

IP-8e

Release Notes

Version 2.16.4

01.04.2025



IP-8e Release Notes

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Version 2.16.4

Fixed Issues

- Synchronous playout SPN based on NTP may be broken (off by some seconds) when the XLR inputs are used as the input source

Version 2.16.3

13.03.2025 – identical to 2.16.3-rc3

New Functionality

- The PTP delay mechanism to use is now configurable
- DAB+ Encoder: added some status values of the FhG MuxEnc and the PAD bitrate to external APIs like SNMP

Changed functionality

- In the case of limited screen width, the new hamburger menu is now more prominent. Additionally, the collapsed menu can be pinned to stay on the screen even in case of limited screen width.
- The PER (packet error/loss rate) for RTP inputs was formerly measured in an interval of 1/10 of the T1 time of the packet loss switch criteria time (where the default is 60s resulting in a default measurement interval of 6s). The measurement does now adapt to the type of incoming stream and its number of packets per second to allow for a resolution of the PER of 0.1% (meaning 1 of 1000 packets). To be able to measure that at least 1000 packets need to be the measurement interval which might result in the PER not updating very frequently.

Fixed Issues

- The silence mode available for elementary stream and Livewire input sources was not working immediately after a reboot or when enabling/disabling it
- Icecast client: enhanced compatibility to streams providing MPEG Layer 2 audio data
- Fix a possible application hangup when loading settings (leading to recovery page)
- DAB+ Encoder: local PAD insertion for DAB DCP output had some problem when inserting DLS+ tags
- Prevent duplicate VLAN IDs on same interface
- RTCP sender report may get broken after some time (breaking automatic codec detection and transport of the global delay information in case of synchronous playout / SFN)
- Enhanced compatibility to some WAV files
- PTP for unicast configuration improved
- Enhanced compatibility of the Icecast client in case of FLAC Icecast server content



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Version 2.16.2

04.02.2025 – identical to 2.16.2-rc1

New Functionality

- Added a new button to the Overview page to reset the counters of all encoders
- Optional DAB+ encoder: the FhgMuxEnc protocol for the connection to a Fraunhofer DAB Mux is now sending automatically a status for "Audio Silence Detection" together with the audio frames. The Fraunhofer DAB Mux is thereby able to determine if the DAB+ encoder feed is erroneous due to the audio silence detection and can switch internally to a backup source.

Changed functionality

- The menu of the web interface does now collapse to a hamburger menu in case screen width is limited
- TS Multiplexer: The "PID removal on bad input" functionality has a new option to stop the TS output completely
- Show Dual Streaming block on Overview even if FEC is enabled

Fixed Issues

- Bunch of internal improvements and optimizations
- A livewire input source may suddenly start to provide twice as much packets as before when the "Silence mode" is activated (only true for Livewire destinations targeting the encoders)
- Headphone output may be distorted
- FLAC codec did not work in combination with SRT
- SRT in caller mode doesn't always connect reliable with the SRT listener
- When changing the ancillary data configuration the device may crash
- xHE-AAC encoder: added parameters to control live loudness and DRC
- UDP ancillary input improved (might not work after reboot when DHCP is enabled)
- When changing a TS/Demux source and switching to the Overview page the device could crash with certain transponders
- Check minimum SRT passphrase length
- TS/SRT decoding was broken



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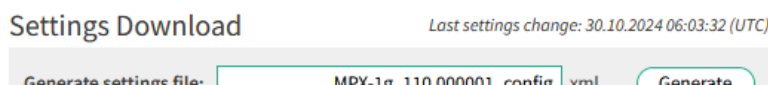
Version 2.16.1

26.11.2024 – identical to 2.16.1-rc4

New Functionality

2.16-1-rc1

- Added a configuration option to file input sources to play only once (no loop)
- Last settings change: The time can be read out as a UNIX timestamp or human readable date. This is also displayed on the Global page:



Changed functionality

2.16-1-rc3

- TS Multiplexer: The “PID removal on bad input” functionality has two different options now. It can not only remove both the PID itself and the signaling via PAT/PMT (what was done until now), but optionally only remove the PID, but still signal it via PAT/PMT.

Fixed Issues

2.16-1-rc4

- Enhanced resilience against audio drops

2.16.1-rc3

- Enhanced stability
- Live Listening: switching the input source while listening may not work reliable
- Loading settings might not activate all input source settings

2.16.1-rc2

- Fix seldom audio drops in the encoded audio
- Avoid alarms of inactive channels due to reduced channel count In “Analog/Digital” device mode
- Fix the possibility to get frequent “Encoder – no input data” alarms without “good” messages In between
- Optional DAB+ encoder: Changed default value of DCP output stream data packet spreading to 50% to avoid problems with the receiving multiplexer

2.16.1-rc1

- Optional DAB+ encoder: DCP output stream did not work reliable with FEC activated
- Automatic codec detection for TS/Demux input sources may not signal correct AAC type
- Fixed possible application crash when loading settings or loading factory settings
- Individual status values for Livewire input sources did not work via external APIs
- TS Multiplexer: "Low bitrate priority" mode did not lead to the lowest possible audio bitrate overhead
- E-aptX decoding inside TS did not work
- Live Listening might be done with wrong audio speed (when switching between different input sources)
- Enhanced compatibility of the optional HLS client/decoder to certain streams



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- Optional AES67 outputs had a generic name in the SAP announcements – now the audio output name configurable via Codec / Input Sources / Interfaces gets used (if set)
- Optional AES67 outputs: improved startup behavior, inhibiting a possible output burst at startup



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Version 2.16

16.10.2024 – identical to 2.16-rc8

New Functionality

2.16-rc7

- TS Multiplexer: added support for SCTE-35 (Digital Program Insertion Cueing Message)
There are new SCTE-35 endpoints/inputs available on the TS Multiplexer configuration page:

Codecs Settings

TS Multiplexer

Payload sources

Encoder audio Data SCTE-35

Input Source

- SCTE-35 Input 1
- SCTE-35 Input 2
- SCTE-35 Input 3
- SCTE-35 Input 4
- SCTE-35 Input 5
- SCTE-35 Input 6
- SCTE-35 Input 7
- SCTE-35 Input 8

Multiplex 1 × +

General

Encoding Standard: DVB

MPEG TS tables: All tables

Auto-calculate required TS bit rate: ON

Bitrate priority: Low latency

PID removal on bad input: OFF

Network ID: 1

Original Network ID: 1

Transport Stream ID: 100

Network name:

TS Payload content

Service ID	Service Name	Service Provider Name	PMT PID	PCR PID	Mode	Payload	PID	Language
1000	Program 1		100	101	PES	Enc 1 Loudness Test1	101	
						SCTE-35 Input 1	102	

Add Payload

Add Service

Splice inserts must be sent via one of the external APIs. Via SNMP the node is virtCslAudioEncoderScte35Spliceinsert (OID 1.3.6.1.4.1.21529.1001.2.3.18.466)

The splice information has to be passed in JSON format and should look like this:

```
{"cue_point": { "event_id": "1073742575", "splice_time": { "year": 2024, "month": 9, "day": "6", "hours": "12", "minutes": "17", "seconds": "37", "mseconds": "210"}, "duration": "149.5" } }
```

(Time information in UTC, more information will follow in the manual)

- Started to add the possibility to get a counter history on the Overview page via a small icon next to the label. It's more or less a shortcut to the extended log, showing you only the entries for the counter of this



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input source. Currently it's added to the "Missed" and "Timed out" counter, others should follow.

IP					
Src address	Src port	Bitrate	Packets/s	Jitter	
	0	0	0	0.0 ms	
Missed	PER	MDI	Timed out	Max size	Buffer
0	0.0 %	0.0:0.000	0	0	0 ms

2.16-rc3

- New silence mode for elementary stream inputs: when enabled, empty packets will be inserted in case of missing stream input (like the "Generate Null samples" option for AES/EBU inputs, when no input is connected)
This will optionally allow an encoder to generate output from an elementary stream input even if no ES Input is present
- Added the possibility to clear the extended log
- Optional HLS streaming was not working properly with some CDN servers due to cache-control

2.16-rc1

- Added optional TSL/SSL encryption for Live Listening (to support access via https from within the web interface)
- Added separate status values for the TS Multiplexer outputs readable via external APIs (see e. g. SNMP MIB)
- Added uptime values for RTP/SRT input streams and RTP/SRT encoder outputs as separate values readable via external APIs

Changed functionality

2.16-rc3

- Icecast server: the maximum number of clients was limited to five. This is increased now to 15 (but only after a "Load factory settings" is done). Alternatively, the maximum number of clients can be changed via the external APIs (the SNMP node is e. g. `cs1AudioEncoderOutputsIcecastTimeout`, OID 1.3.6.1.4.1.21529.1001.2.3.18.20.319.4096.1.11)

2.16-rc1

- Consolidated TCP/IP page to only have one "Save" button
- The event log page does now also refresh automatically when it is initially empty

Fixed Issues

2.16-rc8

- Fixed a problem on the Codec page, sometimes not allowing to switch between the "General" and the "Switch Criteria" tabs in the input source configuration dialogs
- A disabled decoder (switched to Off) could still show a green/valid status on the Overview page instead of the grey/inactive status

2.16-rc7

- Elementary stream output with external clock set to NTP, activated SPN/SFN and DualStreaming with send delay $\neq 0$ can lead to jumps in the AES67 output of a decoding MoIN instance, leading to errors on devices receiving the AES67 output
- Icecast client: https Icecast streams with credentials may have problems with authentication



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- TS Multiplexer: configuring a new multiplex via the web interface can lead to an invalid configuration
- (Limited) support for Internet Explorer 11 was broken

2.16-rc4

- AES/EBU CRC check failed in case the connected input is in consumer mode
- Clearing the extended log did not work (the latest entries reappeared after a few seconds)
- X.509 certificate error log entries moved from event log to extended log
- Fix accidentally shown number in encoder info blocks on Overview page
- Second stream in DualStreaming setup may show missed packets, although they weren't missed
- Added "Remote command" to GPO switch source select

2.16-rc3

- Elementary output streams with enabled SPN/SFN improved in case NTP synchronisation is active
- A change of the PCM multichannel offset of an elementary stream input was not taken into account (only after disabling/enabling the input source or rebooting the device)
- Icecast server: Fixed long timeout (blocking one of the limited number of client slots) in case of broken TCP/IP connection
- Optional DAB+ encoder: Fixed a problem with the VLAN configuration of the DAB DCP outputs
- Optional DAB+ encoder: Fixed a problem in combination with the IZT DAB ContentServer
- Optional DAB+ encoder: Fixed a problem, when the encoder clock adaptation (called "Encoder SRC" in the Profiles settings) was activated
- In case an audio input is switched to AES67 (via Interface Settings / Audio / Signal type), the switch criteria "AES/EBU no signal" will not monitor the AES/EBU status (which makes no sense in case of AES67), but the status of the incoming AES67 stream

2.16-rc2

- CRC check/alarm for AES/EBU inputs did not work when the sample rate converter was enabled
- Optional HLS streaming was not working properly with some CDN servers due to cache-control

2.16-rc1

- Fixed high number of DNS requests with NTP synchronization enabled
- Fixed T1/T2 alarm times not editable
- Fixed possible crash when embedding ancillary data in the encoder with certain codecs
- ProMPEG FEC encoder reports wrong FEC matrix when one FEC port offset is set to 0
- ProMPEG FEC decoder let main input stream fail, if one FEC port offset is set to 0
- Dedicated status values for TS/SRT did not work at all
- "Automatic" decoder config did not work when relying on SAP as the source for the codec detection (e. g. for AES67 inputs)



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Version 2.15.4

01.07.2024

Changed Functionality

- The Log web page does now optionally refresh automatically even when the event log is initially empty

Fixed Issues

- Fixed small problems with the Livewire input source configuration dialog
- "Automatic" decoder configuration did not work when relying on SAP as the source (therefore sample rates different to the default 48 kHz were not recognized and did not work)

Version 2.15.3

25.06.2024

New Functionality

- Optional DAB+ encoder: added UTF-8 support for the PAD DLS charset

Fixed Issues

- The ASI input did still show some problems and CC errors
- The optional Dolby decoder did not work correctly and produced audio glitches
- Fixed some problems with the optional Live Listening feature

Version 2.15.2

07.06.2024

Changed Functionality

- File input did only support files < 2 GB
- Added support for WAV files with BW64/RF64 file format

Fixed Issues

- PTP might not be active after loading a settings file with a corresponding configuration
- Ancillary data sources might not work after loading a settings file with a corresponding configuration
- Icecast server: removed the additional "bitrate" information from the connection request answer (in addition to the "ice-bitrate" information, introduced in 2.15-rc7), as it breaks compatibility with other Icecast clients, e. g. VLC
- Changing the SNMPv3 authentication protocol (MD5 <-> SHA1) did not work



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- After IP address change the SRT decoder (in Listener mode) did not receive any SRT streams
- Corrected the spelling of "Input" in the encoder low level alarm log entries
- Headphone Downmix did not work in Analog/Digital Mode
- Optional DAB+ encoder: input source reassignment for an encoder was not correctly considered when the encoder was also used for a DAB Submux

Version 2.15.1

29.05.2024

New Functionality

- Optional DAB+ encoder: Added DAB Submux functionality
- Optional DAB+ encoder: Added the possibility to update the DLS via SFTP

Fixed Issues

- When doing "Load factory settings", the user account passwords were not reset to default
- When loading a settings file, the user account passwords were not taken from the loaded settings file
- The metronome icon on the Overview page (which should signal external clock usage and its state) was not shown appropriately
- Elementary stream input sources with multicast addresses were re-initialized when saving although no relevant changes were made
- Web interface: session timeout did no longer work
- SRT input stream will be only restarted if parameters were changed



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Version 2.15

12.04.2024

New Functionality

- The ASI input is now operational and shown as additional input source
- Added ping and traceroute possibility to the web interface (Network Settings / TCP/IP / Tools)

Network Settings
↳ TCP/IP

General Tools

Ping

Settings

Destination: heise.de

Interface: Data 1

Count: 5

TTL: 255

Data size (0=default): 0

Start Stop

Output

PING heise.de (193.99.144.80): 56 data bytes

64 bytes from 193.99.144.80: seq=0 ttl=247 time=10.199 ms

64 bytes from 193.99.144.80: seq=1 ttl=247 time=10.105 ms

64 bytes from 193.99.144.80: seq=2 ttl=247 time=9.901 ms

64 bytes from 193.99.144.80: seq=3 ttl=247 time=9.894 ms

64 bytes from 193.99.144.80: seq=4 ttl=247 time=9.773 ms

--- heise.de ping statistics ---

5 packets transmitted, 5 packets received, 0% packet loss

round-trip min/avg/max = 9.773/9.974/10.199 ms

Traceroute

Settings

Destination: heise.de

Interface: Data 1

Max. hops: 5

Time to wait: 3

Start Stop

Output

traceroute to heise.de (193.99.144.80), 5 hops max, 38 byte packets

1 192.168.96.1 (192.168.96.1) 0.212 ms (64) 0.221 ms (64) 0.192 ms (64)

2 mx204-2.ham.purtel.com (185.39.84.9) 7.292 ms (254) 4.488 ms (254) 4.079 ms (254)

3 * * *

4 100.83.140.3 (100.83.140.3) 3.414 ms (252) 3.503 ms (252) 3.845 ms (252)

5 be100.c350.f.de.plusline.net (80.81.193.132) 10.210 ms (251) 10.138 ms (251) 10.077 ms (251)

- Added event log entries when loading a settings file or loading factory settings
- When generating a settings file on the Global page you can now not just only download the file, but also save it to the internal storage. You can also directly change the name of the settings file.

Generate settings file: IP-4c_740.000145_config.xml Generate

Settings file: IP-4c_740.000145_config.xml

Actions: Save to storage Download

- Settings files can also be uploaded via the Storage page

Status
↳ Storage

Internal Storage NFS Storage aws S3 Storage

Data storage

72%

1758.1 MB free of 6282.7 MB

Audio files Settings files

Filename Date (UTC) Size Delete

IP-4c_740.000145_config.xml 12.03.2024 08:53:52 4.4 MB Delete

IP-4c_740.000855_config.xml 12.03.2024 08:54:24 3.8 MB Delete

Upload file Note: Only for file size < 100MB (for bigger files use SFTP)

Supported types: Audio, firmware and settings files.

Browse / Drop file No file selected Upload



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- Once there are settings files on the internal storage the user can select one of the stored files instead of uploading one and load it via the new consolidated Upload/Activate section, thereby having a sort of preset functionality, allowing to quickly switch between different configurations.

- Optional DAB encoder: regarding PAD insertion it's now possible to choose between local PAD and FhG MuxEnc (or none of course)

- Optional DAB encoder, DCP output: Added ETI and STI-D output mode
- Added encoder input low level detection alarm (in addition to the silence detection alarm)

Changed functionality

- The PTP configuration did only allow to configure a single IP interface, providing no backup functionality in case PTP fails on that interface. Now all interfaces configured for streaming data (under Network Settings / Services) will be used for PTP, automatically switching between the interfaces when needed
- New version of the xHE-AAC encoder (04.05.04) with default loudness -24 LUFS
- Updated Fraunhofer libraries to latest versions
- TS Multiplexer: Rework of the private data encapsulation mode configuration. Instead of configuring a global mode the mode can now be configured individually per payload

TS Payload content

As part of this change we also added three more private data encapsulation modes: TS Adaptation, Payload (behind TS Adaptation) and a special Tandberg compatibility mode.



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- RDS Databridge (optional): If no RDS update is received for a period of time, the Databridge can switch the RDS encoder to a "Fallback DSN".

Dynamic update fallback ?

Enable DSN fallback: ☒ ON

Standard DSN: Fallback activation after: s

Fallback DSN:

- The different upload sections on the Global web page (Firmware, Settings, Rights, SSL Certificate) were consolidated to just one section allowing to upload every of the four kinds of files

Upload / Activate Check for Updates ?

Supported files: Firmware, Settings, Rights and SSL-Certificate

No file selected

- The manager user can be granted limited access to the Global web page, allowing just to switch between different configurations
- Cleanup of SRT / TS/SRT Overview info block fields
- Reduced sensitivity of the jog wheel for better user experience when doing configurations via the LCD menu
- Changed Load Factory Settings via LCD menu to new method without reboot
- Loading settings via the web interface will show a warning when trying to load settings from a different device type. The settings can still be loaded, though, as many settings are of general nature.

Fixed Issues

- Web interface: on the Login page the user name field now has the focus on page load to allow immediate typing without the need for the additional click into the field
- Web interface: for elementary stream and TS/IP input sources not applicable configuration options are getting hidden in case of UDP protocol selection
- Web interface: fixed PHP errors on GPI page
- Icecast client: enhanced compatibility to OGG/FLAC streams
- Icecast server: enhanced compatibility to some clients (expecting a "bitrate" information instead of the "ice-bitrate" information)
- RDS Databridge (optional): the EON TA functionality does now respect the EON PI configuration and will not react on EON Ta from a service with a different PI then the configured one
- Fixed Icecast client issues with Icecast streams with ContentType audio/mpegurl
- Optional HLS encoding: fix wrong HLS container formats
- General security and access control improvements
- PTP did only work with domain 0
- Fixed some problems with HE-AACv2 and the Icecast Source Client and Icecast Server
- Enhanced compatibility to AAC encoded in short frame mode
- Optional DAB encoder, DCP output: optimization of PAD interface
- After settings updates (loading a settings file or loading factory settings) input source changes may not be applied
- Under high CPU load IP packets may get lost and reported as missed



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- When doing “Load Factory Settings” and unplugging the power immediately after completion, the new (default) settings might not have been saved
- High CPU load in case of incomplete TCP connections (Icecast/HLS)
- Web interface: the page heading may show an incorrect title when some menu entries are disabled



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Version 2.14

29.01.2024

New Functionality

- When loading settings via the web interface it can now optionally include the TCP/IP configuration
- Added the IP interface link status to the IP interface selection in the different configuration dialogs
- Added speed selection (Auto / 1000 Mbit / 100 Mbit) to the TCP/IP interface configuration.
Sometimes auto negotiation does not work correctly. In this case a manual configuration might help.
- Added silence detection alarm for encoder inputs
- Added missing VLAN support to SRT input/output configurations
- For security reasons the password hashes of the user accounts are no longer stored as MD5, but as bcrypt hashes. The migration will be done automatically.
- Added the possibility to update the firmware locally from USB
- Added the fan status to Status→Device in the web interface (needs System Controller firmware 1.07)
- New (optional) DAB+ encoding functionality
- SRT: added source address and port as well as uptime (time since last connection start) to the decoder status information
- Added a possibility to configure a delay for the GPI Forwarding/Tunneling (per profile)
- Added the possibility to locally load settings from and save settings to USB via the LCD menu.
The settings loading from USB can be done in two different ways – one without and one including the TCP/IP configuration.
- Added the possibility to access the latest 50 event log entries via external API (e. g. SNMP)
- Added SNMP traps for detected local GPI and remote GPI (GPI Forwarding) changes
- Added the possibility to enable an automatic firmware update check, giving a hint in the web interface, that a new firmware is available without the need to manually check it
- Some minor improvements to logging of SRT drops and connect in extended log

Changed functionality

- The individual gain adjustments per input source can now be done with a resolution of 0.01 dB steps
- “Load factory settings” from within the web interface does no longer require a reboot
- Loading settings via the web interface will no longer require a reboot of the device, but will be done very quickly (within a few seconds)

Fixed Issues

- After doing “Load Factory Settings” a newly configured encoder output might not start when activated directly via the Codec page (instead of using the Edit dialog)
- When doing “Load Factory Settings” some settings may not be back to default, but only after a reboot
- When uploading and activating a settings file some settings may not be activated correctly and may need a reboot
- Icecast streams with FLAC codec did not work
- Audio on headphone may be distorted if listening to an audio input with disabled SRC



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- Long Icecast URLs containing the ‘&’ character failed to be imported via the settings file
- The web interface can get unresponsive on the Overview page, when an input source description is longer than one line and changes its content regularly (e. g. in case of file playlists)
- DNS configuration per interface did not work via LCD menu
- Configuration changes done in the Codec dialogs may not get saved, when the dialog is left while the new (individual) Switch Criteria tab is selected
- Extended log entries for missed RTP packets may have empty stream names, if the packet was lost in a redundant (DualStreaming) or FEC stream
- Saving settings to USB may fail, if the USB stick is removed too quickly after the OK message is shown
- Fix visible name field in TS/Demux configuration when in “Service from list” mode
- TS Encoder: audio decoding may drift due to a problem with the PTS timestamps
- TS Encoder: fix a possible mismatch of TS stream type and audio descriptors in case of AAC
- Device could hang up when the audio file belonging to a file input source is not present (e. g. after loading a settings file)
- Fixed a high CPU load problem when many AES67 receive streams are configured
- MPEG TS decoder: enhanced audio decoding compatibility of RTP streams inside MPE
- Ember+ improvement for read-only items (in the virt subfolder)
- Some special characters in the name fields of the input sources etc. could mess up the display of the defined settings in the web interface
- Fix possible crashes with AES67 inputs (due to new extended logging)
- External clock configuration - a reconfiguration from an already configured and valid clock source (e.g. from PTP to PPS) was not handled correctly
- SNMP monitoring status was not reset to “disabled” if all input sources are getting disabled
- SNMP monitoring status / Warning LED / Relay were not reset (e. g. for the “Audio – No Input Data” alarm) if all input sources are getting disabled
- Alarm "No Input Data" did not check if PES/MPE data is present in case of TS/Demux input source
- Fixed "good" check for ES input source switch criteria "No input data" (T1 time was not considered correctly)
- Fixed "good" check for file input sources (were always bad since latest firmware when “No Input Data” switch criteria was activated)
- Fix AAC-ELD(v2) ES decoding not working when Decoder type is explicitly set accordingly
- File playback from playlist was broken since latest firmware 2.13
- VLAN interfaces were not correctly activated after a firmware update
- When updating the recovery bundle the wait page was shown endlessly
- FEC: Fix audio frames being unnecessarily recovered in case of 1x4 matrix
- Optional TS MPE Encoder: enhanced compatibility to certain 3rd party decoders

Known bugs

- xHE-AAC: Ancillary data and GPIO Forwarding does not work



IP-8e Release Notes

Version 2.13.4

27.07.2023

New Functionality

- Added Mono Downmix option to headphone output
- Added an option to disable the RTCP output in parallel to an RTP output stream
- Added a configurable silence mode to Livewire input sources for the encoders, inserting empty packets in case of missing Livewire stream input, thereby having continuous encoder output

Changed functionality

- Reported Livewire GPI/GPO states are now high in default state and low in triggered state
- SRC/DST name size increased for Livewire Routing Protocol (LWRP) from 15 to 25 characters

Fixed Issues

- Invert setting in GPO configuration had no influence on reported GPO state via Livewire
- HLS encoder: small bugfix for FLAC and ISOBMFF

Version 2.13.3

29.06.2023

Fixed Issues

- Fixed PCR accuracy problems on ASI output
- Sample frequency detection for AES/EBU audio input channels higher than 1 improved in case the sample rate converter at the audio input is switched off (resulting in audio distortions)
- RTP receiver information in case of dual streaming improved

Version 2.13.2

26.06.2023

Fixed Issues

- Fix possible crash when changing the sample rate of an audio input
- RIST receiver improved for low latency transmissions, avoiding sometimes quite aggressive retransmission requests
- RIST receiver improved for dual streams with encoder send delay
- Switching VLAN activation without modification doesn't work
- Control services are not correctly setup in VLAN environment
- When a sampling rate for an audio input in analog mode is set to a value different than 48kHz, the decoded audio is distorted



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- File input source from NFS source does not perform auto-reconnect on XML settings import

Version 2.13.1

08.06.2023

Fixed Issues

- Changing an active gateway address for a VLAN requires a reboot to become valid
- VLAN modifications could influence the main interface, too

Version 2.13

02.06.2023

New Functionality

- Added Mono Downmix option to headphone output
- The event log web page will now update automatically when new events arise while having the page open
- First steps to more advanced / extended logging:
For some events, which were previously only counted (like missed packets) there's now the possibility to check the point in time when these events occur. There's a new "Extended Log" tab on the Log web page, which shows these events. Currently we added events for RTP Rx start, RTP missed packets, RTP unrecovered packets, RTP Rx timeout and SRT connect/disconnect.
Like the counters this extended log is volatile, meaning it is cleared after a reboot. If syslog is enabled, these events will however also be sent out to syslog, making them externally persistent.
- Added optional support for NFS storage (for audio files; in addition to the internal storage)
- Added optional support for AWS S3 storage (for HLS Push encoding)
- Added interface (and VLAN) selection to the SNMP trap manager configuration
- HLS encoder: HLS container format is now configurable
- Added gain configuration to all input sources
- Added playlist support (m3u, m3u8, pls) to the file input source
- Added VLAN support to the Icecast client input source
- Major Livewire integration enhancements:
 - Added optional Livewire Sources for the audio inputs, providing its physical XLR audio inputs as Livewire audio streams.
 - The Livewire input sources for the Encoder section will now be a fixed number of Livewire Destinations (as much as encoders are available), thereby allowing the configuration via LWRP (corresponding to the fixed number of Livewire Destinations for the Audio Decoder section).
 - GPIO Tunneling has now the option to tunnel the Livewire GPO state instead of the physical GPI state.
 - Added Livewire level meters for Sources and Destinations (enhance compatibility to e. g. Pathfinder)
 - Added support for the Livewire GPIO snake mode, allowing to link the GPOs of the IP-8e to the GPIs of a different Livewire device.
 - Added status tab to Overview page for optional Livewire Sources



IP-8e Release Notes

- Added PIN lock option for LCD menu
- HLS encoder: added support for FLAC and Opus
- Added VLAN interface status

Changed functionality

- The Icecast Server did still answer with “ICY 200 OK” to a connection request and only for certain user agents / browsers with “HTTP/1.0 200 OK”. For some time now the ICY answer is however deprecated and should no longer be used, so we do now always answer to a connection request with the HTTP answer, thereby hopefully improving the general compatibility of the Icecast server to certain clients.
- Improved the file upload via the Storage page
The upload limit was increased to 100 MB and the error handling was improved.
- The “NTP Server Quality” parameter options are renamed from “Internet” and “Local” to “Logging” and “Clock Source” to better describe the usage and thereby the internal evaluation parameters used. With NTP expert settings enabled you can now also change the NTP quality settings “RMS Offset” and “Skew” (used as evaluation parameters).
NTP validation is now much faster on device startup or NTP activation.

Fixed Issues

- NFS handling improved
- HLS server push improved for S3 storage
- HLS encoder: Fixed issues with xHE audio on stream start with Safari and iTunes
- Fixed Livewire routing protocol LWRP not working after changing the IP address of the device
- RIST encoder retransmission improvements in case of SFN
- Fixed possible timestamp display problems in new Extended Log
- VLAN output streams will be now refreshed after config changes to VLAN parameters
- Changes to the headphone config via web interface were not applied
- The new “Auto Refresh” of the event and extended log web page sometimes didn’t work
- Livewire stream names with more than 15 characters could crash the system if the Livewire Routing Protocol LWRP is active
- SNMP get for virtual IP address nodes did always return 0.0.0.0
(e. g. virtCsllpcfgtempCtrlIp, OID 1.3.6.1.4.1.21529.1001.35.2.42.43.1)
- Encoder did not react on SRC on/off in audio input config (potentially changing input sample rate)
- RIST receiver improved for RTP streams, where the encoder has RIST disabled (could lead to audio buffer not building up)
- RIST didn’t work in dual streaming setup for the redundant line
- RTP: Overall packet lost counter could be too large in rare cases
- HLS encoder: fixed a problem with some HLS clients reporting faulty HLS segments (especially with the AAC codecs)
- Fixed compatibility issues when MM01 is the audio encoder and RTP packet fragmentation is activated via “RTP max payload”



IP-8e Release Notes

- MPEG/TS decoder: enabled the possibility to decode audio streams not announced via PAT/PMT automatically without the need to set the codec type manually
- MPEG TS decoder: fixed compatibility of private data TS decoding if the PID is not announced via PAT/PMT
- MPEG/TS decoder: enhanced audio decoding compatibility of RTP streams inside MPE
- Added support to transcode TS/Demux input sources with private data in Pipe mode
- Optional Live Listening feature did not work with Safari browser
- Fixed ancillary data descriptor in MPEG TS encoding for better compatibility
- A configured source IP for SSM (Source Specific Multicast) could not get deleted
- When switching between different web interface menu items the page will now always scroll back to top
- DTE baudrate changes were not applied immediately, but only after a reboot – Fixed
- Ancillary data was not shown on the Overview page for XLR audio input sources
- Fixed a possible crash with NTP synchronization and “Bind to interface” option enabled
- Fixed an issue with syslog messages stop working after a reboot
- Fixed a possible crash on the Overview page, when RTP with dual streaming, VLAN and multicast is enabled
- Fix GPIO Tunneling info not shown for encoder if no ancillary data source is selected
- Fix xHE-AAC problems with low bitrates
- Icecast input source handling improved for faulty meta data from some Icecast servers
- Improved Icecast client compatibility to “bursty” Icecast streams similar to HLS
- IGMP binding improved for RTP multicast (in case of interfaces with identical addresses)
- Livewire: increased general compatibility with Pathfinder
- AES67 input: fix problem with mono input streams
- Fixed MPEG TS signaling for MPEG2-AAC
- Improved transcoding of MPEG TS with ancillary data
- SAP: fixed crash with too many announced streams
- HLS/Icecast Encoder: Fixed a possible deadlock when TLS/SSL encryption is enabled

Known bugs

- xHE-AAC: Ancillary data and GPIO Forwarding does not work



IP-8e Release Notes

Version 2.11

12.09.2022

New Functionality

- Added VLAN support to the Livewire routing, advertisement and GPIO configuration
- Added support for Multichannel AES67/Ravenna input streams – it's possible to select the channel(s) to use from it
- Added support for Multichannel AES67/Ravenna input streams – it's possible to select the channel(s) to use from it
- Added support for FEC according to RFC 2733 (e. g. used by Digigram IQOYA)
- Added support for user authentication to Icecast client
- Added the additional possibility to configure a DNS server per IP interface (including VLAN)
- Added the possibility to upload MP4 and M4A files from within the web interface via Status / Storage
- Added "copyToClipboard" icon to IP status on the Overview page to copy the IP address
- In case the physical inputs are configured as Digital AES/EBU, the sample rate is shown in the level meter blocks at the top of the Overview page

Changed functionality

- SAP: the device is now listening to SAP announcements on all interfaces
- SAP announcements are now done on the interfaces, where the streams are sent (including VLANs)
- Ember+ can now be used simultaneously via more than one interface including VLAN(s). The configuration is done via Network Settings / Services. The switch to disable Ember+ completely is removed – instead disable Ember+ for all interfaces via Network Settings / Services.

Fixed Issues

- MPEG TS Encoder: Better PTS generation, thereby improving compatibility to certain decoder devices (e. g. Ateme DR5000 or SA D9846)
- Ancillary data decoding didn't work with Opus codec
- Switching off all TS/IP multiplexer outputs did switch off a possible active ASI output, too
- Audio silence detection may produce log entries named AES67
- NTP based synchronized playout was broken in version 2.10
- SAP: Enhanced compatibility to Dante Controller
- HE-AAC and MPEG-4 Audio Descriptor: the encoder will now signal level 2 if the sample rate is <= 24 kHz and level 3 above (according to ISO/IEC 14496-3:2019, Table 1.11)
- HLS encoder: decoder switched to mono mode with xHE HLS stream
- HLS encoder: AAC-LC (e.g. 192kbit/s) was not working with Safari as decoder
- Enhanced compatibility of the Icecast decoder to TCP-only streams (without Icecast information)
- Changing e. g. a file input source used by an encoder (changing the configured file) lets the encoder stop encoding



IP-8e Release Notes

- Web interface font rendering on non-Windows devices was suboptimal, showing some visual glitches – now it should be more or less equally on all devices

Known bugs

- n/a



IP-8e Release Notes

Version 2.10

17.06.2022

New Functionality

- Completely revised ancillary data handling for full flexibility
The fixed linkage between ancillary inputs and corresponding audio inputs is entirely removed. Instead a variable number of UDP ancillary inputs can be configured besides the available DTE inputs.
When configuring the Encoders on the Codec page, each configured audio input source can be accompanied by one of the ancillary input sources to be encoded together using the given profile. In that way the ancillary input sources are no longer limited to the physical XLR audio input sources, but can now be used together with any audio input source, e. g. Livewire, AES67 or Icecast. In addition there's a special "Pipe" ancillary input selection for the Encoder. Using the Pipe mode the ancillary data of the audio input source (including possible GPI Forwarding) will be preserved and transcoded together with the audio.
- Ancillary data and GPI Forwarding state available on Overview page
If ancillary data or GPI Forwarding information is contained in an input source, the Audio info blocks on the Overview page will allow to inspect that data. This is especially useful in a transcoding setup, where the ancillary data would otherwise not be available for examination.
- Optional audio PID removal from TS Multiplexer in case of audio input loss (to e. g. trigger external backup)
- Added new ancillary input alarms for the dedicated inputs
- Added ancillary data decoding support to Icecast input sources
- Added the possibility to bind the optional syslog output to a certain interface
- Added optional support for ASI input (available with the satellite tuner or standalone)
- Enhanced optional MPE encoding capability of TS Multiplexer to support RTP, too
- Added xHE-AAC support to optional HLS encoder
- In case external clock synchronization is active, the transcoding elementary stream output of an unsynchronized RTP stream used as encoder input source can now be synchronized to the external clock (activate the "Synchronous Playout" switch in the RTP elementary stream output settings)
- Ember+: added support for "subscribe" mode

Changed functionality

- RIST: the additional bandwidth used by the RIST encoder can be limited now
- Increased compatibility to not RFC2250 standard compliant MPEG audio streams (e. g. from Telos iPort)
- SAP is now also available with the Livewire license and no longer limited to the Ravenna license
- NTP is now enabled per default (with "pool.ntp.org")

Fixed Issues

- A memory leak caused a reboot of the device after a few hours up to a few days in case PTP clock synchronization was enabled
- TS/Demux: fetching the service list could sometimes fail to get all service names



IP-8e Release Notes

- When starting up in reduced channel count mode (to enable Analog input/output usage in case more than 4 channels are licensed), switching back to “Digital only” did not directly work and a reboot was required
- Livewire source select was wrong in Livewire input source configuration dialog
- Livewire was always able to set GPO without respecting switch source setting
- Changing the manual decoder config of a TS/Demux source did not trigger a reconfiguration
- Fixed a crash when changing the TS/IP input source used by a Demux source in MPE mode
- When only the port of an elementary stream output was changed, the SAP/SDP announcement wasn’t updated
- MPEG TS encoder: improved compatibility to certain decoders (PTS offset reduced)
- display problem with umlauts etc. in service name of TS/Demux sources in config list
- SAP information wasn’t updated on external clock configuration change
- ASI output was disturbed when only an encoder profile was changed
- The optional SIRC Data Channel was included in the TS only every 2nd time on TS Multiplexer changes
- MPEG TS decoder: further enhancements to private data decoding
- SNMP: SRT connectionState could return undefined value
- PHP error messages on TS/IP and TS/SRT input source tabs in case TS decoder license is missing
- Link to ASI configuration page was broken on TS Multiplexer page / Multiplexer Outputs / ASI Output

Known bugs

- n/a



IP-8e Release Notes

Version 2.08

07.02.2022

New Functionality

- Added syslog support (event log entries can be forwarded to up to three syslog servers)
- Added VLAN support to Icecast source client and Icecast server encoder output
- Added HTTPS support for Icecast input via proxy server
- Added optional MPE encoding capability to TS Multiplexer
- Added optional ADTS (MPEG-2) TS transport format support for HE-AAC and HE-AACv2

Changed functionality

- Increased possible maximum global delay to 10 seconds (only possible with NTP based synchronized playout)

Fixed Issues

- Fix problems with Icecast input via proxy server
- SPN/SFN/Stream4Sure did not work with AES67 as audio input, but only with the XLR hardware audio inputs
- SNMP/Ember+ status information for TS input sources improved
- SNMP: sysObjectID (.1.3.6.1.2.1.1.2) was always answered with .1.3.6.1.4.1.21529 (twowcom), no matter which device it is. It does now indicate the device in use.
- Added possibility to download the user manual from within the web interface (if the device has access to the internet)
- Improved "Check for updates" functionality from within the web interface

Known bugs

- n/a



IP-8e Release Notes

Version 2.07

28.12.2021

New Functionality

- New TS Demux configuration option via service selection
Instead of having to enter the audio PID manually, you can now switch to the new configuration mode “Service from list (fixed PID)”. It will show a list of all available unscrambled audio services in the chosen TS source and a sub selection of the audio PIDs belonging to that service (some services carry more than one audio PID). If the service list is not yet available (the TS source is not yet configured for any decoder or encoder) there’s a refresh button next to the service list, which will temporarily configure the chosen TS source (tune in case of SAT source) and collects the service information.
This new additional configuration option via service list is also added to the Private Data decoding.
Limitation: To not interrupt an already configured TS decoding, it will not be possible to get the service list of a SAT source, which is using the same RF input as the already configured and used TS source. So if you want to get the service lists of several SAT transponders (SAT/TS sources) you should fetch the service lists prior to configure one of that TS/Demux sources for one of the Decoders or Encoders.
- New TS Demux configuration option “Service from list (auto PID)”
In contrast to the above explained new configuration mode “Service from list (fixed PID)” the “auto PID” mode will also show a list of unscrambled audio services to select from by service name, but will not allow to configure/select the audio PID belonging to that service. The device will automatically choose the first available audio PID belonging to that service and will automatically follow any audio PID change that is announced via the PMT. That’s useful for services which change the audio PID for a certain time slot during the day, e. g. for a slot with a different language.
- New TS Demux private data configuration option via service selection
Corresponding to the new audio configuration via service selection it is now also possible to configure the private data PID via a selection from a list of available services carrying private data.
- Backup configuration for the external clock
It’s now possible to configure two backup external clock sources besides the main external clock source together with corresponding switch criterias. NTP is added as a possible source (besides 1PPS and PTP).
- Support for bit transparent transport of 32bit AES3 signals according to Ravenna AM824 / SMPTE ST 2110-31
- Added Ancillary Data timeout alarm (for both input and output)
- Added AES3 CRC error alarm
- Added TLS/SSL support for optional HLS encoding
- Added TLS/SSL support for Icecast client (allowing to receive streams via HTTPS)
- Added TLS/SSL support to Icecast server encoder output
- Added support for burst on connect to Icecast server encoder output
- Added proxy server configuration for Icecast input streams (on TCP/IP page)
- Added VLAN support to PTP interface configuration



IP-8e Release Notes

Changed functionality

- Increased the number of possible presets per input source type from 16 to 64
- For TS/Demux sources the name shown in the Source Assignment section and on the Overview is no longer the name of the TS source, but either the service name or the TS/Demux name (if in Manual mode)
- Rework of the TS service list on the Overview page
It will now show a lock symbol for scrambled services and we added symbols to better differentiate audio, video and other streams.
- When hovering over a TS/Demux source with the mouse cursor, the associated TS source is now highlighted (the other way around – hovering over a TS source and highlighting associated TS/Demux sources – did already work)
- Icecast server output streams now use the given name of the output settings as stream name (icy-name) instead of the fixed icy-name "2wcom Live Stream"
- The metadata (StreamTitle) used for the Icecast server output streams can now be set via the external APIs, e. g. SNMP (where it is the node virtCslAudioEncoderOutputsIcecastMetastr, OID .1.3.6.1.4.1.21529.1001.35.2.3.18.20.319.4096.1.5)
- Removed the need to enable ancillary data usage for XLR inputs (actually there's no need for it – the usage can still be enabled/disabled via the profile)

Fixed Issues

- Elementary Stream decoding in DualStreaming mode may exhibit audio errors even if the combined stream has no errors
- RIST decoder accuracy improved - sometimes RIST requested packets unnecessarily
- MPEG TS decoder: fixed several issues with not decodable Layer 2 audio streams and audio decoding not starting after configuration changes (needing Decoder Off/On cycle to start up)
- MPEG TS decoder: when changing the audio PID to decode of a TS/Demux source, the change will not get active directly (needing Decoder Off/On cycle to get active)
- MPEG TS decoder: changing the RF input of a SAT source was not applied instantly (needing Decoder Off/On cycle to get active)
- MPEG TS decoder: improved character table support for e. g. service names
- Fixed a possible crash when changing the configuration of a SAT/TS input source and afterwards a TS/Demux source, which is using that SAT/TS source
- When deleting a TS/Demux source, the corresponding TS source delete button was not re-enabled
- 'keep-alive' detection for optional HLS encoding
- PTP was not available as external clock source if the Livewire license is present, but the Ravenna license is missing
- Headphone output not working correctly with 44.1kHz audio
- Elementary Stream configuration dialog: The list of available streams via Ravenna/SAP is sometimes wrong
- Elementary Stream configuration dialog: When copying the settings of an available stream (announced via Ravenna/SAP), the Save button won't get active

Known bugs

- n/a



IP-8e Release Notes

Version 2.06

06.10.2021

New Functionality

- Added support for SNMPv3
- Added support for PTP Unicast
- Added optional HLS encoding capability
- Added Unrecovered counter to RIST info block for the Overview
- Added BER value to SAT Tuner info block for the Overview
- Rx (Receive) and Tx (Transmit) bitrates added to LCD status screen

Changed functionality

- Rework of the System Settings / User web page
- When the transponder input frequency method is used in the SAT configuration dialog, it's no longer necessary to configure the frequency range (Low Band / High Band)

Fixed Issues

- MPEG TS decoder: fixed a possible memory leak (in case of PIDs referenced by multiple programs), resulting in a crash of the device after some time
- AAC decoding in MPEG TS did impose an audio delay of 2 to 3 seconds
- In DualStreaming Elementary Stream setup, only the send delay of the first stream was shown on the Codec and the Overview page
- Increased disconnect/reconnect timeout for SRT connections in caller mode with no audio from 5 to 30 seconds
- DNS and Gateway entries in the TCP/IP settings couldn't get deleted by entering 0.0.0.0

Known bugs

- n/a



IP-8e Release Notes

Version 2.05

25.08.2021

New Functionality

- NTP synchronized playout
- Support for Differentiated Services DSCP tagging
- Switchable audio device mode
If you have more than 4 channels licensed, until now the device turned into a Digital only device, meaning the audio inputs and outputs can no longer be switched to Analog mode.
For such a device you now have the option to switch to Analog/Digital mode. You will thereby reduce the useable channel count to 4, allowing to configure only four audio outputs and eight encoders, but you will regain the possibility to switch the inputs and outputs to Analog mode.
- Configurable maximum payload size for Elementary Streams, increasing the number of packets per second (to e. g. reduce the latency implied by FEC usage)
- Rx (Receive) and Tx (Transmit) bitrates added to Overview page (per interface)
- Optional support for SIRC (Satellite In-Band Remote Control)
- MPEG TS encoder: selectable mode for private data insertion (ES - stream type 0x89 or UECP/RDS – stream type 0x80)
- RIST support added to SIP
- Configuration option for the stream ID added to SRT caller mode
- Configuration option for maximum reorder tolerance added to SRT input sources
- Web Login/Logout actions are now logged in the EventLog
- When the connection to the device is interrupted, a dialog will pop up

Changed functionality

- FEC encoder: setting the column port offset to 0 will now enable a row only FEC
- In RTP DualStreaming setup (without FEC), the Missed counter of the first stream was in fact the overall missed counter (after combining the two streams). This is corrected – that Missed counter will now reflect the missed packets of only the first stream. Instead there's a new block "Dual streaming" showing the Unrecovered/Missed counter of the combined streams.
- In RTP DualStreaming setup (without FEC), the PER (packet error rate) of the first stream was in fact the overall PER (after combining the two streams). This is corrected – that PER value will now reflect the PER of only the first stream. Instead the overall PER is now shown in the new "Dual streaming" block.
- In RTP setup with FEC, the Missed counter did not reflect the missed packets of the audio stream, but the number of packets not available after FEC correction (it was identical to the Unrecovered counter in the FEC block). This is corrected – the Missed counter will now reflect the missed packets of the audio stream (without FEC correction).
- In RTP setup with FEC, the PER did reflect the overall PER after FEC correction and not just the one of the audio stream. This is corrected – that PER value will now reflect the PER of only the audio stream. Instead the overall PER is now added to the "FEC" block.



IP-8e Release Notes

Fixed Issues

- PTPv2 support was accidentally broken since firmware V2.03 (seldom audio dropouts)
- Livewire Routing Protocol (LWRP) was broken since firmware V2.00
- Ancillary data decoding in MP3 did not work
- PCM 24 bit elementary streams with ancillary data and FEC could cause audio distortions
- A crash could happen when using an encoder send delay for RTP streams together with PCM or Opus encoded data, which has ancillary data and/or GPIO Forwarding active
- Automatic reconnect in SRT input source caller mode, when no audio packets are received for some time
- SRT bitrate gauge did not show the current bitrate, but a mean value over the whole connection period
- PER calculation failed for SRT connections (resulting in always being 0.0% PER)
- Jitter measurement for first RTP stream in DualStreaming setup was wrong
- Elementary stream decoding may fail to start up correctly when a higher jitter is present (the dejitter buffer level is 0 and the audio buffer will not build up).
- RIST info on Overview page was missing for TS/IP input sources
- RIST info on Overview page was wrong for input sources: the Requested counter did show the number of retransmitted packets and the Retransmitted counter was always 0
- SNMP node virtCslAudioDecoderInputsourcesRtpStatusFecunrecoveredcnt (OID .1.3.6.1.4.1.21529.1001.35.2.3.4.5.10.37.4096.1.23) was wrong and did not reflect the unrecoverable packets after FEC (but did reflect the number of internal FEC errors)
- MPEG TS decoder improved for high number of audio frames per PES
- MPEG TS encoder: SRT output didn't work if the entry was copied (only the first one did work)
- MPEG TS encoder: RTP output did not allow to select Ctrl interface, even if it was enabled for streaming via Services menu
- After pressing "Save" on the Ancillary Data configuration page, the embedding of ancillary data into a private TS PID did not work any more
- Decoder Streaming Input alarm shows yellow led, even if only Main source is configured and faulty (which should result in a red led)
- Distortions on headphone output
- The audio output may stuck completely in very rare cases, which could only be recovered by rebooting the device
- Ctrl interface was not selectable for ancillary output even if enabled via Services menu
- Ctrl interface was not selectable for TS Multiplexer outputs even if enabled via Services menu
- The table height of the event table on the Log page is now automatically the max. window height
- AES sync lost counter was incorrect

Known bugs

- With PTPv2 it's still possible, that a small buffer drift does occur (around 1-2 ms per day). Currently under investigation
- Disable and enable an encoder stream (which is using SFN or external clock activated for audio output) within 10 seconds could lead into audio errors on decoder. Fixed in 2.06, workaround: reboot decoder or disable encoder for at least 10 seconds and enable it again.



IP-8e Release Notes

Version 2.04

07.06.2021

New Functionality

- Finally added full (optional) Stream4Sure functionality (compatible to MM01 and MM08E)
- Added RIST (Reliable Internet Stream Transport, <https://www.rist.tv/>) support as an alternative to the already supported SRT protocol for a more standard based (RTP and SMPTE-2022) option to reliably transport your audio while being fully interoperable with standard RTP.
We currently support the RIST Simple Profile, which in contrast to SRT already supports multicast, but on the other hand does not support encryption like SRT does.
RIST will not require a new license, but will automatically be available if you have the SRT Encoder/Decoder license.
- Added VLAN support to the Ctrl interface

Changed functionality

- n/a

Fixed Issues

- MPEG TS encoder improved for input sources with a higher jitter
- MPEG TS encoder: External PCR will be signalled as "PCR" and not as "PES Priv. Data"
- MPEG TS encoder: improved compatibility with some decoders (TS stuffing bytes handling)
- Improved transcoding from MPEG TS to MPEG TS
- FEC encoder improved for RFC3640 and RFC4598, RFC4184 (marker bit, Dolby and MPEG4 AAC). The problem could manifest on decoder side as audio buffer running empty with AAC encoding and FEC enabled.
- The device could crash when switching the RF Input for a MPE satellite input source
- New NTP status page is OK now, even if the device has no Ravenna or SFN license
- When toggling the selected TS Multiplex on the Overview page, the output table did not update correspondingly
- For a newly added TS Multiplex, the "Auto-calculate required TS bit rate" toggle switch did not work (did not show the Bit rate input field in case "Auto-calculate" was switched off)

Known bugs

- n/a



IP-8e Release Notes

Version 2.03

10.05.2021

Important!

- This update recommends an updated recovery system. Please update the device using the file **recovery_2.00.upd**. A reboot of the device is not necessary.

New Functionality

- Ancillary Data input does support now input via UDP, too
- A double click on a row in the Profiles, Input Sources or Encoder Outputs table will open the corresponding Edit dialog
- New Ancillary Data status page
- New Services configuration, allowing to enable/disable HTTP, HTTPS, SFTP, SNMP and Streaming Data access/usage for each network interface (including VLANs).
This does also mean, that you can now grant access via HTTP(S) or SNMP via the Data interface – that's no longer limited to the Ctrl interface.
- The Ctrl interface can now also be used for Streaming Data like input source and encoder outputs (has to be enabled via the new Services configuration)
- New TCP/IP configuration page, allowing up to 10 VLANs for Data1 and Data2. Gateways can now be configured for VLANs, too.
- Added source specific multicast configuration to Elementary Streams dialog
- Added NTP status page

Changed functionality

- Ancillary Data handling
The Ancillary Data handling setup has changed. Formerly the configuration (should DTE input be used in conjunction with an audio input) was done as part of the Input Source configuration dialog(s).
To clean this up and to add UDP support, the configuration for the Ancillary Data handling was moved to a new Ancillary Data menu item.
- New NTP implementation (allowing up to four servers to be configured), giving higher precision
- Several packages have been updated (see OSA – Open Source Acknowledgment – on the Global page) to fix e. g. security issues

Fixed Issues

- FEC Encoder now supports the maximum of 64 elementary output streams
- Encoder Outputs: it was not possible to have more than 16 encoder outputs for the same encoder (now all 64 encoder outputs can have the same encoder source)
- MPEG TS Encoder: Bit rate overhead reduced
- MPEG TS Encoder improved for ancillary data in private PID
- MPEG TS Decoder crash fixed on special EIT tables and large SDT tables



IP-8e Release Notes

- Deleting all payloads of a service did not work - after page reload one service is kept
- The device could crash when starting or stopping RTP output streams
- Removed obsolete MPEG TS Packetizing format from Multiplexer settings
- Wrong analogue level display in web interface in cases where the device is digital only
- Added missing interface and VLAN info to input sources and encoder output on Overview page
- Inactive SAVE button in the Livewire+ dialog after changing only the LW source

Known bugs

- The new NTP status page does only work, if the device has the Ravenna or SFN license



IP-8e Release Notes

Version 2.01

10.03.2021

New Functionality

- Icecast Server Encoder Output added
In addition to the Icecast Source Client Encoder Output it's now also possible to enable an Icecast Server as an encoder output. Each Icecast Server will allow a maximum of 5 client connections.
- DSCP configuration for Elementary Stream Encoder Output added

Changed functionality

- The configuration for the usage of an external clock source (1PPS, PTP) has been changed. In former firmware versions the usage of the external clock source had to be enabled for the specific input sources. As this led to some confusion, the configuration has been re-designed.
The usage of the internal or external clock source is now configured on the Audio/XLR page, as the clock source is in fact a characteristic of the audio inputs.

Fixed Issues

- PSU alarms were not available (since firmware V2.00)
- Fixed a crash when changing the interface of an active SRT input source
- After device boot/reboot, some input sources may not start up correctly, when DHCP is used on the Data interfaces. The input source itself will work, but no audio is decoded / played out
- SRT input sources will now (again) be able to follow an SRT encoder, which does encode a file (this was broken in firmware V2.00)
- MPEG TS Encoder: PCR insertion improved for streams in "Low bitrate overhead" mode
- The Jitter information for the redundant part of an RTP output stream was wrong
- Improved Livewire compatibility (e. g. to Omnia One)
- The "Reset Counters" button on the Overview page did not reset the RTP output statistics for a redundant stream
- The "Reset Counters" button on the Overview page did sometimes not reset the FEC decoder status



IP-8e Release Notes

Version 2.00

09.02.2021

New Functionality

- Reference Input: the audio clock control does now allow to configure one of the audio inputs to act as the reference input (clock source) for the outputs (or other inputs)
- Added missing separate status information (via SNMP/Ember+) for all input sources types and all encoder output types
- Configuration of VLAN added for Livewire input sources
- Configuration of the dejitter buffer level added for Livewire input sources
- Added No Input Data alarm for streaming inputs
- SDP files can be downloaded for the output streams
- MPEG TS Encoder: allow referencing an external PCR (earlier firmware versions always used one of the audio streams for the PCR)

Fixed Issues

- PSU alarms are not available, if the device is equipped with hot swappable power supplies
- The (SAT)TS-Bitrate was no longer valid, but always displayed as 0
- TS Sync and C/N switch criteria for SAT(TS) input sources did not work
- IP input source status information was accessible using the decoder index (audio output index / decoder rank) via SNMP/Ember+. Therefore it was not possible to get IP status information, if the input source is only used for an encoder.
Now the index to use is the virtual input source index, following the visual order in the web
- Ancillary data output stopped working after switching to a backup source and didn't recover even if the main source is available again, but all input sources had to be disabled and enabled to get it working again
- DTE Ancillary input data is not routed to DTE output, if corresponding Audio Input is used as input source in the Decoder (Audio Output)
- SRT decoder was not restarted if the configured profile to use has been changed
- Activating NTP time synchronization does cause an audio glitch at synchronization time when using SRT
- Icecast Source Client encoder output did not allow to configure ports below 1024, but sometimes port 80 is used, so now every port starting with 1 is allowed
- Icecast Source Client did not work with AAC codecs
- MPEG TS Encoder: Optimized IP packet size for low bitrate
- SNMP Trap status binding delivers wrong value
- In rare cases the device may crash when changing the input source

Changed behavior

- SNMP Traps: ATTENTION – the old SNMP traps are no longer available!
We made a move from device specific traps to general traps if applicable.
Example:
notificationVirtIp4cDeviceStatusMonitoringSilenceaudio1in (OID .1.3.6.1.4.1.21529.1001.0.7)



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is now available as

notificationVirtCslMonitoringStatusAudioInputsXlrsilencedetection
(OID .1.3.6.1.4.1.21529.1001.0.1098907649)

Please note however, that in contrast to the traps the old status information, which can be polled, is still available, e.g. virtIp4cDeviceStatusMonitoringSilenceaudio1in (OID .1.3.6.1.4.1.21529.1001.35.1.23.37.57.7).

This old information is however marked as Obsolete in the SNMP MIB, as in the future only the new status information will be available

(e.g. virtCslMonitoringStatusAudioInputsXlrsilencedetection, OID .1.3.6.1.4.1.21529.1001.35.2.57.37.3.313.4096.1.1)

Known bugs

- None



IP-8e Release Notes

Version 1.31.4

30.11.2020

New Functionality

- None

Fixed Issues

- Fixed a routing problem when Ctrl and Data 1 or Data 2 were in the same subnet. This issue could result in the traffic being sent over the gateway instead of the subnet. Depending on the network setup this could cause unreachable devices.

Known bugs

- PTP synchronization incorrect after reset (turn off, turn on)
Workaround: Reboot the device after configuring your PTP setup
- When using SAP all multicast parameters are set correctly, but the audio parameters are not used from the session description.
Workaround: setup the audio parameters manually when you configure the reception of a SAP announced stream.



IP-8e Release Notes

Version 1.31.3

24.11.2020

New Functionality

- None

Fixed Issues

- Level meter did not work correctly, showing wrong (variant) values for a fixed level signal
- Wrong LCD display, if 5 channels are activated

Known bugs

- None



IP-8e Release Notes

Version 1.31.2

05.11.2020

New Functionality

- None

Fixed Issues

- Device not reachable because of wrong or missing gateway
- IP VLAN changes are not updated after configuration - only after reboot
- Redundancy stream handling in SFN mode improved
- Headphone output did not work in SFN mode
- Enhanced compatibility to some special Icecast streams

Known bugs

- None



IP-8e Release Notes

Version 1.31.1

05.11.2020

New Functionality

- None

Fixed Issues

- Log page shows error message about wrong localtime
- Device not reachable because of missing gateway. Has been fixed by changing DHCP metrics.
- Ravenna compatibility problems (e.g. to Lawo devices)

Known bugs

- None



IP-8e Release Notes

Version 1.31

28.10.2020

New Functionality

- Web Interface: Profiles are now moved to a separate tab, separated from the Input Sources

Fixed Issues

- Audio output may stop after several days or weeks (although input source(s) are still running)
- A network configuration using different gateways for the different interfaces (CTRL, DATA1, DATA2) did not work properly (policy based routing)
- Web interface is not available in seldom cases after a reboot (if DHCP is used for the CTRL interface)
- PSU failure alarm may produce false positive alarms
- "AES/EBU No signal" alarm did not work
- TS Decoder: audio decoding stops after changing demux configuration in MPTS setup
- SIP: Re-dial active connection after reboot if reconnect is enabled
- SFN mode: output disturbed in DualStreaming setup, if only redundant stream is received
- Web interface / Codec: VLAN may be set to 0 instead of right value in edit dialog
- VLAN may get set unintentionally, if interface is changed from one with VLAN to one without VLAN
- Web Interface: Overview page is empty (more or less), when logged in as user
- TS Encoder: enhanced ATSC compliance

Known bugs

- None



IP-8e Release Notes

Version 1.30

15.09.2020

New Functionality

- TS Encoder: added configuration possibility for the encoding standard (DVB/ATSC) and language codes
- On the TS Multiplexer web page, changes done are monitored now, too and the Save buttons will only get enabled when changes are present. Faulty values are not allowed to be saved.
- Added configuration possibility (on Audio XLR page) for the critical level marker threshold for the level meters on the overview page (threshold/level, at which the level will be shown as orange instead of green)
- PTPv2 support is now fully functional (provided the Ravenna license is present)
- Added support for the optional FM Dualtuner module

Fixed Issues

- SAP service behavior on DHCP changes improved; sometime SAP was not available after changing the network interface to DHCP
- RTP jitter buffer performance and accuracy improved PSU failure alarm may produce false positive alarms
- Problems with MP3 VBR decoding
- Wrong RTP timestamps for G.722 encoding
- Ancillary data output not working in case of SRT input source
- Ancillary data not working with E-aPTX and 32 kHz sample rate
- DHCP leases and TCP/IP reconfiguration improved

Known bugs

- none