



Advanced commentary codec 4 positions with AoIP, ISDN, 3G/4G and Ravenna AES67





















# **RAVENNA** AES67 built-in

- + AoIP over wired Ethernet
- + AoIP mobile over 3G/4G-LTE Optional
- + ISDN 2B Optional
- + Ravenna interface AES 67 built-in Dante compatible - Optional
- + Mobile by USB Optional
- + Wi-Fi by USB Optional
- + Double mono codec
- + N/ACIP compliant (UER Tech 3326)
- + High grade pre-amplification
- + Up to 4 commentary positions
- + Up to 11 audio inputs
- + Flexible monitoring, routing and volume control
- + 2 auxiliary line inputs, 1 AES input Optional

- + Color TFT 7" touchscreen display
- + Dedicated and assignable backlit pushbuttons and potentiometers
- + USB audio device interface (mono)
- + Remote control via IP for all settinas
- + Remote access over Internet

Ideal for TV and Radio sports reporting, ScoopTeam is a commentary unit integrating a codec for live transmission of one or two audio signals in high audio quality (up to 20 kHz bandwidth).

The **two audio channels** can be set with a **single destination** (transmission in "dual mono" format), or two separate destinations (double codec).

ScoopTeam mainly targets IP networks, but it can also use an ISDN line. Alternative transmission modes are available as options: AoIP over 3G/4G, Wi-Fi by USB, and mobile voice (including HD Voice capability).

Moreover, the premium version allow you to transmit via a Ravenna Interface (AES67, Dante compatible) and offers two additional analogs inputs and one AES input.

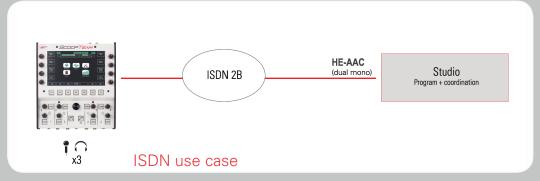
ScoopTeam features many coding algorithms. Their availability depends on the configuration for transmission: network used and single/double codec mode. (See inside for details)

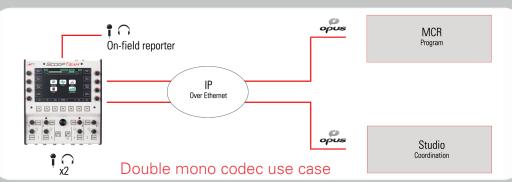
Stereo algorithms (MPEG Layer 2, AAC, Opus, Linear) allow you to use ScoopTeam to transmit a stereo stream by using the coordination bus for mixing the right channel.

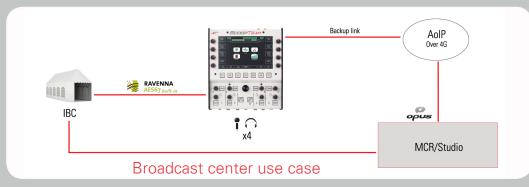
The interface is as simple to use as possible, with assignable buttons and knobs for settings and volume adjustments.











Network/Algorithm	AoIP	ISDN	Mobile - HD Voice	
G711	Mono	Mono	-	
GSM, AMR	-	-	Mono	
AMR-WB / G722.2	-	-	Mono	
G722 SRT, H221, H242	Mono	Mono	-	
MPEG Audio Layer II	Mono, Dual mono, Stéréo	Mono, Dual mono, Stéréo	-	
MPEG AAC-LC, HE-AAC	Mono, Dual mono	Mono, Dual mono, Stéréo	-	
OPUS	Mono, Stéréo	-	-	
Linear L16, L20, L24	Mono, Dual mono	-	-	

## **AUDIO INTERFACES**

- + 2 microphone inputs with phantom power, HP filter, limiter
- + 1 mic/line input with phantom power, HP filter, limiter
- + 1 mic/line input with HP filter, limiter
- + 2 auxiliary line inputs, 1 AES input Optional
- + 4 stereo headphone sockets for monitoring, flexible routing and volume control
- + 1 line output
- + USB audio device interface (mono)
- + Ravenna interface (2 input channels / 2 output channels) Optional

## USFR INTERFACE

- + Color TFT 7" touchscreen display
- + User input: touchscreen display + rotary encoder
- + Dedicated and assignable backlit pushbuttons
- + Dedicated and assignable potentiometers
- + Programmable set-up/dial memories

## REMOTE CONTROL AND SUPERVISION

- + Remote control via IP for all settings
- + Remote access over Internet
- + 2 Ethernet interfaces 10BaseT / 100BaseT (shared with audio transmission)
- + GPIO: 2 inputs / 2 outputs

# ACCESSORIES (not included)

- + Flight case
- + Car DC adapter cable
- + Spare AC/DC adapter
- + Ni-MH battery set
- + Multiband antennas

# TRANSMISSION INTERFACES

#### Ethernet/IP transmission interface

- 2 Ethernet interfaces, 10BaseT / 100BaseT (shared with remote control)
- + Audio transmission (AoIP) in unicast mode: SIP or "Direct RTP"
- + Net bit rate 12 to 256 kbit/s (depending on coding algorithm, linear coding excluded)

#### ISDN transmission interface - Optional

- + One S0 (BRI) interface. Protocols: ETSI, NI-1, NTT
- Single codec 64 or 128 kbit/s, or double 64 kbit/s codec
- + 5AS auto configuration on incoming calls

#### Mobile network access - Optional

- + Integrated 2G/3G/3G+/LTE network access module, 2 antenna sockets
- Voice mode: standard telephone or "HD Voice" (7 kHz with AMR-WB)
- Packet data mode: IP protocol, SIP or "Direct RTP", net bit rate 12 to 256 kbit/s (depending on coding algorithm)
- + External 3G/LTE module connection via USB socket (data mode only) (optional)

### **GENERAL**

- Power: external DC 12V, or 6xAA-type rechargeable batteries (run time > 1 h), charger included
- + Size 212x236x100 mm (WxDxH)
- + Weight < 2,2 kg
- USB host interfaces for optional extensions: Wi-Fi
- + Operating temperature range 0°- 45°C

# **VERSIONS**

Feature/ Version	IP	ISDN	Mobile by USB	Wifi by USB	Built-in Mo- bile (3G/4G- LTE)	Ravenna (AES 67 built- in)	2 Aux. line inputs + 1 AES input
Standard	Yes	Opt.	Opt.	Opt.	-	-	-
Premium	Yes	Opt.	Opt.	Opt.	Opt.	Opt.	Yes