

# D 40:4L | D 20:4L | D 10:4L





### Lab.gruppen D Series: The Integration Superpower

The Lab.gruppen D Series provides exceptional performance and expanded flexibility in audio amplification for challenging systems integration applications in stadia, arenas, convention centers and other advanced or demanding installed sound installations.

Extending downwards in power and based on the proven, innovative and green amplifier technologies of the flagship D Series, these D Series slimline models also include Rational Power Management (RPM™) a proprietary Lab.gruppen technology that rationalises power allocation and potentially reduces amplifier inventory. The Lake-variant D Series

models benefit from the proven package of onboard Lake Processing and Dante™ with AES67 digital audio networking, and also offer integration potential with many 3rd party matrix and proprietary DSP systems via dedicated middleware. Equipment specification, commissioning (including configuring RPM and other unique amplifier technologies) and on-going control and system monitoring are managed via the innovative CAFÉ™ software, running on Mac or PC. D Series features include redundant audio inputs as well as on board surveillance and load monitoring to fulfill the requirements of mission critical voice evacuation compliance

#### D Series Slimline Features and Benefits

- ► Four-channel amplifiers available in three power models
  - ► D 40:4 4000 W of total output power (4 x 1000 W nominal)
  - ► D 20:4 2000 W of total output power (4 x 500 W nominal)
  - ► D 10:4 1000 W of total output power (4 x 250 W nominal)
- ► Channel your power Rational Power Management (RPM)
  - ► True flexibility in allocating power output across channels to match requirements, enabling more efficient use of amplifier inventory
  - Any channel can be significantly scaled up in power from its average quarter of the total available power.
- ► Exceptional power density with 4 high power channels in a slimline 1 rack unit form factor

- ► Dedicated on-board surveillance & load monitoring system
- Unique power platform
  - Universal regulated switching power supply with internal energy storage and best in class power factor correction
  - Breaker emulation limiter with configurable breaker profile and current
  - ► Efficient IDEEA class D amplifier modules with low impedance or direct 70 and 100 V drive.
- CAFÉ (Configuring Amplifiers For the Environment) Software incorporating ESPTM (Equipment Specification Predictor) for design, system and equipment planning, installation and commissioning.

### **Lake Processing Technologies**

- Lake's exclusive classic/linear-phase/FIR speaker processing platform with four throughputs
- ► Group control with Raised Cosine™ MESA EQ™ asymmetric filters
- ► LimiterMax<sup>™</sup> peak and RMS limiters
- ► Extensive loudspeaker preset database (LoadLibrary™)
- Comprehensive clocking management system with low latency sample rate conversion
- ► Full support for Dante Controller

- Multiple and redundant inputs with programmable failover
  - ► Four "Lake Class" analog inputs with Iso-Float™ ground isolation
  - ► Two AES3 digital inputs (4 audio channels)
  - ► Eight dual-redundant Dante network audio inputs with AES67 support
- Comprehensive 3rd party protocol for integration potential with 3rd party matrix systems via purpose-developed middleware



## D Series: Technology Overview

The D Series from Lab.gruppen offers an unprecedented combination of sustained high output, impeccable sonic performance, configuration flexibility, and real-world efficiencies for reduced installation and operating costs. D Series brings the world's most innovative, capable and proven amplifier technology to virtually any high specification installation project, regardless of preferred DSP platform or specific matrix components.

#### **Proven Lab.gruppen Technologies**

Reliability and durability remain the bedrock criteria for any installation amplifier, and in this regard the D Series rigorously maintains Lab.gruppen's industry-leading reputation. The amplifier output stages are the Lab.gruppen IntelliDrive Energy Efficient Amplifier (IDEEA). The Class D output stage is highly efficient, while the inherently bridged

topology allows a multitude of the rated power output per channel for RPM and stable inherent hi-impedance operation. D Series also offers a full suite of protection features, including thermal "show-must-go-on" limiting, short circuit protection, excessive average current limiting, sustained VHF (very high frequencies) protection, DC protection and voltage- and current-clip limiting. None of the limiters introduce slow, long term gain changes that can risk altering the balance of a tuned system. A Breaker Emulation Limiter (BEL) prevents power interruption while Under-Voltage Limiting (UVL) allows continued operation despite severe voltage drops (not available on D10 and D20).

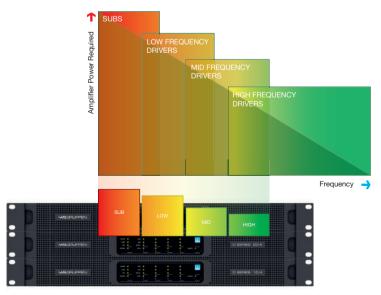
#### Channel your power - Rational Power Management (RPM)

At the core of the D Series platform is Rational Power Management (RPM), a proprietary Lab.gruppen technology that gives system designers and integrators unprecedented freedom to allocate the output power available on each channel for optimum performance with specific load conditions. RPM technology also enables the integrator to minimize initial equipment costs, reduce rack space requirement and improve long-term energy efficiency – all without compromising sonic performance.

With conventional installation amplifiers, it is often necessary to "over-specify" amplifiers to meet the maximum power demand on one channel, leaving excess power capacity wasted on the remaining channels. RPM reduces costly excess capacity by allowing re-allocation of output power capacity among the four channels. RPM can be configured so that any channel can be significantly scaled up from its average quarter of the total available power. With RPM in the D Series, up to the specified single channel power can be used for

power-hungry low-frequency systems while the remaining output power can be allocated as needed for the mid-frequency and high-frequency drivers, or for less demanding zones within a typical large project – such as concessions, concourses, VIP suites and function rooms within a sports arena or stadium.

From within the CAFÉ software, RPM allows the desired power demand to be specified for the different loads in several different ways. RPM then analyses the desired power in relation to load impedance as well as different channel and device constraints. If all desired power levels are within constraints, RPM safeguards the balance and assures that the specified output power will be maintained regardless of demand of on other channels. If a particular zone's input is being driven beyond the specified power levels, RPM aids in limiting that zone to make sure the power is available for other zones. If the desired total power is in excess of what the power model can deliver, RPM can facilitate that the limitation is shared equally among the channels.



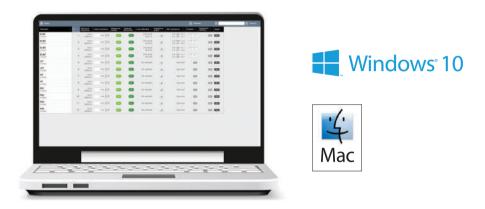
D Series - Amp channels power adjusted to match the loudspeaker requirements



#### **CAFÉ and RPM for Green Credentials**

D Series is configured and monitored using Lab.gruppen's CAFÉ (Configuring Amplifiers For the Environment) software suite. In addition to providing comprehensive system surveillance and configuration of RPM and other amplifier features such as ISVPL and Breaker Emulation Limiter (BEL), CAFÉ also includes valuable help to save the environment. In combination with the RPM configuration CAFÉ can accurately predict, based on the true SPL and speaker requirements of the individual loads for the given project, estimations of average mains current draw and generated heat in BTU. With D Series' innovative power supply technologies (true Power Factor

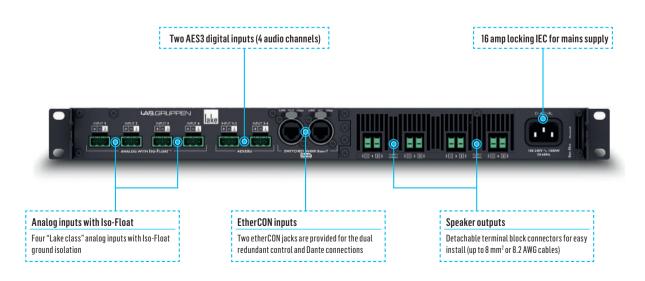
Correction and internal energy storage) the required mains draw is already best in class in relation to burst power output, but in combination with the BEL the mains draw can also be safeguarded to the predicted level. The end result is precise mains management and thermal control, which allows more accurate (rather than overspecified) provision of mains distribution, cabling and cooling. This technology suite not only saves on installation costs, it also reduces lifetime running costs and minimizes environmental impact. It also reduces demands on UPS systems in "mission critical" voice evacuation systems in arenas and stadia.



#### **CAFÉ and Equipment Specification Predictor (ESP)**

CAFÉ also features an innovative design aid – the Equipment Specification Predictor (ESP). ESP examines the system SPL and speaker requirements for a given project and aids in transforming that data into circuit and amplifier channel requirements. On a system

level, ESP supplies a recommendation for optimized placement of channels into amplifiers for the most cost effective solution. The recommendation includes model and quantities of D Series required with most rational use of amplifiers, minimizing wasted headroom.





#### Lake Processing

D Series Lake versions provide extraordinary input flexibility, the legendary power of exclusive Lake processing algorithms, comprehensive control and load monitoring via Lake Controller, and seamless integration into Dante digital audio networks with AES67 support. In addition, by employing third-party middleware, D Series Lake versions can be integrated into on other widely used networked digital matrix systems.

All three D Series Lake models incorporate four full-featured Lake Processing modules, with four discrete channels of audio throughput input to output. Audio signals are selectable from four channels of analog (with Iso-float ground isolation), four channels via AES3 digital inputs and eight dual redundant Dante networked digital inputs. Input signals are individually selectable for each channel, with programmable failover to to a lower prioritized input.

The full-featured, on-board Lake processor includes group control with Raised Cosine MESA EQ asymmetric filters to match the responses of many loudspeaker systems. LimiterMax peak and RMS limiters set the industry standard for loudspeaker protection and sonic transparency.

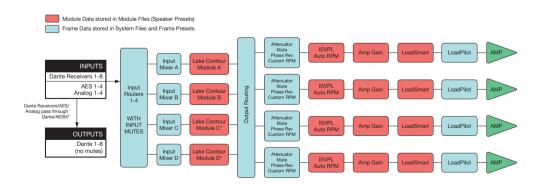
The included Lake Controller software provides a unified interface for control of Lake functions and for comprehensive monitoring of both amplifier status and connected loudspeaker loads. Optimized for a wireless tablet PC, Lake Controller is easy and intuitive to operate, with the "feel" of real-time analog faders and controls. Lake Controller also features seamless integration with third party, real-time sound system measurement, optimization, and control software packages. Users can measure spectrum and transfer function and adjust system EQ at the same time, using the same user interface.



#### **Front Panel: Lake Versions**

The front panel provides controls for power STANDBY/ON and for ampli er channel mute as well as bidirectional select functionallity between device and software. It also includes multi-color LED

indicators which display important status information for the D Series Lake platform unit as well as for each individual output channel.



#### **System Block**

The input section (inputs, input router and input mixer) allows for mixing capabilities as well as redundant and prioritized inputs with automatic switch-over in case of signal failure. Up to four Lake Processing modules provide user EQ and loudspeaker processing,

including LimiterMax limiting. Each power output channel provides individual channel processing, including ISVPL limiter, RPM and load monitoring.



# **Specifications**

Company	D 40:4L	D 20:4L	D 10:4L		
General Processing / Network	Lake / Dante	Lake / Dante	Lake / Dante		
Number of amplifier channels	4	4	4		
Total burst power all channels (share among channels with RPM)	4000 W	2000 W	1000 W		
Max. Output Power (all ch.'s driven) 1)					
2 ohms	800 W	500 W	250 W		
2.67 ohms	1000 W	500 W	250 W		
4 ohms	1000 W	500 W	250 W		
8 ohms	1000 W	500 W	250 W		
16 ohms	700 W	425 W	250 W		
Hi-Z 25 V	500 W	500 W	250 W		
Hi-Z 70 V	1000 W	500 W	250 W		
Hi-Z 100 V	1000 W	250 W	175 W		
Max output power single channel (all models) 1)					
2 ohms	850 W	850 W	850 W		
2.67 ohms	1150 W	1150 W	1000 W		
4 ohms	1750 W	1600 W	1000 W		
8 ohms	1350 W	1100 W	1000 W		
16 ohms	700 W	600 W	600 W		
Hi-Z 25 V	500 W	500 W	500 W		
Hi-Z 70 V	1450 W	1400 W	1000 W		
Hi-Z 100 V	2000 W	1000 W	700 W		
Amplifier output modules (all models, all channels)					
Peak output voltage	150 Vpk	142 Vpk	142 Vpk		
Max output current	30 Apk	30 Apk	30 Apk		
Rational Power Management (RPM)		r" of the product freely among cha	nnels. Any channel can scale up to the "sin		
	channel max power"  145 Vpk	100 Vpk	CE Val		
Default voltage limitation (can be lifted with RPM configuration)  Protection features			65 Vpk ection (VHF), Direct Current Protection (D		
Frotection readules		Current-Clip Limiter, Voltage Clip I			
Amplifier platform Inter Sample Voltage Peak Limiter (ISVPL)	Configurable Book volta	as threshold and profile			
	Configurable Peak volta	•			
Amplifier gain	Digital configurable amplifier gain 22 - 44 dB				
Pilot tone generation and analysis	Yes				
Load impedance analysis	Yes				
Temperature control	Temperature regulated f	ans (front-to-rear airflow) and show	must go on limitation (ATL, PTL)		
Audio Performance (Amplifier platform with digital input)					
THD + N 20 Hz - 20 kHz for 1 W	< 0.05 %				
THD + N at 1 kHz and 1 dB below clipping	< 0.04 %				
Dynamic range	> 112 dB				
Channel separation (Crosstalk) at 1 kHz	> 70 dB				
Frequency response (1 W into 8 ohm, 20 Hz - 20 kHz)	+/- 0.05 dB				
Internal sample rate / Data path	96 kHz / 32 bit floating	DOINT			
Product propagation delay AES 96 kHz / analog input	1.61 / 1.68 ms				
Lake processing					
Loudspeaker processing			, delay, LimiterMax™ - peak and RMS limite		
System tuning		ed Cosine™ MESA EQ™ asymmet	ic filters		
Input redundancy / Matrix		edundancy / 4 input mixers			
System integration	Comprehensive 3rd par	y protocol over UDP Ethernet			
Dante Audio Network					
	8 x 8 (shared with AES67	)			
Dante I/O	Flexible topology / Supports Daisy-chained and Dual redundant networks				
	Flexible tonology / Sunn	orts Daisy-chained and Dual redund	. 03 11		
Network topology / redundancy			ant networks		
Dante I/O Network topology / redundancy Sample rates / transport Network latency	Flexible topology / Supp 48, 96 kHz / Uni + Multio 0.25, 0.5, 1.0, 2.0, 5.0 m	cast	ant networks		
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Network topology / redundancy Sample rates / transport Network latency  AES67 Audio Network I/O Network topology / redundancy Sample rates / transport Network latency Analog inputs	48, 96 kHz / Uni + Multir 0.25, 0.5, 1.0, 2.0, 5.0 m 8x8 (shared with Dante) Flexible topology / Suppo 48 kHz / Multicast 2 ms	cast s	rant networks		
Network topology / redundancy Sample rates / transport Network latency  AES67 Audio Network I/O Network topology / redundancy Sample rates / transport Network latency Analog inputs Inputs	48, 96 kHz / Uni + Multir 0.25, 0.5, 1.0, 2.0, 5.0 m 8x8 (shared with Dante) Flexible topology / Suppo 48 kHz / Multicast 2 ms	cast s s orts daisy-chained networks	rant networks		
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Network topology / redundancy Sample rates / transport Network latency  AES67 Audio Network I/O Network topology / redundancy Sample rates / transport Network latency Analog inputs Inputs Maximum input / digital reference Sampling rate / resolution	48, 96 kHz / Uni + Multic 0.25, 0.5, 1.0, 2.0, 5.0 m 8x8 (shared with Dante) Flexible topology / Support 48 kHz / Multicast 2 ms 4 high quality inputs with + 26 dBu / +21 dBu 96 kHz / 24 bit	cast s s orts daisy-chained networks	rant networks		
Network topology / redundancy Sample rates / transport Network latency  AES67 Audio Network I/O Network topology / redundancy Sample rates / transport Network latency Analog inputs Inputs Maximum input / digital reference Sampling rate / resolution Input impedance balanced / unbalanced	48, 96 kHz / Uni + Multic 0.25, 0.5, 1.0, 2.0, 5.0 m 8x8 (shared with Dante) Flexible topology / Support 48 kHz / Multicast 2 ms 4 high quality inputs witt + 26 dBu / +21 dBu 96 kHz / 24 bit 20 / 10 kOhm	cast s s orts daisy-chained networks	rant networks		
Network topology / redundancy Sample rates / transport Network latency AES67 Audio Network	48, 96 kHz / Uni + Multic 0.25, 0.5, 1.0, 2.0, 5.0 m 8x8 (shared with Dante) Flexible topology / Support 48 kHz / Multicast 2 ms 4 high quality inputs with + 26 dBu / +21 dBu 96 kHz / 24 bit	cast s s orts daisy-chained networks	ratit networks		
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# **Specifications**

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Back panel interface					
Analog inputs		nnectors analog input with +, - and			
AES inputs		4 x Terminal block connectors analog input with +, - and ground			
Output connectors	4 x 2 pole Terminal block connectors rated at 1000 V / 41 A (exceeding amplifier capacity) Can take up to 8 mm2 (8.2 AWG) cables				
Ethernet ports	2 x EtherCon RJ45 Lake Controller, Dante controller and/or DLM (3rd party protocol)				
Detachable mains cord	Detachable locking 3-pin IEC rated at 250 V / 16 A				
Front panel user interface					
System status indication	3 x tri-color LED. FRAME, TEMP, PSU for device status indication				
Channel status indication	3 x tri-color LED per channel. Status indication separated for channel LOAD, AMP, SIGNAL status				
Mute	Per channel touch button for MUTE control and tri-color LED for indication				
Power	Touch button for ON/	Touch button for ON/STANDBY control and tri-color LED for power state indication Touch button and LED for bi-directional device software select functionality			
Select	Touch button and LE				
Mains Power					
Nominal voltage	100 - 240 V AC 50 - 6				
Operating voltage	70 - 265 V AC 45 - 66 Hz				
Mains wall plug	Selectable on order				
	CEE 7/7 "Schuko" 23	CEE 7/7 "Schuko" 230 V / 16 A,			
	NEMA 5-15 125 V / 15	NEMA 5-15 125 V / 15 A.			
	BS1363-A 250 V / 13 A,				
	IA16A3 250 V / 16 A,				
	AU/NZ 250 V / 10 A.				
	JP 125 V / 15 A.				
	OPCS-CCC 250 V / 10 A				
	CPCS-CCC 250 V / 1	UA			
Power supply features					
Soft start / Inrush power	Yes				
Power factor correction					
Regulated switch mode power supply (R.SMPS)		> 0.98 for mains power > 300 W			
	Yes Configurable mains current threshold and breaker profile (profile only D 40)				
Breaker Emulation Limiter (BEL)					
BEL max current threshold	15 A	7 A	7 A		
Power Average Limiter (PAL)	Yes				
Under Voltage Limiter (UVL)	Yes	No	No		
Mains undervoltage and overvoltage protection and mains glitch tolerance	Yes				
Adaptive Rail Control	Optional green mode for improved efficiency at low output levels				
Dimensions					
Rack rail to rear panel	W: 483 mm (19") H: 4	4 mm (1 U), D: 425 mm (16.7")			
Overall depth front-rear support	D: 463 mm				
Overall department real support	D. 400 IIIII				
Weight	8.3 kg (18.3 lbs)	7.9 kg (17.3 lbs)	7.8 kg (17.1 lbs)		
Finish		nassis with grey painted steel front			
1 111011	Diack painted steel of	addio with grey painted steel front	with detachable grille		
Approvals	CE ETI (ANGI/III C	SA), PSE, RCM, BIS, CCC			
Approvate	OE, ETL (ANSI/UL, O	DAJ, FOL, NOIVI, DIO, CCC			

Note 1): Lab.gruppen burst power (1 kHz, 25 ms burst power @ 150 BPM, 12 dB Crest factor)

All specifications are subject to change without notice.

