## 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 record it







For the broadcasters offering streaming services, using this technology instead of the standard server-client workflow would result in:

- 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
  - 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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ventes@aeta-audio.com

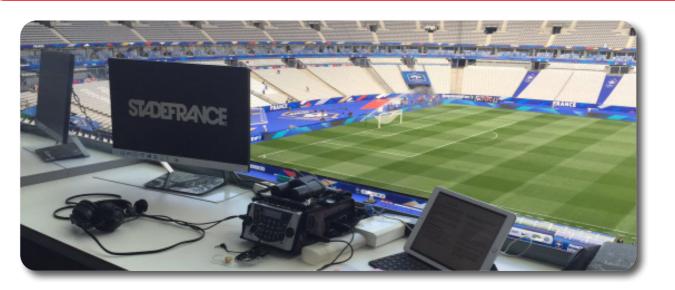
**AFTER SALES SERVICE** sav@aeta-audio.com

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.



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

# SIMPLICITY /::



- + Remote Access: 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**



- 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
- 2 years warranty

# **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





- 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 control

# X2 STEREO





# SCOOPYFLEX



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

- 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

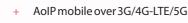
# SCOOPFONE P











- AoIP over wired Ethernet
- Mobile voice over 3G with HD Voice<sup>™</sup> (7 kHz) and VoLTE

## + 1 balanced mic/line input - 1 balanced line inputs - 2 headphone outputs

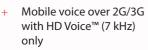
- + Easy update by USB and HTML page
- > 5H autonomy with Rechargeable NiMH batteries, 6xAA, integrated charger

# SCOOPFONE [10]









# SCOOPFONE P







- AoIP over wired Ethernet only
- Rackable version available

# **SCOOPFONE**



## 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

# **Studio units**

# MULTI SCOOP



### Audio channels X16

### Full Redundancy:

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

## 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

Up to 8 IP / IP & ISDN modules (with

### Hot-swap modules (sold separatley):

- double mono codec)
- Up to 2 AES 67 modules RAVENNA/
- Up to 2 Control modules

## Easy supervision

- Full control from control modules with separated networks
- Supplied with ScoopManager for simplified codecs management

**STEREO** 

- EmbeddedHTMLserver(LAN)foreach modules
- SNMP



- + AES 67 (RAVENNA or DANTE)

Rackmount studio codec in 1/2 RU

- Mobile voice with HD Voice™ (7 kHz) and VoLTE

### Audio channels X2

## **STEREO**

- + x2 Ethernet
- AoIP mobile over 3G/4G-LTE/5G

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

**STEREO** 

MONO

MONO

# 



- + 2 versions: analog or digital I/O
- AoIP over wired Ethernet

### Audio channels X2

- Stereo analog or digital outputs (XLR 3 pins)
- Compact design (1/3 of 19")
- Low power consumption
- + Power over Ethernet (48V)

## SCOOPFONE 56 SCOOPFONE 46 SCOOPFONE P

## 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
- high-pass filters, selectable gain

Audio processing:

Remote control through embedded HTML web page (LAN)



## Rackmount version

- Two versions: Ethernet or Relays (GPIO)
- 3 units in a 1U
- Supports HD Voice and GSM service

Professional wireless phone in rack mount

**OLED** display

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

