MIMO4040CDN

DIGITAL MATRIXES

Conferencing Digital Matrix with AEC & DANTE™/AES67 Interface



PRODUCT OVERVIEW

MIMO4040CDN is a conferencing digital matrix with native processing of 40 inputs and 40 outputs (8 analogue inputs, 8 analogue outputs, 32 DANTE™/AES67 digital inputs and 32 DANTE™/AES67 digital outputs). Incorporating audio over IP technology, MIMO4040CDN allows signal transmission using a Local Ethernet network, suppressing cabling distance limitations. MIMO4040CDN comes with a CONFERENCE tailored firmware, including Automixer, Feedback Killer and Acoustic Echo Cancelling functions (*). It features accessories from WPNET Series connected by Ethernet interface and PoE DC supply ready. This ensures flexible and efficient installations. MIMO4040CDN is also capable of executing EclerNet projects embedded into its processor and electronics engine, acting as a UCP (User Control Panels) server within an EclerNet devices network.

Each microphone added to an AEC room (available under AEC tab in ENM SW) will decrease the remaining standard MATRIX size by 4 input channels and 4 output channels (available under the MATRIX tab in ENM SW). Please, see <u>EclerNet Manager Software user manual</u> for further details.

(*) Acoustic Echo Cancelling processing for the MIMO4040CDN is available by means of software licensing, upon request. Contact you Ecler distributor for further information.

ECLER TECHNICAL DATA SHEET

KEY FEATURES

- CONFERENCE firmware only (no STANDARD firmware available)
- 40x40 audio input / output ports available
- 40 input channels x 40 output channels digital audio native matrix
- Audio input / output available ports:
 - 8 MIC / LINE analogue audio inputs
 - 8 LINE analogue audio outputs
 - 32 DANTE [™]/AES67 digital input channels
 - 32 DANTE [™]/AES67 digital output channels
- 8 GPI ports
- 8 GPO ports
- 2 DANTE[™]/AES67 interfaces (primary and secondary), available for configuration of redundant DANTE [™]/AES67 networks
- 1 Ethernet programming and control interface (EclerNet, UCP, TP-NET and WPNET series control panels)
- 1 serial control port, RS-232 (TP-NET)
- Embedded project manager and server of the project's integrated UCP panels
- Connection of all peripherals directly via EtherNet network, and without local wiring (remote controls WPNET series, WP22DN wall panel DANTE™/AES67 interface, etc.)

APPLICATIONS

- Commercial
- Hospitality
- Education
- Corporate
- Sports and wellness

ACCESSORIES & COMPATIBLE DEVICES

- WPNET4KV
- WPNET8K
- WPNETEX
- WP22DN
- WPNETTOUCH
- PAGENETDN







PAGENETDN (Paging not available just compatible as standard microphone and aux input Dante transmitters and UCP Control)

TECHNICAL SPECIFICATIONS

DSP		
001	CPU	Quad core 64bits 1GHz
	Sampling rate	48 kHz
	Latency analog IN to analog OUT	<4.3 ms.
Converters		
	Resolution	24 bit
	Dynamic range	AD:110dB, DA: 115dB
Analogue		
	8 Input / Output	Terminal block (Symmetrical)
	2 monitor output	Terminal block (Symmetrical)
	Headphones connector	Jack ¼
	Analogue input headroom	+27dBV = +30dBu
	Max. output level	+18dBV = +21dBu
	Input sensitivity @0dBV out	from -50dBV to +10dBV in 0.5dB steps



Input impedance	Balanced, >4k Ω
Phantom power	+42VDC, 5mA max. software switched
· · · · · ·	>200mW/200 Ω
Headphones	5Hz to 24kHz
Frequency response (-3dB)	
Flatness	better than ±0.1dB
THD+Noise @ 1kHz, 0dBV input (line)	<0.004%
THD+Noise @ 1kHz, -40dBV input (mic.)	<0.008%
Output Noise floor FFT (20Hz - 20kHz)	better than 115dB
Interchannel crosstalk (20Hz - 20kHz)	better than 90dB (100dB typ.)
Channel Leakage (20Hz - 20kHz)	better than 100dB (115dB typ.)
CMRR 20Hz- 20kHz	65dB typ.
DANTE™/AES67 Audio interface	
DANTE™/AES67 Network Tx / Rx channels	32/32
Latency	1 / 2 / 5 ms (selectable)
Connector	1xRJ45 primary, 1xRJ45 secondary
Cable length between devices	100m CAT5e/CAT6
Processing	
Input level (x40)	Range: from Off to 0 dB
	Mute: Yes
	Signal Polarity reverse: Yes
	Metering: VU+clip pre & post fader
Output level (x40)	Range: from Off to 0 dB
	Mute: Yes
	Solo: Yes
	Signal Polarity reverse: Yes
	Metering: VU+clip pre & post fader
Output gain (x40)	Range: from 0 to +6 dB
Input delay (x40)	from 0 to 1000 ms. Units: sec/ms/m/cm.
Output delay (x40)	from 0 to 1000 ms. Units: sec/ms/m/cm.
Parametric EQ types (4max/input -	Bypass / On-Off all channels
8max/output)	Param Eq. Freq: 20Hz-20kHz;
	Gain: -60/+12 dB
	Q: 0.3 to 200
	Low & High Shelf 6/12 dB/oct
	Low & High Pass 6/12 dB/oct
	All Pass 1/2 order
High & Low pass output X-over filters (x40)	Bypass On-Off
	Butterworth in 6/12/18/24 dB/oct
	Bessel in 12/18/24 dB/oct
	Linkwitz-Riley in 12/24 dB/oct
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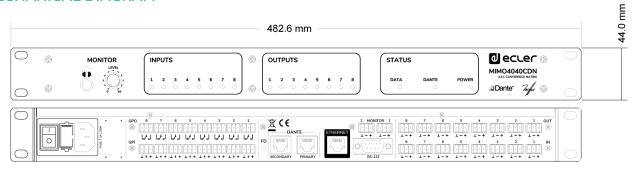
Input noise gate (x40)	Bypass On-Off
	Threshold: from –80 dBV to +18 dBV
	Depth: 0 dB to 80 dB
	Attack time: from 0,1 ms. to 500 ms.
	Hold time: from 10 ms. to 3000 ms.
	Release time: from 10 ms. to 1000 ms.
Input compressor / limiter (x40)	Bypass On-Off
	Threshold: from –36 dBV to +18 dBV
	Knee: hard / soft
	Ratio: inf:1 (limiter)
	Attack time: from 0,1 ms. to 500 ms.
	Release time: from 10 ms. to 1000 ms.
	Make up gain: from 0 to +10 dB
Input Frequency Shifter (x40)	Per input. ON / OFF function
(Feedback Loop Reducer)	
Output Limiter (x40)	Bypass On-Off
	Threshold: from –36 dBV to +18 dBV
	Attack time: from 0,1 ms. to 500 ms.
	Release time: from 10 ms. to 1000 ms.
Built in Signal Generator	Sine: from 20 Hz to 20 kHz
	Polarity: from 20 Hz to 20 kHz
	White noise
	Pink noise
Stereo Linking	Adjacent input / output channels
	Linked processing
	Matrix routing linked
Mix Matrix	Size: 40x40
	Analogue in/out ports: 8x8
	DANTE™/AES67 network in/out ports: 32x32
	Vol: Input, Output, Crosspoint
	Mute: Set/Clear individual, row, column, all
	Input /output Mono/stereo selector
	Meter: Input /output VU and clip
Ducker (x25)	Input: IN1 to IN40
	Priorities: 25 (1 max, 25 min)
	Depth: 0 dB to 80 dB
	Attack time: from 5 ms. to 2000 ms.
	Release / Hold time: from 50 ms. to 3000 ms.
AEC audio input channels	Up to 8, subject to software licensing

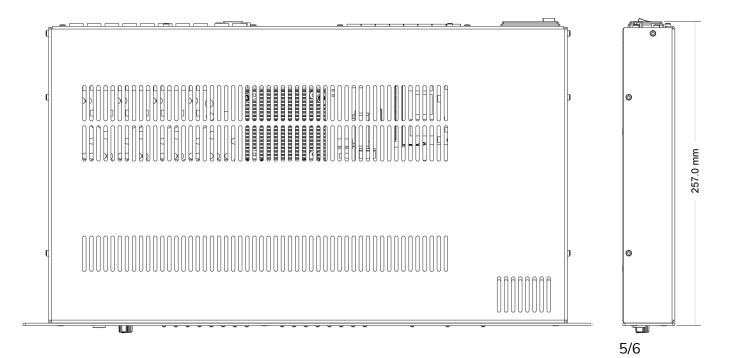
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Mechanical	
Dimensions	482,6 x 44,0 x 266,5mm / 19.0"x 1.7" x 10,5"
Weight	3,25kg / 7.17 lb.
Power supply	
Mains	90-264VCA 47-63Hz
Power consumption	30VA
Miscellaneous	
Management Connectivity	Ethernet Base-Tx 10 /100Mb, 1GB Auto X-Ove
	CAT5e or better
GPI	8, from 0 to 12VDC or TTL level
GPO	8, 3 poles isolated relay; 1A, 48VDC max.
Aux. Power Supply for Remotes & GPI	+12VDC, 1.2A max. (short circuit protected)
Time and date retention (battery)	> 3 months
Programming and control application	·
EclerNet Manager	From v6.00
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MECHANICAL DIAGRAM

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A & E SPECIFICATIONS

The digital matrix with native processing shall provide up to 40 x 40 audio channels individually configurable over Eclernet Manager Software. It shall support 8 analogue I/O and 32 digital I/O channels configurable as either DANTE[™] or AES67 formatted networked audio. It shall include 2 DANTE[™]/AES67 interfaces (primary and secondary), available for configuration of redundant DANTE[™]/AES67 networks.

The digital matrix shall manage external control interfaces such as remote control touch screens, digital audio wall panel interfaces, remote controls panels and networked amplifiers.

Programming and remote management shall be available via Ethernet using EclerNet Manager software (either point-to-point, with direct CAT5/CAT6 cable, or from an Ethernet network connection). Remote control via custom control panels UCP (User Control Panels). iOS and Android remote control app available (Ecler UCP V2). Remote control from third-party systems shall be available using TP-NET control protocol through Ethernet o RS-232 ports.

On the front panel, the matrix shall include Power, DANTE™/AES67 and Data status LEDs, Inputs and outputs signal level indicator, monitor output jack and monitor level knob. On the rear panel, the matrix shall include 8 analogue I/O and 2 monitor outputs (euroblock connector), 8 GPI and 8 GPO general purpose controls (euroblock connector), RS-232 port (DB9 connector), Ethernet, Primary and Secondary DANTE™/AES67 RJ-45 ports.

All internal processing shall be digital (DSP). The DSP shall include Matrix router-mixer, from any input to any output (analog and/or digital DANTETM/AES67) with adjustable crosspoint level, treatment of channels in mono or stereo mode, level, mute, vumeters and phase adjustment in inputs and outputs, internal signal generator (sinusoidal signal, pink noise, white noise, polarity test), parametric EQ, delays, noise gate, compressor on input channels, compressor / limiter on outputs, priorities (ducking) between input channels, virtual and physical paging consoles. Configuration memory management (presets). Events scheduled on a calendar basis.

The digital matrix shall include Automixer and Feedback killer features by default, as well as a maximum of 8 AEC microphone processing channels that can be enabled by software licenses

Audio conversion shall be 24-bit, 48 kHz. The dynamic range shall not be lower than 110 dB, A-weighted with a maximum input level of +27 dBu and maximum output level of +21 dBu.

The dimensions of the matrix shall be $482 \times 44 \times 266,5$ mm. The weight shall be 3,25 Kg.

The digital matrix shall be the ECLER MIMO4040CDN



All product characteristics are subject to variation due to production tolerances. **NEEC AUDIO BARCELONA S.L.** reserves the right to make changes or improvements in the design or manufacturing that may affect these product specifications.

For technical requests address to your supplier, distributor or fill the contact form in our website, at Support / Technical Request.

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