Amplifiers
Software
As the name implies a d&b audiotechnik system is not just a loudspeaker. Nor is it merely a sum of the components: loudspeakers, amplifiers, signal processors, networking, software and accessories. Right from the outset the d&b audiotechnik approach was to build integrated sound reinforcement systems that actually are more than the combination of parts: an entirety where each fits all. Every element is tightly specified, precisely aligned and carefully matched to achieve maximum efficiency. For ease of use, all the user-definable parameters are incorporated, allowing the possibility of adjustment, either directly, via remote control surfaces, or integrated within wider networks. Neutral sound characteristics leave the user all the freedom needed to realize whatever the brief. At the same time d&b offers finance, service and support, a knowledgeable distribution network, education and training as well as technical information, so the same optimal acoustic result is achieved consistently by every system anywhere, at any time. In reality: the d&b System reality.
The integrated d&b Workflow improves efficiency all the way from the start of a project through planning and simulation to control of the final result. In the first stage of the d&b Workflow, a 3D model of a venue is created using d&b ArrayCalc, where loudspeakers can be virtually placed, positioned, and configured. When this is completed the system performance can be simulated. Once the mechanical settings of a line array have been finalized the optional ArrayProcessing function within ArrayCalc can be applied. ArrayProcessing uses powerful filter algorithms to optimize the level and tonal balance of a line array across the entire audience area. Once the desired performance has been achieved, the amplifiers can be configured and signal routing defined. ArrayCalc then generates rigging plans, parts lists, and graphics for use in the final proposal. In the next stage of the d&b Workflow, the file built by ArrayCalc can be opened by the R1 Remote control software where a graphical user interface for the complete system is automatically generated. It is now that all of the settings defined in ArrayCalc can be sent to the amplifiers, and signal distribution configuration can be sent to Signal Engines or Audio Network Bridges.
The d&b ArrayCalc simulation software

The d&b ArrayCalc simulation software is the prediction tool for d&b systems. This is a comprehensive toolbox for all tasks associated with acoustic design, performance prediction, alignment, rigging and safety parameters. For safety reasons d&b line arrays must be designed using the d&b ArrayCalc simulation software. d&b ArrayCalc is available as a native stand-alone application for both Microsoft Windows 1 (Win7 or later) and Mac OS X 2 (10.12 or later) operating systems and is available at www.dbaudio.com, along with further information and video tutorials.

Simulation

Listening planes can be defined in the venue tab, creating a three dimensional representation of any audience area in a given venue. Special functions assist in obtaining accurate dimensions with laser distance finders and inclinometers. Flown arrays or subwoofer columns can be defined in a project file as single hangs or in pairs. Point source loudspeakers can also be fully integrated as well as a ground stacked SUB arrays. Level over distance is calculated for each source with high resolution in real-time, for either band limited or broadband input signals. The comprehensive simulation precisely models the actual performance of the system, taking into account input level, all system configuration options (such as CUF, CPL, HFC or INFRA), limiter headroom and air absorption. Acoustic obstacles, such as video screens, can be added to a model. Acoustic shadowing, whether by these obstacles, or a balcony overhang, is taken into consideration. The load status of all array rigging components is calculated accurately and displayed to determine whether a given array is within the load tolerances. Subwoofer array design is assisted by coverage and polar plot prediction. A specialized algorithm allows the user to specify subwoofer positions and a coverage angle, which is then converted into appropriate delay settings that result in the desired dispersion. The alignment tab enables different sources to be time aligned to one another, as well as showing arrival times and Sound Pressure Levels at a definable reference point on one of the audience areas. For alignment of the flown system with the ground stacked SUB array, the phase response of both the SUB array and a flown source is calculated at a definable reference point. The level distribution resulting from the interaction of all active sources can be mapped onto the audience areas in a three-dimensional view. EASE and DXF data export capabilities are also available.

System setup

A rigging plot with all necessary coordinates, dimensions and weights of arrays is generated for export and printing and a parts list, detailing all components required. The d&b ArrayCalc Viewer app presents this key information for positioning and

ArrayProcessing

The optional ArrayProcessing function applies powerful filter algorithms to optimize the tonal (spectral) and level (spatial) performance of a line array column over the audience area defined by its mechanical vertical coverage angle. Within the d&b ArrayCalc simulation software, spectral and level performance targets over the listening areas can be defined while specific level drops or offsets can be applied to certain areas, to assign reduced level zones. ArrayProcessing applies a combination of FIR and IIR filters to each individual cabinet in an array to achieve the targeted performance, with an additional latency of only 5.9 ms. This significantly improves the linearity of the response over distance as well as seamlessly correcting for air absorption. In addition, ArrayProcessing employs the same frequency response targets for all d&b line arrays, to ensure all systems share a common tonality. This provides consistent sonic results regardless of array length or splay settings. The resulting coverage is enhanced with spectral consistency and defined level drops while maintaining the overall system directivity to cover longer distances or steep listening areas effectively.

d&b Soundscape

The d&b ArrayCalc Software is also used for planning and simulation of d&b Soundscape. An Early reflections area can be defined in the Venue view, for which the En-Space software will calculate additional convolutions to better represent the early reflections of the original venue. Additionally Positioning areas can also be defined for use in combination with the En-Scene software, which are used as a reference plane for positioning Sound objects either from R1 or via external controllers. The d&b Soundscape loudspeaker system is also planned, configured and simulated in ArrayCalc, with loudspeakers assigned to Function groups, for which the Soundscape algorithms calculate individual transfer functions according to their function within the system. DS100 Signal Engine devices can be added to a project, together with the necessary DS10 Audio network bridges. Signal routing can be defined for the whole signal path, from DS100 to loudspeaker, through the DS10 and amplifier devices. A Dante Controller preset file can be automatically generated in ArrayCalc, including the complete Dante routing for the whole system. This preset file can be later loaded in Dante Controller, so that no manual Dante patching is required.
The d&b NoizCalc immission modelling software

The d&b software uses international standards to model the far field noise immission from multiple complex and coherently emitting sources such as line arrays and subwoofer arrays. More and more, gaining permission and licenses to stage live open air events requires an official statement with a prediction of how noise could impact on the surrounding area. NoizCalc takes all complex loudspeaker data and a reference point from the d&b ArrayCalc simulation software and calculates the sound propagation and relative attenuation values towards the far field for a certain scenario with particular meteorological conditions for one or more d&b loudspeaker systems.

A 3D terrain map imported from Google Maps or Street View displays the calculated immission in the areas surrounding the audience listening zones. This visual representation shows the actual system performance in the far field, enabling users to optimize for listeners while satisfying local noise restrictions and affiliate regulations.

To ensure reliable results, NoizCalc includes all complex data concerning the addition and subtraction of sound waves, including phase information to describe the combination and interaction effects within a loudspeaker system consisting of multiple line arrays, subwoofer arrays and delay systems.

NoizCalc models immissions in the far field according to the internationally accepted ISO 9613-2 or Nord2000 calculation standards. Ground characteristics can be set depending on the absorbency or reflectivity of surfaces, while areas with volume attenuating properties can be defined. Buildings can be included, and the maximum reflection order option adjusts how many reflections are calculated. Parameters for humidity, air pressure and temperature ensure that the correct air absorption figures are accounted. The ISO 9613-2 standard requires limited meteorological information and assumes a worst-case scenario.

The more sophisticated propagation model, Nord2000 enables a more precise handling of meteorological conditions allowing the user to model with prevailing wind information. The d&b NoizCalc immission modelling software is available at www.dbaudio.com for registered download, along with further information and video tutorials. It was developed in collaboration with SoundPLAN, a specialist software developer for environmental noise prediction.

Calculation

All complex loudspeaker data and the reference point is taken from ArrayCalc. NoizCalc then displays the modelled effects of the sound propagation and the resulting immission from the sound reinforcement system in the far field.

Modelling

Using 3D geographical data, mitigation areas such as woodland and solid obstacles can be added and modelled. NoizCalc displays the immission on a terrain map, presenting the calculated Sound Pressure Levels in dBA applying the selected frequency spectrum using either the ISO 9613-2 or Nord2000 standards. If Nord2000 is selected, additional meteorological data including wind direction, wind speed and temperature gradients can be included in the calculation.

Optimization

NoizCalc is intended to optimize the planning and design of a loudspeaker system. Any potential noise issues can be addressed by virtually modifying the system design, stage orientation or system settings, to achieve remarkable results for the audience with full consideration for the far field.

Monitoring

The results map displays the calculation in the far field according to the selected standard and associated parameters. Additional time histogram prediction and monitoring positions may still be required for official purposes. The NoizCalc results map will show the sound propagation and attenuation over distance. The system technician can then monitor the level at the reference point and assess the actual results at the monitoring positions while adapting specific meteorological or spectral variations. Comparing the calculated results with the actual measurement at the reference point will indicate how the system should be adjusted to meet offsite noise restrictions.

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The d&b R1 Remote control software

The remote control capability of the d&b Remote network enables central control and monitoring of a complete d&b loudspeaker system from anywhere in the network, be it from a laptop in the control room, at the mix position, or on a wireless tablet computer in the auditorium. In the typical user workflow, the d&b Remote network takes settings optimized in the ArrayCalc simulation software and applies these to all the d&b devices within the network. In mobile situations R1 Remote control software provides extensive functionalities for storing and recalling system settings, enabling setups to be repeated as and when required. Project files can be adjusted for use with different equipment at another location. d&b System check verifies that the system performs within a predefined condition, while the Array verification function automatically identifies the physical position of a loudspeaker in an array to check that the system is cabled correctly. For permanent installations, system integrators can define settings that are correctly configured. When grouped together, a button or fader enables central control and monitoring of a complete d&b system. The remote control capability of the d&b Remote network takes settings optimized in the ArrayCalc simulation software and applies these to all the d&b devices within the network. In mobile situations R1 Remote control software provides extensive functionalities for storing and recalling system settings, enabling setups to be repeated as and when required. Project files can be adjusted for use with different equipment at another location. d&b System check verifies that the system performs within a predefined condition, while the Array verification function automatically identifies the physical position of a loudspeaker in an array to check that the system is cabled correctly. For permanent installations, system integrators can define settings that are correctly configured. When grouped together, a button or fader enables central control and monitoring of a complete d&b system.

R1 Remote control software

The R1 Remote control software uses the same project file created in ArrayCalc to automatically generate an intuitive graphical user interface including complete details of the simulated system. This user interface can be customized by the user, in order to fit particular user needs. Changes to the project file can be performed at any time in both ArrayCalc and R1. This workflow removes the need to manually transfer data from one software program to the other. All features, functions and controls from d&b devices can be remotely controlled and/or monitored using R1 Remote control software. It allows each channel of the amplifiers as well as loudspeakers specific function switches, such as CUT/HFA/HFC, CIP, and Array Processing. An offline mode is provided for preparation in advance of an event, without the need for physical devices being present or connected. The Home view provides an overview of all views in R1 and access to all user defined remote views. Each user definable Remote view can be populated with control functions of the system and can be optimized for different screen resolutions, either for large monitors or for smaller tablet devices.

DS10 and DS20 Meta data

The DS10 and DS20 Audio network bridge sends Meta data to the d&b amplifiers via the AES32 channel streams. This information can be read remotely from R1, and includes channel subscription information, such as channel labels and network and redundancy status. Additionally, cabling information is also provided, which makes it possible to find possible cabling mistakes in the audio signal chain.

d&b Soundscape

After system planning and simulation in ArrayCalc, the d&b Soundscape is completely configured, monitored and controlled in R1. The DS100 Signal Engine matrix can be operated manually by the controls on the Device view tabs or, alternatively, by controls in a Remote view. The Matrix crosspoint control provides an array of level and delay controls for a user defined range of matrix crosspoints. Input and Output processing is also available, with a combination of Polarity, Level and Delay available on all inputs and outputs. Additionally, an 8-band parametric EQ is provided on all inputs, and a 16-band parametric EQ on all outputs. Special Positioning remote views can be created for En-Scene systems, in which Sound objects can be individually placed within the venue. En-Space sampled spaces can be selected within R1 from the available library provided with the En-Space software license. En-Space levels for inputs and outputs can also be controlled within the Device or Remote views, and additional EQ for each of the four zones of the space is provided.

Service functions

R1 enables the simultaneous firmware update of multiple devices from a central location. The software will automatically search the d&b website and on demand, downloads the latest available firmware versions and R1 Remote control software updates. Defined settings can be created, saved on a computer and loaded into the devices, for example to ensure that configuration switches are set to a known status, or the user definable equalization is set flat. Settings can be copied to additional or spare devices. For service purposes, information may be read from a device, concerning its condition during operation and errors reported. When additional support is required, the error report can be saved and sent to the d&b service departments for further assessment and diagnosis.

1 Microsoft Windows is a registered trademark or trademark of Microsoft Corporation in the United States and/or other countries
2 Mac OS X is a trademark of Apple Inc., registered in the U.S. and other countries
The d&b Amplifiers and Software

The D20 amplifier and D80 amplifier can be remote controlled using the industry standard OCA protocol via Ethernet and through the established CAN-Bus, which can also control the D6 amplifier as well. d&b amplifiers are controlled using the d&b R1 Remote control software, which is available on both Windows and MAC operating systems. This Remote user interface can control the D20 and D80 amplifiers via Ethernet (OCA) and the D6 amplifier through CAN-Bus networks simultaneously. The D20 and D80 amplifiers also provide a web interface, offering access to individual D20 and D80 amplifiers through a web browser.

d&b Remote network – OCA via Ethernet
The D20 and D80 amplifiers can be remotely controlled via a standard Ethernet network, providing higher bandwidth and quicker response compared to the CAN-Bus network. This latest generation of d&b amplifiers are fitted with two etherCON® connectors, allowing simple networks to be set up, without requiring an extra switch. The industry standard Open Control Architecture (OCA) protocol is used, created by the OCA Alliance of which d&b is a founding member. For further information please refer to the d&b TI 310 Ethernet networking, which is available for download at www.dbaudio.com.

d&b Remote network – CAN-Bus
The d&b amplifiers can be integrated within the CAN-Bus network and are fitted with two REMOTE/CAN connectors (RJ 45) to link the CAN-Bus signal and enable daisy chaining. The network may contain any combination of up to a total of 504 devices. It is connected to a PC or MAC running R1 Remote control software V2, using R60 USB to CAN, or R70 Ethernet to CAN interfaces. While the CAN-Bus network covers distances up to 600 m the Ethernet connection to the R70 can be made using standard Ethernet technologies, including wireless or fibre optic networks. For further information please refer to the TI 312 d&b Remote network, which is available for download at www.dbaudio.com.
The DS10 and DS20 Audio network bridges

DS10 Audio network bridge
The DS10 Audio network bridge interfaces between Dante networks and AES3 digital audio signals, while also providing distribution of Ethernet control data. Positioned within the signal chain in front of the amplifiers, this 1 RU device expands the d&b system approach. Each unit can deliver up to sixteen Dante network channels via AES3 digital signal outputs. Additionally, four AES3 input channels provide access to the Dante audio network for applications such as a break-in from a Front of House console. The DS10 incorporates an integrated 5-port switch, offering a primary and redundant network for the Dante protocol, as well as advanced functions such as Multicast Filtering and VLAN modes. Using the DS10 Audio network bridge, audio signals and remote control data can be combined using a single Ethernet cable.

DS20 Audio network bridge
The DS20 Audio network bridge supports the open standards-based Milan protocol rather than Dante. Milan (Media integrated local area networking) is a high level interoperability solution based on Audio Video Bridging (AVB) technology. The main advantages are deterministic behaviour (zero network congestion); improved reliability; optimum synchronization and hassle free network setup, as no special settings, such as QoS, need to be set within the switches to ensure delivery.

The DS100 Signal Engine

The DS10 Audio network bridge front view
The DS10 Audio network bridge rear view
The DS20 Audio network bridge front view
The DS20 Audio network bridge rear view
The DS100 Signal Engine front view
The DS100 Signal Engine rear view

The d&b Soundscape
The d&b Soundscape is the acoustic environment in context, as perceived, experienced and understood. It defines acoustic environments in which people feel performances, speeches and presentations, real or abstract. It introduces another dimension to the d&b system approach, aligning the aural and visual perception, either realistically, or imaginatively. The d&b Soundscape also includes functionality which can modify the acoustic environment by imposing a different reverberation signature in the audience area, whether inside or outside.

DS100 Signal Engine
The DS100 Signal Engine is the platform underneath the Soundscape, based on a specialized rack mount 3 RU audio processor with Audinate Dante networking. It provides a 64 x 64 audio matrix with level and delay adjustments at all cross points.

The DS100 is a versatile tool for use within complex audio systems to route and distribute multiple audio channels to numerous amplifiers driving loudspeaker positions and zones, show relay and break out rooms. The networking capabilities with a Dante enabled processor are significant, particularly for busy, multi-room complexes.

The DS100 completely integrates with the overall d&b system approach, including loudspeakers, amplifiers, rigging, transport and networking accessories and the DS10 Audio network bridge, which interfaces between Dante audio networking and the AES3 inputs of the d&b amplifiers. The DS100 and DS10 both operate on standard networking technologies, making them flexible and efficient. The complete system is designed and optimized in the d&b ArrayCalc simulation software, and controlled via the d&b R1 Remote control software.

The comprehensive input processing provides gain, EQ, delay and polarity switches, enabling the user to combine all types of input signals to create a mix from a wide variety of sources. Extended processing capabilities are also provided on every output.
The d&b amplifiers are designed specifically to power d&b loudspeakers and are the beating heart of the d&b System Reality. As such, they incorporate Digital Signal Processing for comprehensive loudspeaker management, switchable filter functions, remote capabilities and user-definable controls, to fulfill the exact needs of each application.

Every loudspeaker configuration combines comprehensive system limiting, equalization and crossover settings to ensure consistent results and optimal performance. d&b amplifiers offer different output configurations for different loudspeaker setups, including Dual Channel mode, for passive setups, Mix TOP/SUB mode, in which two channels are driven through a single output connector, and 2-Way Active mode, which also sends the output of two channels down one connector to drive appropriate loudspeakers actively.

The d&b switch functions provide selected filters to precisely tailor a wide variety of setups to their applications. Examples of these switch functions are the CSA (Cardioid Subwoofer Array) and HFC (High Frequency Compensation) modes. CSA increases low frequency directivity control by minimising energy transmission towards the rear while HFC compensates for air absorption for loudspeakers covering far field listening positions. In addition to these functions, d&b amplifiers offer a comprehensive set of specific filters such as CUT, a cut mode for TOP loudspeakers when used with d&b subwoofers; CPL, to compensate for the coupling effect between loudspeakers in close proximity to other loudspeakers or hard objects and HFA mode, to attenuate the high frequencies of a loudspeaker to mimic the effect of far field listening. d&b amplifiers offer extended, user-definable equalization and delay capabilities, eliminating the need for external processing devices in the signal chain.

Sophisticated protection circuits modelling thermal and mechanical driver behaviour are provided, resulting in the sustained reliability of d&b systems. These amplifiers also have the functionality to enable system status monitoring and protection features, increasing the longevity of d&b systems. These features are designed for incorporation in applications specified to the requirements of sound systems for emergency purposes. A password protected LOCK function prevents unauthorized changes. The AutoStandby option automatically changes the D20 and D80 amplifiers to Standby mode after a predefined time without any signal to individually specified inputs. The AutoMakeup function reactivates the amplifier when an input signal is active.

A powerCON® mains connector socket is fitted on the rear panel. The switched mode power supply of each amplifier incorporates mains overvoltage protection, inrush current limiting and loudspeaker protection at start up. Temperature and signal controlled fans cool the internal assemblies. d&b amplifiers offer analog and digital AES3 signal inputs, with link outputs for each channel. The AES3 link output carries a refreshed signal, while a power fail relay is incorporated to prevent interruption of the signal chain, in the event of a power failure. All d&b amplifiers integrate with the d&b Remote network to enable the remote control and management of systems from anywhere within a network. Firmware updates containing new loudspeaker configurations or additional functions can be loaded to the amplifiers via the d&b Remote network.

For permanently installed applications, the d&b 10D and 30D amplifiers are recommended. Further information is available in the d&b xD Installation Amplifier and Software brochure, which is available to download from www.dbaudio.com.
The d&b amplifiers

Input routing
The D20 amplifier and D80 amplifier provide four analog inputs, four digital AES3 channels and analog and digital link outputs, all on XLR connectors. The digital input pairs contain independent Sample Rate Converters (SRC). Each input features a separate input gain, meaning sources with various input sensitivities can be used without requiring an external mixing device. Individual inputs can be summed and routed to any of the four outputs.

System status monitoring
d&b amplifiers incorporate specific functions to check, monitor and control a d&b system. The System check feature can verify the status of a complete reinforcement system by measuring the impedance of the connected loudspeakers. Each driver in the system is checked and compared to the default impedance values for each particular cabinet. This calibration ensures that each loudspeaker performs within a tolerance band, and indicates possible damage to system components. The Input monitoring function can detect an incoming pilot signal regardless of whether this input is routed to an output. The Load monitoring function can detect a possible loudspeaker malfunction. This feature automatically checks load impedance to calculate individual driver status and report any failure. The Fallback and Override features within the D20 and D80 amplifiers ensure that any secondary or emergency signal is transmitted when required. The Fallback function ensures that the device will switch from the program material to an alternative source via a different input. The Override feature ensures that any general message or emergency signal takes priority. Fallback and Override can be triggered manually or automatically. These functions ensure the d&b amplifiers are suitable for applications specified to the requirements of evacuation systems for emergency purposes.

Integrated Web Remote interface
The D20 and D80 amplifiers can be controlled via the integrated web interface, which enables the Remote control of a single device using a browser window. Once the amplifier is connected to the d&b Remote network, an intuitive user interface becomes available after the IP address of the device is entered into the address bar of a web browser. Each amplifier in a system is managed individually, meaning that a new window or tab will be required for each device. The amplifiers can be accessed using a mobile device if the network features a wireless access point.
## The d&b amplifiers

### Comparison of the d&b amplifiers

<table>
<thead>
<tr>
<th></th>
<th>D6</th>
<th>D20</th>
<th>D80</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>User interface</strong></td>
<td>Encoder/SC display</td>
<td>Encoder/colour TFT touchscreen</td>
<td>Encoder/colour TFT touchscreen</td>
</tr>
<tr>
<td><strong>Output channels</strong></td>
<td>2</td>
<td>4</td>
<td>4</td>
</tr>
<tr>
<td><strong>Input channels</strong></td>
<td>2 x AES3 or 2 x analog</td>
<td>4 x AES3 or 4 x analog or 2 x AES3 and 2 x analog</td>
<td>4 x AES3 or 4 x analog or 2 x AES3 and 2 x analog</td>
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<tr>
<td><strong>Latency</strong></td>
<td>0.3 msec</td>
<td>0.3 msec</td>
<td>0.3 msec</td>
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<tr>
<td><strong>User equalizers (per channel)</strong></td>
<td>4-band</td>
<td>2 x 16-band</td>
<td>2 x 16-band</td>
</tr>
<tr>
<td><strong>Delay</strong></td>
<td>340 msec/116.9 m</td>
<td>10 sec/3440 m</td>
<td>10 sec/3440 m</td>
</tr>
<tr>
<td><strong>Rated output power (THD+N &lt; 0.5%, 12 dB crest factor)</strong></td>
<td>2 x 350 W into 8 ohms</td>
<td>4 x 800 W into 8 ohms</td>
<td>4 x 2000 W into 8 ohms</td>
</tr>
<tr>
<td></td>
<td>2 x 600 W into 4 ohms</td>
<td>4 x 1600 W into 4 ohms</td>
<td>4 x 4000 W into 4 ohms</td>
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<tr>
<td><strong>Output routing</strong></td>
<td>Dual Channel w/o B1 and B2</td>
<td>Dual Channel, Mix TOP/SUB 2-Way Active</td>
<td>Dual Channel, Mix TOP/SUB 2-Way Active</td>
</tr>
<tr>
<td><strong>Output connectors</strong></td>
<td>NL4</td>
<td>NL4 plus central NL8</td>
<td>NL4/EP5 plus central NL8</td>
</tr>
<tr>
<td><strong>Cable compensation</strong></td>
<td>None</td>
<td>LoadMatch</td>
<td>LoadMatch</td>
</tr>
<tr>
<td><strong>Power supply</strong></td>
<td>Autonomous switched mode power supply with active PFC</td>
<td>Universal range switched mode power supply with active PFC</td>
<td>Autonomous switched mode power supply with active PFC</td>
</tr>
<tr>
<td><strong>Mains voltage</strong></td>
<td>100 - 120/220 - 240, 50 - 60 Hz</td>
<td>100 - 240 V, 50 - 60 Hz</td>
<td>100 - 127/208 - 240 V, 50 - 60 Hz</td>
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<tr>
<td><strong>Weight (kg/lb)</strong></td>
<td>8/17.6</td>
<td>10.8/23.8</td>
<td>19/42</td>
</tr>
<tr>
<td><strong>Dimensions</strong></td>
<td>2 RU x 19&quot; x 353 mm</td>
<td>2 RU x 19&quot; x 460 mm</td>
<td>2 RU x 19&quot; x 330 mm</td>
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<tr>
<td><strong>Remote</strong></td>
<td>CAN</td>
<td>OCA via Ethernet/CAN</td>
<td>OCA via Ethernet/CAN</td>
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<tr>
<td><strong>Airflow</strong></td>
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<td><img src="image2.png" alt="Image" /></td>
<td><img src="image3.png" alt="Image" /></td>
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</table>
The D6 amplifier

The 2 RU two channel lightweight D6 amplifier delivers medium power into low impedance loads between 4 and 16 ohms and is ideally suited for use in both mobile and installation environments. The D6 contains setups for d&b loudspeakers and a linear mode; exceptions are 2-Way Active mode, V-Series and B22-SUB. The signal delay capability enables user definable settings of up to 340 msec (=100 m/328 ft) to be applied independently to each channel. The same applies to the 4-band parametric equalizer, providing optional boost/cut or notch filtering.

The D6 incorporates a digital rotary encoder and a LC display to configure the amplifier. The D6 incorporates Class D amplifiers utilizing a switched mode power supply with active Power Factor Correction (PFC), suitable for mains voltages 100 V/120 V/200 V/240 V, 50 - 60 Hz and maintains a stable output when used with weak or unstable mains supplies. It is supplied with two NL4 loudspeaker output connectors.

**Control and indicators**
- **POWER**: Mains power switch
- **SCROLL/EDIT**: Digital rotary encoder
- **Display**: Liquid Crystal Display (LCD)/120 x 32 pixel
- **ISP, GR, OVL A/B**: LED indicators

**Digital Signal Processing**
- **Equalizer**: 4-band PEQ/Notch
- **Latency analog and digital inputs**: 0.3 msec
- **Delay setting**: 0.3 - 340 msec with 0.1 msec detents
- **Configurations**: current d&b loudspeakers and linear mode
- **Function switches**: d&b loudspeaker specific circuits
- **Frequency generator**: Pink noise or Sine wave
- **Sampling rate**: 96 kHz/27 Bit ADC/24 Bit DAC

**Connectors**
- **INPUT ANALOG (A1, A2)**: 3 pin XLR female
- **ANALOG LINK (A1, A2)**: 3 pin XLR male
- **INPUT DIGITAL (D1, D2)**: 3 pin XLR female AES3
- **DIGITAL LINK (Output)**: 3 pin XLR male
- **Sampling rate**: 48 kHz/96 kHz
- **OUT CHANNEL A/B**: NL4
- **REMOTE**: 2 x RJ 45 parallel, CAN-Bus
- **SERVICE**: USB type B
- **Mains connector**: powerCON®

**Data (linear setting with subsonic filter)**
- **Maximum output power per channel (THD + N < 0.5 %, both channels driven)**
  - CF = 6 dB at 4/8 ohms: 2 x 600/350 W
  - CF = 12 dB at 4/8 ohms: 2 x 600/350 W
- **S/N ratio (unweighted, RMS)**: > 110 dB

**Power supply**
- **Auto-sensing switched mode power supply with active Power Factor Correction (PFC)**
- **Rated mains voltage**
  - High range: 208 - 240 V, 50 - 60 Hz
  - Low range: 100 - 127 V, 50 - 60 Hz

**Dimensions, weight**
- **Height x width x depth**: 2 RU x 19” x 353 mm/13.9”
- **Weight**: 8 kg (17.6 lb)
The D20 amplifier

The 2 RU four channel D20 amplifier is ideally suited to mobile applications which require low to medium Sound Pressure Level (SPL) capabilities. The D20 features the same Digital Signal Processing (DSP) platform as the latest generation of d&b amplifiers, containing configurations for all d&b loudspeakers except the 2-Way Active loudspeakers, and a linear mode. The signal delay capability enables user definable settings of up to 10 s (= 3440 m/11286 ft), which can be applied independently to each channel. The same applies to the two 16-band equalizers, providing optional parametric, asymmetric, shelving or notch filtering. The D20 incorporates a colour TFT touchscreen, offering quick access to the menu structure, while the rotary encoder can be used for fine adjustment. The front panel and the integrated touchscreen are tilted up for easy operation when the amplifier is below eye level. The equal ratio of signal input to amplifier output channels increases application flexibility particularly for monitor, frontfill or effect channel use. The LoadMatch function integrated within the D20 amplifier electrically compensates for the properties of loudspeaker cable used. The D20 incorporates Class D amplifiers utilizing a power supply with active Power Factor Correction (PFC) suitable for mains voltages 100 V - 240 V, 50 - 60 Hz and maintains a stable output when used with weak or unstable mains supplies. An NL8 provides all outputs on a single connector for loudspeaker multicores. Dual channel mode is driven through four individual NL4 connectors, while Mix TOP/SUB and 2-Way Active mode output configurations for applicable d&b loudspeakers are available via two dedicated NL4 connectors.

Control and indicators

- POWER - Mains power switch
- SCROLL/EDIT - Digital rotary encoder
- Display - Colour TFT touchscreen, 3.5”/320 x 240 pixel

Digital Signal Processing

- Equalizer - 2 x 16-band PEQ/notch/shelving/asymmetric
- Latency - analog and digital inputs - 0.3 msec
- Delay setting - 0.3 - 10000 msec
- Configurations - current d&b loudspeakers and linear mode except 2-Way Active loudspeakers
- Function switches - d&b loudspeaker specific circuits
- Frequency generator - Pink noise or Sine wave
- Sampling rate - 96 kHz/27 Bit ADC/24 Bit DAC

Connectors

- ANALOG LINK [A1 - A4] - 3 pin XLR male
- INPUT DIGITAL [D1/2, D3/4] - 3 pin XLR female AES3
- DIGITAL LINK (Output) - 3 pin XLR male
- SPEAKER OUTPUTS A/B/C/D - NL4
- Mix TOP/SUB/2-Way Active output A/B and C/D - NL4
- 4 CHANNEL OUTPUT - NL8
- CAN - 2 x RJ 45 parallel
- ETHERNET - 2 x etherCON®, 10/100 Mbit Ethernet

Data (linear setting with subsonic filter)

- Maximum output power per channel (THD + N < 0.5 %, all channels driven) -
  - CF = 6 dB at 4/8 ohms - 4 x 1000/800 W
  - CF = 12 dB at 4/8 ohms - 4 x 1600/800 W
- S/N ratio (unweighted, RMS) - > 104 dB
- Analog input - > 104 dB
- Digital input - > 106 dB

Power supply

Universal range switched mode power supply with active Power Factor Correction (PFC)

Rated mains voltage - 100 - 240 V, 50 - 60 Hz

Dimensions, weight

- Height x width x depth - 2 RU x 19” x 460 mm/18.1”
- Weight - 10.8 kg (23.8 lb)
The 2 RU four channel D80 amplifier is a high power density amplifier, ideally suited for use in both mobile and installation environments. It contains setups for all d&b loudspeakers and a linear mode. The signal delay capability enables user definable settings of up to 10 s (= 3440 m/11286 ft) to be applied independently to each channel. The same applies to the two 16-band equalizers, providing optional parametric, asymmetric, shelving or notch filtering. The R1 Remote control section of this brochure gives a full explanation of the equalization section of the D80 amplifier.

The D80 incorporates a colour TFT touchscreen, offering quick access to the menu structure, while the rotary encoder can be used for fine adjustment. The front panel and the integrated touchscreen of the D80 amplifier is tilted up for ease of operation when the amplifier is below eye level.

The equal ratio of signal input to amplifier output channels increases the application flexibility particularly for monitor, frontfill or effect channel use. The LoadMatch function integrated within the D80 amplifier electrically compensates for the properties of loudspeaker cable used. The D80 incorporates Class D amplifiers utilizing a switched mode power supply with active Power Factor Correction (PFC) suitable for mains voltages 100 V/127 V, 50 - 60 Hz and 208 V/240 V, 50 - 60 Hz and maintains a stable output when used with weak or unstable mains supplies. An NL8 provides all outputs on a single connector, while individual outputs are optionally NL4 for EPS connectors.

### Control and indicators
- POWER: Mains power switch
- SCROLL/EDIT: Digital rotary encoder
- Display: Colour TFT touchscreen, 3.5”/320 x 240 pixel

### Digital Signal Processing
- Equalizer: 2 x 16-band PEQ/notch/shelving/asymmetric
- Latency analog and digital inputs: 0.3 msec
- Delay setting: 0.3 - 10000 msec
- Configurations: current d&b loudspeakers and linear mode
- Function switches: d&b loudspeaker specific circuits
- Frequency generator: Pink noise or Sine wave
- Sampling rate: 96 kHz/27 Bit ADC/24 Bit DAC

### Connectors
- INPUT ANALOG (A1 - A4): 3 pin XLR female
- ANALOG LINK (A1 - A4): 3 pin XLR male
- INPUT DIGITAL (D1/2, D3/4): 3 pin XLR female AES3
- DIGITAL LINK (Output): 3 pin XLR male
- SPEAKER OUTPUTS A/B/C/D: NL4, optional EPS
- 4 CHANNEL OUTPUT: NL8
- CAN: 2 x RJ 45 parallel
- ETHERNET: 2 x etherCON1, 10/100 Mbit Ethernet
- PowerCON-HC1

### Data (linear setting with subsonic filter)
- Maximum output power per channel (THD + N < 0.5 %, all channels driven):
  - CF = 6 dB at 4/8 ohms: 4 x 2600/2000 W
  - CF = 12 dB at 4/8 ohms: 4 x 4000/2000 W
- S/N ratio (unweighted, RMS): > 110 dBr
- Analog input: > 114 dBr
- Digital input: > 114 dBr

### Power supply
- Autosensing switched mode power supply with active Power Factor Correction (PFC)
- Rated mains voltage: High range: 208 - 240 V, 50 - 60 Hz
- Low range: 100 - 127 V, 50 - 60 Hz

### Dimensions, weight
- Height x width x depth: 2 RU x 19” x 530.5 mm/20.9”
- Weight: 19 kg (42 lb)
The D20 and D80 Touring rack assemblies are designed as fully equipped and prewired system racks, providing mains power distribution, connector interfaces and all internal cabling for three D20 or D80 amplifiers. They are both equipped with a 32 A CEE mains power connector, a mains distribution device with a 32 A mains link, and a loudspeaker connector panel.

All internal audio and remote connections are fully prewired. A loom comprising two AES3 cables and one CAT5/etherCON® is supplied, providing patch cabling for a network connection and four AES3 channels. The D20 and D80 Touring rack assemblies are provided with either a DS10 or DS20 Audio network bridge, which comes supplied with all necessary cabling, or a 1 RU drawer.

The 10 RU Touring racks house a 19” internal shockmount steel frame accommodating three D20 or D80 amplifiers and the requisite connection panels as detailed on the next page. They feature two sliding doors, a perspex window at the front, four wheels, six handles and recessed stacking moulds at the top.

The DS10 Audio network bridge and DS20 Audio network bridge provide 16 AES3 outputs via Dante or Milan transport protocols respectively, and are intended to integrate directly into a D80 Touring rack assembly. The front panels of the audio network bridges are designed to align with the input panel of the D80 Touring rack assembly. The DS10 and DS20 can be powered by one of the auxiliary mains outputs. All necessary patch cabling and links are supplied with a D80 Touring rack assembly including a DS10 or DS20.

The I/O panel provides the input connectors of the first amplifier while the other two amplifiers are linked within the rack. The INPUT section allows both analog and digital audio signals to be fed while the INPUT LINK section provides the link output connectors of the last amplifier. The REMOTE section allows the daisy chaining of system racks within a remote network using the enclosed rack link cable. ETH 1 provides the upper etherCON® connector of the first amplifier, ETH 2 provides the bottom etherCON® connector of the last amplifier.

CAN input provides the CAN input of the first amplifier while the other two amplifiers are linked within the rack. The last CAN-Bus device of a CAN-Bus segment can be terminated by the TERMINATE switch.

The Mains power distributor panels are designed and dimensioned to provide and distribute the mains power supply necessary for the three D20/D80 amplifiers. They also serve as a loudspeaker connection panel. The D20 Touring rack features three NL8 sockets to allow connection to a total of twelve amplifier channels. The D80 Touring rack includes three NL8 and one additional LKA25 multicore connectors. Three electrically interlocked auxiliary mains outputs (powerCON® 16 A sockets) are provided. They are intended for the connection of low current devices such as notebooks or additional Ethernet switches. Three phase mains indicators are provided.
The 3 x D20 and 3 x D80 Touring rack assemblies

3 x D20 Touring rack

Z5560.xxx
Total weight including amplifiers:
98.5 kg [217 lb]

Z5561.xxx
Total weight including amplifiers:
97.5 kg [215 lb]

3 x D80 Touring rack

Z5570.xxx
Total weight including amplifiers:
128 kg [282 lb]

Z5571.xxx
Total weight including amplifiers:
127 kg [280 lb]

A maximum of two D20 and D80 Touring rack assemblies can be linked to the 32 A CEE 5P mains supply.

Key
1 - Touring rack 10 RU
2 - Rack drawer or DS10/DS20
3 - I/O panel
4 - D20/D80 amplifiers
5 - Mains power distributor
6 - Rack link

Sliding doors at front and back allow for quick and effective deployment on site.
The 6 x D80 Touring rack assembly

The d&b 6 x D80 Touring rack assembly and the d&b 3 x D80 Touring rack assembly are intended for large scale sound reinforcement applications. They are designed as a complete prewired system amplification rack providing mains power distribution, connector interfaces and all internal cabling for D80 amplifiers.

The 19" internal shockmount steel frame also accommodates a d&b DS10 or DS20 Audio network bridge and an I/O panel providing both analog and digital audio signals as well as four network connectors for either Ethernet or CAN-Bus remote Networking.

The d&b Audio network bridges interface between audio transport networks and AES3 digital audio signals while also providing distribution of Ethernet control data. The DS10 supports Dante networks, while the DS20 is used for the open standards-based Milan protocol.

The Audio network bridges each incorporate an integrated 5-port switch, offering a primary and redundant network. The DS20 5-port switch is fully AVB enabled, while the DS10 offers special functions such as Multicast Filtering and VLAN modes.

The d&b 6 x D80 Touring rack assembly comes with a 2 RU loudspeaker connector panel providing six NL8 (4 channel) and two LKA25 (12 channel) loudspeaker outputs. The loudspeaker connector panel is also equipped with a Power over Ethernet (POE) socket to connect and power a d&b ArraySight inclinometer. The d&b ArraySight inclinometer system is intended to vertically aim an entire array in its operation position. It provides precision angle measurement over a wide measuring range of ±90 degrees. It utilizes an ultra bright green laser to provide a visible indication for aiming the array. The d&b ArraySight sender unit also sends temperature and humidity information to the R1 Remote control software via OCA/AES70.

The necessary patch cabling for the DS10 and DS20 Audio network bridges and I/O panel are prewired. The digital outputs of the DS10 (OUT 1/2 and OUT 3/4) are linked to the corresponding input sockets (A2 D1/2 and A4 D3/4) of the I/O panel to feed the digital audio signals to the amplifiers. In addition, the ETH 3 connector of the DS10 is linked to the ETH 1 connector of the I/O panel to provide access to the amplifiers.

The Z5578 Loudspeaker connector panel acts as a connecting interface providing all output channels of each amplifier on its dedicated NL8 socket (4 channels) and two LKA25 connector sockets (12 channels each), which directly integrate into the d&b MC24 multicore system.

The Z5577 Mains power distributor 32 A CEE is designed and dimensioned to provide and distribute the mains power supply necessary for one Z5576.000 Touring rack assembly.

The 3 x D80 Touring rack assembly provides three NL8 (4 channel) and one LKA25 (12 channel) loudspeaker output. Both touring racks are available with a 32 A CEE mains power distribution or 30 A NEMA mains power distribution panel.

The Z5577.001 Mains power distributor is supplied with two NEMA 30 A 3ØY 120/208 VAC sockets. Each connector socket provides mains power for three D80 amplifiers.
The 6 x D80 Touring rack assembly

Z5576.000
Total weight including amplifiers:
210 kg (463 lb)

Key
1 - Touring rack 18 RU
   with shock mounted 19” frame
2 - Mains power distributor
3 - LS Connector panel 2 x LKA 25/6 x NL8
4 - D80 amplifiers
5 - I/O Panel
6 - DS10/DS20 Audio network bridge
7 - POE (Power Over Ethernet) injector device
8 - Rack link

Sliding doors at front and back allow for quick and effective deployment on site.

Z5576 D80 Touring rack assembly dimensions in mm [inch]
The cables and adapters for Dual Channel mode

Amplifiers in Dual Channel mode

- 1 x D80 amplifier
  OUT: NL8
- 1 x D20 amplifier
  OUT: NL8
- 3 x D80 Touring rack
  OUT: LKA25F/3 x NL8
- 3 x D20 Touring rack
  OUT: 3 x NL8

Cables and Adapters

- Z5327.000: Adapter LKA25M to 12 x NLT4M
- Z5344.002: Adapter LKA25M to 4 x NL8M
- Z5344.001: Breakoutbox NL8 F/M to 6 x NL8
- Z5344.000: Adapter NL8F to 4 x NL4
- Z5347.001: Breakoutbox NL8 F/M to 6 x NL4
- Z5340.002: Adapter NL7F to 4 x NL74M
- Z2297.xxx: MC4SD Cable EP5
- Z5340.xxx: MC4 Cable NL24 F/M
- Z2299.xxx: MC2.5 Cable NL4 F/F
- Z5328.xxx: MC24 LKA25 F/M Multicore
- Z5343.xxx: MC8 Cable NL8 F/M
- Z5344.001: Adapter NL7F to 4 x NL4
- Z5347.001: Breakoutbox NL8 F/M to 6 x NL4

Amplifiers in Dual Channel mode

- Z5560/Z5561.xxx: 3 x D20 Touring rack
  OUT: 3 x NL8
- Z5570/Z5571.xxx: 3 x D80 Touring rack
  OUT: LKA25F/3 x NL8

The cables and adapters for Dual Channel mode
The cables and adapters for Dual Channel mode

Amplifiers in Dual Channel mode

- Z5376.xxx 6 x D80 Touring rack assembly
  OUT: 3 x LKA35/6 x NL8
- Z5383.xxx MC24 LKA25 F/M Multicore
- Z5342.xxx MC3 Cable NL8 F/M
- Z5347.000 Breakoutbox NLTS F/M to 6 x NL8
- Z5347.001 Breakoutbox NLTS F/M to 6 x NL4
- Z5347.000 Breakoutbox NLTS F/M to 6 x EP5
- Z5340.xxx MC4 Cable NL4 F/M
- Z5343.xxx MC8 Cable NL8 F/M
- Z5343.xxx MC8 Cable NL8 F/M
- Z5297.xxx MC60 Cable EPS
- Z5327.000 Adapter LKA35M to 12 x NL14M
The cables and adapters for 2-Way Active and Mix TOP/SUB modes

Amplifiers in 2-Way Active and Mix TOP/SUB modes

- **D20**
- **D80**

**Z5570/Z5571.xxx**
3 x D80 Touring rack
OUT: LKA25F/3 x NL8

**Z5600/Z5601.xxx**
3 x D30 Touring rack
OUT: 3 x NL8

**Z5570/Z5571.xxx**
3 x D80 Touring rack
OUT: LKA25F/3 x NL8

**Z5560/Z5561.xxx**
3 x D30 Touring rack
OUT: 3 x NL8

**1 x D80 amplifier**
OUT: NL8

**1 x D30 amplifier**
OUT: NL8

**Z5340.xxx**
Adapter NLT8F to 2 x NL4

**Z5347.001**
Breakout box NLT8 F/M to 6 x NL4

**Z5346.000**
Adapter NLT8F to 2 x NL8

**Z5346.000**
Adapter NLT8F to 2 x EP5F

**Z5347.000**
Breakout box NLT8 F/M to 6 x EP5

**Z5343.xxx**
MC8 Cable NLT8 F/M

**Z5345.xxx**
MC8 Cable NLT8 F/M

**Z5332.000**
Adapter LKA25M to 6 x NL4M

**Z5338.xxx**
MC24 LKA25 F/M Multicore

**Z5330.000**
Adapter LKA25M to 6 x NL4M

**Z5347.000**
Breakout box NLT8 F/M to 6 x EP5

**Z5340.xxx**
MC4 Cable NL24 F/M

**Z5342.xxx**
MC4 Cable NL24 F/M

**Z5347.000**
Breakout box NLT8 F/M to 6 x NL4
The cables and adapters for 2-Way Active and Mix TOP/SUB modes

Amplifiers in Dual 2-Way Active mode and Mix TOP/SUB modes

Z5576
D80 Touring rack assembly
OUT: 2 x LKA25F/6 x NL8

Z5326.xxx
MC26 LKA26 F/M Multicore

Z5343.xxx
MC8 Cable NL8 F/M

Z5340.xxx
MC4 Cable NL4 F/M

Z5347.000
Breakoutbox NL8 F/M to 6 x NL4

Z5347.001
Breakoutbox NL8 F/M to 6 x NL4

Z2297.xxx
MC4SD Cable EP5

Z5342.xxx
MC8 Cable NL8 F/M

Z5355.000
Adapter LKA25M to 6 x NL4M

Z5343.xxx
MC8 Cable NL8 F/M

Z5340.xxx
MC4 Cable NL4 F/M

Z5347.000
Breakoutbox NL8 F/M to 6 x EPS
## The amplifiers and software product overview

### Processing and distribution
- **Z4010.000** D510 Audio network bridge
- **Z4011.000** D520 Audio network bridge
- **Z4100.000** DS100 Signal Engine

### Amplifiers
- **Z2700.000** D6 amplifier NL4
- **Z2700.400** D6 amplifier China NL4
- **Z2700.500** D6 amplifier USA NL4
- **Z2750.000** D20 amplifier NL4
- **Z2750.400** D20 amplifier China NL4
- **Z2750.500** D20 amplifier USA NL4
- **Z2710.000** D80 amplifier EP5
- **Z2710.001** D80 amplifier NL4
- **Z2710.400** D80 amplifier China EP5
- **Z2710.401** D80 amplifier China NL4
- **Z2710.501** D80 amplifier USA NL4

### Amplifier rack assemblies
- **Z5560.xxx** 3 x D20 Touring rack
- **Z5561.xxx** 3 x D20 Touring rack (includes DS10)
- **Z5570.xxx** 3 x D80 Touring rack
- **Z5571.xxx** 3 x D80 Touring rack (includes DS10)
- **Z5576.xxx** 6 x D80 Touring rack (includes DS10)
- **Z5332.xxx** I/O Panel 19", 1 RU
- **Z5577.000** Mains panel 2RU, 32A CEE
- **Z5577.001** Mains Panel 2HE, 2x30A NEMA
- **Z5578.000** LS panel 2xLKA25/6xNL8
- **Z5566.000** Mains/LS Panel NL8/LKA25 CEE
- **Z5566.001** Mains/LS Panel NL8/LKA25 NEMA

### Racks
- **E7480.000** D20 Touring rack 2 RU, 19" SD, shock mounted, handles, window
- **E7480.000** D20 Touring rack 2 RU, 19" SD, shock mounted, handles, window
- **E7480.000** D5100 Touring rack 3 RU, 19" SD, shock mounted, handles

### Remote network
- **Z6118.000** R60 USB to CAN interface
- **Z6124.000** R70 Ethernet to CAN interface