

D40 amplifier

The D40 amplifier represents a new performance level of four channel Class D amplifiers using Digital Signal Processing (DSP) to incorporate loudspeaker specific configurations and user definable setups, equalization and delay functions.

The D40 amplifier has a dynamic range of 116 dB (unweighted) and is designed to drive high voltage d&b loudspeakers while providing comprehensive management and protection capabilities.

The user interface of the amplifier consists of a 4.3" (480 x 272 pix.) color touchscreen in conjunction with a digital rotary encoder providing comprehensive information of the device configuration and enhanced status monitoring.

The D40 includes enhanced energy saving features, power efficiency and Automatic Wake up for environmentally responsible and sustainable Green Building requirements.

Powerful signal processing extends the level of functionality of the onboard features. These include a range of loudspeaker-specific filter functions plus two user-definable16-band EQs, which facilitate system tuning. The delay capability covers a range of up to 10 seconds. The DSP unit of the amplifier has a fixed latency of 0.3 ms.

The amplifier enables up to eight input channels and provides four analog inputs as well as four AES3 channels with corresponding link outputs. Each input channel can be routed to any of the output channels.

The D40 amplifier features a flexible Fallback structure, ensuring the transmission of a secondary signal when required.

For applicable d&b loudspeakers, d&b LoadMatch enables the D40 amplifier to preserve tonal balance when cable lengths of up to 70 m (230 ft) are used.

The D40 utilizes a switch mode power supply with automatic mains range selection and active Power Factor Correction (PFC) to produce a clean current draw and ensure stable and efficient performance under adverse mains conditions. Remote control and full system integration are realized using the d&b ArrayCalc simulation software and R1 Remote control software.

The D40 amplifier includes two Ethernet ports (1 Gbit/100 Mbit) on RJ 45 etherCON[®] connectors (PRI/SEC) with the OCA/AES70 protocol incorporated for the upper (PRI) connector socket and star wiring topology.

The bottom (SEC) connector socket is currently disabled.



Operating conditions

Operating temperature (*continuous/**short-term)
10 °C +40*/+50** °C (+14 °F +104*/+122** °F)
Storage temperature20 °C +70 °C (-4 °F +158 °F)
Humidity (rel.), non-condensating

Power supply

Switched mode power supply with automatic mains range selection and active Power Factor Correction (PFC).

Mains connector	powerCON® TRUE 1 TOP
Rated mains voltage (High range)	
Rated mains current (High range)	
Rated mains voltage (Low range)	100 - 127 V, 50 - 60 Hz
Rated mains current (Low range)	

Protection circuits

Mains and power supply: Overvoltage and undervoltage, inrush current limiter, internal fuse.

Output: Overcurrent, DC offset, HF voltage limiter, pop-noise suppression.

Cooling: Temperature-dependent fan, self-resetting overtemperature protection.

Power consumption (typical values)

Standby	
Idling	
Peak output	

Audio power outputs*

SPEAKER OUTPUTS A/B/C/D	4 x NL4
4 CHANNEL OUTPUT	1 x NL8
Maximum output voltage/current	180 $V_{peak}/35 A_{peak}$
Output power rating EIA-426B noise CF 12 dB	4 x 2000 W/8 Ω
	4 x 2400 W/4 Ω
Sine wave 1 kHz, long term, +40 °C (+104 °F).	4 x 250 W/4 Ω
Frequency response (-1 dB, Linear mode)	35 Hz - 25 kHz
Gain (Linear mode @ 0 dB)	31 dB

Output noise/Dynamic range

Output noise (BW 20 kHz)/dynamic range (BW 20 kHz, reference 180 V_u)

/ _{RMS} /111 dB
/ _{RMS} /114 dB
/ _{RMS} /116 dB
/ _{RMS} /119 dB

THD+N / Crosstalk

THD+N (unweighted, 20 - 20 kHz)	
4x 250 W/8 ohms	< -86 dB/0.005 %
4x 250 W/4 ohms	<pre>-83 dB/0.007 %</pre>
Crosstalk (20 – 20 kHz)	
· · · · · · · · · · · · · · · · · · ·	4x 250 W into 8/4 Ω

Analog inputs and outputs

INPUT A1 - A4	3 pin XLR female
Pin assignment	1 = GND, 2 = pos., 3 = neg.
Input impedance	. 32 k Ω , electronically balanced
CMRR @ 100 Hz/1 kHz / 10 kHz	>80/>80/>70 dB
Maximum input level (balanced/unbala	anced)+25/+18 dBu
	+27.3 dBu @ 0 dBFS
LINK A1 - A4, parallel to input	3 pin XLR male
Pin assignment	1 = GND, 2 = pos., 3 = neg.

Digital inputs and outputs

IN - D1/2, D3/4	
Pin assignment	1 = GND, 2 = AES Signal, 3 = AES Signal
Input impedance	110 Ω, transformer balanced
Sampling frequency	
Word length	
OUT - D1/2, D3/4	3 pin XLR male
	electronically balanced
Output modes	. Mains on: analog signal buffering (refresh)

Digital Signal Processing

System start-up time	< 21 sec.
Time to tone (Standby/Ready	Standby)< 3/< 1 sec.
Time to tone (Off/Wake on A	udio)< 21 /< 4 sec.
Conversion	
Latency analog/digital (AES)	input0.3/0.3 msec.
A/D conversion	
Internal processing	Combination of high-resolution fixed point
	and floating point processing
Equalizer	two user definable 16-band equalizers
Filte	r types: PEQ/Notch/HiShlv/LoShlv/Asym
Delay	0.3 msec 10 sec.
Frequency generator	.Pink noise or Sine wave 10 Hz – 20 kHz

Network (PRI/SEC)

Connector type	
PRI	Remote control via R1, Star topology
SEC	Currently disabled



Controls and indicators

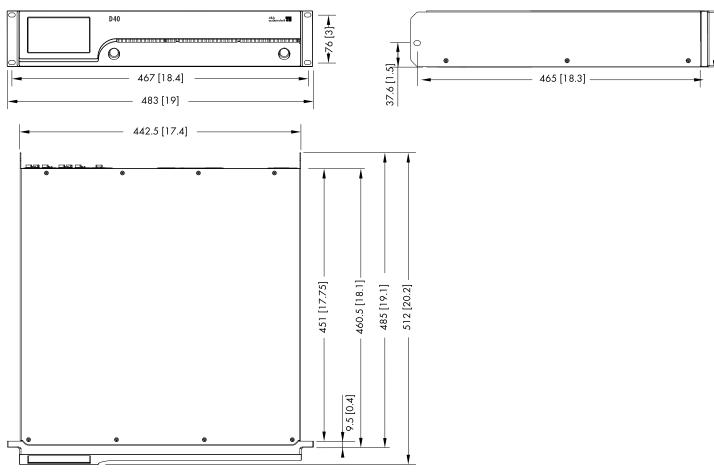
POWER	Mains power switch
SCROLL/EDIT	Digital rotary encoder
TFT color touchscreen	

Fan noise emission

Rack mounted, measured on axis, 1 m (3.3 ft) to from	it panel, A-weighting.
Min./Max. RPM	30/50 dB(A)
Ambient tempero	ature 23 °C/73.4 °F

Dimensions and weight

Height x width x depth	
Weight	13.8 kg/30.4 lb



D40 dimension in mm [inch]

Features and benefits

- High voltage output suitable for driving applicable d&b loudspeakers, including the KSL System
- Increases overall system efficiency
- Advanced voltage management
- Energy-efficient for minimal ecological impact
- Enhanced DSP capabilities accommodates future system architecture

Applications

- Musical theatre rental
- Corporate and industrial rental



Architectural specifications

The amplifier shall be four channel incorporating digital signal processors (DSP) to provide loudspeaker specific configurations and functions and dedicated protection circuits. It shall be equipped with digital and analog signal inputs as well as link outputs, remote control and monitoring capabilities via Ethernet (OCA/AES70). User interface shall be via a 4.3" (480 x 272 pix.) color touchscreen in conjunction with a digital rotary encoder while remote control shall be via dedicated remote control software.

Four analog input and link output connectors shall be provided. Two digital input connectors shall be provided, each accepting a 2 channel digital (AES3) audio signal.

Analog inputs shall be electronically balanced with an input impedance of 32 kOhms.

The digital inputs shall be transformer balanced with an input impedance of 110 ohms while the digital link output shall be electronically balanced providing analog signal buffering (refresh) and power fail relay (Bypass).

Signal processing shall utilize 96 kHz sampling rate while the latency shall not exceed 0.3 msec.

The output connector options shall be 4 x NL4 for Dual Channel mode and 2 x NL4 for Mix TOP/SUB and/or 2-Way Active modes. A four channel out connector (NL8) shall be provided for loudspeaker multicore purposes.

Output configurations shall be selectable for Dual Channel, Mix TOP/SUB and 2-Way Active modes.

It shall incorporate two user definable 16-band equalizers for independent application to each channel allowing parametric filters, notch, hi- and lo- shelf filters as well as asymmetric filters.

A signal delay capability of up to 10 sec. shall be incorporated for independent application to each channel.

It shall contain a signal generator offering pink noise or sine wave program.

Compensation for cable length shall be incorporated to improve impulse response.

Load monitoring and System check functions shall be included to ascertain the status of the loudspeaker impedance. Load monitoring shall allow impedance monitoring to determine the status of an LF or HF driver in systems with multiple elements, even if these are crossed over passively.

Input monitoring shall be included to allow detection of incoming pilot signals.

A Fallback function shall be available to enable the definition of primary (Regular) and secondary (Fallback) signal paths for analog and digital input signals with two different modes (Manual or Auto). It shall ensure that any secondary or emergency signal fed to the Fallback inputs is transmitted when required.

A Override function shall be available to allow a dedicated analog input to be set as a major signal path with highest priority for general messages or emergency services.

An AutoStandby function shall automatically switch the amplifier to Standby mode after a predefined time when the incoming signal level at the individually specified inputs drops below a defined threshold. The function shall be independent of the mute status of the respective channels. An AutoWakeup function shall automatically repower the amplifier, when an input signal is present and exceeds a defined threshold.

A switched mode power supply shall be incorporated and shall allow automatic mains range selection of 100 to 127 V AC and 208 to 240 V AC, 50 - 60 Hz mains power supply voltages.

Active power factor correction (PFC) shall be incorporated to provide a clean and efficient sinusoidal current draw.

Mains voltage monitoring, mains inrush current limiter, self-resetting overtemperature, under- and overvoltage protection shall be incorporated.

It shall have temperature and signal controlled fans for cooling the internal assemblies.

The power amplifier channels shall have ground fault protection, output pop-noise suppression, DC offset protection, output HF voltage limitation, output current limitation/protection and self-resetting overtemperature protection.

The output power shall be 4 x 2000/2400 W into 8/4 ohms (EIA-426-B signal with a crest factor (CF) of 12 dB, all channels driven) while the maximum output voltage shall be at least 180 V_{peak} and the maximum output current shall be 35 A_{peak}. THD+N (20 Hz - 20 kHz) shall be < -86 dB/0.005% and the Crosstalk (20 Hz - 20 kHz) shall be < -70 dBr while the dynamic range (SNR - digital input unweighted) shall be at least 116 dBr.

The dimensions (H x W x D) shall not exceed 2RU x 19" x 465 mm (18.3") and shall weigh no more than 13.8 kg (30.4 lb).

The amplifier shall be the D40 by: d&b audiotechnik GmbH & Co. KG.

