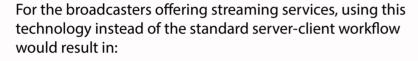
AETA partners with the French company Ace Media Tools to globally help on Ines software suite consisting of:

- + Ines VTS for the management of hardware codecs as well as its own software codecs
- + Ines RTC for establishing spot RTC links
- + Ines Solo for being able to establish a telecom link and record it









- + No saturation of the streaming severs during large events
- + Cost savings globally
 - · At CDN provider services
 - Investment costs of the infrastructure
- Have a greener footprint with a possibility to measure the saving that has been done
- Streaming more data with the same infrastructure
- Less power consumption at server level
- Less hardware and space consumption



FACTORY SIP

- + Redunded server with back-up process
- + Total availability with 24/7 moni
- + TÜV certified, ISO27001 datacenter
- + Included in every AETA codec

SIP PREMIUM

- Portability
- + Incoming and outgoing call blocks
- Speed and group dialling
- + Self management of passwords
- Quarterly call report









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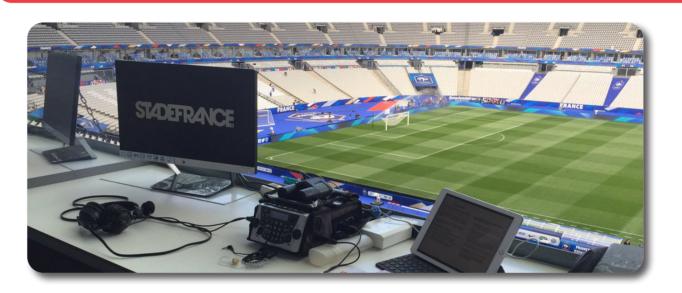
AFTER SALES SERVICE sav@aeta-audio.com

www.aeta-audio.com



CATALOG

Live Reports Codecs - Rackmount Codecs - Software - Services



SIMPLE AND SAFE AUDIO LINKS

All AETA codecs offer a large panel of great functionalities.

SOLUTIONS 2



- + Rack mount codec
- + Software codec
- SIP server
- Remote Access
- Codec management system
- OTT streaming solution

MCR Software Suite

SIMPLICITY ,



- Auto jitter buffer
- + Remote Acess: Remote control via internet
- + Embedded HTML server for easy updates and access to all parameters
- + SIP or Direct RTP
- Factory SIP accounts included
- + Simultaneous registration on 2 SIP servers
- + Auto redial
- + Memorizable presets
- + GPIO
- + Full Duplex
- + Program & coordination
- + N/ACIP compliant (EBU Tech 3326)
- Best in class audio
- + AES67 RAVENNA or DANTE

SAFETY



- + Auto bit rate OPUS
- + Double streaming and bonding FEC: Safe connection with packet duplication
- + Adjustable OPUS bitrate (12-256 kbit/s) during communication to fit available bandwidth, without audio glitch
- + State of the art wireless connection
- + Internal 4G and 5G modules
- + Fall back in HDVoice ™ mode, VoLTE
- + SNMP
- Redundant power supply
- + Redundant AETA SIP server

Field units

SCOOPTEAM



Commentary codec

- + Dual codec AoIP
- + Up to 7 differents networks
- + 2x Ethernet
- 2x Integrated 3G/4G-LTE/5G
- + Mobile voice over 3G with HD Voice™ (7 kHz) and VoLTE
- + ISDN up to 2B
- + 2x AES67 interface RAVENNA or DANTE
- + Wi-Fi by USB
- Additional 4G by USB

STEREO



- + Up to 4 commentary positions
- + Up to 11 audio inputs
- 2 auxiliary line inputs, 1 AES input
- 7" touchscreen display
- Flexible monitoring, routing and volume

SCOOPYFLEX



Live reports codec

- + Recording / editing features
- + 2x Ethernet
- + Up to 7 different networks
- + 2x Integrated 3G/4G-LTE/5G
- Mobile voice with HD Voice™ (7 kHz) and VoLTE
- Wi-Fi integrated





- + Up to 4 balanced mic/line inputs
- + 2 balanced line outputs
- + Up to 4 headphones outputs
- + 2x AES 67 (RAVENNA or DANTE)
- USB audio device interface (stereo)
- DC or battery > 5H autonomy with Liion battery, integrated charger
- + 1× AES 3 imput

SCOOPFONE 56 SCOOPFONE 46



Professional mobile phone

- + AoIP mobile over 3G/4G-LTE/5G
- AoIP over wired Ethernet
- Mobile voice over 3G with HD Voice[™] (7 kHz) and VoLTE





- + 1 balanced mic/line input 1 balanced line inputs - 2 headphone outputs
- + Easy update by USB and HTML page
- + 2 GPIO
- + > 5H autonomy with Rechargeable NiMH batteries, 6xAA, integrated

SCOOPFONE [10]







Mobile voice over 2G/3G with HD Voice™ (7 kHz)

SCOOPFONE P





MONO

 $\sqrt{x_1}$ $\sqrt{x_1}$



- AoIP over wired Ethernet only
- Rackable version available

eScoopfone



Smartphone codec application for broadcasters

- AoIP via 3G-4G
- OPUS with adjustable bit rate +
- SIP connectivity
- SIP status indicator
- Network quality meters
- Balance Tx / Rx
- Headset and mic gain
- Bargraphs headset and mic
- GPIO

MULTI SCOOP



Audio channels X16

Full Redundancy:

- Power, control and AES67 modules
- Separated audio and control networks for

Modular multi-channel codec with IP. ISDN and AES67

STEREO

Up to 16 audio channels in 3U

Studio units

Flexible audio interfaces (up to 16 in 3U): analog, AES3, AES67

Hot-swap modules (sold separatley):

- Up to 8 IP / IP & ISDN modules (with double mono codec)
- Up to 2 AES 67 modules RAVENNA/ DANTE
- Up to 2 Control modules

Easy supervision

- + Full control from control modules with separated networks
- Supplied with ScoopManager for simplified codecs management
- Embedded HTML server (LAN) for each modules
- SNMP



- + AES 67 (RAVENNA or DANTE)
- AoIP mobile over 3G/4G-LTE/5G
- Mobile voice with HD Voice™
- Audio channels X2

STEREO

STEREO

MONO

MONO

- + x2 Ethernet

Rackmount studio codec in 1/2 RU

- (7 kHz) and VoLTE
- Balanced analogue inputs and outputs
- Digital audio inputs/outputs AES/EBU
- + 2 GPIO
- + Redundant power supply (230V +12 V DC)





Point to point AoIP codec

- + 2 versions: analog or digital I/O AoIP over wired Ethernet
- Audio channels X2
- Stereo analog or digital outputs (XLR 3
- pins) 2 GPIO
- + Compact design (1/3 of 19")
- Low power consumption
- + Power over Ethernet (48V)



Rackmount version

- Professional wireless phone in rack
- Up to 3 units in a 1U 19" rack
- Supports HD Voice and data service
- **OLED** display

- + Professional audio interfaces
- + Audio processing:
- high-pass filters, selectable gain Remote control through embedded HTML web page (LAN)



Rackmount version

- + Two versions: Ethernet or Relays (GPIO)
- Professional wireless phone in rack mount
- 3 units in a 1U
- Supports HD Voice and GSM service
- **OLED** display

- Professional audio interfaces
- Audio processing: high-pass filters, selec-
- Remote control through embedded HTML web page (LAN)
- + DTMF send (to IVR servers)

