AETA partners with the French company Ace Media Tools to globally help on lnes 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







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



REMOTE ACCESS

+ Access the HTML webpage of the unit and all sethings from anywhere

REMOTE ACCESS PORTAL

This HTML tool allows you to:

- + Have global visibility over all your codecs
- + See straight away the status of each codec
- + Filter out sub-groups
- + Identify codecs that need to be updated





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CATALOG



SIMPLE AND SAFE AUDIO LINKS

All AETA codecs offer a large panel of great functionalities.



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

+



Live Reports Codecs - Rackmount Codecs - Software - Services

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



- + Auto bit rate OPUS
- Double streaming and bon-+ ding 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 +

SCOOPTEAM



SCOOPYFLEX



SCOOPFONE 56 SCOOPFONE 4G

















Field units

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

Live reports codec

+ 2x Ethernet

and VoLTE

Wi-Fi integrated

+ Recording / editing features

+ Up to 7 different networks

+ 2x Integrated 3G/4G-LTE/5G

Professional mobile phone

+ AoIP mobile over 3G/4G-LTE/5G

+ Mobile voice with HD Voice[™] (7 kHz)



- + Up to 4 commentary positions
- Up to 11 audio inputs
- 2 auxiliary line inputs, 1 AES input

Up to 4 balanced mic/line inputs

USB audio device interface (stereo)

ion battery, integrated charger

DC or battery > 5H autonomy with Li-

2 balanced line outputs

+ Up to 4 headphones outputs

+ 2x AES 67 (RAVENNA or DANTE)

7" touchscreen display

STEREO

+ 1× AES 3 imput

Flexible monitoring, routing and volume control





X4

1 X2



MULTI SCOOP



Audio channels X16

Full Redundancy:

- + Power, control and AES67 modules
- Separated audio and control networks for more safety











+ AoIP via 3G-4G Balance Tx / Rx + OPUS with adjustable bit rate +

Headset and mic gain

Ethernel 1

- Bargraphs headset and mic
- GPIO
- SIP status indicator + Network quality meters

SIP connectivity

+

MONO

AoIP over wired

Rackable version available

Ethernet only

- J X1 1 X1





- 1 balanced mic/line input 1 balanced line inputs - 2 headphone outputs
- + Easy update by USB and HTML page
- + > 5H autonomy with Rechargeable
- NiMH batteries, 6xAA, integrated
- + 2 GPIO
- AoIP over wired Ethernet
- Mobile voice over 3G with HD Voice[™] (7 kHz) and VoLTE charger MONO MONO



Modular multi-channel codec with IP, ISDN and AES67





Up to 16 audio channels in 3U

+ 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
- EmbeddedHTMLserver(LAN)foreach modules
- SNMP



Rackmount studio codec in 1/2 RU

- + x2 Ethernet
- + AES 67 (RAVENNA or DANTE)
- AoIP mobile over 3G/4G-LTE/5G
- Mobile voice with HD Voice™ (7 kHz) and VoLTE
- Audio channels X2

+ 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

- Stereo analog or digital outputs (XLR 3 pins)
- + 2 versions: analog or digital I/O + AoIP over wired Ethernet
- Audio channels X2
- 2 GPIO
- Compact design (1/3 of 19")
- Low power consumption
- Power over Ethernet (48V)
- + Professional wireless phone in rack
- Up to 3 units in a 1U 19" rack
- Supports HD Voice and data service
- Professional audio interfaces Audio processing: high-pass filters, selectable gain
- Remote control through embedded HTML web page (LAN)



MONO

- + 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, selectable gain
- Remote control through embedded HTML web page (LAN)
- DTMF send (to IVR servers) +





STEREO









