

Amadeus Core – Data sheet

Product	
Available models	Amadeus Core 128
	Amadeus Core 32
Hardware	
Terminals	DANTE Primary/Secondary: 2x Network Control
reminais	(Ethernet R.145): Option: MADI Optical In/Out
Power supply	100-240Vac/ 8-4 A/ 700W/ 50-60Hz
Power supply (redundant)	100-240Vac/ 8-4 A/ 700W/ 50-60Hz
Installation	19" Rack mounting; 4 RU; Box Dimensions 520mm
	(H) x 205mm (W) x 509mm (D)
Net Weight	15 kg
Server Grade Hardware Components	yes
CPU based	yes
Audio Specifications	
I/O Channels (Dante or MADI)	Core 128: 128 ln / 128 Out;
	Core 32: 32 In / 32 Out;
I/O Configuration	Core 128: 64 Immersive Audio Inputs (Option), 64
	Microphone Inputs, 128 Loudspeaker Outputs;
	Core 32: 16 3D Audio Inputs (Option), 16
On white Date	Microphone Inputs, 32 Loudspeaker Outputs;
Sample Rate	48 KHZ
Bitrate	I/O: 24bit, Signal Processing 32bit floating point
Synchronisation	
Internal Processing Specifications	
Input Stage	Parametric equalizers, level, dynamic processing,
	delays, input grouping, mute/solo, feedback
	suppression for extreme acoustical settings (e.g.:
Ontionaly Immorphy Audia Inputa	Cathedral like acoustics)
	No input processing
Acoustic module	 Natural sound. Every foom is unique with its natural acoustic signature. Each Amadous
	Core is individually tailored for the room by a
	unique internal 3D model
	unique, internal SD model. √ Hybrid evetam: A regenerative approach for
	 myblic system. A regenerative approach for natural reverboration approach in Line
	and an every eration enhancement, in-Line
	approach for increase of loudness indoors of
	generation of the acoustics outdoors

	 Signal Processing: Double Matrix Approach and additional decorrelated algorithmic reverb based on neutral FDNs independent for each channel (Feedback Delay Networks) Room shaping: Control of sound energy distribution within the room Time shaping: Early reflections based on a 3D Model of the room providing the correct acoustical geometry; room size can be increased or single walls or the ceiling can be moved based on a geometry-dependent scaling Spectral shaping: The spectral balance can be shaped on a master level and for individual or a group of microphones and loudspeakers, but also for different parts of the reflection pattern Reverb shaping: Acoustical parameters can be shaped nearly independently through fine- tuning of the regenerative part and the use of the algorithmic reverb which is fed by the time- correct delays Density shaping: Double or sparse matrix approach to increase or decrease density of reflections
Option: Immersive Audio Module	 Immersive Audio using state-of-the-art 3- Dimensional WFS and Amplitude panning, Control through the Amadeus 3D-Pan Plugin running on all major DAWs Control by external devices using OSC Active Acoustics and Immersive Audio run in parallel WFS provides sweet spot independent sound source localization for every audience seat Delay interpolation keeps movements sounding smooth and clean Increase or decrease width or blur by focussing sound on a single speaker or include all speakers Virtual Surround Sound: place fixed virtual sources to simulate 7.1, 9.1.4 or Dolby Atmos loudspeaker layouts Recording and Playback of soundscapes Bass Management
Output Stage	Parametric equalizers, level, delays, output grouping, mute/solo (e.g.: cathedral like acoustics), speaker mute/solo, speaker test oscillator
GUI	 Easy to use Graphic User Interface for quick control of presets, main parameters, individual microphone and loudspeaker channels and group control Simplified web-based control interface for the client with preset selection and meter control, 2D representation of the venue External control for media controllers through OSC (AMX, Crestron, Max/MSP etc.)

General	
Warranty	2 years



Amadeus Acoustics GmbH, Franzensgasse 3/5, 1050 Vienna, Austria

Technical information contained in this document is subject to change without prior notice. Should you need further clarification or information, please contact us.