: .rps
Save preset	Opens a dialog to restore the device settings from a file. Filetype: .rps

Hyperlinks:

- Settings (p 20)
- Load preset (p 21)
- Save preset

Settings



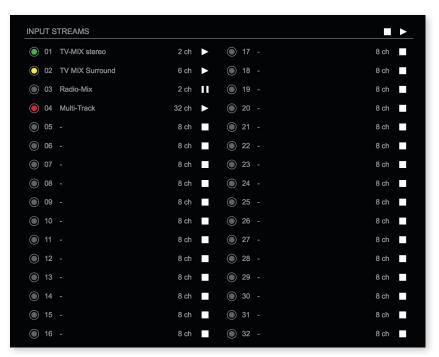
AoIP Module SW	Module's software version. It is updated together with hardware version via network.
AoIP Module HW	Module's bitstream version. It is updated together with software version via network.
AoIP Module Update	Opens a dialog for selection of the update file - see "RAV2 - Firmware Update" on page 53.
AoIP Module Reboot	Restart of the AoIP module. Confirmation required. Audio transmission will be interrupted.
Language	Menu language (english, german).
Manufacturer Settings Reset	Restore device settings to factory defaults. Confirmation required.

Load Preset



The device configuration can be stored to a single file (.rps). Restoring the configuration a dialog prompts for selection of individual settings.

This enhances flexibility at setup changes when a particular adjustment shall be preserved or just a single adjustment shall be restored.



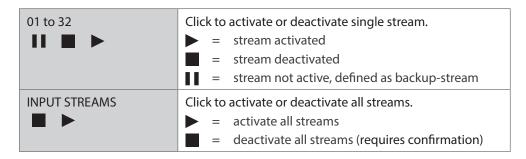
Status - Input Streams

The module can subscribe up to 32 streams. The overview displays the basic information of each stream. The input stream name can be set manually (discovery protocol: manually, see page p 30) overriding the SDP's stream name information.

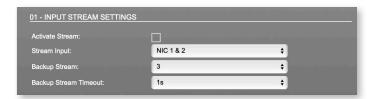
A backup stream can be defined as source after an adjustable timeout. A central active / inactive switch allows to toggle the stream state of all input streams at once.

01 to 32	State of inco	min	g streams
	O(OFF)	=	stream not activated
	(ON)	=	stream activated, receiving data
	O(ON)	=	stream activated, receiving data via
			one NIC only (input redundancy)
	(blinking)	=	stream activated, not receiving data
			(unicast, connection not established)
01 to 32	Name of stre	eam	gathered from SDP or set manually
Name	in the stream	n set	ttings dialog.
01 to 32	Number of a	nudio	channels transported by the stream
xx ch			





Backup Streams



Example:

Backup stream (input 3) that will act as source in the audio matrix if the current session (input 1) fails. Switch-over occurs after the defined timeout (1s). Stream 3 is marked accordingly in the status view



Input 1 failed and Input 3 becomes active after the timeout.



NOTE

In case the main input fails the main stream is stopped (IGMP LEAVE) before the backup stream is being activated. This behaviour ensures that the required network bandwidth does not increase in case of a failure.



Hyperlinks:

• Name (p 26)

Mouse over:

• LED - indicating stream state

NOTE



Source-Specific Multicast (SSM) support for IGMP v3, v2 and v1 (SSM via protocol only in IGMP v3, SSM via internal filtering is applied for IGMP v2 and v1) - see "Source Specific Multicast" on page 30.

01 - INPUT STREAM SETTINGS Activate Stream: Stream Input: NIC 2 **‡** \$ Backup Stream: disabled Backup Stream Timeout: \$ 1s Stream name: Stage A Stream state: not connected 50 Stream state offset min (samples): Stream state ip address src NIC 1: Stream state ip address src NIC 2: Offset fine: 128 (2.67 ms) \$ Offset in samples: Start channel: Discovery protocol: RTSP (RAVENNA Session) Session NIC 1: Session NIC 2: Stage A@NIC 2 \$

Input Stream Settings

Up to 32 input streams can be subscribed. Each stream is organized in a 'RAVENNA session' (SDP = Session Description Protocol) that describes the stream parameters (audio channels, audio format, etc.).

The stream settings allow to adjust the processing of the received audio data (offset, signal routing). The receiving of stream data starts once the stream has been enabled.

The settings displayed vary depending on the selected discovery protocol.



TIP

A sample offset of at least doubled packet time (samples per frame) is recommended

Example: Samples per frame = 16 (0.333 ms) Offset $\geq 32 (0.667 \text{ ms})$

It may be helpful to alter the stream discovery protocol if an expected stream can't be discovered by the device.

Selects one or both NICs used for stream input. Both NICs means input redundancy. Backup Stream Selects a backup stream that will act as source in the audio matrix if the current session fails. Switch-over occurs after the defined timeout. Defines timeout [1 s to 120 s] before switch-over to backup stream. Stream name Name of stream gathered from SDP Information about stream state: connected not connected receiving data read successfull error Stream state Measured value (maximum). A high value indicates that the media offset of the source might not match the adjusted media offset of the device. Stream state Measured value (minimum). The offset should not become negative. Stream state ip address src NIC 1 Multicast address of input stream subscribed at NIC 1. Unicast transmission: IP address of sender. Multicast address of input stream subscribed at NIC 2. Unicast transmission: IP address of sender. Modules output delay of received audio data (input buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol SAP = Session Announcement Protocol Session NIC 1 Selection of discovered streams at NIC 1	Activate stream	Stores parameters and activates or deactivates the receiving of audio data. (Unicast: additionally the negotiation of the connection)
audio matrix if the current session fails. Switch-over occurs after the defined timeout. Backup Stream Timeout Defines timeout [1 s to 120 s] before switch-over to backup stream. Stream name Name of stream gathered from SDP Stream state Information about stream state:	Stream input	·
Stream name Name of stream gathered from SDP Stream state Information about stream state:	Backup Stream	audio matrix if the current session fails. Switch-over
Stream state Information about stream state: connected not connected receiving data read successfull error Stream state message Status info related to stream state. Stream state offset max Measured value (maximum). A high value indicates that the media offset of the source might not match the adjusted media offset of the device. Stream state Measured value (minimum). The offset should not become negative. Stream state ip address src NIC 1 Multicast address of input stream subscribed at NIC 1. Unicast transmission: IP address of sender. Stream state ip address src NIC 2 Multicast address of input stream subscribed at NIC 2. Unicast transmission: IP address of sender. Offset fine Enables adjustment of offset in increments of one sample. Offset in samples Modules output delay of received audio data (input buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Session NIC 1 Selection of discovered streams at NIC 1	Backup Stream Timeout	
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Stream state ip address src NIC 1 Multicast address of input stream subscribed at NIC 1. Unicast transmission: IP address of sender. Multicast address of input stream subscribed at NIC 2. Unicast transmission: IP address of sender. Multicast address of input stream subscribed at NIC 2. Unicast transmission: IP address of sender. Enables adjustment of offset in increments of one sample. Offset fine Modules output delay of received audio data (input buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Selection of discovered streams at NIC 1		that the media offset of the source might not match
Stream state ip address src NIC 2 Multicast address of input stream subscribed at NIC 2. Unicast transmission: IP address of sender. Enables adjustment of offset in increments of one sample. Offset in samples Modules output delay of received audio data (input buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Selection of discovered streams at NIC 1		
Start channel Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol Session NIC 1 Enables adjustment of offset in increments of one sample. Discovered audio data (input buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Session Announcement Protocol Selection of discovered streams at NIC 1	·	
Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Selection of discovered streams at NIC 1	· ·	
buffer). Start channel Assignment of first stream channel in the audio matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Selection of discovered streams at NIC 1	Offset fine	
matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing matrix. Discovery protocol Connection protocol or manual setup. RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Selection of discovered streams at NIC 1	Offset in samples	i i
RTSP = Real Time Streaming Protocol SAP = Session Announcement Protocol Session NIC 1 Selection of discovered streams at NIC 1	Start channel	matrix. E.g. stream with two channels, starting at channel 3 is available at channel 3 & 4 of the routing
	Discovery protocol	RTSP = Real Time Streaming Protocol
Session NIC 2 Selection of discovered streams at NIC 2	Session NIC 1	Selection of discovered streams at NIC 1
	Session NIC 2	Selection of discovered streams at NIC 2

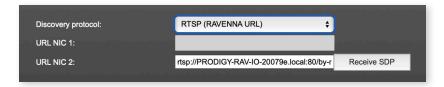
Stream Discovery in AoIP environments is a colorful mixture of different mechanisms. To serve a successful stream management RAV2 provides a bunch of options, not making operation easier but effective.



Discovery RTSP (Session)



Discovery RTSP (URL)



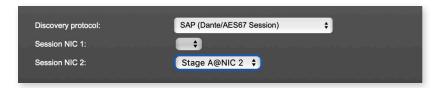
URL	URL (Uniform Ressource Locator) of the session of
	the device that is serving streams.
	Examples:
	rtsp://192.168.74.44/by-id/1
	or
	rtsp://PRODIGY-RAV-IO.local:80/by-name/Stage_A
Receive SDP	Recalls the stream configuration of the defined session(s).



NOTE

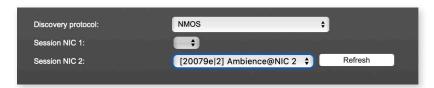
In case the automatic stream announcement and discovery of RAVENNA streams fails or cannot be used in a given network, the stream's SDP file can also be obtained via an RTSP URL.

Discovery SAP



SAP is used in Dante environments.

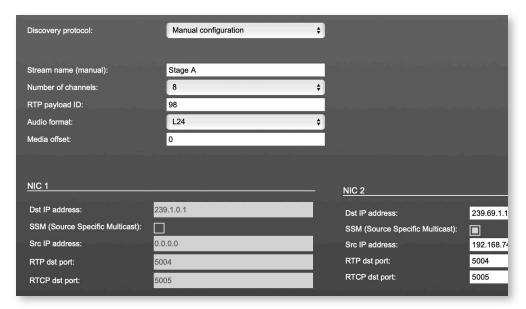
Discovery NMOS



Session	[MAC Address of sender] stream name @NIC
Refresh	Initiates a scan for available streams.

NMOS is suited for use in SMPTE ST 2110 environments.

Manual Setup



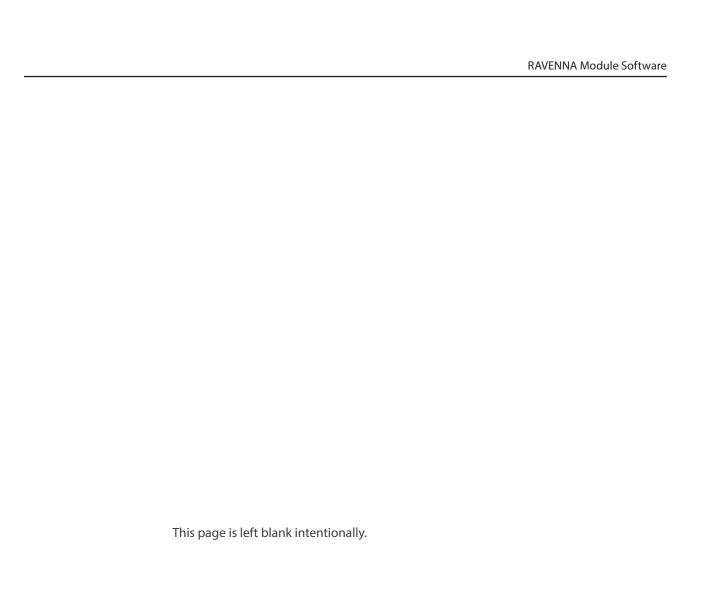
Stream name (manual)	Stream name for display in status view and matrix. Can be specified individually, different than the name gathered from the SDP.
Number of channels	Number of audio channels in the stream
RTP-Payload-ID	RTP-Payload-ID of the audio stream (Real-Time Transport Protocol). Describes the format of the transported content.
Audio Format	Stream's audio format (L16 / L24 / L32 / AM824)
Media Offset	Offset between stream's timestamp and PTP-clock
Dst IP address	Multicast IP address of audio stream
SSM	Activate Source Specific Multicast filter for this stream.*
Src IP address	IP address of sending device.*
RTP dst port	Stream's destination port for RTP
RTCP dst port	Stream's destination port for RTCP (Real-Time Control Protocol)

^{*} An RTP packet contains the IP address of the sender (source IP) and the stream's multicast address (destination IP). With SSM activated the receiver only accepts RTP packets of a certain destination IP that are originated by a sender with the specified source IP.

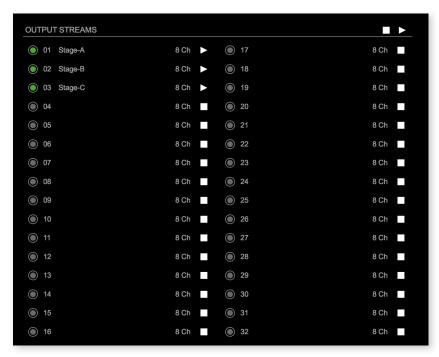


NOTE

RTP Payload ID must match between sender and receiver.







The device can send up to 32 streams. The overview displays the basic information of each stream.

01 to 32	State of outgoing streams (OFF) = stream not activated (ON) = stream activated, sending data (ON) = stream activated, stream output via both NICs selected, but one NIC is not linked to the network.
01 to 32 Name	Name of stream defined in the settings
01 to 32 xx ch	Number of audio channels transported by the stream
01 to 32	Activate or deactivate stream. = stream activated = stream deactivated
OUTPUT STREAMS	Click to activate or deactivate all streams. = activate all streams = deactivate all streams (requires confirmation)

Hyperlinks:

• Name (p 34)

Mouse over:

· LED - indicating stream state

TIP



AES67 Streams

To create output streams for interoperability in AES67 environments please consult the information document of our partner DirectOut: <u>Info - AES67 Streams</u>.

TIP



SMPTE 2110-30 / -31 Streams

To create output streams for interoperability in SMPTE ST 2110 environments please consult the information document of our partner DirectOut: Info-ST2110-30 Streams.

Both documents are available at our partner DirectOut: http://academy.directout.eu.

02 - OUTPUT STREAM SETTINGS Activate Stream: Stream Output: NIC 2 **‡** Stream name (ASCII): Ambience RTSP URL (HTTP tunnel) (by-name): RTSP URL (HTTP tunnel) (by-id): rtsp://PRODIGY-RAV-IO-20079e.local:80/by-name/Ambiertsp://PRODIGY-RAV-IO-20079e.local:80/by-id/2 rtsp://PRODIGY-RAV-IO-20079e.local/by-name/Ambience rtsp://PRODIGY-RAV-IO-20079e.local/by-id/2 RTSP URL (by-name): RTSP URL (by-id): SDP: v=0 o=- 2 2482002937 IN IP4 192.168.74.44 s=Ambience t=0 0 m=audio 5004 RTP/AVP 98 Unicast: RTP payload ID: 98 45 (0.94 ms) Samples per Frame (packet time): **‡** AM824 **‡** Audio format: 9 **‡** Start channel: Number of channels: \$ NIC 1 NIC 2 RTP dst port: 5004 5004 RTP dst port: 5005 5005 RTCP dst port: RTCP dst port: Dst IP address (IPv4): Dst IP address (IPv4): 239.69.1.3 239.69.1.4

Output Stream Settings

Up to 32 output streams can be sent to the network. Each stream is organized in a session (SDP = Session Description Protocol) that describes the stream parameters (audio channels, audio format, etc.).

Each stream may be labelled with an individual stream name (ASCII) which is useful for enhanced comfort at organizing the setup.

The stream settings allow to adjust the processing of the sent audio data (blocks per frame, format, signal routing, ...). The sending of stream data starts once the stream has been enabled.

Once the stream is active, the SDP data is displayed and may be copied from the window or downloaded via http://<rav-io>/sdp.html?ID=<stream no.>.

Activate stream	Stores parameters and activates or deactivates the receiving of audio data. (Unicast: additionally the negotiation of the connection)
Stream Output	Selects one or both NICs used for stream output. Both NICs means output redundancy.
Stream name (ASCII)	Individually defined name of an output stream. It is used in the URL which is indicated in different ways below.*
RTSP URL (HTTP tunnel) (by-name) / (by id)	Current used RTSP -URL of stream with HTTP port used for RTSP, stream name or stream id.
RTSP URL (by-name) / (by id)	Current used RTSP -URL of stream with stream name or stream id.
SDP	SDP data of the active stream.
Unicast	If activated, the stream is sent in unicast mode.**
RTP payload ID	Stream's payload id
Samples per Frame	Number of blocks containing payload (audio) per ethernet frame - see packet time on p 26.
Audio format	Stream's audio format (L16 / L24 / L32 / AM824) ***
Start channel	Assignment of first stream channel from the audio matrix. E.g. stream with eight channels, starting at channel 3 is fed from channel 3 to 10 of the routing matrix.
Number of channels	Number of audio channels in the stream.
RTP dst port	Stream's destination port for RTP
RTCP dst port	Stream's destination port for RTCP (Real-Time Control Protocol)
Dst IP address (IPv4)	Stream's IP address for multicast (should be unique for each stream).

^{*} Only ASCII characters are allowed.

^{**} A unicast stream can only be received by one device. If a device is already receiving the stream, further connection calls by other clients are answered with, service unavailable (503). The release time after disconnect or interruption of the client's connection amounts to about 2 minutes.

^{***} L16 = 16 bit audio / L24 = 24 bit audio / L32 = 32 bit audio / AM824 = standardized according to IEC 61883, allows AES3 transparent transmission (SMPTE ST 2110-31).

Advanced - Overview

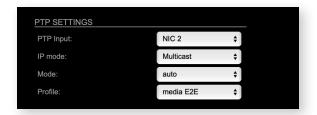


The tab 'ADVANCED' is divided into several sections:

- PTP SETTINGS
- PTP PROFILE
 - CURRENT
 - SETTINGS
- CURRENT PTP MASTER
- PTP STATISTIC
- PTP CLOCK SETTINGS
- NETWORK ADVANCED SETTINGS
- PTP JITTER

- definition of PTP source, mode and profile
- definition of a customized PTP profile
- monitoring PTP characteristics
- monitoring device's PTP state, jitter and delay
- definition of adaption algorithms to reduce jitter
- definition of network and QoS characteristics
- graphical display of measured PTP jitter

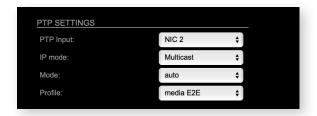
Advanced - PTP Settings



PTP Input	Selects one or both network ports used for PTP input. Both ports means input redundancy. *		
IP Mode	Multicast	=	Sync messages and delay request are sent as multicast message to every node within the network.
	Hybrid	=	Sync messages are sent as multicast, delay requests are sent as unicast messages directly to the Grandmaster or Boundary Clock.**
	Unicast	=	Sync messages are sent as unicast, delay requests are sent as unicast messages directly to the Grandmaster or Boundary Clock.***

- * Using redundant PTP-operation a switch-over is triggered not only at signal loss of the Grandmaster but depends on the quality of the PTP clock.

 Changes (e.g. clock class) are observed permanently and the algorithm decides for the best signal present.
- ** Hybrid Mode reduces the workload for all nodes in the network as they do not receive the (unnecessary) delay requests from other devices anymore.
- *** Unicast Mode may help when multicast routing is not possible within the network. As an opposite to the Hybrid Mode it increases the workload of the grandmaster since sync messages must be sent to each single slave individually.



Mode	auto	 PTP-clock master / slave configuration is auto negotiated between devices in the network. Module's master / slave state may change automatically.
	slave only	 PTP-clock slave configuration is preferred. Module clocks to another device in the network
	preferred master	 PTP-clock master configuration is preferred. Module acts as network grandmaster. Priority values are adjusted automatically to ensure Grandmaster status. *
	master only	= PTP-clock master is forced. **
Profile	-	d PTP profile (default E2E, default P2P, P2P) or activates customized PTP profile.

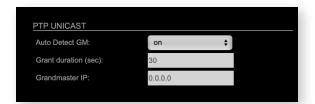
- * If more than one device announces as PTP-clock master the network Grandmaster is determined following the Best Master Clock Algorithm (BMCA).
- ** 'Master only' configures the device to act as Unicast Grandmaster. This setting is available only with PTP Mode set to 'unicast'



NOTE

PTP profile,customized' allows for individual adjustment of the PTP parameters. If profile is set to,media' or,default' the PTP parameters cannot be altered and are displayed only. Factory default setting is PTP Media Profile E2E.

Advanced - PTP Unicast



Auto Detect GM	on = enables the automatic detection of the grandmaster * off = IP address of grandmaster needs to be defined manually
Grant duration (sec)	Time period during which the slave receives sync messages from the grandmaster.**
Grandmaster IP	IP address of the grandmaster. ***

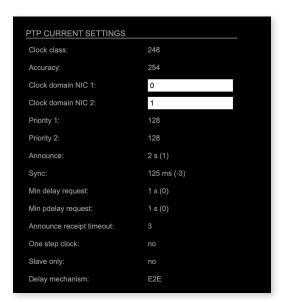
- * 'Auto Detect GM' is a proprietary function and might not be supported by 3rd party GMs.
- ** Depending on the temporary workload of the grandmaster the negotiation may fail.
- *** This value is used only with 'Auto Detect GM' set to <off>.

About PTP Unicast

Since the BMCA is not available with PTP unicast, the PTP properties of the devices require some extra configuration.

Example:

Grandmaster	IP Mode Unicast, Mode Master only
Slave(s)	IP Mode Unicast, Mode Slave Only,
	Auto Detect GM ON, Grant Duration 30 sec



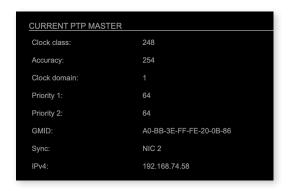
Advanced - PTP Profile Customized Settings

The settings become available with PTP profile set to 'customized'.

Clock class	PTP-clock's class according to IEEE 1588 [read only]
Accuracy	PTP-clock's accuracy according to IEEE 1588 [read only]
Clock domain NIC 1	PTP-clock's domain at NIC 1
Clock domain NIC 2	PTP-clock's domain at NIC 2
Priority 1	Priority setting for master announcement (the smaller the value the higher the priority)
Priority 2	If value 'Priority1' (and other PTP-clock parameters) of more than one device in the network match: Priority setting for master announcement (the smaller the value the higher the priority)
Announce	Intervall of sending announce-packets for autonegotiation.
Sync	Intervall of sending sync-packets to the PTP-clock slaves in the network.
Min delay request	Intervall of sending End-To-End packets of PTP-clock slave to PTP-clock master. To determine the offset slave-to-master.
Min pdelay request	Intervall of sending Peer-To-Peer packets between two PTP-clocks. To determine the offset master-to-slave and slave-to-master.
Announce receipt timeout	Number of missed announce-packets (threshold) to reinitialize the negotiation of PTP-clock master.

One step clock	Timestamp of PTP-clock is integrated in PTP-sync- packets. No follow-up packets are sent. No = Two step clock is used
Slave only	Yes = PTP-clock is always slave.
Delay mechanism	E2E - Offset slave-to-master is determined by End-To- End packets. P2P - Offset master-to-slave and slave-to-master is determined by Peer-To-Peer packets.

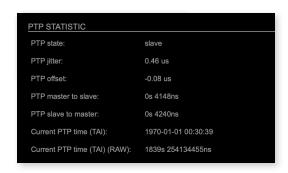
Advanced - Current PTP Master



Monitoring display only.

Clock class	PTP-clock's class according to IEEE 1588
Accuracy	PTP-clock's accuracy according to IEEE 1588
Clock domain	PTP-clock's domain at selected NIC
Priority 1	Priority setting for master announcement (the smaller the value the higher the priority)
Priority 2	If value 'Priority1' (and other PTP-clock parameters) of more than one device in the network match: Priority setting for master announcement (the smaller the value the higher the priority)
GMID	ID of current Grandmaster
Sync	Selected NIC for PTP clock
IPv4	IP address of Grandmaster

Advanced - PTP Statistic



Monitoring display only.

PTP state	Information about current PTP-clock state: intialize error deactivated receiving data pre master master passive not calibrated slave
PTP jitter	PTP-clock jitter in microseconds (μs)
PTP offset	Offset relative to PTP-clock master
PTP master to slave	Absolute offset master-to-slave in nanoseconds
PTP slave to master	Absolute offset slave-to-master in nanoseconds
Current PTP time (TAI):	Date and time information from GPS source*
Current PTP time (TAI) (RAW):	RAW TAI from GPS source*

* Temps Atomique International - if no GPS source is available for PTP timestamping,the date / time display starts at 1970-01-01 / 00:00:00 after every reboot of the device.

Advanced - PTP Clock Setting



No PTP Switch 1 Gbit/s	Adapted PTP-clock algorithm to reduce clock jitter using 1 GB network switches without PTP support. Max. number of 1 Gbit/s switches: less than 10
No PTP Switch 100 Mbit/s	Adapted PTP-clock algorithm to reduce clock jitter using 100 MB network switches without PTP support. Max. number of 100 Mbit/s switches: 1

NETWORK ADVANCED SETTINGS IGMP NIC 1: auto IGMP NIC 2: auto TCP port HTTP: 80 554 TCP port RTSP: TTL RTP packets: 128 AF41 (0x22) DSCP RTP packets: DSCP PTP packets: Multi stream rx: yes RX/TX MDNS announcement: RX/TX SAP announcement: **‡** Network settings: Apply

Advanced - Network Advanced Settings

IGMP NIC 1	Definition or auto-select of IGMP version used to connect to a multicast router at NIC 1.
IGMP NIC 2	Definition or auto-select of IGMP version used to connect to a multicast router at NIC 2
TCP port HTTP	TCP port for HTTP
TCP port RTSP	TCP port for RTSP
TTL RTP packets	Time-To-Live of RTP packets - default: 128
DSCP RTP packets	DSCP marking of QoS of RTP packets - default: AF41
DSCP PTP packets	DSCP marking for QoS of PTP packets - default: CS6*
Multi stream rx	If activated, the device allows to subscribe to the same multicast stream more than one time - default: off
MDNS announcement	Announcement of streams via MDNS can be controlled to optimize network traffic or CPU load. Values: Off, RX, TX or RX/TX **
SAP announcement	Announcement of streams via SAP can be controlled to optimize network traffic or CPU load. Values: Off, RX , TX or RX/TX **
Network settings Apply	Confirms and saves changes being made. Reboot required.

^{*} AES67 specifies EF, but some implementations use EF for Audio streaming.

To avoid overlapping of RTP and PTP packets in the same queue CS6 has been chosen as default.

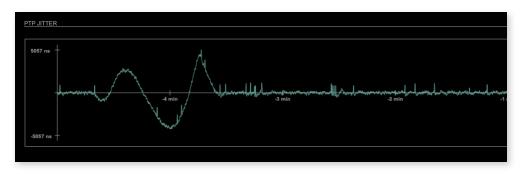
^{**} RX = receive, TX = transmit, RX/TX = receive and transmit

NOTE



Source-Specific Multicast (SSM) support for IGMP v3, v2 and v1 (SSM via protocol only in IGMP v3, SSM via internal filtering is applied for IGMP v2 and v1) - see "Source Specific Multicast" on page 30.

Advanced - PTP Jitter



Graphical display of measured PTP jitter.

NOTE



An error message next to Jitter measurement is displayed if delay requests are not being answered by Grandmaster.

NMOS - Overview



NMOS provides a family of specifications related to networked media for professional applications. It is produced by the Advanced Media Workflow Association (AMWA).

Support for NMOS is introduced with the AoIP Module version SW 0.17 / HW 0.46 according to the specifications:

- IS-04 Discovery & Registration
- IS-05 Device Connection Management

IS-04 allows control and monitoring applications to find the resources on a network. Resources include Nodes, Devices, Senders, Receivers, Sources, Flows...

IS-05 provides a transport-independent way of connecting Media Nodes.

More information: https://specs.amwa.tv/nmos/

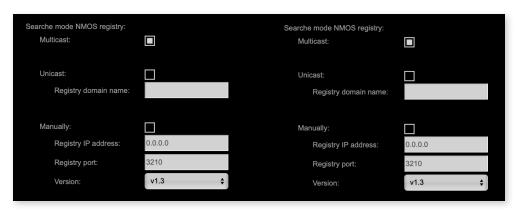
NMOS port - NIC1 & NIC2

The port entries for NIC1 and NIC2 are pre-configured by default. Modifications are possible but not necessary.



NMOS port	Port address. Reboot required after modification.
(NIC1 + NIC2)	

Search mode NMOS registry



Multicast	use mDNS to determine and connect to the registry server
Unicast	use DNS-SD to connect to the registry server
Registry domain name	DNS resolvable domain name of the registry server
Manually	
Registry IP address	
Registry port	
Version	Support of NMOS API version

NMOS - Additional Settings

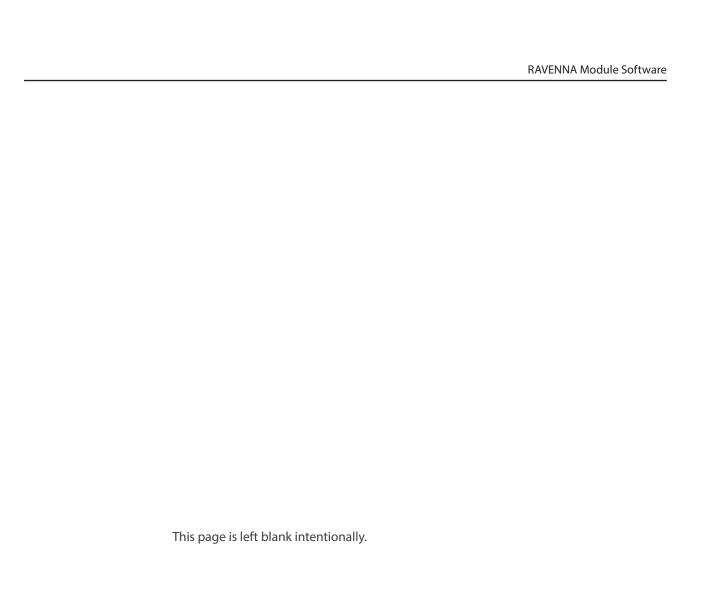


Disable stream during config	Automatically disable and re-enable streams when settings are changed via NMOS (recommended)
Seed id	Unique identifier, subordered entities are derived from the seed id.
Generate new seed id Generate	Generates a new unique identifier. Reboot required.

NMOS uses a logical data model based on the JT-NM Reference Architecture to add identity, relationships and time-based information to content and broadcast equipment. Hierarchical relationships group related entities, with each entity having its own identifier.

The identifiers are persisted across restarts of the device in order to make them useful over a period of time longer than a single production deployment.

New identifiers may be generated manually if required.



Logging



The tab 'LOGGING' displays logging depending on the 'Log Settings'. The logging can be enabled individually for different protocols, each of with an adjustable filter. An adjustable log level specifies the information detail of each entry.

To save a log the content of the view can be copied and pasted to a text document.

Log Level

0	log data
1	level and log data
2	protocol, level and log data
3	protocol, process-id of requesting process, process-id of running process, level and log data
4	protocol, process-id of requesting process, process-id of running process, level, processor time in ticks and log data
5	protocol, process-id of requesting process, process-id of running process, level, processor time in ticks, file name and line and log data

Protocol Types

, ,	
ARP	Address Resolution Protocol
BASE	Basic operation of module
DHCP	Dynamic Host Configuration Protocol
DNS	Domain Name System
FLASH	Process for updating the module
IGMP	Internet Group Management Protocol
MDNS	Multicast Domain Name System
NMOS	Network Media Open Specification
PTP	Precision Time Protocol
RS232	Serial Protocol
RTCP	Real Time Control Protocol
SAP	Session Announcement Protocol
ТСР	Transmission Control Protocol
Zeroconf	Zero Configuration Protocol

Log Filter

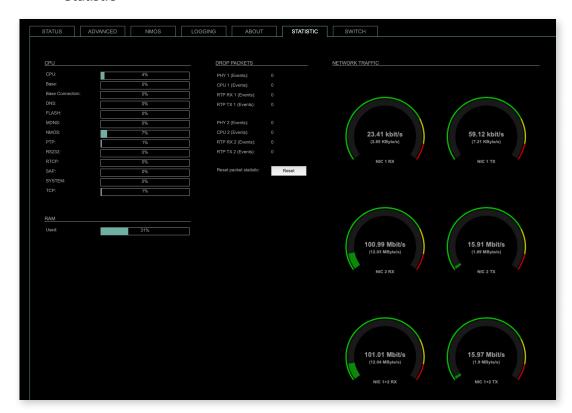
NONE	logging disabled
ERROR	error occurred
WARNING	warnings - condition that may lead to unwanted behavior or an error
INFO 1	log info* + warning + error
INFO 2	log info* + warning + error
INFO 3	log info* + warning + error
INFO 4	log info* + warning + error

^{*} increasing amount of log info starting from,INFO 1'

Log Operation

Save log	Downloads the current log entries to a text-file (log.txt).	
Clear log	Deletes all log entries without further prompt.	
Scroll lock	Interrupts automatic scrolling of the list view to allow copying the content to a text file via copy & paste. If scrolling is stopped for a longer period of time the display may not list all entries.	

Statistic



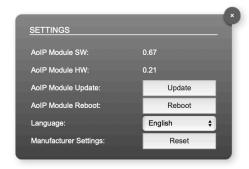
The tab 'STATISTIC' displays an overview of the CPU load of the particular processes, an error counter and a monitor display to indicate the incoming (RX) and outgoing (TX) network traffic on both network ports individually.

See "Protocol Types" on page 51.

RAV2 - Firmware Update

The RAV2 module is updated via network.

Open the control page of the module and navigate to the tab STATUS and click SETTINGS in the top right corner (p 20).



Click 'Update' and browse to the update file after unzipping first. Example: rav_io_hw_0_21_sw_0_62.update

Folllow the instructions displayed.



WARNING!

It is strongly recommended to backup the device configuration (Save Preset) before running any update.

Service Information

1. Software and firmware updates

Download software updates from our website. No registration required.

http://www.avt-nbg.de

Navigate to Downloads > Software.

2. Support

Our support is available on working days:

Monday to Friday from 09:00 - 16:30 CET.

Support portal: https://avt-nbg.zammad.com

Email: support@avt-nbg.de Phone number: +49 911 5271-110



To deal with your problem efficiently please note down the factory number of the unit as well as the software version that you use.

The factory number is visible in the software under Administration > Registration.

If you bought the system via your local dealer, please contact them first.

3. Repairs

If your unit is defective, please contact us before sending in the device.

To send in the system please fill in the included Service Request and send the unit to the following address:

AVT Audio Video Technologies GmbH

- Repairs -

Nordostpark 91 90411 NÜRNBERG

GERMANY

4. WEEE (Directive on Waste Electrical and Electronic Equipment)

Due to Directive 2012/19/EU on waste disposal, this device must be recycled. All electrical and electronic equipment must be disposed of separately from general household waste via approved collection points or disposal companies. The proper disposal and separate collection of old electrical and electronic equipment serves to prevent possible damage to the environment and health. The device contains valuable raw materials that can be recycled.



For proper recycling, send the device to us:

AVT Audio Video Technologies GmbH
- Recycling Nordostpark 91
90411 NÜRNBERG
GERMANY

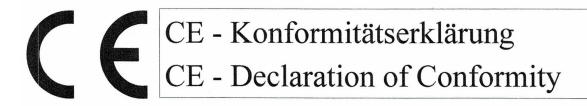
WEEE Reg. No. DE83099164

Only prepaid parcels will be accepted!

These instructions only apply to appliances installed and sold in countries of the European Union. In countries outside the European Union, other regulations may apply to the disposal of electrical and electronic equipment.



Always recycle packaging material and electrical appliances or their components through authorised collection points or disposal companies.



Das Erzeugnis entspricht den grundlegenden Anforderungen der zutreffenden EG-Richtlinien. Ein in den Richtlinien vorgesehenes Konformitätsbewertungsverfahren wurde durchgeführt.

The product conforms to the basic requirements of the relevant EC directives. A conformity assessment method as provided for in the directives has been performed.

Hersteller:

Manufacturer:

DirectOut GmbH Leipziger Str. 32 09648 Mittweida Germany

Gerätebezeichnung: Device designation:

RAV2

Zum Nachweis entsprechend For verification in accordance with

RL2014/30/EU RL2011/65/EU RL2015/863EU

wurden folgende harmonisierte Normen herangezogen: the following standards were applied:

EN 62368-1:2021 EN 50581:2012 (RoHS)

Die oben genannte Firma hält die folgende technische Dokumentation zur Einsicht bereit:

- Bedienungsanleitung
- Konstruktionspläne
- Prüfunterlagen
- sonstige technische Dokumentation

The company mentioned above keeps the following technical documentation ready for inspection:

- operating instructions
- design diagrams
- test documents
- other technical documentation

Diese Erklärung wird verantwortlich für den Hersteller abgegeben durch This declaration is given for the manufacturer by

Claudio Becker-Foss, CEO/CTO DirectOut GmbH Mittweida, 31.07.2022



Leipziger Str. 32 09648 Mittweida Germany

DirectOut GmbH info@directout.eu www.directout.eu P +49 3727 56 65 -100 F +49 3727 56 65 -101