Data Sheet TouchControl 5









TouchControl 5



Touch Screen • Flexible Screen Layout • Pushable Rotary Knob • AoIP I/O • 32 Channels • Speaker Calibration • SPL • Chart Fader • PPM/TP • Multichannel • Immersive • Loudness • LRA • RTA • Prem. Metering • VSC • Monitor Control • Bass Management

TouchControl 5 is a compact AoIP (Audio over IP) based monitor controller, using the monitor control engine of the well-known SurroundControl. It supports AoIP standards (Dante® or RAVENNA®/ AES67/ST 2110) and can be powered over ethernet. It features instant control over a high channel

count: up to 32 audio channels can be controlled with one single knob. Beside others, it provides a high-quality microphone input, speaker output, headphone output and features e. g. mute/solo, metering, SPL measurement and talkback.

Graphical User Interface

The TouchControl 5's graphical user interface is controlled simply by the touch of your finger and the pushable rotary knob. The integrated instruments can be scaled, randomly positioned and combined for optimum utilization of the available screen space.

With its IP address and the comprehensive Web App, TouchControl 5 can be adapted to your individual needs within the AoIP network (Dante® or RAVENNA®).

The Device

Hardware

- 5" capacitive touch screen 16:9 TFT (1280 x 720 pixel) with multitouch functionality
- 32-channel audio over IP interface for Dante® or RAVEN-NA® audio networks (RJ-45 ethernet)
- Power supply via ethernet connection (PoE power over ethernet, IEEE802.3af compliant)
- Integrated Microphone for SPL measurement & talkback
- Studio-grade 48 V phantom powered high-quality microphone input (XLR)
- Powerful headphone output e. g. for monitoring a userdefined downmix (6.3 mm Stereo jack)
- Analog 2-channel stereo loudspeaker output (Line Out 3.5 mm Stereo jack)
- Control via touch screen and pushable rotary knob 1 3
- Freely scalable and positionable applications and instruments
- Up to 31 presets selectable

Software

- Device configuration via IP address and Web App within the Dante® or RAVENNA® network (web-based interface)
- Toolbox with simple TruePeak meter and up to four on-screen faders (slider) for up to 32 channels, Talkback application
- Support for Stereo, Surround, Immersive and Multichannel formats for up to 32 channels incl. 5.1 and 7.1.4 formats
- Loudness & SPL functions acc. to all common standards, Loudness Chart and Loudness Range instrument (LRA)
- Comprehensive functions for Monitor Control (like Solo, Cut, Phase, DIM, Mute) and loudspeaker level calibration, 8-band equalizer, SPL measurement, Bass Management and up to 4 sources and 4 destinations (loudspeaker sets)
- Premium Metering with Multiformat-PPM and TP meter incl. comprehensive scales and Moving Coil needle instruments, Audio Vectorscope and Real Time Analyzer





320517ND (Dante®) | 320518ND (RAVENNA®)





Essential Features

TouchControl 5 is equipped with a comprehensive software package. Beside the control functions, the software provides various applications and instruments that can be used individually depending on the area of application. Core of the system is the pushable rotary knob and the Monitoring application, which you can use to calibrate, control and monitor your monitoring system. With the Metering application, which can be positioned up to four times, you can carry out extensive measurement tasks. And with the Leveling application, which can also be used four times, you can control the level of individual formats or channel groups.

Not forgetting the Talkbalk application, which can be used to address any node in the audio network via the built-in or an external microphone. And when placed up to four times, Intercom communication between several TouchControl 5 units is possible.

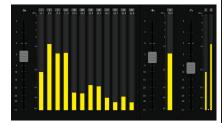
Monitor Control



This application includes downmix, DSP-supported speaker calibration (with internal or external microphone), SPL display, selection of up to 4 sources and up to 4 speaker sets, solo, cut, phase, mono, dim, mute and test tone generator. The SPL display (bar graph and/or numerical) shows the sound pressure level measured live in your room.

The downmix instrument outputs surround or immersive mixes in stereo and mono to speakers, headphones or any AoIP channel. Each speaker can be muted or soloed or its phase can be rotated.

Leveling



This application is used for independent control (leveling) of individual channels or channel groups with up to 4 faders and simple TP meters, which can be combined with the rotary knob for relative level control of several channel groups.

Loudspeaker Level Calibration



The speakers in the setup can be calibrated with regards to level, delay and EQ. The support for measurement microphones, the built-in test tone generator and the live input SPL meter form the tool combination for calibrating the speaker levels.

Loudspeaker DSP section



Each speaker has a DSP section for ultimate control. Delay, gain and phase can be set individually, and a 8-band EQ is available for each speaker in each setup.

Talkback

This application uses the built-in or an external microphone as an intercom microphone to address each node in the audio network including Intercom functionality. Up to 4 instances are possible.

Bass Management



Up to two LFE channels with adjustable crossover frequencies can be adapted to the requirements with the bass management. Full-range speakers and smaller speakers can be finely tuned to each other and operated in the same system. HP filters for bass-controlled speakers, all-pass filters for other speakers and an additional SUB output channel for formats without LFE are also available.

Metering



This application provides the familiar RTW Premium metering functions and instruments: Multiformat PPM, TP meter, Moving Coil needle instruments, loudness measurement and calculation, loudness range, chart, SPL display, SPL sum value calculation, Audio Vectorscope and Real Time Analyzer. Up to 4 instances are possible.

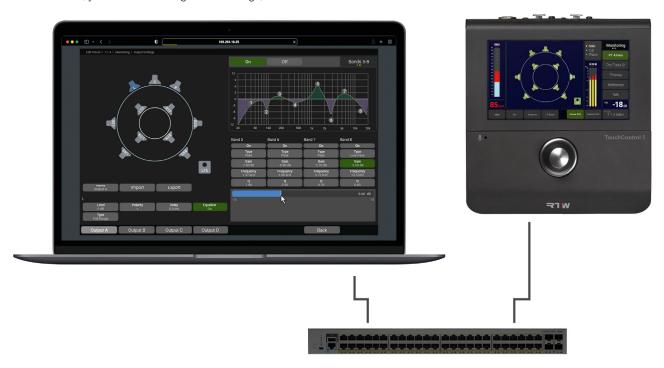
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Essential Features (continued)

Web-based Interface

The TouchControl 5 is a network-based device. It is therefore also set up via the network, using the IP address of the device and a standard web browser in the same Dante® or RAVENNA® AoIP network. With the user interface (WebApp) displayed in the browser, you can make the general settings, create and

manage up to 31 of your own presets, create your own screen views and much more. You can also control access to the device and restrict operation to certain functions to prevent unwanted use.



Extensive Routing

The routing matrix is used to determine the audio channels to be used as inputs and outputs in the preset. The 32 channels assigned in the Dante Controller™ resp. the RAVENNA® manager and the physical inputs and outputs offer many possibilities. For example, the same channels can be used for monitoring and metering. Or they can be split up so that metering is independent of monitoring.

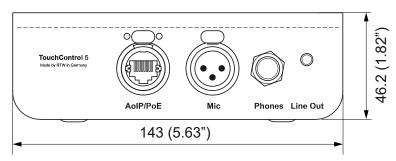
Own Display Views

TouchControl 5 allows you to design up to three of your own display views. Independently from the applications, you can place any instrument in any view and determine its size and ratio. Several instruments can also be rotated to adapt them to your own requirements. Buttons can be placed anywhere on the screen, whereby the buttons in the sidebar are available on all display views of the preset.

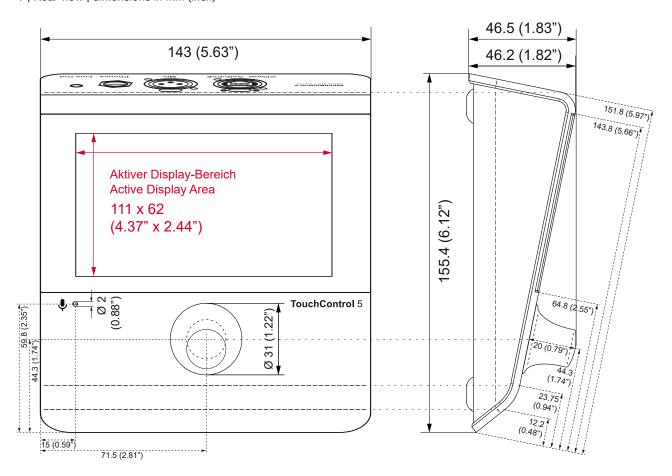


Dimensions

TouchControl 5 Desk-top Unit (320517ND/320518ND)



1 | Rear view | dimensions in mm (inch)



2 | Front view/top view | dimensions in mm (inch)

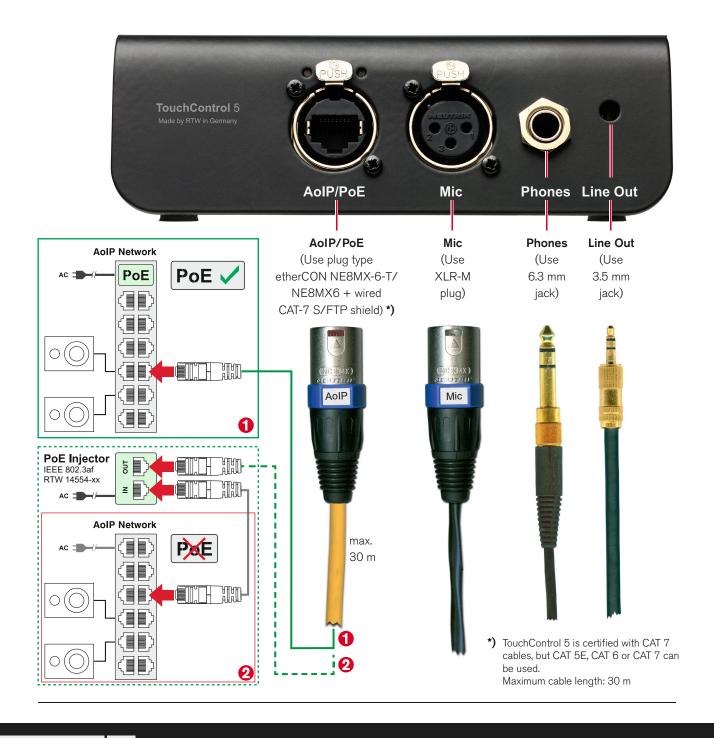
3 | Side view | dimensions in mm (inch)

Connections

Connectors

NOTE - The power supply of the TouchControl 5 is done via the network connection (AoIP/PoE) and the Dante® (320517ND) resp. RAVENNA® (320518ND) AoIP network without any additional cable, if this network has the Power over Ethernet functionality (PoE - IEEE 802.3af-compliant) 1.

If your switch does not provide Power Over Ethernet, an IEEE 802.3af-compliant ethernet power injector such as the RTW 14554-xx is required for power supply 2.



Specifications

System

General

Power requirements: Power over Ethernet (PoE - IEEE 802.3af-

> compliant) 12 W maximum

Power consumption:

Display:

5" capacitive touch display 16:9 wit multitouch

funtion (1280 x 720 pixel)

1 x RJ-45: LAN/Ethernet built-in socket Connectors:

NE8FD type for Dante® audio over IP and power supply (PoE - IEEE 802.3af-compliant) 1 x 3-pin XLR-F (microphone input, switchable phantom powered 48 V - Mic), 3 kOhm 1 x 1/4 inch Stereo jack (6.3 mm headphone

output - Phones)

1 x 3,5 mm Stereio jack (analog loudspeaker

output - Line Out) Dimensions (W x H x D): 143 x 46.5 x 155.4 mm

approx. 830 g Weiaht: Operating temperature: +5° to +35° C

Functions (Availability depends on selected application)

- Operation with touch sensitive display and pushable rotary knob
- Instruments and controls can freely be scaled and positioned
- Monitor Control and SPL measurement for up to 4 sources/destinations
- Integrated and calibrated microphone for SPL measurment and talkback
- Studio-grade 48 V phantom powered highquality microphone input (XLR)
- Powerful 1/4 inch Headphones output with the option of monitoring a user-defined downmix (6.3 mm Stereo jack)
- Analog 3.5mm speaker output with the option of monitoring a user-defined downmix (3.5 mm Stereo jack)
- Multiformat PPM and TP meter for level metering of up to 32 channels with in different configurations (Mono, Stereo, Surround, Immersive or Multichannel)
- Loudness-Meter: ITU-R BS.1770-4/1771. EBU R128, ATSC A/85, ARIB, OP-59, AGCOM, CALM Act, LEQ(M), TASA, SAWA, Custom mode
- Dialog Gated Loudness
- Loudness Test Time Control
- Loudness Range instrument (LRA)
- Loudness Chart (Loudness over time)
- SPL meter
- Test signal generator
- Moving Coil (BR, VU, Loudness, BBC mode)
- Spot Correlator in the Stereo bargraph display
- Audio Vectorscope, Real Time Analyzer
- Downmix with adjustable coefficients
- Numerical displays
- Up to 4 on-screen fader and simple TP Meter for the simultaneous control of up to 32 input

- Bass Management for up to 4 loudspeaker sets
- · Level calibration for each individual loudspeaker in each of the output sets
- 8-band equalizer for each loudspeaker
- Immersive-Setups (5.1.2, 5.1.4, 5.1.6, 7.1.2, 7.1.4, 7.1.6, 9.1.2, 9.1.4, 9.1.6, 22.2, Array)
- Up to 32 presets selectable (31 user-definable, 1 write-protected with standard settings)
- Configuration of the device via IP address and Web App in the AoIP network

Digital Inputs

Inputs:

32 Audio over IP inputs (network channels, Dante® or RAVENNA® depending on the device type) via the RJ-45 built-in socket NE8FD type 44.1, 48, 88.2, 96 kHz

Sample rates: Word width: 16, 24, 32 bit

Digital Outputs

Outputs:

32 Audio over IP outputs (network channels, Dante® or RAVENNA® depending on the device type) via the RJ-45 built-in socket NE8FD type referenced to digital inputs or internal clock

Sample rates:

Latency Internal device latency: 3 ms (independent from sample rate)

Minimum network latency: • Dante®: 1 ms (Dante Controller™)

RAVENNA®: 0.25ms

Be aware that latency also depends on other network devices, such as switches and other networked products.

Pushable Rotary Knob

Function:

- Turn to control the volume with definable listening level (individual, initial, maximum)
- · Push to trigger a defined action
- On the Calibration screen: Rotate to select options for activated buttons
- On the Calibration screen: Push to deactivate all selected speakers

Mute, DIM, Recall Reference Volume. Allocation of a function in the WebApp

Volume:

Push function:

- user-defined:

- maximum:

- at power on:

- at preset recall:

78 dBC/dBA; adjustable in the range from 10 to 100 dBC/dBA in steps of 0.5 dB or off. Volume can be called up at any time by touching the Reference button placed on the screen or by pushing the rotary knob when the corresponding push function is defined

100 dBC/dBA; adjustable in the range from 60 to 100 dBC/dBA in steps of 0.5 dB or off

last set monitoring level, reference monitoring level or none (Silence)

last set monitoring level, reference monitoring level or none (Silence)

absolute: dBC oder dBA or

 relative (dB): 0 dB refers to the selected reference level (e. g. 78 dBA)

Application Leveling

Used for independent control (leveling) of individual channels or channel groups with up to 4 on-screen sliders (faders), which can be combined with the rotary knob for relative level control of several channel groups. Up to 4 instances are possible.

Functions:

- Digital fader for the level control of up to 32 channels in different formats
- Up to 4 faders simultaneously possible
- Change of the different levels of selected faders by the use of the rotary knob
- Simple TruePeak meter with fixed labeling and scale
- Spot Correlator in Stereo mode

Digital fader

0 dB *); adjustable from -infinity to +6 dB in Fader range:

steps of 1 dB

TP-Meter

- Display: Up to 4 TP meter, coupled with faders

- Scale: TP60: +3..-60 dB

*) Default values are in bold.

Application Talkback

Provides the option of using the built-in or an external microphone as an intercom microphone that can address any node in the audio network. Up to 4 instances are possible.

Talkback

Function:

- Instrument for using the internal or external microphone for Talkback
- Adjustment via input/output routing
- With enabled Monitoring application DIM is

Hold function. Talkback source: as switch (hold) or push button (momentary) Internal microphone (MIC), external microphone (XLR) or any AoIP audio channel

Internal source:

- Internal Mic Gain:

0 dB; adjustable in the range from 0 to +40 dB in steps of 0.5 dB

External source:

0 dB; adjustable in the range from -12 to +12- Talkback level trim:

dB in steps of 0.5 dB (inputs of other applications inside the preset can be influenced)

- High-Pass-Filter:

Off or On - High-Pass-Frequenz: 120 Hz; adjustable in the range from 80 to 250

> Hz in steps of 1 Hz Off or On

- Phantom Power:

- XLR-Gain: 0 dB; adjustable in the range from -8 to 60 dB

in steps of 0.5 dB

Application Metering

Provides the familiar RTW Premium metering functions (multi-format PPM and TP meter, moving coil, RTA, audio vectorscope) and the functions for loudness calculation, loudness range display, chart, SPL display and sum SPL value calculation, vectorscope and real time analyzer. Up to 4 instances are possible.

General

Input sources: Output destinations: Formats:

- Mono - Stereo:
- Surround:

- Immersive:

32 network channels, Mic In, Internal Mic 32 channels, Headphone Out, Line Out Mono, Stereo, Surround, Immersive, Multichannel several single channel signals selectable several 2-channel Stereo pair selectable **5.1**; LCR, LCM, 4.0, 5.0, 5.1, 6.0, 6.1, 7.0, 7.1 selectable

5.1.4; 5.1.2, 5.1.4, 5.1.6, 7.1.2, 7.1.4, 7.1.6, 9.1.2, 9.1.4, 9.1.6, 22.2, Array selectable

- Multichannel: 8; 1 to 32 single channels in one instrument selectable

PPM

Display type: Bargraph; Bargraph (for all formats) or Moving

Coil needle instrument (for Stereo format)

- Display: Peak level
 - Peakhold (depending on type)
 - · Numerical value of the display
 - Digital Over
 - Gain (+20 dB, +40 dB acc. to standard),
 - Peakhold on/off (depending on type)
 - Memory
 - Reset (Memory/Peakhold)

Digital Peakmeter (PPM)/TP Meter

Display type:

Functions:

Bargraph, variously combinable with loudness display

Orientation: Word width: vertical; vertical or horizontal selectable

24 bit Digital Scales:

- TP60: +3 .. -60 dB (default) • TP20: +3 .. −20 dB
- Dig60: 0 .. -60 dB (Attack: Sample)
- Dig40: +20 .. -40 dB (Attack: Sample)
- Dig20: 0 .. –20 dB (Attack: Sample)
- Dig0: +18 .. 0 dB (Attack: Sample)
- Dig18: +18 .. -18 dB (Attack: Sample)
- ARD9: +9 .. -60 dB (Attack: 10 ms)
- DIN5: +5 .. -50 dB (Attack: 10 ms)
- DIN10: +10 .. -50 dB (Attack: 10 ms)
- Nordic: +12 .. -42 dB (Attack: 10 ms)
- BR IIa: 7 .. 1 (Attack: 20 ms)
- BR IIa ext: 7..1 (Attack: 20 ms)
- BR IIb: +12 .. -12 dB (Attack: 20 ms)
- BRIIb ext: +12..-12 dB (Attack: 20 ms) Zoom10: +10 .. -10 (Attack: 10 ms)

Zoom1: +1 .. -1 (Attack: 10 ms)

Scale marker: Off; switchable in the range from -30 to 0 dB in | PPM + Loudness:

steps of 0.5 dB or Off

-9 dB; adjustable in the range from 0 to Headroom: -20 dB in steps of 1 dB (not available for Dig40, Dig0, Dig18, ARD9)

fixed with reference 997 Hz for:

Dig40:+20..-40dB: 0 dB fixed at -20 dBFS, Headroom up to +20 dB at 0 dBFS

 Dig0:+18..0dB: 0 dB fixed at -18 dBFS, Headroom up to +18 dB at 0 dBFS

Dig18:+18..-18dB: 0 dB fixed at -18 dBFS, Headroom up to +18 dB at 0 dBFS

ARD9:+9..-60dB: 0 dB fixed at -9 dBFS, Headroom up to +9 dB at 0 dBFS

Operation field: adjustable in the range from 0 to -20 dB in

steps of 1 dB

Integration time (Attack): acc. to corresponding standard or (partly)

selectable: Sample, 20 ms, 10 ms, 1 ms, 0,1 ms,

British BRII scales also 150 ms Gain: +20 dB, +40 dB acc. to standard

High-pass filter: Off; 5 Hz, 10 Hz, 20 Hz or Off selectable (not

for TP scales)

Peakhold display: Off; 1 s, 2 s, 4 s, 10 s, 20 s, 30 s, manual reset

or Off selectable 1 s or manual

Over indicator hold time: Over indicator PPM

> - Threshold: Full Scale, Full Scale -1LSB, Full Scale -2LSB,

-0.1 dBFS, -0.5 dBFS, -1 dBFS, -2 dBFS, -3

dBFS

1 to 15 samples - Attack time: - Word width: 16 to 24 bit, selectable

Over indicator TruePeak

- Threshold: -1 dBTP; adjustable in the range from -3 to +3

dBTP in steps of 0.1 dBTP

Moving Coil Instruments

(only available in Stereo mode)

Display type: PPM (L/R, M/S), VU, Loudness, PPM + Loud-

ness (L/R; M, S or I), selectable

PPM.

- Channel arrangement: Dual, Dual + M/S horizontal, Dual + M/S vertical, Stereo horizontal, Stereo vertical

BR IIa: 7..1 (default) - Scales:

 BR IIb: +12..-12 dB - Integration time: 10 ms; Sample, 0,1 ms, 1 ms, 10 ms, 20 ms,

150 ms selectable

- Headroom Ref: -10 dB; adjustable in the range from 0 to

-20 dB in steps of 1 dB

only available, if M/S type is selected: M3, M6 $\,$ - S mode: - Peak indicator: off; Peak, True Peak, BR Peak, off selectable

- BR Peak Threshold:

BR IIa: adjustable in the range from 4 to 7 dB

in steps of 0.25

BR IIb: adjustable in the range from 0 to

12 dB in steps of 1 dB

VU:

- Channel arrangement: Stereo horizontal, Stereo vertical

- Scale: VU (-20 to + 3 dB)

0 dB; adjustable in the range from 0 to 12 dB in - Lead:

steps of 1 dB

off; Peak, TruePeak, off selectable - Peak indicator:

Loudness:

- Channel arrangement: Dual, Stereo horizontal, Stereo vertical

- Scales: acc. to Loudness settings - Integration time: acc. to standard off, no selection - Peak indicator:

- Channel arrangement: Dual-PPM (as described above) with additional

Loudness display (BBC mode) for M, S or I

(selectable) in one instrument

 PPM: see above - Scales:

> ■ Loudness: +9 to -9 LU fixed (center of the scale represents the Target Level of the

selected Loudness standard)

Numerical display: switchable in all modes

Stereo Correlator

Display: Bargraph, additional spot indicator between

PPM bargraphs -1 r to 0 to +1 r

Scale range: -1 r to -0.1 r Standard color setting: red:

• white: 0 r (-0.1 r to +0.1 r) • green: +0.1 r to +1 r

Attack/release time: 1.0 s/2.5 s

Audio Vectorscope

2-channel display of the phase interaction of any Function:

two channels on a rotated coordinate system

(Lissajous figure)

Inputs: Any pair selectable for the display

Mode: L/R or M/S, switchable Grid display: Solid or Dotted

Colors: 24 colors each for background, background grid,

grid, waveform

Loudness & SPL

Loudness and SPL measurements acc. to all relevant worldwide standards and guidelines like ITU-R BS.1770-4/1771-1, EBU R128, ATSC A/85, ARIB, OP-59, AGCOM, CALM Act, LEQ(M), TASA, SAWA, SPL, including Dialog Gated Loudness and Loudness Range.

General

Functions:

 Loudness bargraph displays of the single channels, can be combined with PPM in

 Loudness Sum: Momentary, Shortterm and Integrated of all channels of a format

Test time control

 Dialog-based loudness measurement with own summing and numeric instruments

 Numerical display of the sum, maximum, LRA and duration values

Loudness Range instruments (LRA, LRA D)

SPL

• Loudness sum of the channels in selectable combination of the values:

- M bargraph (Momentary - summation of momentary loudness values of all channels for a short span of time)

S bargraph (Shortterm - loudness summation value of an adjustable dynamic time

I bargraph (Integrated - long term loudness value infinite or manual control)

adjustable tolerance range for M, S, I

· Separate Loudness Sum display for dialogbased measurements

• Dialog/ no dialog indicator

Bargraph orientation:

Bargrarph display:

vertical or horizontal selectable

Numerical display: M, S, I, LRA, TP Max, Time; available values de-

pending on selected Loudness standard: M, S, I, LRA, TPmax, Mmax, Smax, Time, PLR, LDR, S dialog, I Dialog, LRA Dialog, Dialog

Amount

Area-dependent settings

Europe: EBU R128
United Kingdom: EBU R128
North/South America: ITU 1771
Australia: OP-59
Asia: ARIB

Standard-dependent settings:

In the defined loudness standards, specific parameters are fixed that cannot be changed or can only be changed in part. The setting ranges for changeable parameters (1) can be looked up under the corresponding designation in the "Customer-specific Loudness Mode" section.

ITU-BS.1771

Scales: ITU+9: +9..-18 LU, ITU0: 0..-30 LKFS

Weighting filter: ITU BS.1770 (k)
Target Level: 1) -24 LKFS
Momentary: 400 ms
Shortterm: 1) 3 s

Integrated Silence Gate: -70,0 LKFS, switchable Integrated Relative Gate: -10 LU, switchable

Tolerances

- Over: 1) -2 dBTP
- Headroom: 1) -9 dB
- M, S, I high: 1) +2 LU
- M, S, I low: 1) -2 LU

EBU-R128

Scales: **EBU +9: +9..-18 LU**, EBU+3: +3..-18 LU,

EBU+18: +18.-36 LU, EBU+9a: 14.-41 LUFS,

EBU +18a: -5..-59 LUFS

Weighting filter: ITU BS.1770 (k)
Target Level: ') -23 LUFS
Momentary: 400 ms
Shortterm: 3 s
Integrated Silence Gate: -70,0 LUFS
Integrated Relative Gate: -10 LU

Tolerances

- Over: 1) -1 dBTP
- Headroom: 1) -9 dB
- M, S, I high: 1) +1 LU
- M, S, I low: 1) -1 LU

ATSC-A/85, CALM Act, OP-59

Scales: ITU+9: +9..-18 LU, **ATSC0: 0..-60 LKFS**,

ATSC0a: 0..-30 LKFS
Weighting filter: ITU BS.1770 (k)
Target Level: 1) -24 LKFS

Momentary: 400 ms Shortterm: 1) 3 s

Integrated Silence Gate: -70,0 LKFS, switchable Integrated Relative Gate: -10 LU, switchable

Tolerances

- Over: 1) -2 dBTP - Headroom: 1) -9 dB - M, S, I high: 1) +2 LU - M, S, I low: 1) -2 LU

ARIB

 Scale:
 ATSC0: 0..-60 LKFS

 Weighting filter:
 ITU BS.1770 (k)

 Target Level: ¹)
 -24 LKFS

 Momentary:
 400 ms

 Shortterm:
 3 s

Integrated Silence Gate: -70,0 LKFS, switchable
Integrated Relative Gate: -10 LU, switchable

Tolerances

- Over: 1) -1 dBTP
- Headroom: 1) -9 dB
- M, S, I high: 1) 0 LU
- M, S, I low: 1) 0 LU

AGCOM

Scales: EBU +9: +9..-18 LU, EBU+3: +3..-18 LU,

EBU+18: +18.-36 LU, EBU+9a: 14.-41 LUFS, EBU +18a: -5.-59 LUFS, ITUO: 0.-30 LKFS, **ATSCO: 0.-60 LKFS**, ATSCOa: 0.-30 LKFS

Weighting filter: ITU BS.1770 (k)
Target Level: 1) -24 LKFS
Momentary: 400 ms
Shortterm: 1) 3 s
Integrated Silence Gate: -70,0 LKFS
Integrated Relative Gate: -10 LU

Tolerances

- Over: 1) -2 dBTP
- Headroom: 1) -9 dB
- M, S, I high: 1) +2 LU
- M, S, I low: 1) -2 LU

Streaming

Scales: EBU +9: +9..-18 LU, EBU+3: +3..-18 LU,

EBU+18: +18..-36 LU, EBU+9a: 14..-41 LUFS, EBU +18a: -5..-59 LUFS, ITUO: 0..-30 LKFS, ATSCO: 0..-60 LKFS, ATSCOa: 0..-30 LKFS

Weighting filter: ITU BS.1770 (k)
Target Level: ') -15 LUFS
Momentary: 400 ms
Shortterm: 3 s
Integrated Silence Gate: -70,0 LUFS
Integrated Relative Gate: -8 LU

Tolerances

- Over: 1) -5 dBTP
- Headroom: 1) -9 dB
- M, S, I high: 1) +0,5 LU
- M, S, I low: 1) -0,5 LU

Customer-specific Loudness Mode (Custom)

Scales: 2)

Loudness scales:

EBU+9: +9 .. -18 LU
EBU+3: +3 .. -18 LU
EBU+18: +18 .. -36 LU
EBU+9a: 14 .. -41 LUFS

EBU+18a: -5 .. -59 LUFSEBU0: 0 .. -60 LUFS

■ ITU+9: +9 .. –18 LU (Loudness Units)

ITU0: 0 .. -30 LKFSATSC0: 0 .. -60 LKFSATSC0a: 0 .. -30 LKFS

Weighting filter: K; A, C, CCIR, k filter acc. to ITU BS.1770, or

switchable (None)

Target Level: 2) - 23 LUFS; adjustable in the range from -10

to -30 LUFS in steps of 1 LUFS

-24 LKFS; adjustable in the range from −10 to −30 LKFS in steps of 1 LKFS

Momentary: 2)

- Window Time (SQR): 400 ms; adjustable in the range from 200 ms to

1000 ms in steps of 100 ms

- Integration Time (IIR): IEC 125 ms Fast, 250 ms (IRT), 500 ms, 750

ms, IEC 1000 ms Slow, 1500 ms, 2000 ms

selectable

Shortterm: 2)

- Integration Time: 3 s; time window adjustable from 1 to 20 s in

steps of 1 s

Integrated: 2)

- Silence Gate: • -70.0 LUFS; adjustable in the range from

-80.0 to -40.0 LUFS in steps of 0.5 LUFS, switchable

-70.0 LKFS; adjustable in the range form

-80.0 to -40.0 LKFS in steps of 0.5 LKFS,

-10.0 LU; adjustable in the range from -40.0 to - Relative Gate:

0 LU in steps of 0.5 LU, switchable

Level adjustment for the summation: 2)

• 0.0 dB (L, R, C); adjustable between -6 and

+6 dB in steps of 0.5 dB

+1.5 dB (LS, RS, LSR, RSR), adjustable between -6 and +6 dB in steps of 0.5 dB

• Off (LFE); Off, 0 dB, 10 dB selectable

2) Limited availability of settings depending on the Loudness standard used

Tolerances (different presets depending on the Loudness standard used):

- TP Over Sensitivity: -1,0 dBTP; adjustable in the range from 0 to

-4 dBTP in steps of 0.1 dBTP

- TP Headroom: -9.0 dB; adjustable in the range from 0 to

-20 dB in steps of 0.1 dB

- M High: +1.0 LU; M tolerance above Target Level,

adjustable in the range from 0 to 10 LU in steps

- M Low: -1.0 LU; M tolerance below Target Level, adjus-

table in the range from 0 to -12 LU in steps of

0.1 LU

- S High: +1.0 LU; S tolerance above Target Level, adjus-

table in the range from 0 to 10 LU in steps of

0.1 LU

- S Low: -1.0 LU; S tolerance below Target Level, adjus-

table in the range from 0 to -12 LU in steps of

0.1 LU

- I High: +1.0 LU; I tolerance above Target Level, adjus-

table in the range from 0 to 10 LU in steps of

- I Low: -1.0 LU; I tolerance below Target Level, adjus-

table in the range from 0 to -12 LU in steps of

0.1 LU

LEQ(M)

Scales: TASA, SAWA

Weighting filter: linear, A (Leq(A)), C, CCIR Leq(M), ITU BS.1770 (k)

Reference level: 78 dB; adjustable in the range from 68 to 88 dB

in steps of 1 dB

IEC 1000 ms slow Integration time:

Shortterm: 3 s Integrated Silence Gate: Off Off Integrated Relative Gate:

Tolerances

- Over: 1) -2 dBTP

- Headroom: 1) -9 dB - M, S, I high: 1) +1 LU - M, S, I low: 1) -1 LU

TASA

Scales: TASA

linear, A (Leq(A)), C, CCIR Leq(M), ITU BS.1770 (k) Weighting filter:

Reference level: 85 dB

Integration time: IEC 1000 ms slow

Shortterm: 3sIntegrated Silence Gate: Off Integrated Relative Gate: Off

Tolerances - Over: 1) -2 dBTP - Headroom: 1) -9 dB - M, S, I high: 1) +1 LU - M, S, I low: 1) -1 I U

SAWA

Scales: SAWA

Weighting filter: linear, A (Leq(A)), C, CCIR Leq(M), ITU BS.1770 (k)

Reference level: 82 dB

Integration time: IEC 1000 ms slow

Shortterm: 3sIntegrated Silence Gate: Off Integrated Relative Gate: Off

Tolerances

- Over: 1) -2 dBTP - Headroom: 1) -9 dB - M, S, I high: 1) +1 LU - M, S, I low: 1) -1 LU

1) Setting range see "Customer-specific Loudness Mode"

For more standards, see the corresponding article on our blog page on the Internet: rtw.com/en/standards (https://rtw.com/index.php?id=1609)

SPL

Display: Bargraph for summation of channels Orientation: vertical or horizontal selectable

Weighting: A; Linear (None), A, C, CCIR, K (ITU BS.1770)

selectable

M Integration time: IIR: 125 ms (fast) oder IIR: 1000 ms (Slow)

selectable

Target level: 78 dB; adjustable in the range from 68 to 88 dB

in steps of 1 dB

Reference point: 78 dB; adjustable in the range from 68 to 88 dB

in steps of 1 dB

Dialog Gated

ITU+9: +9..-18 LU, ITU0: 0..-30 LKFS

Weighting filter: ITU BS.1770 (k)

Target Level: -24 LKFS; adjustable in the range from -10 to

-30 LKFS in steps of 1 LKFS

Momentary:

- Window Time (SQR): 400 ms; adjustable in the range from 200 ms to

1000 ms in steps of 100 ms

- Integration Time (IIR): IEC 125 ms Fast, 250 ms, 500 ms, 750 ms, IEC

1000 ms Slow, 1500 ms, 2000 ms selectable

Shortterm:

- Integration Time: 3 s; time window adjustable from 1 to 20 s in

steps of 1 s

Integrated Gate:

Absolute Threshold: -70.0 LKFS; adjustable in the range from

-80.0 to -40.0 LKFS in steps of 0.5 LKFS,

switchable

- Relative Threshold: -10.0 LU; adjustable in the range from -40.0 to

0 LU in steps of 0.5 LU, switchable

Dialog Gated:		- MD, SD, ID High:	+2.0 LU; I tolerance above dialog gated Target
- Target Level:	-27 LKFS ; adjustable in the range from −30.0		Level, adjustable in the range from 0 to 10 LU in
-	to -10.0 LKFS in steps of 1 LKFS		steps of 0.1 LU
- Threshold:	-15 %; adjustable in the range from 0 to 100 %	- MD, SD, ID Low:	-2.0 LU; I tolerance below dialog gated Target
	in steps of 1 %		Level, adjustable in the range from 0 to −10 LU
- Absolute Threshold:	-70.0 LKFS ; adjustable in the range from -80.0		in steps of 0.1 LU
	to -40.0 LKFS in steps of 0.5-LKFS, switchable		-
 Relative Threshold: 	−10.0 LU ; adjustable in the range from −40.0 to	Loudness Test Time Co	
5	0 LU in steps of 0.5 LU, switchable		tomatic, semi-automatic or manual loudness
- Dialog channels:	L, R, C; each available channel selectable	measurements.	
Tolerances:	+10 III. M talaranca abaya Targat Laval	Start: - Functions:	Autostart after preset lead, autostart with
- M, S High:	+1.0 LU ; M tolerance above Target Level, adjustable in the range from 0 to 10 LU in steps	- Functions:	Autostart after preset load , autostart with gate, autostart with gate and autoreset, manually
	of 0.1 LU		via keys.
- M, S Low:	-1.0 LU; M tolerance below Target Level, adjus-	- Level for gate:	-70,0 LUFS/LKFS; adjustable in the range
, 0 2011	table in the range from 0 to -10 LU in steps of	20101101 gato	from -85 to -10 LUFS/LKFS in steps of 0.5
	0.1 LU		LUFS/LKFS
- I High:	+0.5 LU; S tolerance above Target Level, adjus-	Stop:	
<u>o</u>	table in the range from 0 to 10 LU in steps of	- Functions:	manual control only, autostop with gate, auto-
	0.1 LU		stop with gate and time.
- I Low:	-0.5 LU; S tolerance below Target Level, adjus-	- Level for gate:	-70,0 LUFS/LKFS; adjustable in the range
	table in the range from 0 to -10 LU in steps of		from -85 to -10 LUFS/LKFS in steps of 0.5
	0.1 LU		LUFS/LKFS
- MD, SD High:	+1.0 LU; I tolerance above dialog gated Target	- Time for gate:	1 s; adjustable in the range from 1 to 15 s in
	Level, adjustable in the range from 0 to 10 LU in		steps of 1 s
	steps of 0.1 LU		
- MD, SD Low:	-1.0 LU; I tolerance below dialog gated Target	l auduasa Danas In	other and (LDA LDA D)
	Level, adjustable in the range from 0 to -10 LU	Display:	strument (LRA, LRA D)
- ID High:	in steps of 0.1 LU +0.5 LU; S tolerance above dialog gated Target	Display:	Graphical display of the Loudness Range of the I measurement
- ID High.	Level, adjustable in the range from 0 to 10 LU in	Mode:	LRA Bar; LRA Bar, MagicLRA, MagicLRA + I,
	steps of 0.1 LU	Wiode.	MagicLRA + I + Num selectable
- ID Low:	-0.5 LU; S tolerance below dialog gated Target	Scale range:	10 LU ; 6 LU, 10 LU, 20 LU, 30 LU selectable
15 20	Level, adjustable in the range from 0 to -10 LU	LRA low range:	2 LU ; adjustable in the range from 0 to 30 LU in
	in steps of 0.1 LU		steps of 0.5 LU
	•	Comfort zone:	4 LU; adjustable in the range from 0 to 30 LU in
Netflix			steps of 0.5 LU
Scales:	ITU+9: +918 LU, ITU0: 030 LKFS	LRA high range:	depends on the selected scale range and the
Weighting filter:	ITU BS.1770 (k)		spread of the comfort zone
Target Level:	-24 LKFS	Colors:	selectable for each range, 32 predefined colors
Momentary:	400 ms		
Shortterm:	0 -	Laudness Chart Inc	tu mont
 Integration Time: Integrated Gate: 	3 s	Loudness Chart Ins	 Horizontal running bargraphs with individually
- Absolute Threshold:	_7001KES	i unctions.	definable colors evaluate the common quality
- Relative Threshold:	-10.0 LN 3		of Loudness values TP, M, S, I acc. to selected
Dialog Gated:	10.0 L0		standard
- Target Level:	-27 LKFS		 Progress of a measurement (value over time)
- Threshold:	-15 %; adjustable in the range from 0 to 100 %		of one of the four selectable values M, S, I or
	in steps of 1 %		TP drawn as graph in a coordinate system
- Absolute Threshold:	-70.0 LKFS ; adjustable in the range from -80.0		 Vertical bargraph for the selected value
	to -40.0 LKFS in steps of 0.5-LKFS, switchable		 Adjustable time ranges
- Relative Threshold:	−10.0 LU ; adjustable in the range from −40.0 to		 TP scale and operation range selectable
	0 LU in steps of 0.5 LU, switchable	Display:	Bargraph:
- Dialog channels:	L, R, C; each available channel selectable		Color change of the running bargraph indica-
Toleranzen			tes the section the loudness value is moving
- M, S High:	+1.0 LU; M tolerance above Target Level,		in: normal range, operation range, Headroom,
	adjustable in the range from 0 to 10 LU in steps of 0.1 LU		Over (availability depending on selected value)
- M, S Low:	-1.0 LU; M tolerance below Target Level, adjus-		Chart-Graph:
- IVI, 3 LOW.	table in the range from 0 to -10 LU in steps of		Continuously drawn graph (value over time)
	0.1 LU		of one value as line with colored filling corres-
- I High:	+0.5 LU; S tolerance above Target Level, adjus-		ponding to the color selection of the horzontal
	table in the range from 0 to 10 LU in steps of		bargraphs, added with Tolerance Indicator or
	0.1 LU		position of the relative gate (if selected)
- I Low:	-0.5 LU; S tolerance below Target Level, adjus-		Buttons for the selection of the loudness
	table in the range from 0 to -10 LU in steps of		value and the time range
	0.1 LU		

TP-Skala:

TP-Arbeitsbereich:

Time Range: Time grid adjustment for the coordinate system

and the horizontal bargraphs: Auto, 10 s, 30 s, 1 min, 5 min, 10 min, 30 min, 1 h, 2 h selectable **TP60: +3 .. -60 dB**, TP20: +3 .. -20 dB **0 dB**; einstellbar im Bereich von 0 bis −20 dB in

1-dB-Schritten

Colors: 32 individually selectable for normal range,

operation range and Headroom

Spectrum Analyzer (Real Time Analyzer - RTA)

Spectral distribution display of the frequency range of single channels, channel pairs or groups.

Functions: Input selection

Peak hold on/off, color selection

A, C, Linear weighting Integration time

Set reference Scaling

 Frequency range Bargraph arrangement

Display-Hold

Buttons for on-screen control of the frequency range, the channel selection, the scale steps, the reference range and the Peak hold reset

Input sources: selectable: all channels without LF, all channels,

individual channels

Frequency range: Norm: 20 Hz to 20 kHz,

additional H band > 20 kHz switchable

LE: 5 Hz to 5 kHz

Number of bands: 1/3-octave: 31 bands 1/6-octave: 61 bands

• 1/12-octave: 120 bands Linear; Linear, A, C selectable

Weighting filter: Peak hold indicator: 1 s, 2 s, 4 s, 10 s, 30 s, manual reset or off

Measuring range: 45 dB max. Reference range: -36 to +36 dB Scaling: Steps with 3, 6 or 9 dB

Integration time (ballistics): Impulse, Fast, Slow, Peak (10 ms)

LF Bleed

Orientation:

Function: Display of the amount of unwanted high-fre-

quency audio signals in the LFE channel above

a user-definable frequency

LFE bargraph with TP60 scale (-60 to 0 dBFS) Display: Frequency: **120 Hz**; selectable in the range from 20 Hz to 250 Hz in steps of 1 Hz

vertical or horizontal

Colors: 32 colors selectable for background and indivi-

dual sections of the bargraph display

Application Monitoring

Full Mono to extensive Surround and Immersive control (Monitor Control), bass management, speaker level calibration and volume level monitoring with numerical display, SPL value calculation or direct measurement with internal microphone.

General

Functions: Instrument for monitoring Mono, Stereo, Surround and Immersive signals

Arrangement of up to 32 loudspeakers

· Multifunctional rotary knob for controlling the volume level and other functions

· Definable loudspeaker functions Solo, Cut, Phase

DIM and MUTE function

Numerical display

Calculation of the SPL value of the electrical

Measured SPL value via internal or external microphone

 4 switchable inputs A/B/C/D (via one button or via separate buttons), can be labeled individually

 4 switchable loudspeaker sets A/B/C/D (via one button or via separate buttons), can be labeled individually

 Downmix instrument (coupled with A-input) with customizable coefficients

B/C/D outputs same as A or Stereo or Off

· Instrument for switching between loudspeakers and headphones (Phones output can be routed to the outputs or a Dante® resp. RAVENNA® connection)

· Bass management to operate full range and small speakers in the same system, support of two LFE channels

Two dedicated bass management outpus

 Support of Array loudspeakers acc. to Dolby[®] guidelines

LF-Boost function switchable

· Surr-Att function switchable

• HP filter for all bass managed speakers and All-pass filter for others

· For formats without an LFE, an additional SUB output channel is added

• Input-Instrument with input selection for the use of Intercom functionality between several TouchControl 5

• Button with toggle function for the selection of the loudspeaker modes Solo, Cut or Phase

Stylized circular speaker representation acc. to the selected audio format (ITU circle)

· Control of the selected speaker function by single or multiple tapping of the corresponding speaker symbols (toggle or multiple)

 Color indication of the loudspeaker state (green: active, red: inactive, outline: phase inverted)

Switchable SPL or Loudness display

Bargraph for summation of channels

the reference point, e. g. 78 dBA)

vertical or horizontal selectable

 Toggle buttons for input A/B/C/D, output A/B/C/D, Mono, Dim, Mute, LF-Boost, Surr-Att, Phones, Downmix, Reference

Absolute (dBC, dBA) or relative (0 dB refers to

A; Linear (None), A, C, CCIR, K (ITU BS.1770)

Listening volume display

SPL Meter

Display:

Display:

Display SPL value:

Orientation: Weighting:

Integration time:

selectable IIR: 125 ms (fast) oder IIR: 1000 ms (Slow)

selectable Reference point:

78 dB; adjustable in the range from 68 to 88 dB in steps of 1 dB

includes 32 dB Scale range:

Scale:	68 to 100 dB(A) in steps of 2 dB with reference point 78 dB(A)	- Frequency:	300 Hz ; adjustable in the range from 20 Hz to 20 kHz
	 Changes when changing the reference point Changing the reference point does not 	- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in steps of 0.1
	change the reference level	Band 3	
Speakers		- Type:	Peak ; Peak, Low Shelf, High Shelf, Low Pass, Band Pass, High Pass, All Pass 2P, All Pass 4P,
Modes:	Stereo; Mono, Stereo, Surround, Immersive selec-		Notch; selectable
- Surround formats:	table 5.1 ; LCR, LCM, 4.0, 5.0, 5.1, 6.0, 6.1, 7.0, 7.1	- Gain:	0 dB ; adjustable in the range from −18 to +18 dB in steps of 0.1 dB
	selectable	- Frequency:	600 Hz; adjustable in the range from 20 Hz to
- Immersive formats:	5.1.4 ; 5.1.2, 5.1.4, 5.1.6, 7.1.2, 7.1.4, 7.1.6, 9.1.2,	D(O):	20 kHz
Inputs A/B/C/D:	9.1.4, 9.1.6, 22.2, Array selectable	- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in steps of 0.1
- Delay:	0 ms ; adjustable in the range from 0 to 200 ms	Band 4	0.0000 0.1
, -	in steps of 0.1 ms	- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,
- Gain (Trim):	0 dBFS ; adjustable in the range from -60 to +6 dBFS in steps of 0.5 dBFS	.,,,,	Band Pass, High Pass, All Pass 2P, All Pass 4P, Notch; selectable
- Mix In:	Off or On, allows to mix additional signals to the	- Gain:	0 dB ; adjustable in the range from −18 to +18
	selected input, e. g. for Talkback		dB in steps of 0.1 dB
LF-Boost:	On or Off , increases the level of the LF output by +10 dB	- Frequency:	1.2 kHz ; adjustable in the range from 20 Hz to 20 kHz
Sur att.:	On or Off , decreases the level of the surround	- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in
	channels by -3 dB		steps of 0.1
		Band 5	
Individual Loudspeake		- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,
•	le in the network can be assigned to four sets		Band Pass, High Pass, All Pass 2P, All Pass 4P,
(A/B/C/D) and individua	ally adjusted for the respective listening situation.	0 :	Notch; selectable
Name:	max. 3 characters	- Gain:	0 dB ; adjustable in the range from −18 to +18 dB in steps of 0.1 dB
Type:	full range or non-full range	- Frequency:	2.4 kHz ; adjustable in the range from 20 Hz to
Gain:	0 dB ; adjustable in the range from -24 to +12	rroquoney.	20 kHz
	dB in steps of 0.5 dB	- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in
Delay:	0 ms ; adjustable in the range from 0 to 200 ms		steps of 0.1
	in steps of 0.02 ms (units can be ms, m, ft)	Band 6	
Polarity:	+ or –	- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,
BM Level:	Increasing or decreasing the signal level sent from each non-full range speaker to the LFE		Band Pass, High Pass, All Pass 2P, All Pass 4P, Notch; selectable
E " (EO)	output	- Gain:	0 dB ; adjustable in the range from –18 to +18
Equalizer (EQ):	8 bands, can be enabled individually	Г	dB in steps of 0.1 dB
Phase Curve:	On or Off , the phase curve will be added to all bands	- Frequency:	4.8 kHz ; adjustable in the range from 20 Hz to 20 kHz
Music Curve:	Activates the Dolby Atmos® Music Curve, ap-	- Bandwidth (Q):	0.7; adjustable in the range from 0 to 10 in
	plies it to the bands 4 to 8 and activates it with		steps of 0.1
	the corresponding values (bands 1 to 3 remain	Band 7	
	unchanged)	- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,
Individual band settings: Band 1			Band Pass, High Pass, All Pass 2P, All Pass 4P, Notch; selectable
- Type:	Peak ; Peak, Low Shelf, High Shelf, Low Pass, Band Pass, High Pass, All Pass 2P, All Pass 4P,	- Gain:	0 dB ; adjustable in the range from −18 to +18 dB in steps of 0.1 dB
	Notch; selectable	- Frequency:	9.6 kHz; adjustable in the range from 20 Hz to
- Gain:	0 dB ; adjustable in the range from -18 to +18 dB in steps of 0.1 dB	- Bandwidth (Q):	20 kHz 0.7 ; adjustable in the range from 0 to 10 in
- Frequency:	150 Hz ; adjustable in the range from 20 Hz to	- Dandwidth (d).	steps of 0.1
r requericy.	20 kHz	Band 8	steps 01 0.1
- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in	- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,
	steps of 0.1	71	Band Pass, High Pass, All Pass 2P, All Pass 4P,
Band 2			Notch; selectable
- Type:	Peak; Peak, Low Shelf, High Shelf, Low Pass,	- Gain:	0 dB ; adjustable in the range from −18 to +18
	Band Pass, High Pass, All Pass 2P, All Pass 4P,	Erogues a	dB in steps of 0.1 dB
- Gain:	Notch; selectable 0 dB ; adjustable in the range from -18 to +18	- Frequency:	19.2 kHz ; adjustable in the range from 20 Hz to 20 kHz
- Gailli	dB in steps of 0.1 dB	- Bandwidth (Q):	0.7 ; adjustable in the range from 0 to 10 in
	== Stope of off dB	24.4.1411 (4)1	steps of 0.1

Loudspeaker Calibration

Level calibration for each individual loudspeaker in each output set

Edit mode:

Gain:

- Solo: Editing the selected loudspeakers (symbols turn green) or
- Selected: Simultaneous monitoring of activated loudspeakers (green) and editing of the last selected speaker (yellow)

0 dB; adjustable in the range from −24 to +12

dB in steps of 0.5 dB

Delay: 0 ms; adjustable in the range from 0 to 200 ms

in steps of 0.2 ms

Delay unit: Time; Time, Distance Metric, Distance Imperial selectable

Test tone generator: On or Off

Pink Noise; Sine, White Noise, Pink Noise, - Signal:

XOver Tone selectable

- Level: 0 dB; adjustable in the range from −90 to 0 dB in steps of 0.5 dB

SPL calibration meter

- Input:

Internal microphone (Mic) or external microphone (XLR) or any Dante® channel

- Phantom supply:

- XLR Gain:

On or Off 0 dB; adjustable in the range from −8 to +60 dB in steps of 1 dB

- Weighting filter: A or C selectable

- Response: Slow; Fast (125 ms), Slow (1 s) selectable - Display SPL value: Absolute (dB(A)) or relative (0 dB refers to the reference point, e. g. 78 dB(A))

Bass-Management

Active: Frequency:

Yes; Yes, No (enables the bass management) 125 Hz; crossover frequency adjustable in the

range from 63 to 250 Hz

Sub level: Filters:

0; adjustable from -24 to 0 dB in steps of 1 dB Phase Safe; Phase Safe (phase ratios remain unchanged), Lagacy (conventional method), Bypass (manual filtering with EQ) selectable

Loudspeaker Selection Modes

Functions:

- Button for switching through the Solo, Cut, Phase modes
- Definable behavior when tapping the loudspeaker symbols in the circular display

Symbol behavior:

- in operation:
- Radio button style (alternate enable/disable) in Solo mode
- Adding style (select one or several, reset all by fast double tapping) in Cut mode
- in calibration mode:
- Adding mode (select one or several, reset each by tapping again)
- Reset all by pressing the rotary knob

Modes:

Solo; Solo, Cut, Phase successively selectable

Solo mode

- Function:
- · Solo in place (green), all other speakers muted (red)
- Solo mixed to defined loudspeaker
- Double tapping on a loudspeaker symbol solos the corresponding whole channel group
- Target loudspeaker:

Channel (Solo in place); channel, L, R, L+R, Center selectable, signal is always Mono

Cut mode

- Function:
- Cut mutes the selected loudspeaker (red), all other loudspeaker are kept active (green)
- Double tapping on a loudspeaker symbol mutes the corresponding whole channel group

Phase mode

- Function:
- · Phase switches the polarity of the selected channels
- Loudspeaker symbol changes to a green outline
- Channel labeling is marked with ^

DIM (Mute all)

Function:

DIM level:

- Button to reduce the monitoring volume by a predefined value or "Mute all"
- Can be used as momentary or latch/hold switch
- -20 dB; adjustable in the range from -80 to 0 dB in steps of 1 dB or Mute
- Temporarily adjustable by touching and holding the DIM button and turning the rotary knob

DIM Phones output: Off or On

Mute

Function:

- Button for muting the loudspeakers
- Can be used as momentary or latch/hold

Key/button and Volume meter appears red, Display: when Mute function is active

Mute Phones output: Off or On

Mono

Function:

- Button to output the signal in a mono-summed
- Output of a defined loudspeaker (depending on the selected Speakers mode)

Target loudspeaker:

L+R; Center, L, R, L+R, All w/o LFE/Sub selectable, signal is always Mono

Downmix (5.1, 7.1 only)

Functions:

- Downmix instrument is connected upstream of the A inputs
- Downmix output signal (L/R and/or Mono) can be routed into the Dante® network
- Customizable downmix coefficients
- Switching between current speaker setup and L-R monitoring
- Display and/or monitoring
- 3-channel TP Meter (for L/R and M Downmix)
- available for the surround formats 5.1, 7.1

Downmix rules:

- LS/RS summed to L/R
- LSR/RSR summed to L/R
- CS summed to L/R
- Front summed to L/R
- all other loudspeakers summed to L/R Individually customizable downmix coefficients using the sliders in the WebApp

Channel level for Downmix

(available channels depending on the selected Speakers mode)

Channel Gain:

Peakhold display:

-3 dB; adjustable in the range from -12 to 0 dB in steps of 0.5 dB -3 dB; adjustable in the range from -12 to

Mono Downmix sum:

0 dB in steps of 0.5 dB Off; 1 s, 2 s, 4 s, 10 s, 20 s, 30 s, manual reset

or Off selectable

Phones

Function:

Trim:

- Instrument/button for switching between loudspeakers and headphones
- · Phones output can be routed to the outputs or a AoIP connection

0 dB; adjustable in the range from −12 to 0 dB in steps of 0.5 dB

>

Input Instrument (Intercom Functionality)

unctions:

• In combination with the Talkback application, communication between several TouchCon-

trol 5 devices is possible

 Input instrument with input selection and level control for mixing external audio sources into the Monitoring signal

Input Trim control and Mute option for flexible

communication

Displays: • Numerical level display for the selected input

Appears red with Mute function active

Input level: **0.0 dB**; adjustable in the range from -60 to +6

dB in steps of 0.5 dB

Application Tools

Includes additional instruments

Label

Functions: Positionable text field

 Freely selectable colors for text and background (HSB color model, web colors)

Continuously adjustable font size

Font type selectableOrientation selectable

Label: Enter up to 15 characters (default: **Label**)

Background: Color setting for the background according to the HSB color model or entry of web color

to the HSB color model of entry of web

codes (8 digits)

Text: Color setting for the text according to the HSB

color model or entry of web color codes (8

digits)

Font Size: **60 %**; continoulsy adjustable font size in the

range from 20 % to 80 % of the label area

Font Style: Thin; Thin, Thin Italic, Normal, Italic, Bold, Bold

Italic selectable

Orientation: Horizontal or Vertical selectable

Optional Ethernet Power Injector 14554-xx

This IEEE 802.3af-compliant power injector is required when the Dante® AoIP network provides insufficient or no power over Ethernet (PoE).

Manufacturer: Phihong Technology Co., Ltd., No. 568, Fusing

3rd RD., Gueishan District, Taoyuan City, Taiwan
Model: POE15M-1AFE - Single Port Power over Ether-

net (PSE), Gigabit-compatible

Standard: IEEE 802.3af

Input: 100 - 240 V AC, 800 mA, 50 - 60 Hz

Output: 56 V DC, 275 mA, 15.4 W

Performance class: 0

PD power range: 0.44 to 12.94 W PSE power usage: maximum: 15.4 W

Certificates: CE, UKCA, UL (Canada, US), FCC, IC, LPS,

CAN ICES-3(B)/NMB-3(B)

Territorial coverage: North America, Canada, Europe, Great Britain,

Australia/New Zealand

Items of Delivery

TouchControl 5 Dante:

- Dante® based immersive monitor controller with metering.
- User customizable table-top device with 5" touch display, build-in calibrated microphone and 32 Dante®-channels for stereo, surround and immersive speaker formats
- Monitor Controller for up to 4 input and 4 output sets
- · Speaker level calibration
- SPL measurement
- Bass management
- Premium metering (PPM, TP, Moving Coil)
- Audio Vectorscope, Real Time Analyzer
- Loudness, Loudness Chart, SPL and LRA
- Toolbox with leveling and talkback
- Quick start guide

Order no.: 320517ND

TouchControl 5 Ravenna:

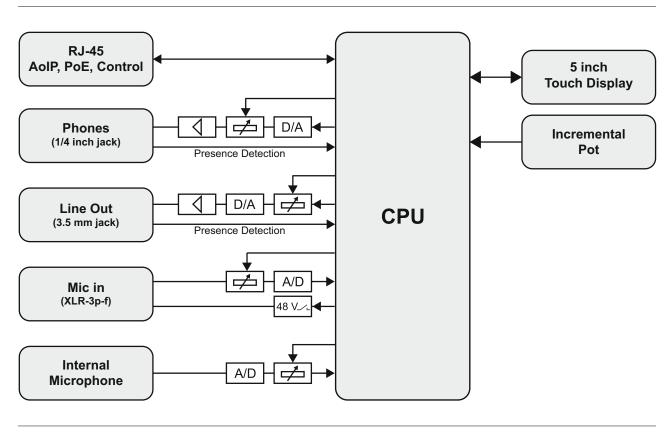
- RAVENNA®/AES67/ST 2110 based immersive monitor controller with metering.
- User customizable table-top device with 5" touch display, build-in calibrated microphone and 32 RAVENNA®-channels for stereo, surround and immersive speaker formats
- Monitor Controller for up to 4 input and 4 output sets
- Speaker level calibration
- SPL measurement
- Bass management
- Premium metering (PPM, TP, Moving Coil)
- Audio Vectorscope, Real Time Analyzer
- Loudness, Loudness Chart, SPL and LRA
- Toolbox with leveling and talkback
- Quick start guide

Order no.: 320518ND

Optional Accessories

- Ethernet Power Injector 14554, PoE tabletop device with corresponding mains cable for different regions:
 - Europe: **14554-EU** (mains cable for Europe or similar)
 - USA: **14554-US** (mains cable for USA or similar)
 - Australia: **14554-AU** (mains cable for Australia or similar)
 - UK: 14554-GB (mains cable for United Kingdom or similar)
 - International: **14554-IN** (includes all cables)
- Metal mounting plate 1166 for mounting with 3/8" holds (e. g. gooseneck, mic stand)

Block Diagram



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