

Model DB910

FULL DUPLEX COMPACT IP AUDIO CODEC

Inspired by DEVA continuous strive for excellence, and based on our long-term formative experience in the field of IP audio broadcasting tools, DEVA product portfolio is extended with yet another strategic solution - the DB910 Full Duplex IP Audio Codec.

A full duplex solution when two tools are in use, the DB910 allows you to send and receive audio at the same time, while guaranteeing fast and easy monitoring of the signal fed to the transmitter. Where two-way transmission is not required, DB910 may be used either as an encoder or a decoder, being fully compatible with DEVA's already widely-known hallmark IP audio codecs.

The real time audio encoding/decoding into a high quality configurable format process is uncompromisingly guaranteed by the powerful DSP processor. Supporting all the mandatory compressed audio streams and lossless, uncompressed PCM stream, the DB910 can be used for a wide range of professional audio applications: Broadcast, Internet Radio, Studio to Transmitter Link and VoIP.

Apart from its main audio source, DB910 supports several IP audio sources and a built-in MP3 Backup Audio Player in addition. With several backup audio sources incorporated, the DB910 will immediately switch between the sources when the audio is lost, and return to the main one when the audio is restored, with no user intervention. The audio content for the MP3 player is recorded on the internal SD card. The backup files can be managed remotely through the built-in FTP Server, via standard FTP client.

Combining the super compact form of the DB90 and some of the most impressive features of the renowned DB9000 line in a single product, the DB910 will meet the requirements of the most demanding broadcasters. This advanced radio gear is equipped with two-way GPIOs and RS-232 serial communication that could be used for telemetry and automation. The front panel audio level bar graphs and LED indicators of the unit allow reading the current state of the device at a glance.

Controlled through a simple and intuitive HTML5 interface and any kind of web browser, be it your PC, smart phone or tablet, the DB910 could be managed remotely or locally.

DB910 is the ultimate all-in-one solution to maintain your competitiveness, combining essential professional applications and key trends.



FEATURES

- Analog Audio Output and Input (4 x ¼" phone jack)
- Digital Audio Output and Input (2 x ¼" phone jack)
- High Quality HE-AAC (v.1 and v.2), MPEG-1 & MPEG-3
- 32 kHz, 44.1 and 48 kHz sample rates support
- Decoding of all standard bitrates and VBR as well
- Full Control and easy setup via a standard web browser
- DHCP Client automatically assigning Network Settings
- IP address pronunciation at startup (through the Phones)

- Auto backup switching in the event of an audio loss
- Full duplex IP audio transmission
- Build-in MP3 Player
- 6 x General Purpose Inputs
- 6 x General Purpose Outputs
- 2 x 6 LED front panel audio level bar graphs
- 2 x 6 GPIO & 4 x Status LEDs, Phones out
- UPnP for easy discovery in Local Networks





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SPECIFICATIONS

Operating Modes		
Two-way Full-duplex TCP, RTP		
Transmitter	Shoutcast/Icecast compatible TCP server, Icecast Source Client, RTP	
Receiver	IP Audio Client, Icecast Source Server, RTP	
File Player	SD Card based player	
Audio Codecs		
Channels	1 or 2	
Sample Rates	32 kHz, 44.1 kHz, 48 kHz	
Codec	MPEG-1 Layer 3; HE-AAC (v.1 & 2); 16-bit PCM	
Bitrates (Enc)	Up to 320 kbps (MPEG-1); Up to 88 kbps (HE-AAC); Up to 1536 kbps (PCM);	
Bitrates (Dec)	All standard bit rates, including VBR	
Audio Backup		
Source	Any of the available sources; Up to 3 configurable backups	
Trigger	Audio Silence Detector	
Threshold	Adjustable, -90dBFs to 0dBFs	
Trigger time	Adjustable, 1s to 240s	
File Player		
Storage	microSD Card	
Codecs	MPEG-1 Layer 3, HE-AAC (v.1 and v.2)	
File types	*.MP3, *.AAC, *.M4A, *.M3U	
Playback Modes	Alphabetical ascending and descending, Shuffle, Playlist and Shuffled playlist	
Remote file	Build-in FTP server	
Analog Audio Input		
Connector	2 x 1/4" (6,3mm) phone jack; Shared with Digital input, software selectable	
Туре	Balanced	
Impedance	600Ω or High Impedance, software selectable	
Level (0 dBFS)	-12dBu to +18dBu, software selectable	
Frequency Resp.	±0.5dB, 10Hz-20kHz	
Distortion	<0.01% THD+N	
Dynamic range	>100 dB	
Analog Audio Output		
Connector	1/4" (6,3mm) phone jack; Shared with Digital output, Software selectable	
Туре	Balanced	
Impedance	600Ω or High Impedance, software selectable	
Level (0 dBFS)	-12dBu to +18dBu, Software selectable	
Frequency Resp.	±0.5dB, 10Hz-20kHz	
Distortion	<0.01% THD+N	
Dynamic range	>100 dB	

Digital Audio In	put	
Type	AES/EBU (IEC 60958)	
Connector	1/4" (6,3mm) phone jack; Shared with Left	
	Analog input, software selectable	
Sample rate	32kHz, 44.1kHz and 48 kHz	
Digital Audio Output		
Connectors	1/4" (6,3mm) phone jack; Shared with Left	
	Analog output, software selectable	
Type	AES/EBU (IEC 60958)	
Sample rate	32kHz, 44.1kHz and 48 kHz	
GPIO		
	6 x GPI; 3.3V TTL logic	
Туре	6 x GPO; Open-collector; 20V/100mA	
	1 x RS-232	
Connector	DB-15	
GPIO	Transparent 2-way channel	
DC 222	Transparent 2-way channel or TCP/IP to RS-232	
RS-232	Redirector; 9600 to 115200 bps	
Front Panel		
Level Indicators	2 x 6 LED bar graphs	
GPIO Indicators	2 x 6 LEDs	
Status Indicators	4 LEDs	
Headphones	1/8" (3.5mm) phones jack	
SD Card	microSD card slot	
Remote Control		
Web interface	Built-in web server with HTML5 interface	
SNMP	SNMP 2c	
	CITIVII 25	
Network	D. 45	
Connector	RJ-45	
Туре	10/100Mbps Ethernet	
Discovery	UPnP Support	
Operating cond	itions	
Temperature	-15°C to 55°C	
Humidity	< 95%, non-condensing	
Altitude	0 to 5000m above sea level	
Power requirements		
Connector	Power Jack 5.5mm	
Power supply	12V DC, 1A	
Size and Weigh		
Dimensions		
(W;H;D)	170 x 35 x 167 mm	
Shipping Weight	270 x 54 x 230 mm / 0.9kg	



