APPLICATIONS

- · Professional Audio Networks
- DVB compliant encoding of audio signals
- · Satellite feeds
- Distribution networks for radio stations
- Feeding digital cable networks
- · Studio Transmitter Link
- · Point-to-point connections
- Point-to-multipoint connections (up to 16 destinations)

FEATURES

- · Up to 12 digital stereo channels
- · Up to 8 analog stereo channels
- MPEG TS over IP outputs (unicast or multicast)
- DVB-ASI output (optional)
- Output of elementary streams over IP
- · Digital MPX transport
- Transmission of ancillary data (over IP or serial interface)
- 8 GPIO inputs
- · Remote controllable over IP via SNMP
- RAVENNA / AES67, Dante[®], ST 2110-30, ST 2110-31

Compression algorithms for TS

- MPEG 1/2 Layer 2
- AAC (optional)
- · Enhanced aptX (optional)

Compression algorithms for ES

- Fraunhofer xHE-AAC® (optional)
- AAC (optional)
- Linear PCM
- MPEG-1/2 Layer 2/3
- Enhanced aptX (optional)
- · OPUS
- · G.711, G.722

ENC7000

Multi-Channel Audio Encoder

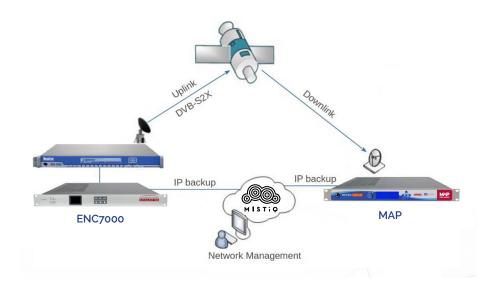
The ENC7000 Audio Encoder features MPEG-4 Advanced Audio Coding (MPEG-4 AAC) and MPEG-1 Layer 2 encoding capabilities. All of the common bit rates and sample rates are offered to enhance the IP delivery of Audio.

The ENC7000 Audio Encoder sets standards for high quality audio encoding. Based on the proven P561 platform, it provides highest signal quality, best build quality and service without compromise. Customers around the world trust our market leading IP Audio Encoder. Either up to 16 digital or up to 8 analog stereo audio channels can be encoded simultaneously. They can then be output as MPEG-2compliant DVB transport streams or elementary streams via the IP interface. Several configurations are available to flexibly adapt the audio input configuration to the network requirements. Each channel can be configured individually.

The **ENC7000** Encoder can be conveniently managed via the integrated web interface

with all common web browsers. The device can also be monitored and managed via SNMP or REST-API. The unit is built in a compact 19" 1 U housing. The transmission of ancillary data and switching contact information (GPIO) is possible with the default interfaces. Optionally, a second data interface can be used as a backup function.

The ENC7000 Encoder is CE and RoHS compliant to meet the demand of users worldwide. The basic unit is licensed for one audio encoder channel and the functions RDS and GPIO. Further channels and options are possible. If you need more than 16 stereo channels, our ENC6000 Encoder is the best choice.





TECHNICAL



AUDIO INPUT	
Digital, AES-3	• XLR, Balanced for stereo input channels 1-8
(max. 12 stereo)	D-sub 25 connector for stereo input channels 9-12 (AES/EBU TASCAM pin assignment)
	• Digital MPX transport
Analog, Balanced	•XLR, Balanced for stereo input channels 1-4
(max. 8 stereo)	D-sub 25 connector for stereo input channels 5-8 (Analog TASCAM pin assignment)
	Audio Frequency Range:
	20 Hz to 20 kHz (± 0.3 dB)
	• THD+N (1 kHz at max. level): < 0.01 % at 1 kHz
	Crosstalk attenuation at 1 kHz; > 100 dB
	•S/N ratio (weighted): > 80 dB
RAVENNA / AES67	Input of up to 16 stereo channels
	Formats: L16, L24, L32 Sampling rate: 32 kHz, 48 kHz
	• Channels: 1 to 64
	· ST 2110-30 and ST 2110-31 compatible
Dante®	
AUDIO COMPRESSION	
Algorithms	• ISO/IEC 1172-3, 13818-3 MPEG-1/2 Layer 2/3
(some require additional licenses)	• ISO/IEC 13818-7 MPEG-2 AAC-LC • ISO/IEC 14496-3 MPEG-4 AAC-LC.
	• HE-AAC V1/2, AAC-LD, AAC-ELD
	Fraunhofer xHE-AAC®
	• Enhanced aptX
	Linear PCM OPUS
	• G.711, G.722
Encoding bit rate	· All bit rates are supported according to
	the standards of the respective algorithms
Modes	Sampling rate: 32 kHz, 48 kHz Stereo, Dual, Mono channel
Ancillary data	Private stream via UECP within the MPEG-2
,	transport stream or embedded in MPEG audio data
	• RS-232 interface
	Breakout cable (optional): 4 or 8 connectors, conversion from
	Sub D 25 to Sub D 9
TRANSPORT PROTOCOLS	
ASI Output (optional)	DVB-ASI (BNC connector)
IP Output	Output of DVB MPEG-2 transport streams
	including service information according to ETSI
	EN 300 468, compliant to "Pro-MPEG Code of Practice #3 release 2" (FEC optional, see below)
	Output of elementary streams
	• FEC according to ST 2022-1 (optional)
	Seamless Protection Switching according to ST 2022 7 (optional)
	ST 2022-7 (optional) SRT (Secure Reliable Transport)
UDP Multicast IP input	Mux with 8 stereo Audio (Analog and Digital) at
I III	SI.No.1
	1 . 1

>1 mbps

NETWORK INTERFACES	NETWORK INTERFACES	
Ethernet	3 separate Ethernet interfaces (IEEE 802.3, RJ45, 100/1000 MBit/s) • 2x Data (elementary / transport streams via IP) • 1x Control (Web interface, SNMP and Ancillary Data)	
VLAN Management		
SYSTEM CONFIGURATION, CONTROL AND MONITORING		
Management / Monitoring	Via Ethernet by accessing the on-system HTTP web server with any Internet browser • REST API • Silence detection (optional) • Via Ethernet with SNMP Traps in case of triggered alarms • Via the front panel keyboard and display	
POWER REQUIREMENTS		
Supply voltage	Integrated switching power supply, input voltage 100 to 240 V AC +/- 10 %, 50 to 60 Hz -48 V DC (optional)	
Redundant power supply (optional)	The optionally available redundant power supply protects the operation of the device and comes with the following functions: • Measurement of the power supply voltages, values are provided via web GUI or SNMP • SNMP trap generation on power supply fail • Activation of switching contacts on power supply fail • Automatic switch-over in case of power supply fail	
Power consumption	< 20 W	
PHYSICAL PARAMETERS		
Chassis	• 19" rack mount cabinet, 1 U	
Size	Width: 483 mm Depth: 360 mm Height: 44 mm	
Weight	4.5 kg	
ENVIRONMENTAL CONDITION	ons	
Operating temperature	0°C to 45°C	
Storage temperature	-20°C to 70°C	
Humidity	20 % to 95 %, non-condensing	
Certifications	DIN EN IEC 62368-1 DIN EN IEC 61000-6-2:2005	
Ambient Temperature	-10 degrees C to 45 degrees C For indoor equipment	

