

SIMPLE AND SAFE AUDIO LINKS

ScoopTeam Special

- Products
- Services
- Software

SCOOP TEAM

The next-generation commentary unit for live sports broadcasts



 X 4
  X 4
 STEREO



2x Ethernet
2x 4G/LTE
Worldwide modules built in



ISDN / 3G

- 4x Microphone positions (2x Mic/Line)
- Color TFT 7" touchscreen display
- Double mono codec over IP
- Flexible routing, monitoring and volume control



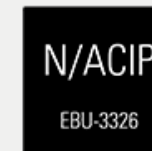
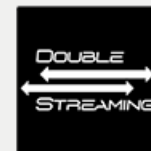
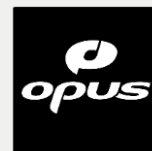
- Products
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SCOOP TEAM

General description

Commentary – Codec – Audio routing device

- As communication device within audio networks (e.g. AES67) located in OB-vans, commentary booths, remote locations, etc.
- Codec for direct transmission of audio to the MCR
 - E.g. Opus, HE-AAC, AAC-LC, MPEG-L2, G.7xx, PCM
- Program and talkback or double mono program
- Audio routing of all inputs and individual headphone mixes
- SIP or Direct RTP
- Local or remote control over the internet (cloud based)
- Plug & Play operation or individual configuration



- Products
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SCOOP TEAM

Input / Output

- 4x Mic.positions (2xMic/line)
 - 2x XLR-3 analogue audio
 - 1x Line out
 - 1x AES-3
 - 2x USB
 - 1x USB Audio
 - 1x GPI/O's
 - 2x Audio Network Ports
- Routing options for all inputs
- AES67 compatible:
- 8 channel Ravenna or
 - 8 channel Dante
 - As communication / audio routing device with talkback within audio networks (in OB Vans, Commentary booths, etc.)
 - or direct transmission of the program to the MCR with talkback



Image shows Premium+ configuration



RAVENNA
AES67 built-in



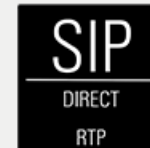
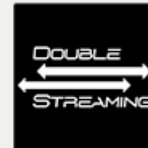
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SCOOP TEAM

Link Modes, Communication, etc.



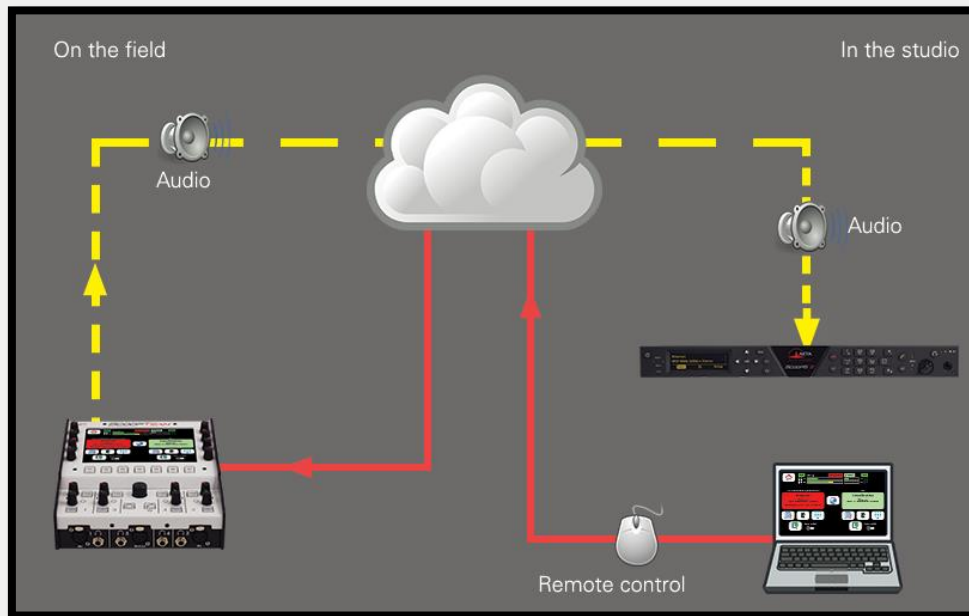
- Auto Bit Rate Opus (Bit rate adjustment on the fly & automatically)
- Auto Jitter Buffer
- Bonding (over two interfaces)
- Double Streaming (to one SIP destination or IP address)
- Double Codec (to two SIP destinations or IP addresses)
- SIP or Direct RTP
- Registration on two different SIP servers
- Forward Error Correction and packet duplication modes
- **N/ACIP compliant, connects to any codec brand**



- Products
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SCOOP TEAM

Modes of operation



- The embedded HTML server can be controlled locally and remotely
- « REMOTE ACCESS » over the internet for full control & monitoring
- RTCP data monitoring (Link quality monitoring)
- Dedicated PC application « myScoopTeam »
- « Remote Access Portal » – cloud based virtual MCR



Embedded
HTML server



CALL PROFILES 2

TRANSTEL ETH
TRANSTEL4G

Deselect

MESSAGES 2

To Local Unit:

Remaining Characters: 320

Send

CONNECTION STATISTICS

	Transmit	Receive
Quality:	98%	98%
Duration:	00:00:28	00:00:27
Total Bytes:	449879	433012
Packets - Total:	1401	1349
Packets - Lost:	0	0
Packets - Discarded:		0
Packets - Duplicated:		0
Packets - Reordered:		0
Jitter - Average:	1 ms	1 ms
Jitter - Max:	2 ms	2 ms
Buffer - Latency:		140 ms

CONNECTION STATE

Status: **Established**
 Network: Ethernet
 Remote Number: 15000@sip.aeta-audio.com:5070
 Coding Algorithm (Tx): OPUS 48kHz 128kbit/s Mono
 Coding Algorithm (Rx): OPUS 48kHz 128kbit/s Mono

Load

Release

CONNECTION PARAMETERS

Bit Rate: 128 kBit/s

Send

CONNECTION STATE (CODEC 2)

Status: **Idle**
 Network: Ethernet
 Remote Number:
 Coding Algorithm (Tx): OPUS 48kHz 64kbit/s Mono
 Coding Algorithm (Rx): OPUS 48kHz 64kbit/s Mono

Load

Dial

Connections

CALL PROFILES

You can select one of the call profiles and click the "Dial" button to set up a connection.


[Manage Profiles.](#)

MESSAGES

With the Info Box field, you can read the last text message that was displayed on the unit. You can also use it to send a note to the display of the unit.


Image shows HTML page example

Image shows HTML page example

STUN 


STUN Mode:

STUN Server:

PROTOCOL 

Default Protocol:

Default SIP Account:

SECONDARY SIP ACCOUNT 

Double Streaming:

Streaming Pattern:

SIP User 2:

SIP Display Name 2:

SIP Auth User 2:


SIP Password 2:

SIP Registrar 2:

Alternate SIP Port 2:


SIP Port 2:

RTP Port 2:

QoS 

RTP DSCP:

Packet Replication:

SIP 

SIP Registration:

SIP Accept Calls:

Only From Server:

SIP User:

SIP Display Name:

SIP Registrar:

SIP Auth User:

SIP Auth Password:


SIP Outbound Proxy:

Keepalive Interval:

Min. Regist. Period:

SIP Port:

RTP Port:

DIRECT RTP 

Accept Calls:

Send Only:

RTP Port:

RTP Timeout:

AoIP Settings

STUN MODE

Allows auto discovery of actual public addressing.

STUN SERVER

You can use `stun.aeta-audio.com` for example.

SIP REGISTRATION

Uncheck to disable registration without having to clear the SIP parameters.

KEEPALIVE INTERVAL

Renew registration at specified period.

SIP AND RTP PORTS

These are the port numbers for the codec itself.

RTP DSCP

If the network supports DiffServ, you can set here the DSCP value (decimal number) assigned to the audio stream transmitted by the codec.

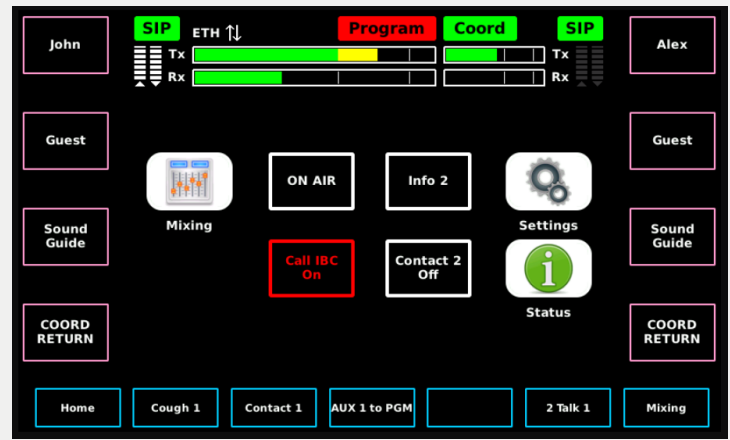
PACKET REPLICATION

0:normal mode without replication / 1:packet replication with short delay / 2:packet replication with interleaving.

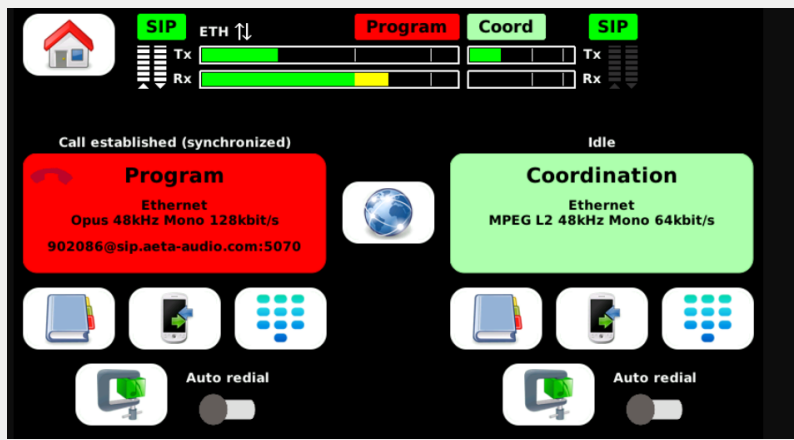
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**Dedicated PC Application
MyScoopTeam**

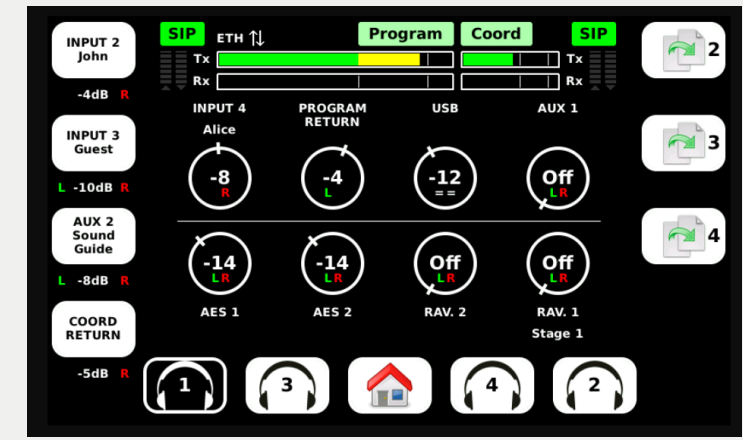
Control all your ScoopTeam settings, remotely and in real time over the internet, including 4G mobile networks



Home Screen



Call Page



Headphone Mix

- Products
- Services
- Software

Remote Access Portal

Your virtual MCR - cloud based
Get an overview on all your codecs on one page



Selected Device

Name:

Group Tag:

Save Open HTML Page

Filter List

Show Connected Only

Group Filter

Export CSV...

	Name	Group Tag	Device Type	SN	Version	MAC	SIP Status and Numbers		Link Status		Network I/F	Options
							1	2	1	2		
1	AmScoopTeam	IBC	ScoopTeam	134	1.09.01	0CC5	-	-	-	-	Ethernet	ISDN / MOBILE / Wi-Fi / 2xAOIP / RA+ / AES67
2	MS1	MCR	MultiScoop/Codec	12	1.03.01	0F00	903840		Released		Ethernet	2xAOIP
3	MS2	MCR	MultiScoop/Codec	13	1.03.01	0F01	903841		Released		Ethernet	2xAOIP
4	MS3	MCR	MultiScoop/Codec	11	1.03.01.GPIO	0EFF	903839	9038392	Established	Idle	Ethernet	2xAOIP
5	MS4	MCR	MultiScoop/Codec	17	1.03.01.RTL	0FC4	904036		Idle		Ethernet	ISDN / 2xAOIP
6	S5 SJ	Office	Scoop5 S	1084	1.0x	0AEC	902796		Released		Mobile (Int.)	ISDN / POTS / MOBILE / Wi-Fi / 2xAOIP
7	Scoop5S 1423	Bissersheim	Scoop5 S	1423	1.14.01	0CCE	903278	9032782	-	-	Mobile (Int.) / Mobile (Int.)	ISDN 4B / MOBILE / 2xAOIP
8	Scoop5S Vilette	Salon Radio	Scoop5 S	1553	1.14.01	0A4D	0033183622934	9034252	-	-	Ethernet	ISDN / MOBILE / 2xAOIP
9	ScoopFoneAms	IBC	ScoopFone 4G	0	1.12	075E	-	-	-	-	-	MOBILE
10	Scoopy+S 1059	Bissersheim	ScoopY+ S	1059	1.13.02	08D8	902264		Idle		Ethernet	ISDN / MOBILE / Wi-Fi / BGAN / 2xAOIP / REC
11	SF4G	Bissersheim	ScoopFone 4G	140	1.11.02	0932	902354		Idle		Mobile (Int.)	MOBILE
12	ST Premium	Bissersheim	ScoopTeam	136	1.09.01	0CCC	903276	9032762	-	-	Mobile (Int.) / Ethernet	ISDN / MOBILE / Wi-Fi / 2xAOIP / RA+ / AES67
13	Studio A	MCR	MultiScoop/Codec	15	1.03.01	0F03	903843	9038432	Released	Released	Ethernet	2xAOIP
14	Talent	Office	µScoop	158	1.10.01	0A66	902662		-	-	Ethernet	2xAOIP
15	µScoop A	Bissersheim	µScoop	163	1.09.02	0A8A	902698		Idle		Ethernet	2xAOIP

Outside

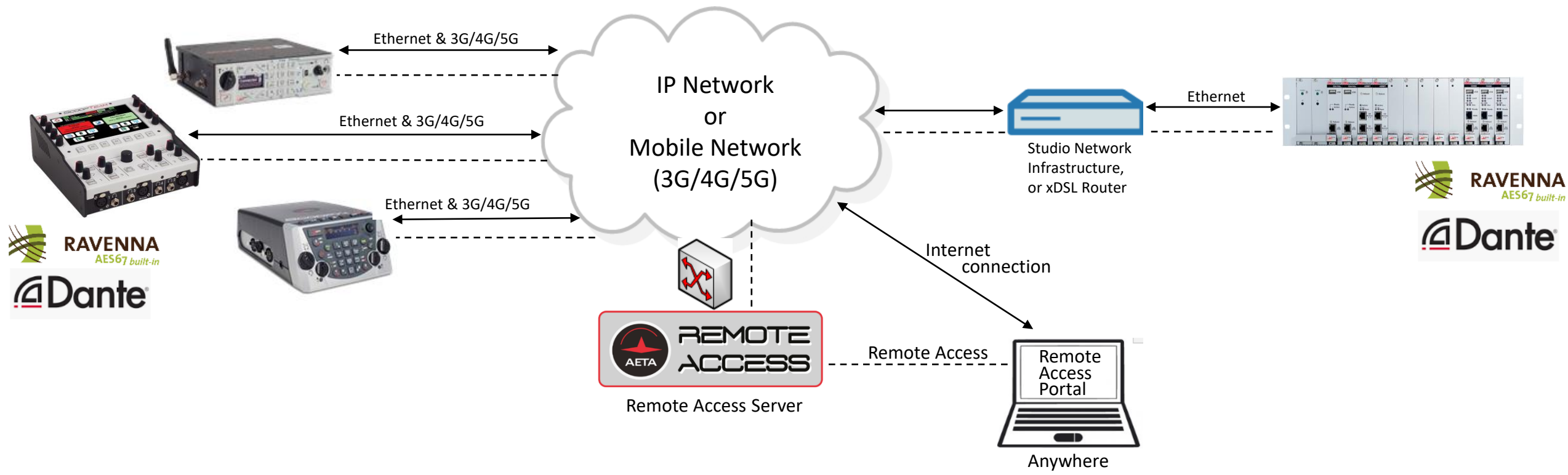
SCOOPTEAM
SCOOPY+ S
SCOOPONE 4G

Remote Access function

Access to the HTML page of the codec over internet

Studio - MCR

MultiSCOOP
SCOOP5 S
μSCOOP



- Products
- Services
- Software



- Products
- Services
- Software

MultiSCOOP

Modular multichannel codec, up to 16 audio connections in 3U

AUDIO CHANNELS X 16

8x STEREO



 **Dante**

 **RAVENNA**
AES67 built-in



Ethernet



ISDN

- **Hot-swap modules** (sold separately)
- **Up to 8 AoIP & ISDN** modules (with double mono codec)
- **Up to 2 AES 67/Ravenna** modules – Dante compatible
- **Up to 2 Control** modules
- **Full redundancy:** power, control and AES67 modules
- **Separate audio and control networks** for more safety

 **SCOOP**
MANAGER

- Products
- Services
- Software

MultiSCOOP



- Products
- Services
- Software

The economical solution to achieve point to point AoIP links



AUDIO CHANNELS X 2

STEREO



Ethernet

- Power over Ethernet 48 V
- Low power platform
- Compact design (1/3 of 19")

2 versions: Analog or digital I/O

- Products
- Services
- Software

μSCOOP



- Products
- Services
- **Software**



- Products
- Services
- Software



Scoop Manager is a professional software solution that allows you to **quickly manage a pool of AETA codecs via an Ethernet LAN**, using an AETA's specific protocol.

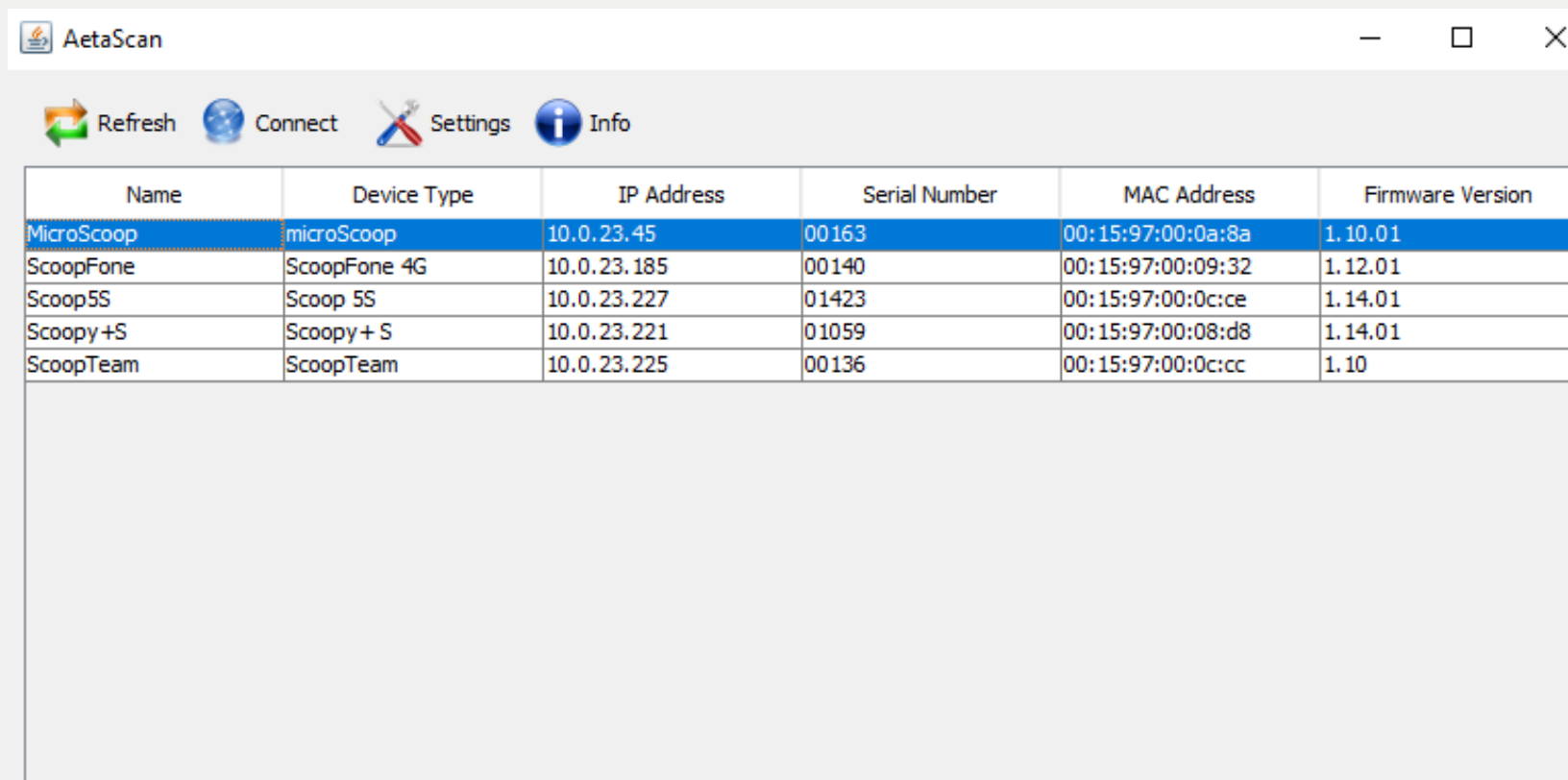


- Supervision of your whole pool of AETA codecs
- Support of various call types (IP/ISDN/Telephone)
- Management of single and double codec modes
- Easy call setup and management
- Multiuser operation
- Easy access to the codecs embedded HTML servers
- Easy handling of all contacts
- Remote party name display
- Phonebook, call history
- Broadcast color patterns
- Configurable alarm sounds



AETA SCAN:

AETA Scan scans your LAN to look for AETA codecs and displays MAC and IP addresses. A double click on the selected codec opens its dedicated web page, enabling the configuration and control of the codec. Download it for free on our website www.aeta-audio.com/support



The screenshot shows the AetaScan application window with a toolbar containing 'Refresh', 'Connect', 'Settings', and 'Info' buttons. Below the toolbar is a table listing detected devices.

Name	Device Type	IP Address	Serial Number	MAC Address	Firmware Version
MicroScoop	microScoop	10.0.23.45	00163	00:15:97:00:0a:8a	1.10.01
ScoopFone	ScoopFone 4G	10.0.23.185	00140	00:15:97:00:09:32	1.12.01
Scoop5S	Scoop 5S	10.0.23.227	01423	00:15:97:00:0c:ce	1.14.01
Scoopy+S	Scoopy+ S	10.0.23.221	01059	00:15:97:00:08:d8	1.14.01
ScoopTeam	ScoopTeam	10.0.23.225	00136	00:15:97:00:0c:cc	1.10

- Products
- **Services**
- Software



- Products
- Services
- Software



All AETA codecs running through IP integrates free and pre-registered factory SIP accounts. These accounts are exclusive for the unit and offer quick and easy connection to the SIP server.

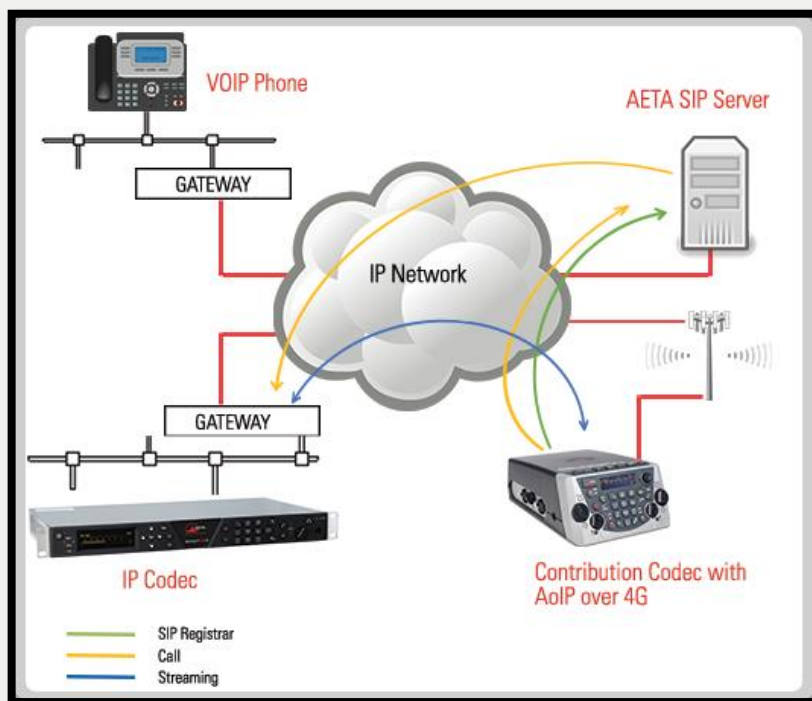


Control all the parameters of your codecs remotely and in real time via Internet, including 4G mobile networks. **AETA Remote Access is available as an option on every AETA codecs running through IP.**

- Products
- Services
- Software



Simplify and secure your AoIP connections



Using a SIP server greatly simplifies the setting up of a connection. AETA has set up a public SIP server, dedicated to Audio over IP broadcasting applications.

Our customers can get accounts and register their audio codecs on this server.

All AETA codecs include a free pre registered SIP account exclusive for the unit.

AETA has also set up a **PREMIUM SIP service with additional features:**

- High availability with 24/7 monitoring.
- Redundant and backed up power.
- Dedicated and reliable equipment.
- Portability - **PREMIUM**
- Optional blocking of incoming and outgoing calls - **PREMIUM**
- Speed Dial and Group Call Numbers - **PREMIUM**
- Autonomous management of passwords - **PREMIUM**
- Quarterly Calls history - **PREMIUM**

The service is hosted in a data center on a dedicated server (TÜV certified, ISO 27001 standard). The service provides the quality and reliability guarantees expectable for a professional service.

How to subscribe to the PREMIUM SIP service

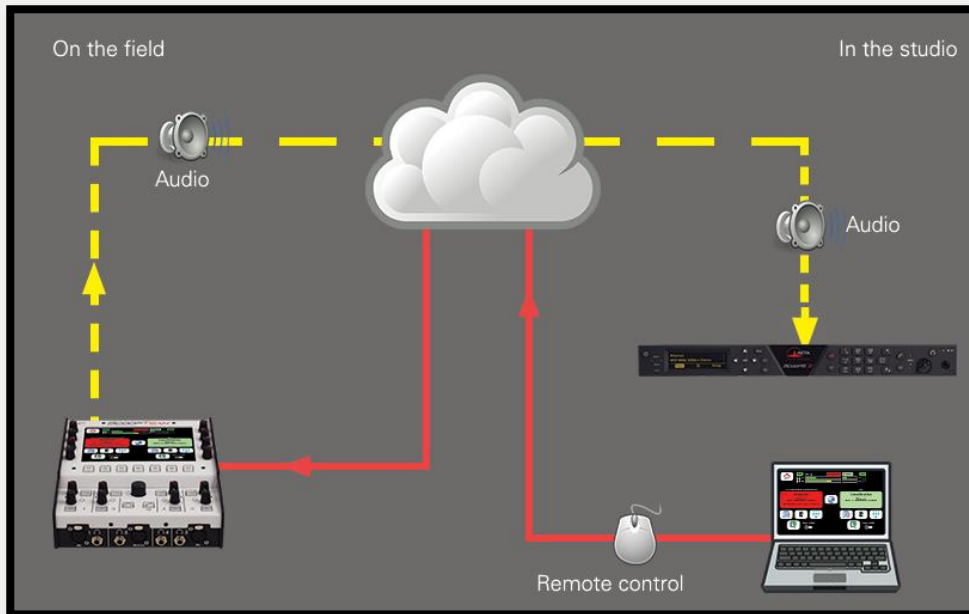
- Subscription of an account for a duration of one year: **contact us**
- Commissioning fees for a SIP account : **contact us**

The fixed subscription allows a device registered on the server to set up connections of unlimited duration with any other device (or agent) also registered on the server.

- Products
- **Services**
- Software



Control all the parameters of your codecs remotely and in real time via Internet, including 4G mobile networks.



AETA Remote Access feature is available for every AETA codec running through IP:

- ScoopTeam
- Scoopy+ S
- ScoopFone 4G
- ScoopFone IP
- MultiScoop
- μScoop
- Scoop5 S
- Scoop5 S-IP

WARNING: In order to get this new feature, it is necessary to install a recent software update, and to subscribe to the Remote Access option.

THANK YOU !

