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Table of Contents



Table of Contents

1. Introduction	5
2. Overview Operation The Screen Display	9
3. Getting Started Power Supply Attaching the Hand Strap Unfolding the Stand Connecting the XL2 Switching the XL2 On and Off Select Measurement Function. Calibration prior Measurement	
4. Sound Level Meter Overview Sound Level Meter - Getting Started RTA Measurement - Getting Started Correction Value KSET Reporting Logging Wav-File Recording Events (optional) Limits	

5. Acoustic Analyzer	64
FFT Analysis + Tolerance	64
Reverberation Time RT60	
Polarity	
Delay Time	
1/12 Octave + Tolerance (optional)	
Speech Intelligibility STIPA (optional)	106
Averaging of STIPA Results	113
STIPA with Ambient Noise Correction	116
6. Audio Analyzer RMS /THD+N Oscilloscope	121
7. Calibration	124
8. Profiles	128
9. Spectral Limits Option (Capture + Tolerances)	135
10. System Settings	147
11. Scheduler	



Table of Contents

154
159
165
167
171
174
177
177
178
190
201
204
209



1. Introduction

Thank you for purchasing the XL2 Audio and Acoustic Analyzer. The XL2 Analyzer forms the unique combination of a state-ofthe-art Sound Level Meter, a comprehensive Acoustic Analyzer as well as a powerful Audio Analyzer. The wide range of functionalities is tailored for:

- Live Sound
- Installed Sound
- Environmental Noise
- PA Hire
- Broadcast
- AVI-Installation
- Cinemas
- Building Acoustics
- Occupational Health
- Quality Control
- Vehicle Noise Inspection
- R&D and Service
- Education and Training

How to Read this Manual

Menu items displayed on the XL2 page screens are shown in this user manual by a bold font, e.g., **SLMeter**, **Parameter**, ...



Introduction

Product Configurations

The following items are included with the respective model:

- XL2 + M4260: XL2 Analyzer
 - M4260 Microphone with pouch
 - Wind protection
 - Microphone holder + Adapter 5/8" 3/8"
 - Test Signal CD
 - Li-Po battery
 - USB cable
 - Hand strap
 - Operating manual
- XL2 + M2210: XL2 Analyzer
 - M2210 Microphone with pouch
 - Wind protection
 - Microphone holder + Adapter 5/8" 3/8"
 - Test Signal CD
 - Li-Po battery
 - USB cable
 - Hand strap
 - Operating manual

- XL2 without
 - Microphone:
- XL2 AnalyzerTest Signal CD
- USB cable
- Li-Po battery
- Hand strap
- Operating manual





XL2 interfaces:







1 XLR Signal Input

Either a balanced audio signal, an NTi Audio measurement microphone or the microphone pre-amplifier MA220 plug directly into the XLR input. The Automated Sensor Detection (ASD) technology reads the electronic data sheet of any connected NTi Audio device after the 48 V microphone phantom power is activated.

2 RCA Input

Unbalanced audio signal input.

③ Voice Note Input

Internal microphone for recording voice notes and measuring polarity and delay time. For polarity measurements, an external microphone may also be used.

4 DC Power Socket

Socket for mains power adapter. Further details in the chapter "Power Supply" in this manual.

5 USB Connector

Mini-B USB connection, for accessing the SD Card.

6 Digital I/O

Programmable digital inputs/outputs.

O SD Card

For storing data, screenshots, voice notes, wav-files.

8 TOSLink Output

24 bit linear PCM audio signal output. For future use; not active at this time.

9 Headphone Output

The XLR/RCA input signals are routed to the headphone output. Connecting headphones mutes the internal speaker. To connect the headphone output directly to a line input, a load impedance < 8 kOhm is required for proper operation. This can be achieved by inserting a 1 kOhm resistance between tip and ground of the output jack.

10 Speaker

The XLR/RCA input signals are routed to the speaker. Press the speaker button O to toggle the speaker on/off. Press and hold the speaker button to access the volume control. Set the speaker level with the rotary wheel O.

11 Tripod Mount

Mechanical mount for attaching the XL2 to a tripod or microphone stand.



- 2 Volume of Speaker and Headphone Output ④
 - Press the button to enable or disable the speaker.
 - The speaker is activated and the speaker or headphone icon shows up in the upper menu bar.
 - Press and hold the speaker button $\textcircled{\bullet}$.
 - A pop-up window for volume is displayed.

(RMS/TH	D XLR
Filter	Z-WEIGHTING
LVLRMS	
THDN	en ois +80 CESSIEND Pre Notch dB
FREQ	1.00000kHz

Operation



1 Page Control

Switches amongst various screens depending on the menu function selected.



• Hold the speaker button ④ down and adjust the level of the speaker and headphone outputs with the rotary wheel ④. Digital gain control prevents gain increases beyond a level that will cause signal clipping. The maximum gain setting depends on the connected input signal.

3 Power & Backlight

The power button (a) switches the instrument on. The XL2 is immediately ready for operation. Holding down the power button (a) for one second switches the XL2 off. Additionally, a brief press of the power button (a) toggles the backlight during operation.

4 Limit 📼

- SLMeter: The limit button lights up green, yellow or red according to the settings in the Limit page. Press the limit button to access the Limit page. For more details visit the chapter "Limit Setting".
- **FFT + Tol:** The limit button lights up green for results within tolerance and red for out-of-tolerance results.
- **Polarity:** The limit button lights up green at positive polarity and red at negative polarity.
- 1/12 Oct + Tol: The limit button lights up green for results within tolerance and red for out of tolerance results.

5 Pause

Pauses the current measurement. Press the pause button II or the start/stop button II to continue the measurement.

6 Start/Stop 🕨

Starts and stops a measurement.

7 Enter 🕑

Confirms a selection.

8 Rotary Wheel

Selects the required measurement function or the individual measurement parameter.

9 ESC esc

Terminates an entry, returns to the top menu level or closes an open window.



The Screen Display

The XL2 displays the actual sound levels also in case of no measurement has been startet. Any displayed averaged level refers to the previous measurement periode. The XL2 shows four lines in case of no previous measurement periode is applicable.

Updating of display

• Numeric values

Updated every 500ms independent of the measurement function. The maximum time span between the end of the measurement period and the first test result display is 500ms.

 Graphs and spectra Updated every 50ms



The display is updated continuously during measurement.



Measurement Results Individual level measurement results.

2 Main Menu

SLMeter/RTA	SPL & RTA Measurement	
FFT + Tol	FFT Analysis with optional Tolerance Management	
RT60	Reverberation Time RT60	
Polarity	Polarity	
Delay Time	DelayTime	
RMS/THD+N	RMS Level and Distortion	
Oscilloscope	Scope	
1/12 Oct + Tol	Spectral Analysis with Tolerance Management (optional)	
STIPA	Speech Intelligibility (optional)	
STIPA Cinema Meter	Speech Intelligibility (optional) Calibration and verification of cinema loudspeaker systems (optional)	
	Calibration and verification of cinema	
Cinema Meter	Calibration and verification of cinema loudspeaker systems (optional)	

3 Page Selector

Toggles between available measurement and result pages within the same function menu. Alternatively use the page button **1**.

4 Input Selector

Select XLR or RCA input connectors as the source.

5 Memory Menu 🗖

The memory menu is used for data storage management. One of the following symbols flashes before or after starting the measurement:

- **9-8-7** Time in seconds until instrument is settled and measurement will start.
- **RUN** Indicates the ongoing measurement.
- **LOG** Indicates the ongoing measurement with data logging.
- AUD Indicates the ongoing measurement with data logging and audio recording.
- **Evt** Indicates an ongoing triggered-event recording.



After a completed measurement, the memory symbol **a**? indicates a non-saved measurement report. Your measurement needs to be stored manually. For more details visit the chapter "Data Management".

6 Speaker/Headphone

Indication of enabled rear speaker or headphone output.

Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on
- 48V Phantom power is switched off.

8 Real-Time Clock

The real-time clock is set in the **System** menu.

9 Battery Symbol

The battery symbol indicates the battery status as follows:

Using rechargeable Li-Po battery:



Level indication 100% (U > 4.0 Volt).



Level indication: 75%: U = 3.9 - 4.0 Volt 50%: U = 3.8 - 3.9 Volt 25%: U = 3.7 - 3.8 Volt



Level indication 0% (U < 3.7 Volt). The battery is almost empty and must be recharged.



The battery is being recharged by the mains power adapter or the USB connection to PC.

Using standard AA batteries:

No level indication as long as U > 4.5 Volt.



Level indication 0% (U < 4.5 Volt). The batteries are almost empty and should be replaced.

Using mains or supply:

No indication of battery status.



Power Supply

The XL2 offers flexible power management options and can be operated either by

- Replaceable, rechargeable lithium-polymer (Li-Po) battery (included with the XL2 Analyzer)
- 4x AA-batteries
- Mains Power Adapter

The new battery is charged to approximately 50% and should be fully charged before use with:

Battery Charger	Charging Time: approx. 3 hours
(optional)	NTi Audio #: 600 000 332
Mains Power	Charging Time: approx. 6 hours
Adapter	Leave the battery inside and switch off
(optional)	the XL2 🕲 for charging.
	NTi Audio #: 600 000 333

USB Power Charging Time: approx. 6 hours from PC Switch off the XL2 (a) for faster charging.

Operation using Mains Power Supply

You can also operate the XL2 with mains power using the NTi Audio DC power supply unit. During operation with this external Mains Power Adapter, it is recommended to leave the batteries inside the instrument.

Mains Power Adapter

- Electrically-isolated, linear DC power supply
- Non-NTi Audio power supplies may have negative effects on measurements.
- A decreased THD+N performance will occur with unbalanced input signals when using switching-type power supplies (approx. 3 dB).
- Damage caused by using an inappropriate external DC supply is not covered by warranty.





Rechargeable Li-Po Battery

- Open the battery cover at the rear of the instrument.
- Insert the rechargeable Li-Po battery with the contacts edge first.
- Close the battery cover.



Switch off the XL2 (3) for faster recharging by mains power adapter or USB connection.

Caution

- Switch the XL2 off prior opening the battery cover in order to prevent any electronic discharges.
- Avoid short-circuits.
- Operate and charge the battery between 0°C and 45°C (32°F 113°F).
- Do not heat the battery above 60°C.
- Do not place the battery in or near fire.
- Do not solder directly on to the battery.
- Do not disassemble the battery.
- Do not insert the battery in reverse polarity.





AA-Batteries

Alternatively, the XL2 can be powered by AA type batteries

- Open the battery cover.
- Insert 4 fully-charged AA batteries, observing correct polarity of the +/- indications in the battery compartment.

The polarity alternates with successive batteries.

• Close the battery cover.



- Switch the XL2 off prior opening the battery cover in order to prevent any electronic discharges.
- Use only similar batteries from the same manufacturer.
- Replace discharged batteries with new ones.
- Do not mix used and new batteries.
- During operation, the battery temperature may increase noticeably. This is not a defect.
- Remove all batteries if the XL2 is not to be used over a long period of time.





Attaching the Hand Strap

To avoid accidentally dropping the XL2, a hand strap is supplied with the instrument.

- Pull the loop of the hand strap through the opening.
- Pull the other end of the hand strap through the loop.
- Pull the hand strap tight.





Unfolding the Stand

A convenient table stand is attached to the rear of the instrument.

• Unfold the stand and rest the XL2 on a flat surface.

Connecting the XL2

Acoustic Measurements

Connect an NTi Audio measurement microphone to the XLR input connector of the XL2.







Audio Measurements: XLR Connection

- Connect the source to the XL2 with an XLR cable.
- Select the XLR input in the menu.

Audio Measurements: RCA Connection

- Connect the source to the XL2 with an RCA (cinch) cable.
- Select the RCA input in the menu.







Switching the XL2 On and Off

Switching On the XL2

Press power B to switch on the XL2.

 $^{\&}$ There is a brief sound of relays and the display illuminates.

Switching Off the XL2

Press power $\textcircled{\sc s}$ and hold it down for one second to switch off the XL2.

Select Measurement Function

- Navigate with the rotary wheel ⁽¹⁾ to the main menu bar.
- Confirm with enter $oldsymbol{\Theta}$ to select the measurement function.

🖇 The main menu window opens



Main menu with enabled options

- Scroll up/down with the rotary wheel ⁽¹⁾ to select the required function and confirm with enter ⁽²⁾.
- The measurement function is selected.



Setting the Parameters with the Rotary Wheel ${f O}$

- Turn the rotary wheel ⁽¹⁾ to navigate within the display screen.
- The selected parameter is highlighted with a black background.
- Confirm with enter @.
- The parameter display flashes and/or available parameters or settings are shown.
- Turn the rotary wheel ⁽¹⁾ to set the parameter or toggle with enter ⁽²⁾ through the settings.
- Confirm with enter @.

Now the parameter is set.

Calibration prior Measurement

We recommend the daily calibration of the XL2 Analyzer with the precision calibrator. This ensures accurate measurement results.

For more details see chapter "Calibration".

The XL2 provides, together with the measurement microphone, a precise sound level meter for monitoring live sound events and/or environmental noise.

For example, Actual, Lmin, Lmax, Leq with frequency weighting A, C and Z and time weighting's F and S can be measured at the same time. All measurement results are simultaneously available. You may log all acquired level information, including real-time information, onto the removable SD Card. To complete the documentation of the measured sound pressure levels, the XL2 offers wav-file recording, as well as the facility to add voice notes for each measurement. The accuracy of the XL2 fulfills IEC 61672-1, class 1. The XL2 measures correction values between the loudest point of the live event and the actual measurement positions in accordance with DIN 15905-5 and SLV 2007.

In parallel with the wide band parameters, the XL2 measures the real-time spectrum either in 1/1 or 1/3 octave-band resolution. The RTA perfectly suits tasks such as optimization of sound systems

Extended Acoustic Pack (optional)

The Extended Acoustic Pack offers the following additional features for sound-level and acoustic measurements:

- SLMeter/RTA function
 - Recording of linear wav-files (24 bit, 48 kHz)
 - Percentiles for wide band and spectrum: LAF1%, LAF5%, LAF10%, LAF50%, LAF90%, LAF95%, LAF99%
 - Sound Exposure Level LAE
 - 100ms logging
 - Event-triggered audio and data recording
 - Time weighting: Impulse (LxI, LxIeq with x= A, C, Z)
 - True peak level in 1/1 and 1/3 octave resolution
 - Clock-Impulse Maximum Level (TaktMax) and values as specified in DIN 45645-1
- FFT function
 - High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz
- RT60 function
 - Reverberation time RT60 in 1/3 octave resolution



The sound level meter offers different pages:

Page Selection using the Page Button

• Press page 🗇 to toggle between the numeric result page and the RTA page.

Page Selection using the Rotary Wheel O

- Select the sound level page 123 with the rotary wheel ⁽¹⁾
- Confirm with enter 🕑



Menu with enabled options

You have selected the individual sound level meter page. 6



Sound Level Meter

Displays the selected broadband sound level results. You can change the font size of the result. Depending on the font size chosen, the XL2 shows 3 or 5 results simultaneously. Individual frequency weighting, time weighting, actual, minimum, maximum and correction values can be chosen for each displayed result.



RTA Real-Time Spectrum

Displays the 1/3 or 1/1 octave spectrum of the selected sound level within the audio band. Additionally the wide band result is shown graphically by a bar.

Correction Value

IKSET This page provides a wizard to measure correction values, which could be helpful for compliant sound level monitoring of live events. It measures the correction values between the actual measurement position and the loudest position accessible by the audience. The XL2 displays and logs the selected sound pressure levels including the correction values, allowing the sound engineer to monitor the sound level at the loudest position in the audience. The measurement meets the requirements of the standards DIN15905-5 and SLV 2007. For details refer to chapter "Correction Value KSFT"





Report Setting

Here you set which sound pressure levels shall be stored in a .txt file after the completed measurement. Choose between:

ALL Stores all sound pressure levels.

Selected Stores up to 10 different individuallydefined sound pressure levels.

For details refer to the chapter "Report".

Logging Page

_09| The XL2 features a powerful sound level meter data logger, which allows you to record all required sound level values during the measurement. In the logging page you set which sound pressure levels shall be logged over time. Choose between:

- ALL Logs all sound pressure levels.
- Selected Logs up to 10 different individually-defined sound pressure levels.

For details refer to the chapter "Logging".



Event Setting (optional)

The event function is available with the optional Extended Acoustic Pack. The XL2 event feature offers the following functionalities:

- Automated event triggered at noise levels above/ below a preset value including setting markers for specific noise categories. Application example: Recording the noise level for LAF > 80 dB.
- Event triggered by external keypress of the XL2 Input Keypad. You can utilize four keys (1-4) to categorize any noise of interest or noise to be later excluded in the post-processing. A typical application is categorizing any annoying industrial noise by persons living in the neighborhood.



Lim Limit Setting

Here you set the function of the limit LED (), thereby highlighting any sound level that exceeds the pre-set limits in orange or red color.

Further, external peripherals are controlled by the optional Serial I/O Interface based on the sound level. such as displaying sound levels on a large external red-orange-green lamp.

For details refer to the chapter "Limit Setting".

25

Sound Level Meter



Overview

Numeric Result Page

The numeric result page 123 displays the selected broadband sound level results. You can change the font size of the result. Depending on the font size chosen, the XL2 shows three or five results simultaneously. Individual frequency weighting, time weighting, actual, minimum, maximum and correction values can be chosen for each displayed result.

Real-Time Analyzer Page

The real-time analyzer page RTA **I** measures and displays the 1/3 or 1/1 octave spectrum from 6.3 Hz to 20 kHz including wideband results. The real-time spectrum RTA is measured in parallel with the A and Z-weighted wideband levels.







1 Sound Level Result 1

All sound levels are measured and logged simultaneously. You select which sound levels should be displayed.

Change Parameter

- Turn the rotary wheel O to select the parameter **Lxx**.
- Press enter 🕑 to open the selection menu and choose the wideband sound pressure level to be displayed.
- Turn the rotary wheel ^(C) to select the required test result parameters and confirm with enter ^(C).

Change Font Size

- Turn the rotary wheel 0 to select the actual test result.
- Press enter ④ 1x, 2x or 3x to set the font size to small, medium or large.

The XL2 shows 3 sound levels on the display if a large font is chosen, otherwise it shows 5 sound levels.

2 Sound Level Result 2 & 3

Follow the setting instructions for sound level result 1.

3 Sound Level Result 4 & 5

To display sound level results 4 & 5, choose a smaller font for all results. Follow the setting instructions for sound level result 1.

(4) Input Range

The XL2 Analyzer provides three input ranges to accommodate the wide range of input signals. The individual ranges are based on the microphone sensitivity setting in the calibration menu of the XL2. For example at a sensitivity S=20mV/Pa the input ranges are

- Lower range: 10 110 dBSPL
- Middle range: 30 130 dBSPL
- Upper range: 50 150 dBSPL

Select the lowest possible input range according to the maximum level expected during the measurement; e.g., if the sound pressure will always be below 110 dBSPL, then select the lowest input range 10 - 110 dBSPL.

6 Run Indication

The run indication shows the measurement status running, paused or stopped. Various measurement settings are locked during ongoing measurements, such as changing the input ranging or the preset measurement time.

Actual Measurement Time

Counts actual measurement time in hrs:min:sec. Supports time modes: continuous, single and (synchronized) repeat.

Timer Mode Continuous

(applicable for standard measurements) All values are recorded and monitored continuously after starting a measurement with start **F**. The actual measurement time is shown.

Timer Mode Single

Automatically stops the measurement after the pre-set measurement time.

- Set the required measurement time.
- Start the measurement 🖭
- The actual measurement time counts back to zero and the measurement ends.
- All measurement results may be recalled.

5 Error Indication

OVR Overload

The error message **OVR** is displayed when:

- The measured sound pressure level exceeds the pre-set measurement range during the measurement period. Once this happens, the **OVR** indication remains displayed for the complete measurement period -> choose the next higher measurement range.
- The measured sound pressure level is just below the maximum level of the connected NTi Audio measurement microphone.

LOW Input level too low

The error message **LOW** is displayed when:

- The measured sound pressure level is below the pre-set measurement range. In this case the results shown are most likely higher than the actual measured sound pressure level -> choose the next lower measurement range.
- The measured sound pressure level is just above the residual noise of the connected NTi Audio measurement microphone, which reduces the measurement accuracy.





- **1** -

Timer Mode Repeat

Provides automated repeated measurements with user-defined, preset measurement time cycles.

- Set the required measurement time.
- Press start 🖭.
- The actual measurement time counts back to zero. When the preset measurement time has elapsed, the measurement time and the measurement results are reset and a new measurement is started. All measurement results of the previous cycle are reset.

For consecutive storage of all measurement cycles, select the parameter **Naming+Saving: auto** in the memory menu. Each individual cycle result is then automatically stored to the SD Card. Timer Mode Repeat Synchronized

467

Provides automated repeated measurements synchronized to the XL2 real-time clock. Press start **I** to begin the measurement. In order to align the selected preset measurement time with the real-time clock, the XL2 shortens the first cycle to match the real-time clock synchronization. All following measurement cycles are synchronized to the real-time clock.

For example, the cycle time setting is 30 minutes and the measurement starts at 7.50 a.m. -> the first test cycle measures from 7.50 - 8.00 a.m. Thereafter a new test cycle starts automatically for 30 minutes. The 30-minute cycles repeat until the measurement is stopped.

The measurements in the synchronized repeat timer mode start exactly on the half or full hour in accordance with DIN 15905.

8 Preset Measurement Time

Adjustment of preset measurement time for single and repeat timer setting.



9 Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on
- 48V Phantom power is switched off.
- 10 Result Symbols / Capture

This field offers two functions:

- Symbols for measurement results 15
 - --- Upper RTA parameter displayed as line.
 - Lower RTA parameter displayed as bargraph.

• Capture

One of the displayed RTA readings may be captured. Any measurement data can then be compared with this captured reference live on the XL2 Analyzer. For example, compare the RTA spectrum of the left and right speakers.

- Select the parameter to be captured.
- Confirm with enter 🕑, to capture the reading.
- Select the upper RTA parameter 15 and choose Capt.
- Confirm with enter @.
- The lower RTA reading can be compared with the previously-captured reference data.

11 Measurement Result

Actual level result of the indicated frequency band. The cursor readout displays the center frequency and the level of the band pointed to by the arrow.

- --- Upper parameter displayed as line.
- Lower parameter displayed as bargraph.



12 Y-Scale setting

- Select the Y-Axis with the rotary wheel 0 and confirm with enter 0.
- Select the zoom factor between **20, 10, 5**, **2.5 dB/div** and confirm with enter **(**.
- Scroll up and down with the rotary wheel O to select the Y-axis range.
- Confirm with enter @.

13 RTA Measurement Result

Real-time analyzer results in 1/1 octave or 1/3 octave band resolution. Adjust the resolution at (7).

14 X-Scale setting

Toggles X-scale range between

- 20 Hz 20 kHz RTA levels including wide band results
- 6.3 Hz 8 kHz RTA levels including wide band results
- 6.3 Hz 20 kHz RTA levels
- Select the X-Axis with the rotary wheel ⁽¹⁾ and confirm with enter ⁽²⁾.
- Toggle with the rotary wheel ⁽¹⁾ between the ranges.
- Confirm with enter .

15 Broadband Results

Actual sound level with time weighting F (=Fast).

- A A-weighted broadband sound pressure level
- **Z** Broadband sound pressure level without any frequency weighting

16 Readout Frequency

You may select any frequency to read out individual levels. The selected frequency is indicated by the cursor arrow.

Choose between the following settings:

- The cursor follows automatically the highest level, e.g. tracing feedback frequencies at live sound.
 - Select the frequency with the rotary wheel ⁽¹⁾
 - Press enter 🕘.
 - You may readout any individual frequency.
 - Return to auto frequency with enter 🕑.
 - The cursor returns to the frequency with highest level bar.



The cursor read out may be set manually to any
 frequency band or wide band level. The read out remains at the selected frequency.

- Select the frequency with the rotary wheel O.
- Press enter 🕘.
- Select the frequency of interest.
- Confirm with enter .

The cursor readout (1) displays the measurement results of the selected frequency band.

17 Setting of Test Result Resolution

Set the RTA result display to 1/1 octave or 1/3 octave band resolution as follows:

- Turn the rotary wheel 0 to select the parameter $\boxed{17}$.
- Press enter to toggle between 1/1 OCT and 1/3 OCT. to set this parameter.



Error Indications

In case the measurement result exceeds the preset measurement range, then limit arrows **T** indicate this overload condition. The arrows **T** are displayed for minimum 1 seconds and as long as the overload exists. Furthermore the once exceeded measurement range triggers the **OVR** indication in the bottom line, which remains displayed for the complete measurement period. These exceeded range indications are reset at starting a new measurement.

Possible causes for exceeding measurement ranges are

- The selected measurement range is lower than the actual level -> select the next higher measurement range or reduce the input signal level as applicable.
- The input level is near the maximum level of the connected NTi Audio measurement microphone.





Select Displayed Test Result

This example describes the setting of the commonly-used actual sound pressure level $L_{\Delta F}$ (frequency weighting A, time weighting F).

- Select the first parameter setting with the rotary wheel ⁽¹⁾.
- Confirm with enter **e**.

ASD microphones provided by NTi Audio and switches the 48V phantom power automatically on as follows: • Connect the measurement microphone to the XL2.

• Switch on the XI 2 🖨

Test Preparations

Ð The 48V phantom power indication in the upper menu bar changes to ASD. The XL2 is ready for acoustic measurements.

The XL2 reads the electronic data sheet of any connected

Sound Level Meter - Getting Started

- Position the XL2 at the measurement location using a microphone stand or a tripod.
- Select the SLMeter function in the measurement menu and togale with page \blacksquare to the numeric result page 123.

All wideband and RTA levels are measured and logged simultaneously. You select which sound levels should be displayed.





Press start **D** to measure and display the sound levels results indicated with ----.





Select Frequency Weighting

- The pop-up window **FREQ WEIGHTING** shows up.
- Select the frequency weighting **A**.



Pop-up windows shown at installed Extended Acoustic Option

• Confirm with enter .

Select Time Weighting

The pop-up window extends with **TIME WEIGHTING**.

• Select the required time weighting; e.g., **F** (=Fast).



• Confirm with enter @.

35

βB

В

CRT 00:00:00

Sound Level Meter

Select Correction Value

- The pop-up window extends with **CORRECTION** settings.
- Select the parameter off.

SLMeter 123 XLR (C) 4 ASD 13:24 (LAF (FRED HEIGHTING) TIME HEIGHTING S (F) I EQ PARAMETER LAEq (Live) (max) (min) T3 T3eq (T5) (T5eq)

LAFMOX CORRENTION

10 RNGE 110

off K1

- Confirm with enter e.
- The pop-up window closes and the measured sound pressure level L_{AF} is displayed.

Select Parameter

- The pop-up window extends with **PARAMETER** settings.
- Select the parameter live.



• Confirm with enter .



Select further Sound Levels

(SLMeter 123 XLR 💼 🗏 🕯 650 13:24 🚛		
Laf		
40).3dB	
LAeq	dB	
LAFma×	dB	
10 RNGE 110	SET:: CNT 00:00:00	

- Select further sound pressure levels as described above; e.g., $L_{\mbox{Aeq}}$ and $L_{\mbox{AFmax}}$

Select Input Range

- Select the lowest possible input range according to the maximum level expected during the measurement. Wrong input ranges are indicated by a flashing **LOW** or **OVR** message in the lower menu bar.
- Select the input range **RNGE** and press enter P.
- Turn the rotary wheel ⁽¹⁾ to set the applicable input range and confirm with enter ⁽²⁾.



The numeric result page and the real-time analyzer page use the same input range.


Start Measurement

- The XL2 is ready to measure the sound levels L_{AF}, L_{Aeq} and L_{AFmax}.
- Press start ▶■.
- The run indication switches to running . The integrated sound pressure level over time L_{Aeq} and the maximum level in the measurement period L_{AFmax} are displayed. The parameter **RUN**, **LOG** (logging is enabled) or **AUD** (audio recording is configured) flashes in the memory menu.

SLMeter 123 X	LR 🗖) 4 ASD 13:24 🖷
Laf	
47	7.7dB
Lfleg	49.6dB
LAFma×	61.8dB
10 RNGE 110	SET:: CNT 00:00:11

Stop the Measurement

• Press stop 🕨 to stop the measurement .





Save the Measurement Result

• Select the memory menu 🗊 with the rotary wheel ⁽¹⁾ and confirm with enter ⁽²⁾.

 $^{\&}$ The memory menu opens. **Save Test** is preselected.





The XL2 stores the numeric sound pressure levels and the real-time analyzer results simultaneously.

 $^{\&}$ The sound pressure level measurement is completed.

Data Post-Processing

A Microsoft Excel form generating an automated report and chart is available as free download for all registered XL2 customers on the support website at http://my.nti-audio.com. (Enable all macros when opening the document.)

RTA Measurement - Getting Started

Test Preparations

The XL2 reads the electronic data sheet of any connected ASD microphones provided by NTi Audio and switches the 48V phantom power automatically on as follows:

- Connect the measurement microphone to the XL2.
- Switch on the XL2 ⁽³⁾.
- The 48V phantom power indication in the upper menu bar changes to ASD. The XL2 is ready for acoustic measurements.
- Position the XL2 at the measurement location using a microphone stand or tripod.
- Select the **SLMeter** function in the measurement menu and toggle with page **1** to the real-time analyzer (RTA) page **1**

All wideband and RTA levels are measured and logged simultaneously. You select the sound levels to be displayed.

RTA Configuration

The XL2 displays two different sound pressure levels at the same time. You configure which test results to display; e.g., L_{ZFmax} and $L_{ZFlive}.$

- Ensure no other measurement is going on. The run indication should display the stop symbol .
- Select the LZF value with the rotary wheel ⁽¹⁾.



• Confirm with enter 🕘.

Sound Level Meter





Select RTA Frequency Weighting

- The pop-up window **FREQ WEIGHTING** shows up.
- Select the frequency weighting Z.



• Confirm with enter 🕑.



Select RTA Time Weighting

The pop-up window extends with **TIME WEIGHTING**.

• Select the required time weighting; e.g., **F** (=Fast).



• Confirm with enter 🕑.

Select Upper/Lower RTA Parameter

• Select the parameter shown to the right of the upper LZF value with the rotary wheel ⁽¹⁾; e.g. **max**.

(SLMeter 📶 XLR 🗖) 4 ASD 13:24

PARAMETER

max EQ Hold

SET

CRT 00:00:00

ltivel

E

11

lmin

Capt

PK

- Press enter •
- Ð The pop-up window **PARAMETER** shows up.
- Select the parameter **max**.

60

40 20

- LZF max

III LZFLive

BD 250

10 RNGE 110



• Follow the same instruction and select the lower RTA parameter live

Select Input Range

- Select the lowest possible input range according to the maximum level expected during the measurement. Wrong input ranges are indicated by a flashing LOW or OVR message in the lower menu bar.
- Select the input range **RNGE** and press enter
- Turn the rotary wheel ⁽¹⁾ to set the applicable input range and confirm with enter Θ





The numeric result page and the real-time analyzer page use the same input range.





Start RTA Measurement

- The XL2 is ready to display the measured sound pressure levels L_{ZFmax} and L_{ZFlive}.
- Press start 🖪.
- The run indication switches to running . The actual sound level L_{ZFlive} and the maximum level L_{AFmax} are displayed. The parameter **RUN**, **LOG** (logging is enabled) or **AUD** (audio recording is configured) flashes in the memory menu.



Stop the RTA Measurement

• Press stop 💌.



The XL2 stores the numeric sound pressure levels and the real-time analyzer results simultaneously.



Save the RTA Measurement Result

• Select the memory menu 🖃 with the rotary wheel 🌑 and confirm with enter 🕑.

The memory menu opens.



• Press enter 🕑 to store the test results as an ASCII-file or select **Save Screenshot** to store the screenshot.



Correction Value KSET

The correction value page offers a measurement wizard for live event monitoring, applicable in case the measurement location differs from the loudest position in the live event area.

Measurement Position

The goal at live event monitoring is to measure the sound level at the loudest position accessible by the audience. Setting up any sound level measurement at this loudest location is typically not practical, thus an alternative measurement position is chosen. The recommended measurement position in order to minimize the influence that audience noise has on the measurement:

- Mount the measurement microphone in front of the main speakers
- Position the XL2 Analyzer at front of house (FOH)
- Connect the XL2 Analyzer and measurement microphone using a professional audio cable



Positioning the measurement microphone at front of house (FOH) may result in audience noise interfering with the measurement results. The audience noise will be further amplified by the measured k-values.

Level Difference

Prior to the start of the live event, the level difference between the loudest and the measurement position is measured. The XL2 Analyzer calculates the level difference automatically based on the individual sound level measurements at both locations. This level difference is included in the sound level reading during the live event, and the XL2 thereby displays and records the sound pressure level of the loudest point.





How to Measure the Correction Value

- Play a pink noise signal at the typical sound pressure level of the live event. (signal source Minirator, NTi Audio Test CD)
- Select the correction page KSET.
- Search the audience area for the location with the highest sound pressure level and position the XL2 Analyzer.
- Turn the rotary wheel ⁽¹⁾ to select the parameter **RUN** next to **Audience** and press enter ⁽²⁾.

(SLMeter Ksst XL	R 🗖 🛛 ASD	13:24 •
Position	L _{Aeq} 5"	LCPeak
Audience 💷 Audience	dB	dB
Measure RUN :	dB	dB
Difference ki CLEAR ALL	0.0dB ki	2 0.0dB
50 RNGE 150		01:00:00 00:00:00

- Wait until the measurement is completed.
- Position the XL2 at the measurement position.
- Turn the rotary wheel ⁽¹⁾ to select the parameter **RUN** next to **Measure** and press enter ⁽²⁾.

(SLMeter Ksst XL	R 🗖) < ASI	0 13:24 🗰
Position	L _{Aeq5"}	LCPeak
Audience 💷 2011-12-09 13:24	91.2dB	115.7dB
Measure RUN	dB	dB
Difference k1 CLEAR ALL	0.0dB k	a 0.0dB
111111 50 RNGE 150	SET COT	00:00:00

- The XL2 measures sound level at the measurement position for 5 seconds. The timer counts down to zero.
- The XL2 measures sound level at the loudest position for 5 seconds. The timer counts down to zero.



- Wait until the measurement is completed.
- The correction values k1 and k2 are calculated and displayed including date and time (in accordance with DIN15905).

Manual Setting of Correction Values

You may fine-tune the correction values k1 and k2 manually. Such fine-tuning adds the remark "Manually Adjusted" in the log file.

(CI M.L., IVIVI		
(SLMeter Kset XL	<u>кра</u> јчк	15D 13:24 (
Position	L _{Aeq} 5"	LCPeak
Avdience RUN 2011-12-09 13:24	91.2dB	115.7dB
Measure 🖾 🕅 2011-12-09 13:24	101.2dB	126.7 dB
Difference ki CLEAR ALL	-10.1dB	ka - 11.0 dB
111111 50 RNGE 150		et:: nt 00:00:00

	(SLMeter Kset XL	R 🗖) 4 AS	D 13:24 🗰	
	Position	L _{Aeq5"}	LCPeak	
	Audience RUN 2011-12-09 13:24	91.2dB	115.7dB	
	Measure 💷 13:24	101.2dB	126.7 dB	
1		-	198.5 dB Isted undo	3
	111111 50 RNGE 150		r:: r 00:00:00	

- Turn the rotary wheel 0 to select the correction value 3 and press enter 0.
- She selected correction value starts flashing.



• Set the correction value with the rotary wheel ⁽¹⁾.

 $^{\circ}$ The note "Manually Adjusted" is displayed at 2.

- To undo the manual setting turn the rotary wheel ⁽¹⁾ to select the parameter **UNDO** at ⁽²⁾
- Confirm with enter .

Reset C orrection Values You may clear all correction values back to zero by pressing the **Clear All** (1) and confirm with enter P.

Display k1 and k2 during Measurement The correction values k1 and k2 can be displayed in the numeric result page during the ongoing sound level measurement instead of any other sound levels.

Wav-file Recording

A simultaneously-recorded wav-file supports post-event reviewing of the cause of excessive sound levels.



Reporting

A report saves the conducted measurements onto the installed SD Card. The XL2 supports setting the individual levels to be stored in the report after the measurement is completed.

	SLMeter Rep XLF) d ASD 13:24 💼
	Report Setting	
1	··· Add Spectra:	AU
2	Report Values: 🛛 LAeg	Selected
<u>3</u>	··· D LAeq 1 LCPKmax	5
	¹ LCPKma×	6
	2	1
	3	8
	4	9

1 Add Spectra

- **No** No RTA measurement reporting.
- Leq The RTA Leq level is stored in the report.

Leq, Lmax, Lmin	, The RTA Leq, Lmin, Lmax levels are stored in the report.	
All	All RTA levels are stored in the report.	
2 Report Values Choose between the following result reporting:		
ALL	Records all available sound pressure levels without correction values.	
Selected	Records a subset of up to 10 different levels, including correction values if you require.	
 Select Report Values with the rotary wheel ⁽¹⁾. Press enter ⁽¹⁾ to toggle between All or Selected. 		
 ③ Selected Report Values By setting Selected at ② up to 10 individual report values can be chosen. Select the first value Lxx with the rotary wheel ③ and press enter ④. 		
& The pop-up window for level selection shows up.		

• Select the reporting level with the rotary wheel 0 and confirm with enter 0.



Logging

The XL2 features a powerful sound level meter data logger, which allows you to record all required sound level values during the selected period of time. All results are logged onto the SD Card. The measurement results can be loaded to a PC for documentation and visualization. The LOG menu offers the detailed setup of the log report.

	SLMeter Log XL	R 💼 🗐 ASD 15:36 🖷
1)	Logging	every n sec
2)	Interval dt:	00:00:01
3	Add Spectra:	No
<u>(</u> 4)	- Log Audio:	On
5	··· Format:	Compressed+AGC
6	Log Values:	Selected
6… 7…	Log Values: ¤ L _{Aeq}	Selected s
6 7		Selected 5 6
6 7	¤ L _{Aeq}	Selected 5 6 7
6 7	1 LAFmax	Selected 5 6 1 8



The optional Extended Acoustic Pack offers 100 ms logging.



1 Logging

Select **Logging** with the rotary wheel O and press enter O to enable the automated logging of test results.

- every n sec The XL2 logs measurement data every interval dt 2.
- every 100ms The XL2 logs measurement data every 100 ms fixed.
- off No Logging
- 2 Interval ∆t

Adjust the logging interval.

3 Add Spectra

Select **Yes** to log additionally the RTA spectrum at each logging interval.

4 Log Audio

The XL2 logs audio data as a wav-file. Choose amongst the following settings:

Off	Audio logging is off
On	Audio logging is activate from mea- surement start to stop.
Events Only	Audio logging of triggered events only.

For more details read the chapters "Wav-File Recording" and "Events".

⑦ Selected Log Values

By setting $\ensuremath{\textbf{Selected}}$ at (5) up to 10 individual log values can be chosen.

• Select the first value \mathbf{Lxx} with the rotary wheel 0 and press enter 0.

\checkmark The pop-up window for level selection shows up.

• Select the logging level with the rotary wheel ⁽¹⁾ and confirm with enter ⁽²⁾.

5 Format

Choose amongst the following audio format recording settings:

- Compressed Compressed audio logging
- Compressed+AGC Compressed audio logging with automated gain control
- 24Bit_48kHz Linear audio logging in 24 bit, 48 kHz resolution (optional with Extended Acoustic Pack)

6 Log Value S etting

Choose between the following settings:

- ALL Logs all available sound pressure levels without correction values.
- **Selected** Logs a subset of up to 10 different sound pressure levels, including correction values if you require.
- Select Log Values with the rotary wheel ⁽¹⁾.
- Press enter 🕑 to toggle between All and Selected.



Wav-File Recording

The XL2 records a wav-file of the measured input signal and stores it on the SD Card. The available formats are:

- **Compressed** (default), using ADPCM compression. A new wav-file is started automatically after 12 hours of recording (typical maximum wav-file size = 512 MByte)
- **Compressed+AGC**, compressed with automated gain control. The gain control increases the level of low-level signals, so that the wav-file is well-leveled during playback on the PC.
- **24Bit_48kHz**, linear wav-file logging in 24 bit, 48kHz resolution with the optional Extended Acoustic Pack. A new wav-file is started automatically after 1 hour of recording (typical maximum wav-file size = 512 MByte)

Broadcast Wave Format BWF

The XL2 stores date and time of the recording within the wav-file (according to EBU TECH 3285). This information is available through professional audio/video tools typically used in broadcast. Sample name of wav-file:

```
MyTest_SLM_000_Audio_FS133.0dB(PK)_00.wav
```



1 MyTest

File name defined by user.

2 SLM

Measurement function.

```
3 000
```

Automatically incrementing file number.

4 Audio_FS133.0dB(PK)

Audio file with full scale peak level. In case of a **Compressed + AGC** recording, the file name reads "AGC"; this file contains corrected level information only.

5 00

For wav-file recording over longer periods, the XL2 splits the audio data into individual wav-files with about 500 MB (compressed audio: 12 hours; linear: 1 hour), keeping the file size small for easier handling on the PC. The number (5) increments for each successive wav-file.



The advantage of wav-file recordings is to identify and document sound sources after the measurement. For example, at a live event an excessive peak level may have been measured and logged. Actually, this peak level was caused by people shouting nearby the measurement microphone, and not by the audio system being monitored. The recorded wav-file assists in verifying this and the test results can then be post-processed.

Pausing Measurement

If any ongoing measurement is paused II, the XL2 continues the recording of wav-files during the paused period. The logged data and audio data can be synchronized using the stored real-time information.

Events

The wav-files are stored in a folder, named, for example,

2011-11-30_SLM_000_AudioEvent_0001-0200. The actual wav-files are named for example, as xxxx_FS133.0dB(PK).wav (xxxx = incrementing number) Another example is environmental noise monitoring: Listening to the recorded wav-file after the measurement may help to determine the predominant sound source. The XL2 Analyzer preserves the original absolute test signal level in the recorded wav-file.

The XL2 Analyzer provides three input ranges to accommodate the wide range of input signals. The dynamic range of the recorded wav-file is set according to the selected input range. For example, at a microphone sensitivity of S= 20 mV/Pa, the full scale peak level is:

Range Name	Range Level	Full scale peak level
Lower range	10 - 110 dBSPL	117.8 dBSPL
Middle range	30 - 130 dBSPL	135.9 dBSPL
Upper range	50 - 150 dBSPL	159.9 dBSPL



Select the lowest possible input range according to the maximum level expected during the measurement; e.g., If the sound pressure will always be below 110 dBSPL, select the lowest input range 10 - 110 dBSPL.



Events (optional)

The event function is available with the XL2 Extended Acoustic Pack Option. The XL2 Analyzer may be configured to record wav-files and additional noise levels only when triggered, instead of the complete measurement duration.

Advantages

Reduces the data volume acquired, thereby

- Simplifying data post-processing of the complete measurement period
- Saving data memory for long-term measurement applications.

Events

Events are triggered either automatically by sound levels above/below a preset value or manually by external keypress using the XL2 Input Keypad.

Event Functions

The XL2 event feature offers the following functionalities:

- Automated event triggered at noise levels above/below a preset value including setting markers for specific noise categories. Application example: Recording the noise level for LAF > 80 dB.
- Event triggered by external keypress of the XL2 Input Keypad. You can utilize four keys (1-4) to categorize any noise of interest or noise to be later excluded in the post-processing. A typical application is categorizing any annoying industrial noise by persons living in the neighborhood.

Data Logging



The XL2 Analyzer logs the data specified in the LOG setting for the complete measurement duration into a log file. The markers and triggered events results are added to the same log file.

The XL2 Input Keypad is available as an accessory. Please see chapter "Options and Accessories" for details.



Trigger Events: on level above/below



Features:

- The levels LAeq, LZeq and LCPeak are recorded for the event duration.
- Audio data are recorded for the triggered event duration only, thus simplifying data post-processing and saving memory space for long-term measurements.
- Setting markers by external keypress of the XL2 Input Keypad.





Trigger Events: on external keypress



	R 面)(ASD 13:52 4 s: on ext.keypress ration: 05s …	(15)
Log Audio: Format:	Events Only Compressed+AGC	
Status:	Stopped	
Event count:	0000 HONITOR 1234	

Features:

- The levels L_{Aeq}, L_{Zeq} and L_{CPeak} are recorded for the event duration.
- Audio data may be recorded for the user-defined event duration (15) after the keypress.
- Retrigger with every repeated keypress within the event duration.



1 Trigger Events

Choose from the following settings:

Off	No event triggering
on level above	Events are triggered at noise levels above the preset value 2 for the start duration 13.
on level below	Events are triggered at noise levels below the preset value 2 for the start duration 13.
on ext. keypress	Events are triggered by external keypress.

2 Start [dB]

Set the level at which the triggered event is started.

Start Duration 13

The event start is triggered when the noise level is higher (or lower) than the preset value and continues for the preset duration.

3 Stop [dB]

Set the level at which the triggered event is stopped.

```
on level Stop level = Start level 2 - Hysteresis 4
above
```

on level Stop level = Start level (2) + Hysteresis (14) below

Stop Duration (12)

The event stop is triggered when the noise level is lower (or higher) than the preset value and continues for the preset duration.

Hysteresis (14)

Set the hysteresis to an appropriate value to prevent rapid switching on and off as the level drifts around the Start level.

4 Level Selection

Define the level type to be monitored.



5 Log Audio

The XL2 logs audio data as a wav-file. Choose from the following settings:

Off	Audio logging is off
On	Audio logging is activated from measurement start to stop.
Events Only	Audio logging of triggered events only.

For more details read the chapter "Wav-File Recording".

6 Format

Choose from the following audio recording formats:

Compressed Compressed audio logging

Compressed+AGC Compressed audio logging with Automated Gain Control (AGC)

24Bit_48kHz Linear audio logging in 24 bit, 48 kHz resolution. (Required for post-processing on the PC with Extended Acoustic Pack)

7 Audio Logging Period

If **Log Audio** (5) is set to **Events Only**, then the duration for the audio recording may be further specified:

Recording	Records a wav-file for the whole trig-
whole event	gered event duration.
Stop recording after	Records a wav-file for the specified period after the event start is trig- gered.

8 Status Information

Displays current trigger status:

- Waiting for trigger
- Armed (during start duration 13)
- Audio + data recording
- Completing log cycle

9 Actual Level

Measurement result of the defined level (4).

10 Event Counter

Counts the number of triggered events that have occurred during the ongoing measurement.



11) Trigger and Marker Monitor

Answers the question: What caused the triggered event?

- LvI Event automatically triggered by level.
- **1** Event triggered or marker added by pressing key 1 of the XL2 Input Keypad.
- 2 Event triggered or marker added by pressing key 2 of the XL2 Input Keypad.
- **3** Event triggered or marker added by pressing key 3 of the XL2 Input Keypad.
- 4 Event triggered or marker added by pressing key 4 of the XL2 Input Keypad.
- 12 Stop Duration See 3.
- 13 Start Duration See 2.

14 Hysteresis See 3.

15 Event Duration at Keypress

Preset the **Min. Event Duration** when a key is pressed. If a key is pressed and held for longer than the preset **Min. Event Duration**, then the triggered event lasts for as long as the key is held.



How to Setup the XL2 for Triggered Event Measurements

- Select the logging page, set Logging On, Interval dt: 00:00:01 and choose the required log values.
- Setup the event page. For example, the screenshot below starts the event recording after L_{AF} exceeds 80 dB for 2 seconds and stops after L_{AF} is lower than 70 dB for 3 seconds.



• Select the memory menu and create a new folder. This folder will then be displayed in the memory menu. All log data and event wav-files are stored in this folder.

 $^{\diamond}$ The XL2 is ready for the triggered event measurement.

Logging Enabled

Triggered-event recording requires the activation of sound level logging. Using the default **Interval dt:** 1 second is recommended.

Auto Save

If Log Audio (5) is set to Events Only, then the XL2 sets per default Naming+Saving: auto in the memory menu. This ensures that the recorded triggered event data is stored on the SD Card.

For convenience, in case more than 20 triggered events have been recorded, the **Au-tosave** -> **Delete** function is disabled, as deleting such multiple wav-files on the XL2 takes a long time.

Wav-files

All triggered-event wav-files are stored in an individual folder. For details see chapter "Log-ging".

61

Sound Level Meter



Limits

The limit page offers two functions:

- Setup of limit LED function; enabling the XL2 to highlight any sound levels that exceed the pre-set limit in orange or red color. The default color is green.
- Limit parameter setup for external Digital I/O Adapter, which connects to the digital I/O interface. The Digital I/O Adapter is an optional accessory for the XL2 analyzer. It serves the control of any external peripherals, such as displaying sound levels that exceed limits on a big external red-orange-green lamp.

	(SLMeter	Lim XLR	B) ()	ASD 14:58 💶	
1	· Limit LE	ED: On			
2)	•	L _{Aeq5"+k1}			
-	Red	> 99.0dB > 95.0dB ≤ 95.0dB			<u> </u>
3	Orange	> 95.0dB			
	Green	≤ 95 . 0dB			
4	I/O Box	: On		STAT: 1234	
5		L _{Req5"+k1}		LAeq5"*k1	(7)
-	0ut 1	> 99.0dB	and	> 99.0dB	
6	0ut 2	> 95.0dB	and	≤ 99 . 0dB	
-	Out 3	≤ 95 . 0dB	and	≤ 95 . 0dB	
	Out 4	≤ 0 . 0dB	and	≦ 0 . 0dB	

1 Limit LED On/Off

Select **Limit LED** with the rotary wheel O and press enter O to enable/disable the limit function.

2 LED - Level 1

Select one sound level for the limit LED indication 🗐.

- Select the sound level field with the rotary wheel ⁽¹⁾
- \bullet Press enter O and select the sound level for the limit LED indication.

3 Selected Level Values

To set the level values for the limit LED 🗐,

- Select the Red or Orange sound level with the rotary wheel ^(C) and press enter ^(C).
- Set the level value with the rotary wheel ⁽¹⁾ and confirm with enter ⁽²⁾.

4 I/O Box On/Off

Select **I/O Box** with the rotary wheel O and press enter O to enable/disable the limit function.



5 I/O - Level 1

Setup the level for the control of e.g. the external Digital I/O Adapter Box.

- Select the level field with the rotary wheel ⁽¹⁾.
- Press enter ④ and select the level for the control of the external Digital I/O Adapter Box..

6 Selected Sound Level Values

To set the level values for the control of the external Digital I/O Box.

- Select the Out 1, Out 2, Out 3 or Out 4 level with the rotary wheel ⁽¹⁾ and press enter ⁽²⁾.
- Set the level value with the rotary wheel 0 and confirm with enter 0.

7 I/O - Level 2

Follow the setting for I/O Sound Level 1. The I/O level 1 may be combined with I/O level 2 using and/or combinations, providing indications in level ranges. Alternatively, other levels can be selected for the control of the external Digital I/O Adapter Box.

8 LED - Level 2

Follow the setting for LED - Level 1.



I/O Box Limits with Triggered Events If triggered-event recording is enabled, the digital output 4 (O**UT4**) is utilized as confirmation feedback to the XL2 Input Keypad, thus the setting is disabled.





5. Acoustic Analyzer

Besides the comprehensive sound level meter function, the XL2 Audio and Acoustic Analyzer offers the following acoustic measurement functions:

- FFT Analysis with optional tolerance function
- Reverberation Time RT60
- Polarity
- Delay time
- 1/12 Octave + Tolerance Analysis (optional)
- Speech intelligibility (optional)

FFT Analysis + Tolerance

The FFT measurement is the ideal tool for visualization of comb filters and narrow band effects. It allows a detailed frequency response investigation of audio and acoustic systems. The XL2 includes an extremely fast, real-time FFT. Optional features are

- High-resolution Zoom-FFT up to 0.4 Hz steps in the frequency range 5 Hz - 20 kHz provided with optional Extended Acoustic Pack or the Spectral Limits Option
- Capture and tolerance function provided with Spectral Limits Option; thus the main menu function reads **FFT + Tol**.





- 1 Result Symbols / Capture & Start Tolerance Mode This field offers two functions:
 - Symbols for measurement results 2
 - --- Upper result displayed as line.
 - Lower result displayed as bargraph.
 - Capture & Start Tolerance Mode The displayed readings may be captured as reference reading C1 to C8 for
 - Comparing measurement results against captured traces with relative or absolute curve display.
 - Creating tolerance masks based on captured reference curves for passed / failed measurements.

Capture EQ	Captures the upper parameter
Capture Live	Captures the lower parameter





Manage	Allows to rename captures, clear re-
captures	corded captures, save captures to SD-
	Card for export from the XL2 or load
	captures from SD-Card for import to
	the XL2.

Start toleranceStarts the tolerance mode for passed/
failed measurements comparing the
actual measurement results against a
tolerance band.

2 Measurement Result

Actual level result of the indicated frequency band. The cursor readout displays the center frequency and the level of the band indicated by the arrow.

- --- Upper parameter displayed as line.
- Lower parameter displayed as bargraph.

- **3 Y-Scale setting**
 - Select the Y-Axis with the rotary wheel 0 and confirm with enter 0.
 - Select the zoom factor between **20, 10, 5**, **2.5 dB/div** and confirm with enter **(**.
 - Scroll up and down with the rotary wheel ⁽¹⁾ to select the Y-axis range.
 - Confirm with enter .

4 Measurement Result

Displays the actual and averaged measurement results.

5 Input Range

The XL2 Analyzer provides three input ranges to accommodate the wide range of input signals. The individual ranges are based on the microphone sensitivity setting in the calibration menu of the XL2. For example at a sensitivity S=20mV/Pa the input ranges are

- Lower range: 10 110 dBSPL
- Middle range: 30 130 dBSPL
- Upper range: 50 150 dBSPL

Select the lowest possible input range according to the maximum level expected during the measurement; e.g., if the sound pressure will be always below 110 dBSPL, then select the lowest input range 10 - 110 dBSPL.

67

FFT Analysis + Tolerance



6 Time Weighting

Offers selectable time weighting of 0.1, 0.2, 0.5 and 1.0 second. Applications:

Short Time	High resolution in time with minimum
Weighting	averaging

Long Time Low resolution in time with longer aver-Weighting aging

7 Run Indication

The run indication shows the measurement status running, paused or stopped. Various measurement settings are locked during ongoing measurements, such as changing the input ranging or the preset measurement time.

For passed/failed measurements with the Spectral Limits option the run indication may display ${\bf A}$ for a preset automated level trigger.

8 Actual Measurement Time

Counts actual measurement time in hrs:min:sec. Supports setting of time modes: continuous and single.

CIT Timer Mode Continuous

(applicable for standard measurements) All values are recorded and monitored continuously after starting a measurement with start **D**. The actual measurement time is shown.

Timer Mode Single

Automatically stops the measurement after the pre-set measurement time.

- Set the required measurement time.
- Start the measurement 🖭.

9 Preset Measurement Time

Adjustment of preset measurement time for single timer setting.



FFT Analysis + Tolerance

10 Readout Frequency

You may select any frequency to read out individual levels. The selected frequency is indicated by the cursor arrow. Choose between the following settings:

- The cursor follows automatically the highest level, e.g. tracing feedback frequencies at live sound.
- Select the frequency with the rotary wheel ⁽¹⁾.
- Press enter 🕑.
- You may readout any individual frequency.
- Return to auto frequency with enter .
- The cursor returns to the frequency with highest level bar.
- ➡ The cursor read out may be set manually to any frequency. The read out remains at the selected frequency.
 - Select the frequency with the rotary wheel O.
 - Press enter 🕑.
 - Select the frequency of interest.
 - Confirm with enter .
 - The cursor readout 2 displays the measurement results of the selected frequency band.

D Zoom Mode

(optional, applicable with the Extended Acoustic Pack or the Spectral Limits Option)

- Select the readout frequency 10 and press enter 4.
- The zoom mode is displayed above the flashing arrow.
- Select the zoom-in frequency with the rotary wheel ${f O}$.
- Press limit i and zoom the linear frequency scale in or out with the rotary wheel .
- Release limit i and scroll the linear frequency scale left or right with the rotary wheel **O**.

12 Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on
- 48V Phantom power is switched off.

FFT Analysis + Tolerance



13 Page Selector X-Scale and Parameter Setting

- **20k** Shows FFT result of the frequency range. 484.38 Hz - 20.594 kHz in a resolution of 141.62 Hz with 142 bins shown on the display.
- 1k7 Shows FFT result of the frequency range.58 Hz 1.722 kHz in a resolution of 11.72 Hz with 142 bins shown on the display.
- 200 Shows FFT result of the frequency range.7 Hz 215.01 Hz in a resolution of 1.47 Hz with 142 bins shown on the display.
- Usr Customized zoom mode (applicable with optional Extended Acoustic Pack or Spectral Limits Option)
 5 Hz - 20 kHz in zoom mode with a minimum resolution of 0.366 Hz and 142 bins displayed.
- **Set** Selection for FFT Windowing:
 - Hann: Default for acoustic measurements
 - **Dolph-Chebyshev**: for analyzing small signals (e.g. harmonics) close to a dominant signal.

The page button 🗇 switches these display modes.



FFT Analysis - Getting Started

Test Preparations

- Connect the measurement microphone to the XL2.
- Switch on the XL2 🕘.
- The 48V phantom power indication in the upper menu bar changes to ASD. The XL2 is ready for acoustic measurements.
- Position the XL2 at the measurement location using a microphone stand or tripod.

Configuration

The XL2 displays two different sound pressure levels at the same time. You configure to display either the levels Live, Max, Min, EQ or captured results.

• Select the upper result parameter with the rotary wheel ${igodot}$.



• Confirm with enter .



Select Upper/Lower Reading

- The pop-up window **Select reading** shows up.
- Select the integrated averaged reading **EQ**. All available selections are Z-weighted (= no weighting).



• Confirm with enter **e**.

The pop-up window **Select reading mode** shows up.



- Select **EQ** for normal absolute display of measurements.
- Follow the same instruction and select the lower FFT reading **Live**.





Select Input Range

- Select the lowest possible input range according to the maximum level expected during the measurement.
- Select the input range **RNGE** and press enter *e*.
- Turn the rotary wheel ⁽¹⁾ to set the applicable input range and confirm with enter ⁽²⁾.



Start and Stop the Measurement

- The XL2 measures the selected sound pressure levels **Live** and **EQ**. All available sound levels are Z-weighted (= no weighting).
- Press start 🔳.
- The run indication switches to running . The actual sound level Live and the averaged level EQ are displayed. The parameter **RUN** flashes in the memory menu.



• Press the stop button 🖭 to stop the measurement.
FFT Analysis + Tolerance



Save the FFT Measurement Result

• Select the memory menu 🖃 with the rotary wheel 🕲 and confirm with enter 🕑.

The memory menu opens.



The FFT measurement is completed.

Capture References and Create Tolerances

The Spectral Limits Option extends the function range of the XL2 with trace capturing, relative curve display and comprehensive tolerance handling for the **FFT** Analysis and the high resolution **1/12 Oct + Tol** spectral analysis.

Features:

- Captures multiple traces in the internal memory
- Comparing measurement results against captured traces with relative or absolute curve display
- Comprehensive tolerance handling
- Creating tolerance masks based on captured reference curves for passed / failed measurements
- Export and import of tolerance and capture files
- True peak level in 1/1 and 1/3 octave resolution
- High-resolution Zoom-FFT up to 0.4 Hz steps in the frequency range 5 Hz 20 kHz

Read the detailed description in the separate chapter "Capture + Tolerance" in this manual.



Reverberation Time RT60

The XL2 measures the energy decay from 63 Hz to 8 kHz using the Schroeder method. Use either an impulse source (e.g., a starter pistol) or a gated pink noise as the test signal.

What is Reverberation Time RT60?

Reverberation time RT60 is the time required for the sound pressure level to decrease by 60 dB after the sound stimulus signal is stopped. Since ambient noise often defeats the measurement of a 60 dB level decrease in practice, it is difficult or even impossible to measure a 60 dB decay in many real life rooms. Thus the applicable standard ISO3382 specifies

- The measurement of a 20 dB level decrease multiplied by 3.
- The final 60 dB decay result shall be labeled T20.

RT60 (T20) = $3 \times \text{decay time of } 20 \text{ dB}$

In detail the reverberation time RT60 is based on a linear leastsquares regression of the measured decay curve from -5 dB down to -25 dB according. It is sufficient to create a test level at least 35 dB above the background noise within each frequency band. Please see the next drawing for details. If the overall RT60 is short (e.g. < 0.3 seconds) the room acoustic is referred to as being "dead"; for example, a heavily furnished room with thick carpets, curtains and upholstered furniture may have such an acoustic character.

If the overall RT60 is long (say more than 2 seconds) the room acoustic is referred to as being "live" and echoic; for example, a large empty room with painted plaster walls and a tiled floor may have such an acoustic character.



VDIO

Reverberation Time RT60











RT60 Cycle Result Page



1 Set Level Markers

Prior to conducting a reverberation time measurement, the environmental noise is measured, and the required energy level of the test signal is determined.

- Select **SET** and press enter to measure the actual environmental noise.
- The required level markers appear in grey color.

2 Input Range

Select between the lower, mid and upper input range. The exact ranges depend on the microphone sensitivity setting in the calibration menu of the XL2.

3 Run Indication

This icon displays the running and stop status of the reverberation time measurement. It is controlled by start/stop



4 Measurement Status STAT

Displays the actual measurement status. Start and stop the reverberation time measurement with start/stop **I**. The following status information is displayed:

- **ARMED** Measurement is waiting for the test signal to exceed at least one measurement trigger marker, then the measurement is triggered automatically.
- **NOISE** A sound stimulus signal above the trigger marker is being received.
- **DECAY** Decay is being measured.
- **PAUSE** The measurement has been paused by pressing pause II.
- **STOP** No RT60 measurement is presently being performed.

5 Average AVRG

Counts the measurement cycles. When using a gated noise test signal, a minimum of 3 sequential cycles is recommended.

6 Actual Real-Time Spectrum

The black bars indicate the actual signal, including background noise. While setting up the reverberation time measurement you should increase the test signal level until the black bars exceed the grey bars completely and the status ④ displays **NOISE**.

7 Level Marker

The grey bars indicate the signal level required for effective reverberation time measurements in each octave band. The marker has a length of 35dB, and can be set by

- Selecting **SET** 1 with the rotary wheel **3**.
- Press enter 🕑 to measure the environmental noise in the silent room.
- 8 Band Status

Indicates a successful reverberation time measurement above each octave band with a tick ${\, {\cal J}}$.

(9) Measurement Resolution

Reverberation time measurement selectable in 1/3 octave or 1/1 octave band resolution.



10 Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on
- 48V Phantom power is switched off.

1 Page Selector RT60

Select between the RT60 run page and the test result page. The page button 🗗 toggles between these displays.



RT60 run page



- RT60 result page. Toggle with the measurement result selection (17) amongst
 - **AVRG** RT60 result page showing average of all cycles and uncertainty factor
 - CYC xx RT60 result page for each cycle
- Last RT60 result page for last cycle
- 12 Y-Axis Reverberation Time

Reverberation time in seconds. The scaling is automatically adjusted.

Reverberation Time RT60

17 Measurement Result Selection

The RT60 measurement function allows consecutive measurements within one test sequence. An averaged test result of all measurements is automatically calculated.

Select ${\rm CYC}$ and scroll with the rotary wheel through the individual test result cycles showing

Last Last Test Result Displays the result of last measurement cycle.

xx Single Test Cycle Results

The individual single test results are marked with **CYC xx**, where **xx** is an incriminating number. You may delete individual test results. The averaged reverberation time result uses only the remaining valid measurements. To delete results in a cycle,

- Select **DEL** with the rotary wheel
- Confirm with enter @.

AVRG Averaged Test Result

The averaged test results of all cycles are calculated and displayed.

13 Uncertainty Factor

The uncertainty factor is displayed in the RT60 average result page. It indicates the uncertainty of the averaged measurement results, reduces as more cycles are measured, and depends on the measured reverberation time and the bandwidth of the individual frequency band; lower bands show a higher uncertainty factor. The number of measurement cycles is displayed at (5)

(14) Overall R everberation Time Test R esult

Readout of reverberation time and uncertainty factor.

Uncertainty factor. For more details see 3.

Reverberation time measurement results.

15 X-Axis

RT60 Octave Bands 63Hz - 8 kHz

16 Cursor Readout

Select the individual frequency band and read out the following numeric measurement results

- Uncertainty factor in % or correlation in %.
- Reverberation time RT60 (T20), of the selected frequency band, in seconds (s).





18 Correlation Factor in %

The correlation factor is 100% for perfectly linear sound pressure level decay after the sound source has ceased. The natural deviation from this linearity results in lower correlation values. The correlation factor is typically 80 - 100%.

(19) Cycle R everberation Time Test Result

Readout of cycle reverberation time and correlation factor (18) when selection (17) is **CYC xx** or **Last**.

- Correlation Factor. Scale is on the right Y-axis. For more details see 18.
- Reverberation time measurement result of individual displayed cycle. Scale is on the left Y-axis.

20 Y-Axis Correlation Factor

The right Y-axis shows the correlation factor in %. The correlation factor Y-axis is displayed when test result selection (7) is **CYC xx** or **Last**.

21 Delete Cycle Results

Individual cycle results may be deleted, thereby excluding them from the calculation of the **AVRG** result.

Test Signals

The sound source shall be as omnidirectional as possible. Use either a gated pink noise or an impulse source as the test signal.

• Gated pink noise

Various gated pink noise test signals with different on/off times are offered by the NTi Audio Test CD included with the XL2 or the Minirator signal generator.

Impulse

The trigger signal is a generated impulse source, such as a starter pistol. Individual measurements, or test sequences with repeated trigger signals, can be carried out.

Ideally the XL2 should measure the reverberation time of several on/off cycles and automatically average the reverberation time results. The uncertainty factor is calculated after a minimum of three measurement cycles.



Test Signal: Pink Noise

Play a pink noise signal through an omni-directional speaker in the room under test. The sound source should be played for a long enough time period to ensure that a balance between injected and absorbed acoustic energy has been reached. For example the sound reflections should be given enough time to reach all reflective surfaces in the room. As a rule of thumb, ensure that the pink noise is played for at least the time period of the estimated RT60 test result. If in doubt, play the Minirator MR-PRO or the Test Signal CD as sound source for at least 5 seconds for each cycle. Each time the source signal stops, the XL2 recognizes this interruption, triggers, measures the decay time and calculates the reverberation time automatically.

Test Signal: Impulse

Fire an impulse sound source in the room under test, e.g. use a starter pistol. The XL2 measures the decay time and calculates the reverberation time automatically. Do not fire the next shot until the previous calculation is complete i.e. until the status (4) displays **ARMED**.

Test Preparations

The XL2 reads the electronic data sheet of any connected ASD microphones provided by NTi Audio and switches the 48V phantom power automatically on as follows:

- Connect the measurement microphone to the XL2.
- Switch on the XL2 .
- The 48V phantom power indication in the upper menu bar changes to ASD. The XL2 is ready for acoustic measurements.
- Position the XL2 at the measurement location using a microphone stand or tripod.
- Select the **RT60** function in the measurement menu and toggle with page 团 to the run page **RUN**.
- Prepare the environment for the measurement. For example mute all sound sources to establish silence.





Reverberation Time RT60

Set Level Markers

- Select the parameter **SET** with the rotary wheel 0 and press enter 0.
- The environmental noise is measured and the grey level markers are set.

(RT60 RUN XLR 🖬) 🛙 ASD 13:24 🖷
"I 🖼 Level Markers 1/3 OCT
021
90
60
₃₀ ┇╻┇╻┇╻┇╽┇┇╻╻┊┑
63 125 250 500 1k 2k 4k 8k
STAT: STOP

Get Ready for the Reverberation Time Measurement

• Protect the ears against high sound pressure levels, as the test signals might be very loud.

Setting the pink noise level:

- Start the pink noise test signal with the appropriate on/off time according to the room. Use an initial low level.
- Increase the test signal level until all level markers for the minimum test signal level are passed. Use an equalizer to push individual band levels.
- Switch off the gated pink noise signal after you have finished setting the pink noise level.



Reverberation Time RT60



Start the Measurement

- Press start **D**. The status indication switches to **ARMED**.
- Enable the test signal; e.g., switch on the gated pink noise signal or fire a starter pistol.
- The black bars have to exceed the grey bars completely.



• The RT60 calculation cycle is complete when the sound pressure level falls below the bottom of the grey level markers in every band.

• Confirmation marks J, indicating successful measurements, are displayed above each band.

Continue the Measurement

Using pink noise:

The gated pink noise test signal continues with the preset on/ off time. The XL2 automatically triggers at each cycle. Complete a minimum of three test cycles to accurately calculate the uncertainty factor.

Using an impulse source:

Use further starter pistol shots to repeat the measurements.

- Press pause III to pause the measurement.
- Prepare the next impulse sound source; e.g., starter pistol.
- Press start 💌 to continue the reverberation time measurement.
- Enable the test signal; e.g., fire the starter pistol.

An automated algorithm calculates the averaged reverb time of all measurement cycles.



Stop the Measurement and Read Out the Result

- Press stop 🔎 after all cycles are completed.
- If applicable, switch off the pink noise test signal.
- Select the result page **RES** with page **I**.



The average reverberation time for each frequency band is displayed in seconds and the measurement uncertainty in %. For more details refer to the chapter "Reverberation Time RT60".



Read Out the Individual Cycle Result

• Select **CYC** with the rotary wheel **O**, press enter **O** and select the individual cycle result with the rotary wheel **O**.



Within each cycle, the reverberation time for each frequency band is displayed in seconds and the correlation factor in %. For more details refer to the chapter "Reverberation Time RT60".



The RT60 reverberation time measurement is completed.



Error Indications

Various error indications are displayed for unsuccessful RT60 measurements. Corrupt cycles can be deleted individually. These results are then excluded from the average calculations.

• LOW LEVL

This is the abbreviation of "low test signal level" during the measurement. Increase the test signal level until the black level markers exceed the grey level markers in every frequency band. Also, ensure that the level falls below the grey level markers, otherwise use a RT60 test signal with longer on/off times.

• CORR<70%

The correlation factor is lower than 70% indicating an unreliable measurement result.

• T>18S

The measured reverberation time exceeds the time limit of 18 seconds. This is commonly caused by either an incorrect analyzer range setting or environmental noise. In the first case, choose another range setting. In the second case, ensure that the environmental noise remains constant during the measurement, and/or **SET** the level markers again while the environmental noise is high.

Polarity



Polarity

Polarity measurement is important in matching the left and right speakers for a good stereo sound image. The polarity function measures the polarity of cables, single speakers and speaker cabinets. For testing, use the polarity signal provided on the NTi Audio Test CD or the Minirator test signal generator.

The polarity of individual speakers or speaker cabinets might change with the frequency. For example, the mid-range speaker polarity may differ from the woofer polarity within the same speaker cabinet. Therefore the polarity detailed result page of the XL2 displays the measured polarity of the individual octave bands from 125 Hz to 8 kHz. This allows in-depth verification of the polarity - frequency relationship.



- Polarity is a simple test within the very complex science of signal phasing. Drivers, speakers and crossovers all cause phase shifts of the audio signal.
- The polarity of various speakers within the same cabinet may differ by design!
- Polarity testing is useful for checking the correct wiring of similar speaker systems.

Positive/Negative Page



Polarity



Detailed Result Page



Polarity result for Minirator connected directly to XL2 Analyzer

1 Input Selection

Select the signal source as follows:

- Select **Signal Source** with the rotary wheel ⁽¹⁾.
- Press enter 🕑 to select either
- VoiceUse the internal voice note microphone of theNoteXL2 for polarity measurements. This selectionMicdisables the rear speaker.
- XLR Measure the acoustic polarity with a measurement microphone. Alternatively, measure the polarity of an electrical signal connected with an XLR audio cable.

RCA Measure the polarity of an electrical signal con-**Input** nected with an RCA audio cable.

2 Polarity Test Result

Displays either **POSITIVE**, **NEGATIVE** or **???** (=undefined). For further visual indication, the limit button illuminates green for **POSITIVE** and red for **NEGATIVE** polarity.

3 Level RMS

Measures the absolute level of the input signal. The unit's dBu, dBV and V are selectable.

Polarity



4 Balance Indicator

Indication of the audio signal balance between pin 2 and 3 on the XLR input for input signals > -34 dBu.



The input signal is balanced.



The signal is unbalanced. The level of pin 2 is higher than pin 3.

The signal is unbalanced. The level of pin 3 is higher than pin 2.

5 Relative Level Indicator

The grey area shows the measured signal energy within the individual frequency bands. The bands with the biggest energy have the most effect on the actual displayed **POSI-TIVE** / **NEGATIVE** polarity result.

6 Polarity Indication

- Polarity of frequency band is positive. The polarity result is in the upper display area; the + area.
- Polarity of frequency band is negative. The polarity result is in the lower display area; the area.

7 X-Axis

Seven octave bands with center frequencies from 125 Hz to 8 kHz. **SUB** displays the polarity of Sub-woofers; frequency range < 100 Hz.

8 Negative Polarity Area

Measurement results area with negative polarity (-). The black line in the middle of relative level indicator (5) displays the measured polarity of the individual frequency band. The dashed line indicates the lower limit of the measurement result area.

(9) Uncertain Polarity Area ???

Polarity results for frequency bands in this area are uncertain. Therefore the polarity result **???** is displayed.

10 Positive Polarity Area

Measurement results area with positive polarity (+). The black line in the middle of relative level indicator (5) displays the measured polarity of the individual frequency band. The dashed line indicates the upper limit of the measurement result area.

Polarity Measurement - Getting Started

For acoustic polarity measurement you can use the internal voice note microphone, or an external measurement microphone, plugged into the XLR input. The polarity result for left and right speaker cabinets should match for a good stereo sound image.

- Feed the speaker cabinet with the polarity test signal of the Minirator.
- Adjust the test level (at Minirator or amplifier) until the test signal is clearly heard.
- Enable the polarity test signal at the left speaker cabinet; mute the right speaker cabinet.
- To activate the internal microphone, select **Signal Source** with the rotary wheel **O** and choose **Voice Note Mic** with enter **O**.
- Measure the polarity of the left speaker cabinet and save the screenshot in the memory menu .
- Enable the polarity test signal at the right speaker cabinet; mute the left speaker cabinet.
- Measure the polarity of the right speaker cabinet.
- Compare the polarity test result of both speaker cabinets.

She polarity measurement is completed.



The sample result shows the polarity as:

- Sub-woofer: Low energy, no sub installed
- Woofer: Positive
- Mid-range: Negative
- Tweeter:
 - Positive (with small levels)

The major part of the signal energy is measured in the mid-range frequency bands. Therefore the overall displayed polarity is **NEGATIVE**.



Delay Time

The delay time measurement function is suitable for accurate configuration of delay line setups, by optimizing the directionality of the signal source. The XL2 measures the delay time between the reference signal and the acoustic signal from the speaker. The acoustic delay settings are displayed, allowing you to easily set the measured delay time into the delay device in the rack.

The NTi Audio delay time test signal is provided either by the Minirator MR-PRO, MR2 or the Test CD included with the XL2.



1 Delay Signal Source

Select the signal source for the delay measurement:

- **CD Player** Using the supplied NTi Audio Test CD; this setting implies the synchronization time bar is set to 100 seconds = 1 min. 40 sec.; after 100 seconds the XL2 has to be synchronized to the delay test signal again.
- **Minirator** Due to compatibility and the accuracy of the Minirator, MR-PRO and MR2 test signals, the XL2 offers an extended time range without synchronization. Only after 300 seconds (5 minutes) does the XL2 have to be synchronized to the delay test signal again.

2 Store Button

Press the store button to record the actual delay \bigcirc of the reference speaker.

3 Reference Delay Time

Individually-stored reference delay time; e.g., speaker A as described in the following chapter "Delay Measurement - Getting Started".



4 Calculated Distance

Distance from measurement position to speaker in meters or feet, based on the defined temperature in $^\circ C$ or $^\circ F$

5 Synchronization Time Bar

The automated synchronization allows delay time measurements without any connected electrical reference signal for 100 seconds using the NTi Audio Test CD or 300 seconds using the Minirator, MR2 or MR-PRO. It displays the time remaining until the next required synchronization.

6 Calculated Delay Time: Store - Actual

Calculated difference between delay time of speaker A and speaker B as shown on the next page. The automated difference calculation simplifies the verification of delay line arrangements, such as those used in larger halls or auditoriums.

7 Actual Delay

Actual measured delay time referring to the electrical reference signal.



Delay Test utilizes VoiceNote Microphone The XL2 uses the internal VoiceNote microphone to measure the acoustic delay. Do not connect any measurement microphone to the XL2 for the delay time measurement.

Rear Speaker

The rear speaker on the XL2 is disabled during delay measurements, thereby avoiding measurement failures. The headphone output is active.



Delay Measurement - Getting Started

The delay time measurement is conducted between the synchronized electrical input signal and the acoustic signal measured by the built-in voice note microphone. In this example, the delay between speakers A and B in the illustrated auditorium is measured.



Test Preparations

- Prepare the delay test signal; use either the
 - NTi Audio Test CD
 - NTi Audio Minirator for analog systems
 - NTi Audio Digirator for digital or Dolby/DTS systems
- Start the delay test signal.
- Connect the generated delay test signal with an audio cable to the RCA or XLR input of the XL2, e.g. take the reference signal from an auxiliary channel of a mixing console.
- Select XLR or RCA input in the upper XL2 menu
- Set the actual environment temperature, to ensure that the distance is displayed correctly later.

Delay XLR	🖬) 480 13:45 📹
Delay Signal Source	Minirator
Act.Delay	ms
0.0ms- Act	ms
Distance @ 20°C	m
00:00 Synchr	onizing OS:OO

• Wait until the XL2 synchronizes to the incoming delay test signal, then the synchronization time bar fills up.



• Disconnect the audio cable, with the synchronizing signal, from the XL2, allowing you to move around freely for the delay measurement. When selecting **Minirator** as the delay signal source, the XL2 has to be synchronized to the signal source again after 5 minutes. When selecting **CD Player** as the delay signal source, the XL2 has to be synchronized to the signal source every 100 seconds.

Measure Reference Speaker A

- Start the delay test signal at speaker A, mute speaker B.
- Position yourself with the XL2 at the measurement position next to speaker B sub left as indicated above. This is the worst-case position in the auditorium with regard to acoustic delay in the sound field. The XL2 uses the VoiceNote microphone for the delay test; do not connect any measurement microphone.



The XL2 measures the delay time of speaker A in reference to the electrical input signal in milliseconds.





Store Reference

- Select the displayed **0.0 ms** below **Act. Delay** with the rotary wheel **(**) and press enter **(**).
- The reference result of speaker A is stored for the difference calculation of delay A - B.
- Stop the delay test signal at speaker A.



Measure Sub Speaker B

- Start the delay test signal at speaker B. Mute speaker A.
- Position yourself with the XL2 at the shown measurement position. The XL2 uses the VoiceNote microphone for the delay test; do not connect any measurement microphone.



The XL2 measures the delay time of speaker B in reference to the electrical input signal in milliseconds.

Result Interpretation

- The resulting time in milliseconds shows the time difference of the sound signal from speaker A and B arriving at the measurement position. Speaker B has to be delayed by the displayed difference result.
- To optimize directionality, add a further 5 ms to the calculated difference. The main signal from speaker A then arrives at the measurement position first. The acoustic signal from sub speaker B arrives 5 ms later. This improves the subjective directionality perception of the listeners.

Distance in meter/feet

The distance results in meter/feet are displayed below for easy verification of the test results accuracy. The readings are based on 330 m/s sound speed at 0°C / 32°F.

Automated Difference Calculation

• The delay time difference between speaker A and B is automatically calculated and shown below the Act. Delay result.

The delay time is measured.

Delay Test utilizes VoiceNote Microphone The XL2 uses the internal VoiceNote microphone to measure the acoustic delay. Do not connect any measurement microphone to the XL2 for the delay time measurement.

Measurement Position

Do not position the XL2 too close to reflecting surfaces, such as walls or floors. The reflections are likely to prevent accurate measurements.



1/12 Octave + Tolerance (optional)

The Spectral Limits Option extends the XL2 function range with an RTA analyzer with spectral resolutions from 1/1 octave down to 1/12th octave. Trace capturing, relative curve display and comprehensive tolerance handling are supported in the **FFT** Analysis and the high resolution **1/12 Oct + Tol** spectral analysis.

Features

- High resolution RTA function "1/12 Oct + Tol" with selectable 1/1, 1/3, 1/6 and 1/12 octave spectral resolution
- Capturing of multiple readings into the internal memory
- Comparing measurement results against captures with relative or absolute curve display
- Comprehensive tolerance handling with tolerance masks based on captures for passed/failed measurements
- Export and import of tolerance and capture files
- Frequency band listening at rear speaker

Applications

- PA-rental: Verifying the frequency response of returned speakers and microphones against reference records ensuring they are back in stock in good working condition.
- Industrial quality testing: Passed/Failed test in the production line or during service of any audible quality criteria, such as with motors, machines, vacuum cleaners, ...
- Cinema: Comparing frequency responses against an ideal response curve according the X-Curve requirements.

Tolerance Handling

The XL2 Analyzer compares spectral measurements against a tolerance band and visualizes exceptions in every frequency band. The passed/failed condition is further visualized by the limit button and forwarded to the I/O interface of the instrument in order to drive an external alarm device such the accessory SPL Stack Light.

Tolerance curves can either be imported from txt-files or directly derived from captured measurements. The XL2 calculates tolerance bands based on

- Single captured results
- Manually generated txt-files on the PC
- Mean average of multiple captures
- Min/Max curves of multiple captures



- 1 Result Symbols / Capture & Start Tolerance Mode This field offers two functions:
 - Symbols for measurement results 2
 - Upper parameter displayed as line.
 - Lower parameter displayed as bargraph.
 - Capture & Start Tolerance Mode The displayed spectral readings may be captured in C1 - C8 for
 - Comparing measurement results against captures with relative or absolute curve display.
 - Creating tolerance masks based on captures for passed / failed measurements.
 - **Capture EQ** Captures the upper parameter
 - Capture Live Captures the lower parameter





Manage	Allows to rename captures, clear re-	
captures	corded captures, save captures to SD-	
	Card for export from the XL2 or load	
	captures from SD-Card for import to	
	the XL2.	
captures	Card for export from the XL2 or load captures from SD-Card for import to	

Start toleranceStarts the tolerance mode for passed/
failed measurements comparing the
actual measurement results against a
tolerance band.

2 Measurement Result

Actual level result of the indicated frequency band. The cursor readout displays the center frequency and the level of the band indicated by the arrow.

- Upper parameter displayed as line.
- Lower parameter displayed as bargraph.

- **3 Y-Scale setting**
 - Select the Y-Axis with the rotary wheel O and confirm with enter O.
 - Select the zoom factor between **20, 10, 5**, **2.5 dB/div** and confirm with enter
 - Scroll up and down with the rotary wheel ⁽¹⁾ to select the Y-axis range.
 - Confirm with enter .

4 Spectral Measurement Result

Spectral results in 1/1, 1/3, 1/6 or 1/12 octave band resolution. Adjust the resolution at (12).

1/12 Octave + Tolerance



5 Input Range

The XL2 Analyzer provides three input ranges to accommodate the wide range of input signals. The individual ranges are based on the microphone sensitivity setting in the calibration menu of the XL2. For example at a sensitivity S=20mV/Pa the input ranges are

- Lower range: 10 110 dBSPL
- Middle range: 30 130 dBSPL
- Upper range: 50 150 dBSPL

Select the lowest possible input range according to the maximum level expected during the measurement; e.g., if the sound pressure will be always below 110 dBSPL, then select the lowest input range 10 - 110 dBSPL.

6 Time Weighting

Offers selectable time weighting of 0.1, 0.2, 0.5 and 1.0 second. Applications:

Short Time Weighting	High resolution in time with minimum averaging
Long Time	Low resolution in time with longer aver-
Weighting	aging

7 Run Indication

The run indication shows the measurement status running, paused or stopped. Various measurement settings are locked during ongoing measurements, such as changing the input ranging or the preset measurement time. For passed/failed measurements with the Spectral Limits option the run indication may display **A** for a preset automated level trigger.

(8) Actual Measurement Time

Counts actual measurement time in hrs:min:sec. Supports setting of time modes: continuous and single.

CIT Timer Mode Continuous

(applicable for standard measurements) All values are recorded and monitored continuously after starting a measurement with start **D**. The actual measurement time is shown.

Timer Mode Single

Automatically stops the measurement after the pre-set measurement time.

- Set the required measurement time.
- Start the measurement 🖭



9 Preset M easurement Time

Adjustment of preset measurement time for single timer setting.

10 Broadband Results

User-selectable display of broadband results:

- **A** A-weighted broadband sound pressure level
- **C** C-weighted broadband sound pressure level
- **Z** Broadband sound pressure level without any frequency weighting
- none

11 Readout Frequency

You may select any frequency to read out individual levels. The selected frequency is indicated by the cursor arrow.

Choose between the following settings:

- The cursor follows automatically the highest level, e.g. tracing feedback frequencies at live sound.
 - Select the frequency with the rotary wheel ⁽¹⁾
 - Press enter 🕑.
 - You may readout any individual frequency.
 - Return to auto frequency with enter 🕑.
 - The cursor returns to the frequency with highest level bar.
- H The cursor read out may be set manually to any frequency band or wideband level. The read out remains at the selected frequency.
 - Select the frequency with the rotary wheel ⁽¹⁾
 - Press enter 🕑.
 - Select the frequency of interest.
 - Confirm with enter 🕑.
 - The cursor readout 2 displays the measurement results of the selected frequency band.

1/12 Octave + Tolerance



12 Setting of Test Result Resolution

Set the spectral result display to 1/1, 1/3, 1/6 or 1/12 octave band resolution as follows:

- Turn the rotary wheel 0 to select the parameter 12.
- Press enter 🕑 to open the selection window.
- Turn the rotary wheel ⁽¹⁾ to select **1/1 OCT**, **1/3 OCT**, **1/6 OCT** or **1/12 OCT**
- Press enter 🕑 to confirm.

13 Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on.
- 48V Phantom power is switched off.

Band Listening

The selected frequency band of the input signal is audible at the rear speaker or headphone output.

- Press and hold the speaker button ④ down. The pop-up window **Volume** is displayed.
- Keep the speaker button (pressed and press enter (at the same time.

The band listening is activated.





1/12 Octave - Getting Started

Test Preparations

The XL2 reads the electronic data sheet of any connected ASD microphones provided by NTi Audio and switches the 48V phantom power automatically on as follows:

- Connect the measurement microphone to the XL2.
- Switch on the XL2 ⁽³⁾.
- The 48V phantom power indication in the upper menu bar changes to ASD. The XL2 is ready for acoustic measurements.
- Position the XL2 at the measurement location using a microphone stand or tripod.
- Select the 1/12 Oct + Tol function in the measurement menu.



The displayed wideband and spectral levels are measured and stored simultaneously.

Configuration

The XL2 displays two different sound pressure levels at the same time. You configure to display either the levels Live, Max, Min, EQ or captured results.

• Select the upper result parameter with the rotary wheel ⁽¹⁾



• Confirm with enter .

1/12 Octave + Tolerance



Select Upper/Lower Reading

- She pop-up window **Select reading** shows up.
- Select the time-averaged sound level **EQ**. All available selections are Z-weighted (= no weighting).



• Confirm with enter •

She pop-up window **Select reading mode** shows up.



- Select **EQ** for normal absolute display of measurements.
- Follow the same instruction and select the lower spectral reading **Live**.





Select Input Range

Select the lowest possible input range according to the maximum level expected during the measurement.

- Select the input range **RNGE** and press enter *e*.
- Turn the rotary wheel ⁽¹⁾ to set the applicable input range and confirm with enter ⁽²⁾.



Start and Stop the Measurement

- The XL2 measures the selected sound pressure levels **Live** and **EQ**. All available sound levels are Z-weighted (= no weighting).
- Press start 🔳.
- The run indication switches to running . The actual sound level Live and the averaged level EQ are displayed. The parameter **RUN** flashes in the memory menu.



• Press the stop button 🕨 to stop the measurement.

She measurement is completed.

1/12 Octave + Tolerance

Save the Measurement Result

• Select the memory menu 🖃 with the rotary wheel 🕲 and confirm with enter 🕑.

🖒 The memory menu opens.



• Press enter
to store the test results as an ASCII-file or

select Save Screenshot to store the screenshot.

Capture References and Create Tolerances

The Spectral Limits Option extends the function range of the XL2 with trace capturing, relative curve display and comprehensive tolerance handling for the **FFT** Analysis and the high resolution **1/12 Oct + Tol** spectral analysis.

Features:

- Capturing of multiple readings into the internal memory
- Comparing measurement results against captures with relative or absolute curve display
- Comprehensive tolerance handling
- Creating tolerance masks based on captures for passed / failed measurements
- Export and import of tolerance and capture files
- True peak level in 1/1 and 1/3 octave resolution
- High-resolution Zoom-FFT up to 0.4 Hz steps in the frequency range 5 Hz 20 kHz

Read the detailed description in the separate chapter "Spectral Limits Option (Capture + Tolerances)" in this manual.





Speech Intelligibility STIPA (optional)

The STIPA analyzer option allows reliable measurement of the speech transmission index within 15 seconds. Besides the single value STI or CIS test result, a detailed view of the modulation indices and individual band level results is provided. The STIPA analyzer meets the latest standard edition 4.0 IEC 60268-16, released in 2011. The XL2 supports also noise corrections, automated averaging of measurements and the older standard editions 2.0 and 3.0.

The speech intelligibility depends on the following system parameters:

- Signal-to-noise ratio
- Psychoacoustic masking effects
- Sound pressure level
- Ambient noise level
- Reverberation time RT60
- Reflections
- Frequency response
- Distortion

The speech intelligibility measurement STIPA is an option for the XL2 Audio and Acoustic Analyzer. Ask the local representative for purchasing details.

STIPA Signal Source

Utilize the applicable STIPA signal source dedicated for your installation configuration:

- NTi Audio
- The NTi Audio TalkBox simulates a person talking TalkBox at a precise acoustic level, enabling the measurement of the complete signal chain including the microphone
 - Place the NTi Audio TalkBox in front of the microphone at the typical position of the talking persons head.
 - Select Track 1 for the STIPA test signal.
 - Select Output Mode to Speaker; you should hear the STIPA test signal.
- The Minirator MR-PRO is used for electrical signal Minirator MR-PRO injection into public address systems that commonly use alarm messages from a hard drive (systems without a microphone).
- The NTi Audio CD "STIPA V1.1", included in the CD Player package, can be used with a professional CD player. For details see the chapter "STIPA Measurement Hints"

The standard STIPA signal is based on a band-limited random noise of a male speech spectrum.

STIPA



STIPA Numeric Result Page

STIPA Table Result Page



STIPA 123 XLR				
FINISHED				
Band 8000 Hz 4000 Hz 2000 Hz 1000 Hz 500 Hz 250 Hz 125 Hz	53.2dB 1.01 0. 59.1dB 0.98 1. 65.1dB 0.98 1. 71.2dB 1.01 0. 77.1dB 1.01 1. 80.9dB 1.04 1.	1F2 .98 / .00 / .01 / .95 / .04 / .02 / .04 /		
•		■ A.		
Bad Po	orFairGood Exl	nt AT		
(11)	12 (13)	14		

STIPA



1 Run Indication

This icon displays the run status of the STIPA measurement.

2 Measurement Result

- Single value speech transmission index result.
- Repeatability of measurement result: Since a dedicated noise test signal is used, the STIPA result may deviate by maximum 0.03 STI (=Max-Min) at the same measurement position.

3 Sound Level LAeq

Shows the time-averaged sound level of the 15 seconds STIPA measurement cycle time.

4 Sound Level LAS

Actual sound pressure level.

5 Analog STIPA Bargraph

Bargraph display and interpretation of the speech intelligibility measurement result

- Good 0.60 0.75 STI
- Fair 0.45 0.60 STI
- Poor 0.30 0.45 STI
- Bad 0.00 0.30 STI

6 Qualification Scale

The STI value is shown as a letter of the qualification scale below, which informs about the typical STI requirements for dedicated applications.

Band	STI Range	Examples of typical uses
A+	> 0.76	recording studios
A	0.72 - 0.76	theatres, speech auditoria, parliaments, courts
В	0.68 - 0.72	theatres, speech auditoria, parliaments, courts
С	0.64 - 0.68	teleconference, theatres
D	0.60 - 0.64	class rooms, concert halls
E	0.56 - 0.60	concert halls, modern churches
F	0.52 - 0.56	PA in shopping malls, public offices, cathedrals
G	0.48 - 0.52	PA in shopping malls, public offices
Н	0.44 - 0.48	PA in difficult acoustic environments
I	0.40 - 0.44	PA in very difficult spaces
J	0.36 - 0.40	not suitable for PA systems
U	< 0.36	not suitable for PA systems
109

STIPA

9 Progress Bar

Measurement status indication; the single STIPA measurements takes 15 seconds. The actual measurement time and measurement status are indicated here.

10 Page Selector

123

Tab

Cor

Select between these pages:



Table result page with modulation indexes and





On-site averaging of results.



The page button 🗗 toggles between the **123** - **Tab** - **Avr** screens. These pages show the speech intelligibility measurement results. The page **Cor** offers setting the noise spectra prior the STI measurement.

(1) Octave Bands

Frequencies 125 Hz - 8 kHz in 1/1 octave band resolution.

12 Sound Pressure Level Lea

Individual time-averaged octave band level Leq.

7 Measurement Unit

The STIPA result is displayed in STI (Speech Transmission Index) or CIS (Common Intelligibility Scale), whereby CIS is calculated as CIS = $1 + \log STI$.

8 Edition of Standard IEC60268-16

- ed4.0 actual edition released in 2011 with continuous level dependent auditory masking function
- ed3.0 old edition released in 2003 with stepped level dependent auditory masking function
- ed2.0 old edition released in 1998 with fixed masking function







STIPA

(13) STIPA Modulation Index MF1, MF2

For good speech intelligibility it is mandatory that the integrity of the transmitted voice signal modulations are preserved. Therefore STIPA is based on measuring the MTF (Modulation Transfer Function). This function quantifies the degree to which the voice modulations are preserved in individual octave bands. STIPA determines the MTF by analyzing the seven frequency bands. Each band is modulated with two frequencies, resulting in the modulation index 1 and index 2. All indexes together combined with psychoacoustic models provide the single-value STIPA result.

14 Error Detection

The STIPA measurement function has a built in error detection which helps you identifying faulty measurements. The error detection checks the following parameters:

- Invalid modulation indices (MF1 or MF2 > 1.3)
- Irregularities during ongoing measurements

Both parameters depend on the amount of impulsive environmental noise, which influences the accuracy of any STIPA measurement.

The STIPA numeric result page displays the measurement result interrupted by the flashing "???" question marks if the error detection finds any problem. Furthermore, octave band results detected with irregularities are marked in the STIPA table result page with a question mark (?). The error indication might be caused by either

- Missing test signal level or
- Impulsive background noise events

If the error detection finds any faults, you should repeat the measurement and compare the STIPA test result with the previous ones.

STIPA

Start STIPA Test Signal

Select the STIPA signal source according your application requirements.

- Switch on the STIPA test signal at the signal source.
- Set the acoustic sound pressure level of the PA system to simulate the typical announcement level; e.g., $L_{\Delta S} = 85 \text{ dB}$.





phantom power automatically on as follows:

STIPA Measurement - Getting Started

The XL2 reads the electronic data sheet of the connected NTi Audio measurement microphone and switches the 48V

• Connect the measurement microphone to the XL2 Analyzer.

The 48V phantom power indication in the upper menu bar

Test Preparations

• Switch on the XI 2

Ð

- Position the XL2 at the measurement location using a microphone stand or tripod.
- Select the STIPA measurement function in the measurement menu
- Prepare the environment for the measurement. For example mute all sound sources to establish silence.

No impulsive noise shall occur during the STIPA measurement as well no speaking or other noise sources should be allowed near the measurement microphone.







Start STIPA Measurement

- Press start 🖭.
- The progress bar switches to **RUNNING**. The test result tendency is shown on the STIPA bargraph, marked with **Bad**, **Poor**, **Fair**, **Good** and **ExInt**.

Stop STIPA Measurement

After the period of 15 seconds the STIPA measurement finishes automatically. The progress bar indications switches to **FINISHED** and the final STIPA test result is displayed.

• Switch off the STIPA test signal.





Store STIPA Measurement Result

- To store the test result, select the memory menu
 with the rotary wheel
 and press enter
 .
- Select **Save Test** to store the test results as an ASCII-file or select **Screenshot** to store the screenshot.
- The speech intelligibility STIPA is measured and stored on the SD Card including sound pressure level and modulation indexes.

Averaging of STIPA Results

The standard IEC 60268-16 recommends averaging two or three subsequent results taken at the same measurement location.

The German Standard VDE 0833-4 requires performing minimum three subsequent measurements for one measurement position in case of STI < 0.63.

The XL2 Analyzer offers automated averaging of two up to eight STIPA results based on these standard requirements.



STIPA

Start Averaging

• Select the averaging page Avr.

(STI-PA Avr XLR) 4 850 15:41 (
	Reading	
1. 2. 3.	STI STI STI	
4. 5.	STI STI	
6. 7.	STI STI	
<u> </u>	STI STI	Δ
Bad Pa	orFairGood	ExInt

- Turn the rotary wheel ⁽¹⁾ to select the parameter **START NEW** and press enter ⁽²⁾.
- The first STIPA measurement starts automatically. It is labeled Cyc 1.

Add Cycles

• Press enter 🕑 to confirm Add Cycle.



- Repeat the measurement at the same position as required.
- The XL2 performs further STIPA measurements and adds them to the list.





STIPA

played.

Finish

• Choose **Finish** to end the averaging.

(STIPA Avr XLR 🗖) 4 850 15:47 🖛 Averaging:
Cycle Reading 1. 0.856STI / 2. 0.832STI / 3. 0.848STI / 4STI 5STI 6STI 7STI 8STI
Ø 0.85sti / ∆0.024
Bad PoorFairGood ExInt A+

The STIPA average and the deviation is displayed for documentation.

STIPA with Ambient Noise Correction

Measuring the speech intelligibility index under realistic environmental conditions is often not feasible; e.g., playing the test signal in a railway station at emergency levels during peak hours will irritate passengers. Additionally, at rush-hour the characteristics of ambient noise might be highly impulsive, while a pre-requisite for accurate STIPA measurements is a negligible impulsivity in the ambient noise. Under such circumstances the STIPA measurement should be shifted to a more suitable time of the day; e.g. night time.

Measurement Sequence

- First measure the ambient noise
- Secondly measure the speech intelligibility STIPA

This sequence simplifies the STIPA measurement as follows: The XL2 immediately displays the STIPA result with ambient noise correction. This result provides a guideline if repeated measurements at the same location with averaging is required. For more details see the hints at the end of this STIPA chapter. **Enable Ambient Noise Correction**

- Select the correction page Cor.
- Turn the rotary wheel ⁽¹⁾ to select the parameter **Ambient Noise Correction** and press enter ⁽²⁾.
- $^{\textcircled{}}$ The correction enables and the XL2 displays this screen:

(STIPA Cor X	LR 🖬 🗍 🕸 ASD 16:20 🖷	
Ambient Noise	Correction: On	
Noise to add:	Band LEQ EDIT 8000Hz 55.2dB	
MEASURE 30s	4000Hz 61.2dB 2000Hz 67.2dB	
LOAD SAVE	1000Hz 73.2dB 500Hz 79.2dB 250Hz 82.9dB 125Hz 82.9dB	
source of data: Default Settings		
Bad PoorFairGood ExInt		

STIPA



Commissioning New PA-Systems

New announcement systems are commissioned e.g. at public areas prior the grand opening. Thus the actual ambient noise caused by the public is not available yet during the STIPA measurement. Here you may simulate the real-life condition with ambient noise data enabled by one of the following modes:

- Utilize a reference noise file, which might be applicable for the actual project
 - Measure the ambient noise at another similar project and store this as reference noise file.
 - Back on the actual project select the parameter **Load** with the rotary wheel ^(C) and press enter ^(C).
 - Select the reference noise file, which shall be utilized as ambient noise correction for your STIPA measurements.
- Edit actual noise data
 - Select the parameter Edit with the rotary wheel O and press enter O.
 - Turn the rotary wheel to adjust the noise level.
 - Press enter e to select the next frequency band.

Next go to "STIPA Measurements". Skip the next step "Measure Ambient Noise".

Measure Ambient Noise

- Position the microphone at the STIPA measurement point.
- Select **Measure** (without any test signal presence).
- The XL2 measures the ambient noise and displays the **LEQ** octave band result.

(STIPA Cor XI	LR 🔳 🛛 ASD 16:30 🖷	
Ambient Noise	Correction: On	
Noise to add:	Band LEQ EDIT	
LIANSUNA 30 s IIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIII	8000Hz 48.6dB 4000Hz 62.1dB 2000Hz 50.6dB 1000Hz 46.8dB 500Hz 48.2dB 250Hz 51.0dB 125Hz 51.0dB	
SOURCE OF DATA: Measured 2012-10-23 10:50		
Bad PoorFairGood ExInt		





STIPA Measurement

- Select the STIPA result page 123.
- Perform the default STIPA measurement.



The XL2 displays the corrected STIPA result in big letters. The actual measured STIPA result is listed below in small letters.

STIPA Post-Processing

In case no ambient noise correction has been carried out onsite with the XL2 Analyzer, then you may post process your measurement data on the PC. The NTi Audio STIPA Reporting Tool combines the STIPA measurement taken at quiet conditions and the actual ambient noise caused by the public, e.g. during day-time. This emulates the expected STIPA value at real-life conditions.

STIPA Reporting Tool

The STIPA Reporting Tool creates measurement reports according to the IEC 60268-16 and VDE0833 standards. Import the data directly from your XL2 including the ambient noise measurements. The corresponding speech intelligibility STI or CIS values are shown.

The STIPA Reporting Tool is free to download on the XL2 support website http://my.nti-audio.com for all registered users. (Enable all macros when opening the document.)

System Requirements:

- PC running Windows XP or Windows 7
- Excel 2007 or Excel 2010 (32 bit or 64 bit versions)

STIPA

Hints: German Standard VDE 0833-4 Requirements

STI > 0.63 One single measurement is sufficient.

- STI < 0.63 Perform three subsequent measurements for this measurement position.
 - If the maximum result deviation of these three measurements is > 0.03 then further three measurements shall be performed.
 - If the maximum result deviation of these measurements is > 0.05 then the cause of this instability shall be evaluated and removed.
 - The arithmetic average of the performed three or six measurements has to be reported.

Utilize the STIPA Reporting Tool for the documentation of your measurements according the standard.

An STI > 0.63 ensures, that the speech intelligibility is higher than 0.5 with a confidence level of 95%.

Hints: Ambient Noise

- The ambient noise has to be sufficiently static during the measurement. Impulsive ambient noise during the measurement, such as speech, causes severe measurement errors. The STIPA result is usually too high.
- Fluctuating noise is detected by measuring the direct STI in the absence of the test signal. Carry out these measurements at least at a representative set of locations. If the STI is too high (e.g. STI > 0,2), the measurement results are likely to be erroneous. Then the STIPA measurement should be carried out without this noise being present. Utilize the ambient noise correction for such instances.
- At locations with varying conditions (e.g., some public areas with few people and other areas with crowds) the worstcase STIPA results should be measured. Consult the local regulations (e.g., the NFPA code in the U.S.) for directives concerning measurement locations and number of required STIPA measurements under which circumstances.





Hints: CD-Player

- Only high-quality CD-Players should be used to reproduce the STIPA test signal as only limited time-shifts (+/- 20 ppm) ensure reliable STIPA test results. Pitch control and shock protection should be disabled. We recommend that only professional CD-Players be used. Verify the time shift of the CD-Player with a 1 kHz test signal:
 - Insert the NTi Audio Test CD into the CD-Player and start track 1, which is the 1 kHz test signal.
 - Connect the XL2 directly to the audio output and measure the signal frequency in RMS/THD+N mode. The displayed frequency should be in the range from 0.99998 kHz to 1.00002 kHz
- STIPA test signals from other test system manufacturers may sound similar but are not compatible. Only the NTi Audio STIPA test signal CD V1.1 or higher should be used in combination with the XL2.

Hints: Measurement

- Select typical locations based upon such regulations, or typically position the microphone at 1 1.2 meters above ground in sitting areas or 1.5 1.8 meters in standing areas (typical positions are not directly in front of the speakers).
- The person taking the measurements should be out of the acoustic field, so as not to affect the measurement results. For this purpose the measurement microphone can be mounted on a microphone stand and connected with the ASD-Cable to the XL2.
- Low STIPA readings can be caused by
 - Excessive sound reverberation, echoes or reflections
 - Poor speaker directivity or speaker coverage
 - Speaker level setting incorrect; e.g. low signal-to-noise ratio.

Audio Analyzer



6. Audio Analyzer

Besides the comprehensive sound level meter and acoustic measurement functions, the XL2 Audio and Acoustic Analyzer also offers the analog audio measurement functions **RMS/THD+N** and **Oscilloscope**.

RMS/THD+N

The XL2 Audio and Acoustic Analyzer measures the parameters Level RMS, THD+N and frequency simultaneously.

	(RMS/THD	XLR 💼) 4 480 13:48 🚛	5
1	Filter	Z-WEIGHTING	0
2	···LVLRMS	18.0 dBu	
3	··· THDN	-93.7 dB	
4	···FREQ	100.001 Hz	

1 Filter

- **Z-Weighting** Frequency Z-weighting with flat frequency response from 20 Hz to 22 kHz. Default measurement setting.
- A-Weighting Frequency A-weighting according IEC 61672 for low noise levels.
- **C-Weighting** Frequency C-weighting according IEC 61672 for high noise levels.
- HP 100Hz Highpass 100 Hz, -200 dB/dec., maximally flat magnitude filter, e.g. -60 dB @ 50 Hz
- HP 400Hz Highpass 400 Hz, -120 dB/dec., maximally flat magnitude filter, attenuates 50/60Hz components of test signal
- **HP 19k**Highpass 19 kHz; e.g., to measure a
20 kHz pilot tone level of a public address
system without disturbing the public.
- **22.4 22.4k** Bandpass filter 22.4 Hz 22.4 kHz (IEC 468-4)



2 Level RMS

Measures the absolute level of the input signal. The unit's dBu, dBV, V and dBSPL are selectable.

3 THD+N

Measures the total harmonic distortion and noise of the input signal. The distortion measurement, expressed in dB, linear or in %, is conducted within the bandwidth 10 Hz - 20 kHz.

4 Frequency in Hz

The XL2 automatically extracts and measures the frequency of the fundamental signal. The XL2 frequency counter technology even reads the frequency correctly for heavily distorted signals.

5 Balance Indicator

Indication of the audio signal balance between pin 2 and 3 on the XLR input for input signals > -34 dBu.



The input signal is balanced.



The signal is unbalanced. The level at pin 2 is higher than at pin 3.

The signal is unbalanced. The level at pin 3 is higher than at pin 2.

PreNotch Listening

The input signal is audible at the rear speaker or headphone output.

- Press and hold down the speaker button O.
- The pop-up window volume is displayed.

D	(<u>RMS/THD) XLR ■</u>) 4 480 13348 ■ 2- <u>∓</u> -3 Filter Z-WEIGHTING
	THDN
	FREQ 1.00000kHz

PostNotch Listening

The main frequency component is filtered out by a notch filter, leaving only the remaining signal audible.

- Press the speaker button ④ to display the volume window.
- Additionally press enter 🕑
- The post notch listening with auto-gain function is activated. The maximum gain setting is 0 dB.

(∢ Vo	lume	0.0dB
-80	OÅB	+80
OTSTEN	9 Pos	t Notch

e.g. using a 100 Hz sine test signal connected to the XL2 input, with the 100 Hz main frequency filtered by the notch filter, the complete frequency spectrum is available at the speaker or headphone output. This feature allows you to listen to the actual distortion signal. You therefore hear a noise from the speaker even without any connected inputs signal!



Oscilloscope

The scope function visualizes the waveform of the electrical input signal. It automatically triggers to the fundamental frequency and selects the scaling of the X-axis (time) and Y-axis (level) accordingly. The fundamental, or most dominant, frequency of the input signals is displayed above the scope screen.







7. Calibration

The XL2 Audio and Acoustic Analyzer meets or exceeds the specifications listed in the chapter "Technical Data".

Instrument Calibration

In order to maintain the high accuracy, annual calibration of the XL2 Analyzer and the measurement microphone is recommended. The calibration service verifies the complete product specifications, differences to the last calibration and measures the frequency response of the measurement microphone. To enquire about calibration service, follow the RMA guide lines at www.nti-audio.com.

Microphone Sensitivity

NTi Audio's ASD measurement microphones and the microphone pre-amplifier MA220 include an electronic data sheet. The Automated Sensor Detection (ASD) of the XL2 recognizes the sensitivity and calibration data of the connected microphone. The electronic data sheet, including the microphone sensitivity, is displayed in the menu **Calibrte**. The sound level meter and calibrator should be exposed to the same environmental conditions prior any calibration for the following recommended periods:

- 10 minutes after a temperature change of 10°C.
- 15 seconds after the static air pressure has changed by 5 kPa.
- 10 minutes after a humidity change of 30% without condensation.

The described calibration method and correction data apply for the following environmental conditions:

- Temperature: -10 °C to +50 °C
- Static air pressure: 65 kPa to 108 kPa
- Humidity: 25 % to 90 % r.H. without dew from -10 °C to +39 °C





1 Phantom Power Supply

- **48V** The XL2 provides 48 V phantom power supply to the connection microphone or sensor.
- ASD An NTi Audio measurement microphone with automated sensor detection ASD is connected. The XL2 reads the electronic data sheet of the microphone and switches the 48V phantom power automatically on
- 48V Phantom power is switched off.

2 Factory Sensitivity

Factory sensitivity of measurement microphone in mV/Pa. The sensor detection of the XL2 reads the sensitivity automatically from the electronic datasheet of the connected NTi Audio measurement microphone or microphone pre-amplifier.

3 User Calibration

Sensitivity setting with an external calibrator.

4 Level of External Calibrator

The reference calibration level is 114 dB. The XL2 supports a level setting between 0 dB and 200 dB.





Calibration

5 Level Type

LZF Default setting.

- **LHP100F** Suppressing any low frequency noise, for example wind during outdoor calibrations.
- (6) Electronic Data Sheet of Measurement Microphone Read out data from the connected microphone.

User Calibration - With External Precision Calibrator User-sensitivity setting for measurement microphones with an external calibrator.

- Turn the rotary wheel I to select the parameter Calibration Level 4 and press enter .
- Set the calibration level according to the calibrator used.
- Generate the specific reference signal with the calibrator onto the microphone.
- Turn the rotary wheel ⁽¹⁾ to select the parameter **RUN** next to **User Calibration** ⁽³⁾ and press enter ⁽²⁾.
- The pop-up window **Calibration: Calibration running** ... shows up and changes to **Calibration: Successfully fin-ished!** for a successfully completed calibration.
- The user calibration with an external calibrator is completed.

User Sensitivity

The XL2 writes the user-sensitivity setting onto the electronic data sheet of the connected NTi Audio measurement microphone or microphone preamplifier. All subsequent measurements automatically utilize this user-sensitivity.

Calibration



User Calibration - Manual Sensitivity Setting

In case no external calibration device is available, set the sensitivity of non-NTi Audio microphones manually according to the product specifications as follows:

- Select **Sensitivity** 2 with the rotary wheel **3**.
- Press enter O and adjust the sensitivity with the rotary wheel O in 0.1 mV/Pa steps. The setting range 100 μ V/Pa 9.99 V/Pa is supported.
- Confirm with enter 🕑.

The manual user calibration is completed.

8. Profiles

Profiles are templates of measurement setups. You may define which profile is loaded at startup, so that the measurements are carried out always with the same setup and instrument settings.

A remarkable advantage of the XL2 is the flexible user interface. The measurement screens may be personalized and only a limited subset of the comprehensive functionality enabled to create simplified operator interfaces. Advanced users may access all menus, settings and store their preferred measurement configurations. For simplicity the XL2 Audio and Acoustic Analyzer offers fixed operator interfaces with specific pre-defined settings. The XL2 can boot up with exactly the required measurement parameters according to the pre-defined instrument configurations. This ensures accurate measurements are achieved in both advanced and basic usage environments.

Profiles - Getting Started

Activate Profile Selection at Startup

- Select **Show profile list** 1 in the **System** menu.
- Press enter 🕑 to activate **Yes**. This will show the profile list at startup.

(System Set	🗖 (ASD 12:59 🖷
Auto power off: Backlight:	10 Min Auto On/Dimmed
Date: Time:	2012-07-27 08::2:53
Action at power Show profile list: Phantom power:	Yes
Speaker: USB mode: Dec. separator:	Enabled Ask on connect
	Auto power off: Backlight: Date: Time: Action at power Phantom power: Speaker:



Preset Measurement Function

- Start the XL2 Analyzer in Full mode.
- Select the typically-used measurement function and adjust the measurement parameters according to the requirements.

For example, measuring the sound levels $L_{AF},\ L_{Aeq}$ and L_{AFmax} for 10 seconds:



Store MyFirstProfile

- Select **Profile** in the main menu and press enter
- The profile menu opens.

SLMeter 123 X Profile Select Exit Profile Manage	8.6dB
L _{fleg}	dB
LAFmax	dB
30 RNGE 130	■ SET 00:00:10 →1 00:00:00



- Select Manage and press enter •.
- The Manage Profile menu opens.





- Select Save Profile and press enter P.
- The Save Profile menu opens.



- Select **Rename** and press enter .
- Set the profile name to **MyFirstProfile**. The maximum name length is 20 characters.



- Select Configure available screens and press enter $\boldsymbol{\Theta}$.



- Select additional available pages/measurement functions for the profile; e.g., RTA **IIII**.
- Press escape esc to return to the **Save Profile** screen.
- Set **Append mode available: yes** in order to have the append data feature available in the memory menu **a**.
- Select **Save** and press enter **e**.
- MyFirstProfile is saved in the internal XL2 memory.

Select Profile at Startup

- Switch on the XL2 ⁽²⁾
- The profile selection is shown on the display.



- Select the profile **MyFirstProfile** and press enter
- The XL2 starts up with the pre-configured settings for sound level monitoring.



Select Profile during Operation

- Select Profile in the main menu and press enter $\textcircled{\textbf{P}}.$
- The profile menu opens.

(SLMeter 123 XL	R 💼) 🛛 ASD 13:24 🗰
Profile	
Exit Profile Manage	.6dB
Lfleg	dB
LAFma×	dB
111111111 30 RNGE 130	■ set 00:00:10 →1 00:00:00

• Choose **Select** and load the required profile.

Run Measurement

The profile name is displayed when a measurement function is highlighted. The main menu offers only the measurement functions enabled by the profile, plus the calibration and profile menu.

• Press start 🖭



- Wait 10 seconds until the measurement is finished.
- To return to the full mode either switch off the XL2 (2) again or select **EXIT Profile** in the main menu.



Export MyFirstProfile to PC

If you have two or more XL2 Audio and Acoustic Analyzers, you may like to transfer the same profiles to the other instruments.

- Select **Profile** in the main menu and press enter
- Select **Manage** and press enter **e**.
- Select Export to SD-Card and press enter @.
- The XL2 generates a folder on the SD Card called "Profiles". All available profiles are exported into this folder.
- Connect the XL2 to the PC.
- Copy/paste the profile called **MyFirstProfile** to the PC.

Import MyFirstProfile from PC

- Connect another XL2 to the PC and generate a folder called "Profiles" on the SD Card.
- Copy the profile called **MyFirstProfile** into the folder "Profiles".
- Start up the XL2, select **Profile** in the main menu and press enter ④.
- Select Manage and press enter e.
- Select Import from SD-Card and press enter P.
- MyFirstProfile is available on the other XL2 Analyzer.



- In the exporting XL2 system menu, select the individual option and toggle with enter to Hidden.
- Store the profile in the XL2 Analyzer.



The XL2 can store up to 20 individual profiles in the internal system memory and offers three different profile types:

Types	Description	
Factory default profiles	The XL2 includes factory default profiles, which are marked with "#" in the beginning of the file name. Sample filename: #DIN15905-5.prfs	
	The factory default profiles are a free download for all registered XL2 cus- tomers on the support website at http://my.nti-audio.com. For details see the chapter "Registration" in this manual.	
	For detailed settings of factory default profiles please see the Appendix 2 in this manual.	
User profiles	User profiles are profiles generated by you on the XL2 Analyzer. Sample filename: MyFirstProfile.prfl	

Secured Secured profiles are locked user profiles with profiles overwrite protection indicated by the file name ending in "xxx.prfs". This file name cannot be overwritten. Sample filename: MyFirstProfile.prfs How to generate secured profiles • Export the profiles to SD Card according to the procedure in the chapter "Profile - Getting Started". • Connect the XL2 to the PC. • Select the user profile in the folder "Profile". • Manually change the ending of the user profile from "xxx.prfl" to "xxx.prfs" • Disconnect the XL2 from the PC. • Import the profiles from the SD Card • Power off/on lttp://www.second.com/

> The secured profile is displayed with a locksymbol in the profile selection window on startup of the XL2.

Capture the EQ Measurement Results

• Select the Capture & Start Tolerance Mode symbol for capturing the measurement result.



• Confirm with enter @.

9. Spectral Limits Option (Capture + Tolerances)

The Spectral Limits Option extends the function range of the XL2 with trace capturing, relative curve display and comprehensive tolerance handling for the **FFT** Analysis and the high resolution **1/12 Oct + Tol** spectral analysis.

Features:

- Capturing of multiple readings into the internal memory
- Comparing measurement results against captures with relative or absolute curve display
- Comprehensive tolerance handling
- Creating tolerance masks based on captures for passed / failed measurements



A pop-up window shows up.



- Select Capture EQ.
- Confirm with enter .

The pop-up window ... and store it to shows up.



- Select C1.
- Confirm with enter @ and set the capture name in the displayed letter selection box e.g. to **Ref**.
- \clubsuit The capture has been stored in the internal XL2 memory.



Edit Capture Manually

- Select the Capture & Start Tolerance Mode symbol.
- Confirm with enter 🕑.
- Select Manage captures.
- Confirm with enter @.
- Select Save to SD card.

She pop-up window **Save captures** shows up.

- Select the capture to be exported, e.g. C1.
- The pop-up window **Save capture C1 to** shows up.
- Confirm with Save.
- The XL2 generates the file MyCapture.txt in a new folder called Captures.
- Edit the sound level data in the capture file. You may delete any unused frequencies. All remaining frequency data has to remain the same. Any frequency changes will be discarded by the xL2.
- Load the new capture file with **Manage captures** back into the internal XL2 memory.

Compare the Live Level against Capture

- Blank the upper parameter by setting ----.
- Select the lower parameter with the rotary wheel ⁽¹⁾.
- Confirm with enter 🕘.
- Select Live and confirm with enter $\textcircled{\bullet}$.
- She pop-up window **Select reading mode** shows up.



- Select Live for displaying the relative difference.
- Confirm with enter 🕑.
- Select **Ref** and confirm with enter P.
- Set the Y-axis accordingly to show the zero-line.



The relative difference of the Live level compared to the captured reference is displayed.

(1712 Oct] III XLR 🗖) ASD 12:43 🗰
<u></u> ,-	dB 1/12 OCT
na Live-Ref 28.0	dB 🐪 2.900kHz
	T
40	
20	
- · · · · · · · · · · · · · · · · · · ·	ALL AND AL
ويعاني أنجاب بأحوا	
-50 La Lal L. Marte .	
16 63 250 1	k 4k 16kAZ
16 63 250 1	.K 4K 16KH2
40 RNGE 140 τ 0.1s ■	SET
40 RNGE 140 C 0.15	CNT 00:00:25

Start Tolerance Mode for Passed/Failed Measurements

• Select the Capture & Start Tolerance Mode symbol with the rotary wheel ⁽¹⁾.



• Confirm with enter .



A pop-up window shows up.



- Select Start tolerance mode.
- Confirm with enter 🕘.

She pop-up window **Make tolerance from** shows up.



- Select Ref to load the earlier captured reference spectrum.
- Confirm with enter 🕑.





Passed Result

The tolerance mode is active for passed/failed measurements.

The passed/failed condition is further visualized by the bicolor "limit" button and forwarded to the I/O interface of the instrument in order to drive an external alarm device such the accessory SPL Stack Light.



1 PASSED / Failed Result

- **PASSED** The actual measurement result is within the tolerance band.
- The actual measurement result is outside of the tolerance band. Exceptions are visualized in any out-of-tolerance frequency band.

2 Tolerance Bandwidth

Adjusts the tolerance bandwidth based on the captured reference spectrum in dB.

3 Level Offset

Provides a level offset in dB to move the tolerance band $\ensuremath{\mathsf{up}}\xspace/\ensuremath{\mathsf{down}}\xspace.$



Failed Result

The XL2 Analyzer compares spectral measurements against a tolerance band and visualizes exceptions 1 in every frequency band.



Select another Tolerance File

- Select the tolerance **Ref** with the rotary wheel ^(C).
- Confirm with enter .
- Select the new tolerance file from the SD Card or the internal XL2 memory.

Exit Tolerance Mode

• Select the Capture & Start Tolerance Mode symbol to exit from the tolerance mode.



- Confirm with enter @.
- Select Exit tolerance mode.
- Confirm with enter @.



Advanced Start Functions

You may start the passed/failed measurement by one of the following possibilities:

- Press the XL2 start button 🖭.
- Activate the automated level trigger in the tolerance file.
- Activate the digital input 1 via the external I/O interface; e.g. a foot switch or a PLC control.

Digital I/O Interface

Result Passed:	Output 1
Result Failed:	Output 3
Start Measurement:	Input 1

Tolerance Management

Tolerance curves can either be imported from txt-files or directly derived from captured measurements. The XL2 calculates tolerance bands based on

- Single captured results
- Manually generated txt-files on the PC
- Mean average of multiple captures
- Min/Max curves of multiple captures

Create Tolerance Files Manually on PC

- Capture a spectrum on the XL2 Analyzer.
- Enter the tolerance mode and generate a tolerance band for the capture.
- Select the Capture & Start Tolerance Mode symbol.
- Confirm with enter 🕑.
- Select **Save tol. to SD Card** to export the capture tolerance file. This file includes the reference capture and the tolerance data. The XL2 generates a txt-file in the folder "Tolerances".
- Open the capture tolerance file with a text editor or MS Excel.
- Edit the tolerance data according your requirements. You may delete any unused frequencies. All remaining frequency data has to remain the same. Any frequency changes will be discarded by the xL2.

- Store your individual tolerance file as txt-file in the XL2 folder "Tolerances".
- Start the XL2 tolerance mode and load the new tolerance file from the SD Card.

j	 All entries in the customized tolerance file need to be tab-separated. The individual tolerance file entries may be in any order; no sequential order required. Tolerance file templates and standardized tolerance files are a free download for all registered XL2 customers on the support website at http://my.nti-audio.com. For details see the chapter "Registration" in this manual.
---	--

#Unit					
dBr					
ubi					
#Mode					
HighLow					
#Columns					
Frequency	Min		Ideal	Max	
,					
#ATolerances					
			undef		
UNDEF	8	30	under		90
#ZTolerances					
UNDEF	8	30	undef		90
#BandTolerar	nces				
20	7	70	undef		80
1000		70	undef		80
1000	-	73	undef		77
4000		73	undef		77
4000		70	undef	-	80
		_			
20000	7	70	undef		80
#LevelOffset					
0					
#HideUnused	Bands				
false					
10130					
#nAllowedViolations					
0					





#Unit	dBr (relative	#ZTolerances		Defines the tolerance band for the Z-weighted wideband level (not mandatory).				
#Mode	Describes tolerance type; the XL2 allows set- ting of the tolerance type independently of the tolerance file. HighLow Utilizes upper and lower toler- ance based on ideal reference.		#Band Tolerances	 Defines the tolerance band for the individual frequencies. The ideal parameter can be set as "undef". The same frequency value on two successive rows defines a tolerance step, e.g.: 				
					Frequency	Min	Ideal	Max
	High	Utilizes only upper tolerance			100	70	75	80
		based on ideal reference.			500	70	75	80
	Low	Utilizes only lower tolerance based on ideal reference.			500	75	80	90
					1000	75	80	90
#Columns	Column headings for further tolerance data separated in different columns: Frequency - Min - Ideal - Max		#LevelOffset	All levels in the tolerance file are relative levels dBr. Thus the level offset provides the relation between the relative level and actual sound pressure level. (not mandatory)				
#ATolerances	Defines the A-weighted		Th	The XL2 allows setting of the level offset independently of the tolerance file.				
#CTolerances	Defines the tolerance band for the C-weighted wideband level (not mandatory).							
Capture + Tolerances



#Hide UnusedBands	•	eter defines which tolerance fre- ge is shown. (not mandatory)	#FreqScale Spacing		e frequency scale spacing in the X- e tolerance data. (not mandatory)
	True	Measurement results of fre- quency bands without any specified tolerance are not dis- played.		lin	Tolerance data displayed with linear frequency spacing, e.g. used in the FFT function. (de- fault setting)
	False	The measurement result of all frequency bands are shown.		log	Tolerance data displayed with logarithmic frequency spacing e.g. used in the 1/12 Oct + Tol
#nAllowed Violations		PASSED result with a maximum n failed frequency bands (not man-			function.
	datory)		#AutoStart	Enables au	uto start function. (not mandatory)
				True	Auto start is active. The run indi- cation displays A . The measure- ment starts automatically after the trigger level is reached.

145

Alternatively the measurement may start manually by pressing

No auto start function. (default

start 🕨

setting)

False





#AutoStartTrig-Sets the sound level in dBZ for an automated gerLevel level trigger; the passed/failed measurement starts as soon as the set level e.g. 95 dB is reached or exceeded. The actual input level can be monitored at the input range bar. (mandatory if #AutoStart is True used otherwise not mandatory)

#AutoStartSet-Defines the settling time, after the auto start

- tlingTime trigger level is exceeded, until the start of the measurement; the time is set in 100 millisecond steps e.g. setting **0.5** equals a settling time of 0.5 seconds. If the input level drops below the auto start trigger level during this period, no measurement is carried out. (not mandatory)
- **#MeasTime** Defines the measurement time in 100 millisecond steps; e.g. setting **1.5** equals a measurement period of 1.5 seconds. (not mandatory)

System Settings

Display Contrast Setting

• Hold down escape esc and turn the rotary wheel ⁽¹⁾ until the desired contrast is reached.

2 Page Selector

Toggles with the page button 🕩 between

- Set System setting
- Inf System information
- Sch Scheduler

3 Auto power off

If no button has been pressed for the defined time period, then the instrument switches off.

- Select Auto power off with the rotary wheel ⁽²⁾.
- Confirm with enter 🕑.
- Set the auto power off time with the rotary wheel ⁽¹⁾.
- Confirm with enter $\textcircled{\bullet}$.

The auto power off mode is disabled when a measurement is running or when an external mains power supply is used.

10. System Settings

The various system settings of the instrument can be adjusted as follows: Turn the rotary wheel O to navigate to the main menu (1) and select **System**. Confirm with enter O.

System Settings Page







System Settings

4 Backlight

Toggle the backlight with the power/backlight button B. The backlight offers three different conditions:

- On
- Dimmed
- Off

Auto On/Off The backlight switches off if no key is pressed for 2 minutes.

AutoThe backlight dimms if no key is pressedOn/Dimmedfor 2 minutes.

- Manual The backlight can be toggled between On, Dimmed and Off.
- Navigate the cursor to the backlight function.
- Toggle with enter 🕑 to set this parameter.

Dimming the backlight reduces the power consumption.

5 Date (yyyy:mm:dd) and Time (hh:mm:ss)

The XL2 includes a real-time clock. All measurements are logged with a date and time stamp.

Set the real-time clock as follows:

- Select the **Date** function 5 with the rotary wheel **O**.
- Confirm with enter 🕑.
- Set the date with the rotary wheel ⁽¹⁾.
- Confirm each setting with enter e.
- Follow the same procedure for Time.

You have set the real-time clock.

6 Show Profile List

You may configure the XL2 to start up with individual preset application profiles. For example, sound level monitoring in accordance with DIN 15905 or SLV 2007.

- **Yes** The instrument starts up with default application profiles. Further details about the application profiles are in the chapter "Profiles".
- **No** The instrument starts up in the last configuration without any application profiles.

149

System Settings



7 Phantom Power

You can select if the phantom power should be activated or disabled when the XL2 is powered up.

- Off The phantom power is switched off when the XL2 is powered up. Useful, for example, when it is common to measure lines.
- Remember The XL2 remembers the setting prior to the last shutdown. (= default setting)

8 Speaker

Enables/disables the rear speaker. For example, it is recommended that the rear speaker be disabled for all acoustic measurements. This prevents measuring the audio feedback sound loop generated through the rear speaker.

9 USB Mode

The following setting defines how the PC recognizes the connected $\ensuremath{\mathsf{XL2}}\xspace$:

Ask on connect	Select either the Mass storage or COM port mode after connecting the XL2 to the PC.
Mass storage	The PC recognizes the XL2 as a mass storage device automatically, allowing you to down-load XL2 measurement reports.
COM port	The PC recognizes the XL2 as a COM port au- tomatically for application of the XL2 Projec- tor and the Remote Measurement Option.

10 Decimal Separator

Set the decimal separator to match the generated measurement reports to the PC settings. This simplifies data post-processing on the PC.

- Navigate to the **Dec. Separator** function (8) with the rotary wheel (9).
- Toggle between " . " and " , " with enter 🕑 to set this parameter.

The decimal separator is set.



System Settings

System Information Page

1) 2)	··Firmware: Serial number:	V2 . 50 A2A-03199-D2
3	·· Options:	
	-,	Installed Installed Installed Installed Installed

11 Firmware

Displays of the firmware version number (1) (to update the firmware to the latest version see the chapter "Firmware Update of XL2").

(12) Serial Number

Displays the instrument's serial number 12.

13 Options

Displays the options installed on the XL2.

Every installed option may be temporarily **Hidden** to generate a profile for other XL2s without this option installed.

- Select the installed option.
- Toggle with enter to set this parameter.

Scheduler



11. Scheduler

The XL2 Analyzer offers automated scheduled measurements without any human interaction. You may schedule a measurement for one or multiple days.

XL2 Actions after pressing Start Schedule

- Selects the preset measurement function.
- Standby until start time
- Starts measurement at start time
- Stops measurement after the preset measurement duration
- Stores measurement data to SD Card (no reconfirmation required)
- Stops schedule or standby until next start time.

Supported measurement functions:

- SLMeter/RTA
- FFT + Tol
- RT60
- 1/12 Oct + Tol
- STIPA





Scheduler

1 Function

Select the measurement functions for the scheduled measurements.

2 Start Date and Time

Set the date for the scheduled measurement(s):

- Select Start with the rotary wheel ⁽¹⁾.
- Confirm with enter @.
- Set the date with the rotary wheel ⁽¹⁾.
- Confirm each setting with enter e.

Set the time for the scheduled measurement:

- Select the **T** symbol to the right of the date with the rotary wheel **O**.
- Confirm with enter @.
- Set the time with the rotary wheel ⁽²⁾.
- Confirm each setting with enter e.

3 Duration

Set the duration of the scheduled measurement(s).

4 Recurrence

- Off Schedules a single measurement.
- **On** Schedules a sequence of measurements with the same start and end time times for the selected week days

5 Days

Select the week days for the recurrence

6 End

Set the number of occurrences of the scheduled measurement. Each single measurement is one occurrence.



7 Start Schedule

Starts the scheduled measurement(s).

The **Scheduler active** window is displayed.



Scheduling - Getting Started

After starting the scheduled measurement, the XL2 waits for the next scheduled start date and time. The instrument buttons are disabled.

Power Supply

Scheduler

It is recommended that the instrument remains powered on until the next start time, using the accessory mains power adapter. If the XL2 is powered off and on again during this time, then the XL2 Analyzer restarts in the **Scheduler** function and continuous the countdown. This applies also for any interruption of mains power supply.

Stopping the Schedule

Press and hold the escape button esc to stop the schedule.

Operation with PC

Starting the schedule switches the USB interface automatically into the COM port mode for remote measurement applications or using the XL2 Projector software.



The XL2 stores all acquired measurement data including realtime information onto the removable SD Card. Additionally, wav-files might be recorded and individual voice notes added for a complete documentation of the measurement.

- Carry out a sound level measurement.
- Select the memory menu 🖃 in the upper menu bar.
- Press enter 🕑. The memory menu opens.



1) Create New Project Folder

Add a new folder with an individual project name. The maximum length of the project name is 16 characters. A fast way for creating new folder names is

- Select an existing folder name with **DIR**
- Select NEW
- Rename the folder and confirm with start **I**, thereby creating a new folder name.

2 Select Project Folder

An overview of all existing project folders is shown. Select a folder to store the results of the measurements.

③ Project Folder Name

All measurements are stored in this project folder.

4 Naming+Saving

Setting of auto or manual data file naming and saving

- auto The results are named and stored automatically in the selected project folder.
- **manual** The question mark shows up in the upper menu bar (a) upon the completed measurement. You can choose to individually name and store each single measurement.

5 Save Test

Stores the measurement in the selected project folder.

6 Save Screenshot

Stores the XL2 screen in the selected project folder.

7 Load Test

Loads back previously stored result data from the SD Card for detailed examination on the instrument screen.

8 Append Mode

The append mode stores one or more results into the same data file, simplifying data analysis and handling on the PC.

Application example:

The speech intelligibility STIPA shall be measured at various locations in the same venue. The append mode allows you to store all individual measurement results into one single data file.

9 Free Memory

Displays the remaining available memory on the SD Card.

Data Management



(10) Write Cache (available with Extended Acoustic Pack)

A low **Write Cache** level indicates a good performing SD Card. The optional Extended Acoustic Pack with simultaneous 100ms logging and linear wav-file recording result in a high data volume written onto the SD Card. Some non-original SD Cards may cause **OVLD** performance, thus loose some measurement data. Verify the **Write Cache** of your SD Card during your measurements. Subsequently an original SD Card shall be used. We recommend SD Cards of the manufacturers SanDisk and Transcend.



Original SD Cards ensure that all measurement data and audio wav-files are stored in good order on the SD Card. Non-original SD Cards may lead to low performance or error messages. Order Information: Spare SD Card, NTi Audio # 600 000 374



In case the SD Card is full, you can insert the original spare SD Card into the XL2 and continue with the measurements. The XL2 generates the projects and system folders automatically.



Memory Structure of SD Card



1 Projects

The folder **Projects** contains subfolders with the stored measurement results. The default subfolder name is "**MyProject**". You can create further individual subfolders.

2 XL2.htm

This file opens the XL2 instrument status page, with

- Serial Number, Firmware
- Installed Options

as well serves for online

- Firmware updates
- Activation of options

3 XL2_SYS

The XL2 system folder contains a file with serial number, firmware version and installed options.

Measurement Data - Getting Started

Enable Data Logging

• Select the **Log** page in the sound level meter function and set the logging parameters; e.g., as follows:

SLMeter Log XL	R 🗖 🗍 🕸 ASD 12:06 🖷
Logging	0n
Interval dt: Add Spectra:	00:00:0: No
Log Audio:	0n
Format:	Compressed+AGC
Log Values:	Selected
Log Values: • L _{Aeg}	Selected 5
	Selected s 6
^D L _{fleq}	Selected 5 6 7
[©] L _{Aeq} 1 L _{AFmax}	Selected 5 6 1 8

The data logging is enabled. The sound pressure level is logged each second.



Select Parameters for Measurement Report

• Select the **Rep** page in the sound level meter function and set the parameters as follows:

SLMeter Rep XLR	R 🔳 🛛 ASD 13:24 🖷
Report Setting	
Add Spectra:	AU
Report Values:	Selected
Report Values: © LReg	Selected s
Report Values: © LAeq 1 LCPKma×	Selected 5 6
□ L _{Req}	Selected 5 6 1
¤ L _{Req} 1 LCPKma×	Selected 5 6 1 8

Complete the Measurement

- Press page Ӣ to return to the numeric result page.
- Press start 🖭. The sound level measurement is now running.
- Press stop 🔎 to complete the sound level measurement.
- The memory question mark **n** shows up in the upper menu bar after the completed measurement.



Open the Memory Menu

• Select the memory menu 🖃 with the rotary wheel 🕲 and confirm with enter 🕑.

The memory menu opens with "Save Test" pre-selected.



- Press enter 🕑 to confirm Save Test.
- The pop-up window **Save Test** shows up.

Select File Name



 Select the user part of the file name. The maximum length of the user part is 12 characters. The right part "_SLM_001" is defined by the XL2 to avoid overwriting of existing measurements, whereby "SLM" is the appendix for the measurement function and "001" is an automatically-incrementing number.





- Select **SAVE** and confirm with enter .
- The measurement data including test configurations and all log files are stored on the SD Card.

Overwrite File

Tick off **Automatic Numbering** to store actual measurement results for later overwriting with new results.

Record Voice Notes

Voice notes are recorded speech annotations for each stored measurement. The XL2 utilizes the internal voice note microphone for the recording.

• Select **Record Voicenote** and press enter



- Select **REC** and press enter €.
- Record the voice note and press enter e to finish.

The voice note has been recorded.





Save Screenshot

• Select the memory menu 🔳 with the rotary wheel 🕲 and confirm with enter 🕑.

The memory menu opens.

- Select Save Screenshot and press enter •.
- Select the file name and confirm **SAVE** with enter $\textcircled{\Theta}$.



🖞 The screenshot is saved to the SD Card.

Name + Save Measurement Data Automatically

You may configure the XL2 to store results automatically to the SD Card upon completion of the measurement, by applying the **Naming+Saving: auto** function. This is particularly useful for longer measurements, such as sound level monitoring spanning several hours. The default file name is the actual date.

• Select Naming+Saving in the memory menu and press enter 🕑 to confirm Naming+Saving: auto.





Measurement Data Logging:

If the battery runs empty while you are measuring with logging and/or repeated-reporting enabled, the XL2 creates the folder RESTORE_AF-TER_POWERFAIL and stores the measurement data up until the switch-off time. The next time you switch on the XL2, the following window informs you that this data exists. Click OK:



The files may contain invalid data at the end of the file, which you may delete to create a valid report. **View Project Folder on the PC**

- Connect the XL2 with the USB cable to the PC.
- The pop-up window **USB Mode** shows up.



• Select Mass Storage and confirm with enter @.



The PC recognizes the XL2 as a mass storage device. The following SD Card content is shown:



• Open the folder "Projects" and the subfolder "MyProject".

🖇 All stored measurement data files are shown.



Overview of Measurement Data Files



1 Test System File

System file for use by XL2 only. It contains the measurement data and test configuration of the last measurement cycle for later reload and view of measurement results in the XL2.

2 Sound Level Log File

The XL2 logs sound levels in pre-defined log time intervals. For more details please see the chapter "Sound Level Meter" -> "Logging".

Transfer Measurement Data to PC

• Drag and drop the required data files to the PC.

Card Reader

Alternatively the SD Card can be inserted into a card reader. This offers a faster data transfer from the SD Card to the PC.

Additional Data Files using Repeat Mode For sound level measurements with timer mode Repeat or Repeat Synchronized the following additional data files are generated:

- MyTest_SLM_000_123_Report_Rep.txt
- MyTest_SLM_000_RTA_Report_Rep.txt

Common xxx_Report.txt files contain the measurement data of the last measurement cycle. The xxx_Report_Rep.txt file contains the results of all measurement cycles in the Repeat or Repeat Synchronized mode.

Microsoft Excel

The generated .txt files are best viewed on the PC using "Open with" -> Microsoft Excel.

③ Measurement Report File

Final report after the completed measurement. For more details please see the chapter "Sound Level Meter" -> "Reporting".

4 Wav-File

The wav-file is the recorded audio file. The index "FS133.0dB(PK)" in file name represents the full scale peak level of the recorded wav-file. For more details please see the chapter "Sound Level Meter" -> "Wav-File Recording."

5 RTA Log File

The XL2 logs the detailed real-time analyzer spectra in predefined time intervals. For more details please see the chapter "Sound Level Meter" -> "Logging".

6 Voice Note

A voice annotation can be recorded for each stored measurement or screenshot. For more details please see the chapter "Data Management" -> "Voice Notes".

7 Screenshot

Picture of the actual XL2 display.







Log File Format

File name: e.g. MyTest_SLM_000_123_Log.txt

XL2 Broadband Logging)	Ν	/lyProjects\My	Test_SLM_000	_123_Log.txt	
# Hardware Configuratio Device Info: Mic Type: Mic Sensitivity:	XL2, SNo. A2	A-02673-D1, F 260, S/N: 1486		ted 2011-04-0	5 13:56	
# Measurement Setup						
Timer mode: Timer set: Log-Interval: k1: k2: kset Date: Range:	continuous :: 00:00:01 0.0 dB 0.0 dB k-Values not r 30 - 130 dB	neasured				
# Time						
Start:	2011-05-15, 1	7:44:06				
End:	2011-05-15, 1	7:44:16				
# Broadband LOG Resu	ilts					
Date	Time	Timer	LAeq_dt	LAeq	LAFmax_dt	LCPKmax_
[YYYY-MM-DD]	[hh:mm:ss]	[hh:mm:ss]	[dB]	[dB]	[dB]	[dB]
15.11.2010	17:44:07	00:00:01	97.0	97.0	102.4	119.0
15.11.2010	17:44:08	00:00:02	85.8	94.3	91.9	105.7
15.11.2010	17:44:09	00:00:03	73.8	92.5	85.2	102.4
15.11.2010	17:44:10	00:00:04	79.0	91.4	85.3	103.9
15.11.2010	17:44:11	00:00:05	72.6	90.4	75.9	94.8
15.11.2010	17:44:12	00:00:06	67.3	89.6	71.4	87.3
15.11.2010	17:44:13	00:00:07	91.2	89.9	95.3	112.3
15.11.2010	17:44:14	00:00:08	92.7	90.4	97.1	113.2
15.11.2010	17:44:15	00:00:09	79.3	89.9	81.6	97.9



Level LAeq_dt "delta t"

Level of actual logging interval. e.g. Logging interval = 1 second, then $LAeq_dt$ is the time averaged level of the past 1 second

Report File Format

File name: e.g. MyTest_SLM_000_123_Report.txt

XL2 S	ound Level Meter E	Broadband Rep	orting MyPro	ojects\MyTest_	SLM_000_12	3_Report.txt
# Harc	dware Configuration	ı				
	Device Info:	XL2, SNo. A2	A-02673-D1, FW2	.20		
	Mic Type: NTi Audio M4260, S/N		260, S/N: 1486, U	ser calibrated	2010-11-05 1	3:56
	Mic Sensitivity:	27.3 mV/Pa				
# Mea	surement Setup					
	Append mode:	OFF				
	Timer mode:	continuous				
	Timer set:	:				
	k1:	0.0 dB				
	k2:	0.0 dB				
	kset Date:	k-Values not r	neasured			
	Range:	30 - 130 dB				
# Broa	dband Results					
	Start		Stop			
	Date	Time	Date	Time	LAeq	LCPKmax
	[YYYY-MM-DD]	[hh:mm:ss]	[YYYY-MM-DD]	[hh:mm:ss]	[dB]	[dB]
	15.11.2010	17:44:06	15.11.2010	17:44:16	89.3	119.0

Data Post-Processing

A Microsoft Excel form generating an automated sound level measurement report and chart is available as a free download for all registered XL2 customers on the support website at http://my.nti-audio.com. (Enable all macros when opening the document.)



Recall Measurements

The straight-forward but powerful **Load Test** and **Save Test** functions serve the dual purpose of conveniently retrieving standard test setups for repeat testing and also of saving stored result data for later recall and examination on screen.

The following measurement functions support the recall function:

- SLMeter, sound level meter
- FFT Analysis
- RT60, reverberation time
- STIPA, speech intelligibility measurement

Load Test

- Select the memory menu 🔳 with the rotary wheel ⁽¹⁾ and confirm with enter ⁽²⁾.
- The memory menu opens.





Select File Name

- Select Load Test and confirm with enter .
- The pop-up window **Load Test** shows up.

(SLMeter 123 XLR 🗖 4 ASD 13:24 🚛
LOAD TEST
FOLDER: DIR MyProject
File: DIR MyTest_SLM_000
30 RNGE 130 SET:: cnt 00:00:10

- Select the project folder and stored data file name with **DIR**.
- Select **LOAD**, confirm with enter

View Measurement Data

🌢 The previously stored measurement data is displayed.





Start the Append Mode

- Select SLMeter and preselect the sound level LAea.
- Select the memory menu 🖻 with the rotary wheel 🎱 and confirm with enter 🕘.

🌢 The memory menu opens.

• Select Start Append Mode and confirm with enter @.



The Start Append Mode window opens.

Append Measurements

The append mode stores the results of several individual measurements into the same data file, thus simplifying measurement result analysis and data handling on the PC.

The following measurement functions support the append mode:

- SLMeter, sound level meter
- STIPA, speech intelligibility measurement

Application example:

The sound level L_{Aeq} shall be measured at various locations in the same venue hall using a pink noise test signal. The append mode allows to store all individual measurement results into one single data file.



Backup the stored measurement data from the XL2 to the PC daily. This prevents any data loss if data has been deleted from the SD Card by mistake.



Select Filename

·	123 XLR B 4 ASD 13:24
L _{Reg}	MyProject
Start Ap	opend Mode
MyTest ⊠ Aut	ENAHE SEARCH SLM_000 omatic Numbering pend is: OFF
	TART CANCEL
LAFma×	SET 00:00:10

- Select the user part of the file name. The maximum length of the user part is 12 characters. The right part "_SLM_001" is defined by the XL2 to avoid overwriting of existing measurements, whereby "SLM" is the appendix for the measurement function and "001" is an automatically-incrementing number.
- Select **START** and confirm with enter P.

The append mode is available.

Take Measurement Results

- Measure the sound level L_{Aeq} on the first position.
- Select the memory menu and Append Data.

(SLM	leter 123 XLR ॖ ?) 4 ASD 13:2	4
(FOLDER:	
Lfle	MyProject	
	File:	
	MyTest_SLM_000_00	
	Auto Append is OFF	
	Append Data	
	Exit Append	
Lon	FREE: 1874.2MByte	
Laf	03.30	JD
LAF	max 103.3	d₿
 30 R	SET 00:0	01:00



Append Further Measurement Results

• Move to the next location in the room, carry out the measurement and follow the described append data procedure.

Exit Append Mode

- Open the memory menu .
- Select **Exit Append** after adding the last measurement.

Store Measurement Results

- The pop-up window **Append Data** shows up.
- Select the location ID and measurement ID, thus each measurement result is stored with an individual description for later documentation.
- Select **Append** with the rotary wheel **(2)** and confirm with enter 🕢

46.6dB LAF 99.0dB LAFmax SET 00:00:10 30 RNGE 130

The measurement data is stored on the SD Card 6

Auto Append

Alternatively the XL2 can append measurement data to the same report file automatically.

• Start the Append mode and set Auto Append is ON in the memory menu or in the pop-up window Start Append Mode.

Skip Append Data

If you select SKIP in the Append Data window, then no measurement data is appended to the report file.







Append After Power Up

You can continue to add measurement results into the same data file at any time later.

- Store the measurements in the append mode and switch the XL2 off and on again (3).
- She pop-up window **Continue append** shows up.



• Select **Yes** and continue adding measurement results to the same report file in the append mode.

Recall Previous Measurements for Append

Any previously stored measurement with the remark **for append** can be recalled and further test records appended.

- Select Load Test and press enter .
- Select the project folder and stored data file name.
- Select **LOAD** confirm with enter **e**.

The pop-up window **Load for Append** shows up.



• Select **Continue** and add further measurements to the same report file in the append mode.

171

13. XL2 Projector

The XL2 Projector displays the XL2 screen in real-time on your PC when connected by USB. The software toolbar enables control of the XL2 Audio and Acoustic Analyzer with the mouse and keyboard. XL2 Projector software is a free download for all registered XL2 customers on the support website at http://my.nti-audio.com.

The XL2 Projector background color synchronizes with the XL2 limit button color $\textcircled{\mbox{\footnotesize \mbox{\footnotesize em}}}.$

USB Communication The XL2 Projector uses the USB connection in COM port mode. Installation instruction:

- Disconnect the XL2 Analyzer from the PC.
- Register your XL2 and download the XL2 Projector software from the XL2 Support Page on http://my.nti-audio.com.
- Unzip the file "XL2 Projector Setup Vxx.zip" on your PC.
- Start the software installation and follow the instructions including driver installation.
- Start the XL2 Projector software.
- Start the XL2 Analyzer and connect it to the PC.
- She pop-up window **USB Mode** shows up.





XL2 Projector





- Select **COM port** on the XL2. Microsoft Windows may recognize the new hardware automatically and start the hardware installation assistant.
- If prompted, select "No connection to Windows Update" and continue with the installation.
- Complete the installation.

SL2 Projector displays the instrument screen in real-time.

Features

- The upper window bar displays the XL2 serial number.
- The menu bar supports remote instrument operation of the XL2 Analyzer from your PC.
- All data stored on the SD Card can be accessed directly from the XL2 Projector software. Complete your measurements and click on the SD Card symbol.

For more details and features click the **?** symbol on the XL2 Projector Software.



14. Monitoring Network

Multiple XL2 Analyzers connect with USB Device Servers to a network for online monitoring of noise or other signals. An instance of the Projector Software may be started for each XL2 Analyzer, with all XL2 screens being simultaneously displayed on the PC monitor. The Projector Software offers the control of each XL2 Analyzer from your PC.



Please contact NTi Audio for application details.

XL2 Projector



15. Remote Measurement

The XL2 Analyzer offers an optional remote measurement function. This allows querying the XL2 measurement data online via the USB interface. Thus customers may program their own measurement application on the computer, e.g. for sound level monitoring or automated measurement tasks. The supported XL2 measurement functions are

- Sound level meter and spectrum analyzer SLMeter/RTA
- Audio analyzer RMS/THD+N

The commands are send in ASCII format through the virtual COM port to the XL2 Audio and Acoustic Analyzer.

Example:

- Command set send to XL2 Analyzer: INIT START MEAS:INIT MEAS:SLM:123? LAF
- Result returned to PC: 53.8 dB,OK

For more details you may download the individual Remote Measurement Reference Manual at www.nti-audio.com/XL2.

Order information NTi Audio #: 600 000 339



16.Microphones

NTi Audio Microphone Overview



The plug-on measurement microphones combined with the XL2 form a comprehensive sound level meter and acoustic analyzer. The microphones are 48 VDC phantom-powered and include an electronic data sheet.

Integrated Preamplifier

The microphone bodies contain a pre-amplifier and require 48 VDC phantom power supply for operation. They combine high dynamic range and wide frequency range with low noise. The measurement microphones can also be connected with an ASD Cable to the XL2 Audio and Acoustic Analyzer for measurements at remote locations or for reduction of acoustic reflections.

Electronic Data Sheet

The NTi Audio's ASD measurement microphones and the microphone preamplifier MA220 include an electronic data sheet. The automated sensor detection (ASD) of the XL2 Analyzer reads this data sheet, enabling the XL2 to automatically recognize the microphone model, sensitivity and calibration data. This ensures correct measurement results.



Connection diagram of measurement microphones with electronic data sheet

Microphone plugs directly in to the XL2

The XL2 automatically reads the ASD electronic data sheet of the connected microphone as follows:

- Connect the measurement microphone to the XL2.
- Switch on the XL2 3.
- The XL2 reads the electronic data sheet of the connected microphone during a brief initialization process prior to the first measurement.



Microphones

Microphone Connection via the ASD Cable

The NTi Audio measurement microphones and the microphone preamplifier MA220 can be connected with an ASD Cable to the XL2 Audio and Acoustic Analyzer for measurements at remote locations or for reducing acoustic reflections. The ASD Cable includes a separate line for transmitting the electronic data sheet to the XL2, thereby not affecting the measurement data.

The electronic data sheet is transmitted via the XLR connector's housing and the shield lead of the ASD Cable. Do not touch these parts during the brief initialization period to ensure the complete data sheet is recognized by the XL2. The automated sensor detection does not disturb any measurements. You may join up to 4x ASD Cables together in series. The ASD technology supports accurate data communication up to a combined cable length of 20 meters (= 65 feet).

Microphone Connection via a professional Audio Cable For distances longer than 20 meter (= 65 feet) use a high quality, low capacitance standard professional audio cable. In this scenario, no separate data line for the transmission of the electronic data sheet is available, thus the microphone sensitivity has to be entered manually into the XL2 Analyzer.

- Use the microphone for the intended purpose only.
- Protect the microphone from contamination by always using the supplied windscreen.
- Never use the microphone in a damp or wet environment.
- Do not jar or drop the microphone.
- Do not remove the microphone protective grid.
- Do not touch the microphone membrane.
- Remove the black dust cap of the 1/2" measurement microphones prior to use.

177

Further Information



17. Further Information

Registration

Register as a customer with NTi Audio and benefit from the following possibilities:

- Keep your product up-to-date Access free firmware and software updates.
- Activate options Enable additional functions for your product.
- Access premium content Access downloads, information and specific support for your product.
- Receive application and product news Sign in for the NTi Audio Newsletter.
- Get fast worldwide support Register your product for fast support.

• Confirm your ownership

Allows us to contact you with important product notifications and monitor your product in case of loss or theft. How to Register

- Open the web page "http://my.nti-audio.com".
- You are prompted to login or create your My NTi Audio Account.
- The web page "My NTi Audio Products" opens.
- Select the product type and enter the serial number of your product.
- Confirm with "Register".
- Now your product is listed in the table "My Products".
- Congratulations, your NTi Audio product is registered.



Tips and Troubleshooting

Overview

- Resetting to Factory Default
- XL2 Starts Up with Limited Functions
- SD Card Errors
- Can I use another SD Card?
- Stored Data or Wav-files not available on SD Card

Resetting to Factory Default

If the XL2 Audio and Acoustic Analyzer reacts unexpectedly, a reset to the factory settings might solve the problem.

- Switch off the XL2 3.
- Hold down escape esc and simultaneously operate On/Off
- The reset confirmation is displayed.

XL2 Starts Up with Limited Functions

The XL2 has been operated the last time in one of the simplified application profiles and the system settings have been changed as follows: **Select Profile** from **Yes** to **No**.

- Follow the above resetting to factory default.
- The XL2 will start up with full functionality.

SD Card Errors

The XL2 Audio and Acoustic Analyzer writes measurement data automatically onto the SD Card during ongoing measurements, thus a functioning SD Card has to be inserted at all times.

Error Messages	Actions to do
Missing SD-Card	Insert the SD Card.
SD-Card is not FAT formatted	Format the SD Card on the PC.
SD-Card is full	The memory of the SD Card is full. Download all data to the PC and emp- ty the SD Card.

179

Further Information



Can I use another SD Card?

Yes, you can use any alternative SD Card, such as offered in the chapter "Further Information: Accessories".

- Switch off the XL2 🕘.
- Insert the SD Card into the XL2 Analyzer.
- Switch on the XL2 🖗.
- The XL2 writes the data structure automatically to the inserted SD Card.

Stored Data or Wav-files not available on SD Card The file system might be corrupted.

- On the PC, right-click on the "NTi Audio XL2" drive, select "Properties" and click "Check Now" as shown below.
- Thereafter all stored data should be available again.

eneral A	AutoPlay Tools Hardware Sharing
Error-ch	ecking
e	This option will check the volume for errors.
	Check Now
Defragm	nentation
6	This option will defragment files on the volume.
	Defragment Now
Backup	
	This option will back up files on the volume.
	Backup Now
	Backup Now



Firmware Update

You will find information about the installed firmware version in the systems settings of the instrument. The firmware revision history is listed on the XL2 support page http://my.nti-audio.com.

Updating the firmware on the XL2 is executed by placing the XL2Vxxx.xx file in the root directory of the XL2. When started, the XL2 automatically completes the firmware update.



1 XL2.htm

This file opens the XL2 instrument status page, for online

- Firmware updates
- Activation of options

Firmware update with XL2 in hand, PC online:

- Start the XL2 and connect it to the PC.
- The XL2 displays the pop-up window USB Mode.
- Select Mass storage. The PC thus recognizes the XL2 as a mass storage device.
- Double-click the file xl2.htm 1. The web screen "XL2 Instrument Status" opens.
- Select "Look for FW Updates". The web screen "XL2 Support Page" opens.
- Compare the firmware version to the latest version available.
- If an update is required, download and save the firmware file XL2Vxxx.xx into the XL2 root directory.
- Remove the USB cable and power up the XL2 3.
- Watch the display and wait until the update is finished.

No PC available

If you have no web access at all, please contact the local NTi Audio partner. You will receive the new firmware file.
181

Further Information



Options and Accessories

The $\ensuremath{\text{System}}$ menu shows the installed options on the XL2. The following options extend the measurement functions of the XL2:

Speech Intelligibility STIPA NTi Audio #: 600 000 338

The XL2 Analyzer measures the speech intelligibility according to the latest revision of standard IEC 60268-16:2011 (edition 4) and older editions. It includes ambient noise correction and automated averaging of measurements. The XL2 displays the measurement results as STI or as CIS results, accompanied by the individual levels and modulation indices of the seven octave bands.

The measurement results are acquired from the dedicated STIPA test signal source:

- NTi Audio TalkBox, acoustic signal generator required for audio systems with voice microphones, thus measuring the complete signal chain
- Minirator MR-PRO, test signal generator required for audio systems with line inputs
- STIPA Test CD (included)

Extended Acoustic Pack NTi Audio #: 600 000 339

The Extended Acoustic Pack offers the following additional features for sound level- and acoustic measurements.

- SLMeter/RTA function
 - Recording of linear wav-files (24 bit, 48 kHz)
 - Percentiles for wide band and spectrum: LAF1%, LAF5%, LAF10%, LAF50%, LAF90%, LAF95%, LAF99%
 - Sound Exposure Level LAE
 - 100ms logging
 - Event-triggered audio and data recording
 - Time weighting: Impulse (LxI, LxIeq with x= A, C, Z)
 - True peak level in 1/1 and 1/3 octave resolution
 - Clock-Impulse Maximum Level (TaktMax) and values as specified in DIN 45645-1
- FFT function
 - High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz
- RT60 function
 - Reverberation time RT60 in 1/3 octave resolution



Remote Measurement Option NTi Audio #: 600 000 375

The Remote Measurement Option allows querying of the XL2 measurement data online via the USB interface. Thus customers may program their own measurement application on the computer, e.g. for sound level monitoring or automated measurement tasks.

The supported XL2 measurement functions are

- Sound level meter and spectrum analyzer SLMeter/RTA
- Audio analyzer RMS/THD+N

Spectral Limits Option NTi Audio #: 600 000 376

The Spectral Limits Option extends the XL2 function range with trace capturing, relative curve display and comprehensive tolerance handling for the FFT Analysis and the new high resolution RTA with spectral resolutions up to 1/12th octave.

- FFT and 1/12 octave function
 - Capturing of multiple readings into the internal memory
 - Comparing measurement results against captures with relative or absolute curve display
 - Comprehensive tolerance handling with tolerance masks based on captures for passed/failed measurements
 - Export and import of tolerance and capture files
- 1/12 octave function
 - High resolution RTA function 1/12 Oct + Tol
 - Selectable 1/1, 1/3, 1/6 and 1/12 octave resolution
 - Frequency band listening at rear speaker
- FFT function

High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz

• SLMeter/RTA function

True peak level in 1/1 and 1/3 octave resolution

Cinema Meter Option NTi Audio #: 600 000 379

The Cinema Meter Option forms the dedicated solution for efficient calibration and repetitive verification of cinema loudspeaker systems according SMPTE ST 202:2010 and SMPTE RP 200:2012. An interactive assistant guides the user through dedicated measurement procedures.



• Create new cinema

Generates measurement templates according the cinema size with dedicated X-curve selection.

• Calibrate cinema

Calibration menu for reference data recording of each channel as well as headroom tests and averaging measurement results of different microphone positions

• Verify Cinema

Verification menu for periodical cinema measurements and comparison against the reference data

• Lock into Verify Mode

Locks the XL2 Analyzer in the Verify Cinema mode. After powering off the instruments starts up directly in the Verify Mode again. The user may unlock the instrument anytime.

• View verification results

Displays measurement deviations of periodical measurement against reference data.

• Exit

Returns back to other XL2 measurement functions.

The Cinema Meter Option includes the Spectral Limits Option. In case you have the Spectral Limits Options already installed, then just go for the Cinema Assistant Option, NTi Audio #: 600 000 378, which will provide you the Cinema Meter functionality in cooperation with the previous installed Spectral Limits Option.



Cinema Meter | STEP 00 | | ASD 13:31 |





Installation of Options

Activation with PC online (recommended):

- Start the XL2 and connect it to the PC.
- The XL2 displays the pop-up window USB Mode.
- Select **Mass storage**. The PC thus recognizes the XL2 as a mass storage device.
- Double-click the file xl2.htm 1. The web screen "XL2 Instrument Status" opens.
- Select "Activate Option". The web page "http://my.nti-audio. com" opens.
- You may login or create your My NTi Audio Account.
- The web page "My Products" opens. The XL2 Analyzer is listed with its Serial Number on this page.
- Enter the license number printed on the rear side of the CD cover and click "Get Activation Key".
- The activation key file is then available for download.
- Copy this activation key file into the root directory of the XL2 memory (=SD Card).
- Remove the USB cable and power up the XL2 ③.
- Congratulations, the purchased option is now permanently enabled on the instrument.



Manual Activation:

(applicable if the XL2 cannot be connected to an online PC)

- Open the web site "http://my.nti-audio.com".
- You may login or create the My NTi Audio Account.
- The web page "My Products" opens. Register the XL2 by choosing Product Name "XL2 Analyzer" and entering the unique Serial Number (e.g., A2A-xxxx-xx), listed in the XL2 battery compartment or shown in the XL2 system settings.
- Click "Register". The XL2 Analyzer is now listed with its Serial Number on this page.
- Enter the license number printed on the rear side of the CD cover and click "Get Activation Key".
- The activation key file is then available for download.
- Copy this activation key file into the root directory of the XL2 memory (=SD Card).
- Remove the USB cable and power up the XL2 ④.
- Congratulations, the option is now enabled on the XL2.



Kindly use the license number on the back of the CD cover. This license number is valid for one XL2 Analyzer.



Accessories



Ever-ready Pouch

The Ever-ready Pouch protects the XL2 during transport and operation. With its convenient belt-clip, the XL2 can be kept close-by for those tasks requiring both hands. The Ever-ready Pouch allows operation of the XL2 while fitted in the pouch. NTi Audio #: 600 000 335

Mains Power Adapter

Mains Power Adapter the XL2 Audio and Acoustic Analyzer with removable plug types. The Mains Power Adapter suits the typical power sockets in Australia, China, Europe, Japan, US and UK. NTi Audio #: 600 000 333

Spare Li-Po Battery

Rechargeable spare battery for portable measurements at any time. NTi Audio #: 600 000 337



48V -> ICP® Adapter

Connects to the XL2 and generates ICP power supply for alternative microphones or sensors.

Specifications:

(@ sensor sensitivity = 50 mV/Pa and output impedance < 100 Ohm)

- Residual Noise Floor typical: 17 dB(A)
- Maximum SPL: 140 dB

NTi Audio #: 600 010 223

ICP® is a registered trademark of PCB Piezotronics.

ASD Cable

The ASD Cable allows for extended connections of the NTi Audio measurement microphones or microphone pre-amplifier MA220, while maintaining the automated sensor detection functionality (ASD). You may chain up to 4x ASD Cables together in tandem.

Length = 5 meter (16 feet) NTi Audio #: 600 000 336







Exel System Case

This compact system case provides the professional transport protection for work in the field. It offers space for the handheld instruments, cables and connectors. NTi Audio #: 600 000 334



Manufacturer Calibration Certificate Individual calibration certificate with serial number, traceable according to the standard EN ISO / IEC 17025. After the purchase we recommend the annual calibration of the instrument. NTi Audio # 600 000 018



Battery Charger

The Battery Charger efficiently recharges the spare battery whilst you are using your XL2 Audio and Acoustic Analyzer. One spare Li-Po battery is included with the battery charger. NTi Audio #: 600 000 332

Digital I/O Adapter PCB

I/O The Digital I/O Adapter serves to control external peripheral devices, such as indicating sound levels that exceed limits on a big external red-orange-green lamp. It connects to the digital I/O interface.



XL2 Input Keypad

The XL2 Input Keypad offers four marker keys to trigger an event recording or categorize any noise during measurements. Requires the optional Extended Acoustic Pack.

NTi Audio #: 600 000 384



XL2 Mounting Adapter

Mechanical adapter piece to mount the XL2 on a microphone stand. NTi Audio #: 600 000 372







Stack Light with I/O Adapter Box The XL2 Audio and Acoustic Analyzer controls the external Stack Light via the Digital I/O Adapter Box. The light has three lamp sections, coloured green, orange and red.

Applications:

 Sound level monitoring at live events or clubs

The Stack Light shows orange for sound levels near the limit and red at levels exceeding the limits.

- Industrial noise pollution monitoring
- Passed / Failed measurements (requires XL2 Spectral Limits Option)

NTi Audio # Stack Light: 600 000 382 Digital I/O Adapter Box: 600 000 381



INTI

0

WANT SINEWAVE

L -20.0 dBu # 2.135 H

Precision Calibrator 94/114 dB, Class 1 The precision calibrator ensures accurate sound level measurements. It generates 94 or 114 dBSPL reference tones @ 1kHz for calibration of 1/2" and 1/4" measurement microphones (compatible with all NTi Audio microphones).

NTi Audio #: 600 000 390

Minirator MR-PRO

The MR-PRO is an extremely powerful analog audio generator designed for the professional engineer. It generates the following analog audio signals:

- Sine Waveforms, freely selectable frequencies and levels up to +18 dBu
- Sweep Signals, any frequency interval up to 1/12 octave
- White Noise, Pink Noise
- Polarity Test Signal
- Delay Test Signal
- Uploaded wav-files (*.wav)

NTi Audio #: 600 000 310







NTi Audio TalkBox

The NTi Audio TalkBox greatly simplifies the acoustic feed of the STIPA intelligibility test source signal into closed sound reinforcement systems. It presents the standardized voice-like acoustic signal emission simulating a human talker in accordance with IEC 60268-16, combined with a TNO-certified speech intelligibility signal at standardized levels. NTi Audio #: 600 000 085 4 GB SD Card (default) 1x included with XL2 NTi Audio #: 600 000 374

16 GB SD Card Individually tested high performance card for data storage. NTi Audio #: 600 000 385

32 GB SD Card Individually tested high performance card for data storage. NTi Audio #: 600 000 386

Warranty Conditions

International warranty

NTi Audio guarantees the function of the XL2 Audio and Acoustic Analyzer and the individual components for a period of one year from the date of sale. During this period, defective instruments will either be repaired free of charge or replaced.

Limitations

These guarantee provisions do not cover damage caused by accidents, transportation, incorrect use, carelessness, accessories or the installation of any parts that were not delivered with the instrument, the loss of parts, connection to the mains voltage, operation with non-specified input voltages, adapter types or incorrectly inserted batteries. In particular, NTi Audio accepts no responsibility for subsequent damage of any kind. The warranty will be voided by the carrying out of repairs or service work by third parties who are not part of an approved NTi Audio Service Centre.

Repairs

If the XL2 is not functioning correctly or is damaged, please contact the local NTi Audio partner for assistance. If the instrument needs to be returned for service, kindly follow the service guidelines at www.nti-audio.com/service.



Calibration Certificate

The XL2 Audio and Acoustic Analyzer has been carefully tested during production and corresponds to the specifications listed in "Technical Data".

NTi Audio recommends annual calibration of the instrument. Kindly follow the service guidelines at www.nti-audio.com/service.

Damage through shocks and moisture

- The protective shock jacket shields the instrument against reasonable impacts that could occur in normal use.
- Do not intentionally subject the instrument to extreme stress!
- Please do not drop the instrument!
- Damage caused by dropping or impact is not covered by warranty.
- Do not use the instrument in damp environments! The instrument can be permanently damaged if exposed to moisture.





CE

Declaration of Conformity

CE / FCC Compliance Statement

We, the manufacturer

NTi Audio AG Im alten Riet 102 9494 Schaan Liechtenstein, Europe

do hereby declare that the XL2, M4260, M2210, MA220 and accessories, comply with the following standards or other standard documents:

EMC: 89/336, 92/31, 93/68 Harmonized standards: EN 61326-1

This declaration will become invalid if modifications to the instrument are carried out without the written approval of NTi Audio.

Date: 1. December 2011 Signature:

Rechan

Position: Technical Director

Information for Disposal and Recycling



Dispose of the instrument in accordance with the legal environmental regulations in the country.

Regulations for the European Union and other European countries with corresponding laws:

The instrument must not be disposed of in the household garbage. At the end of its service life, bring the instrument to a collecting point for electrical recycling in accordance with the local legal regulations.

Other countries outside the European Union:

Contact the respective authorities for the valid environmental regulations in the country.



18. Technical Data XL2

Sound Level	Sound Level Meter		
Product Con- figurations	 XL2 with M2210 microphone Class 1 frequency response in accordance with IEC 61672 Type 1 frequency response in accordance with ANSI S1.4 XL2 with M4260 microphone Class 2 in accordance with IEC 61672 Type 2 in accordance with ANSI S1.4 		
Conforms with Stan- dards	 IEC 61672, IEC 60651, IEC 60804, IEC 61260 class 0, ISO 2969 China: GB/T 3785:2010, GB/T 3241, GB 3096-2008, GB 50526, GB-T_4959-1995 Germany: DIN 15905-5, DIN 45645-2, optional: DIN 45645-1 Japan: JIS C1509-1:2005, JIS C 1513 class 1, JIS C 1514 class 0 Switzerland: SLV US: ANSI S1.4, ANSI S1.43, ANSI S1.11-2004 class 1 International IEC standards are adopted as European standards and the letters IEC are replaced by EN. XL2 conforms to these EN standards. 		
Details	 Measurement bandwidth (-3dB): 2 Hz to 23.6 kHz Level resolution: 0.1 dB Internal noise: 1.3 µV A-Weighted 		
Weighting	 Frequency weighting: A, C, Z Time weighting: Fast, Slow, optional: Impulse 		

Linea suren Range		• XL2 + M2210: 28 dB(A) - 144 dB (IEC 60804) • XL2 + M4260: 35 dB(A) - 144 dB (IEC 60804)
Level	Range	 Three level ranges depending on the sensitivity of the connected measurement microphone with manual setting. M2210, M4260 @ sensitivity = 20 mV/Pa LOW, lower level range: 10 - 110 dBSPL MID, mid-level range: 30 - 130 dBSPL HIGH, upper level range: 50 - 150 dBSPL
Stabil Time	ization	< 10 seconds
Overa Range 42 m ¹	e @ S =	 A weighted level: 23.1 - 139.0 dBSPL C weighted level: 23.6 - 139.0 dBSPL



Residual	• Frequency we	ighting A			Measure-
noise in [dB] @ S =	Level range	L _{eq}	L _{peak}		ments
42 mV/Pa	LOW	4	17		
of XL2 with-	MID	18	31		
out mea- surement	HIGH	43	55		
microphone	• Frequency we	ighting C		'	
	Level range	L _{eq}	L _{peak}		
	LOW	3	16		
	MID	17	30		
	HIGH	41	55		Real-Time
	• Frequency we	ighting Z			Analyzer RTA
	Level range	L _{eq}	L _{peak}		
	LOW	7	20		
	MID	21	34		
	HIGH	46	58		
Integration Time	• Minimum: 1 se				Remote
lime	• Maximum: 100	J nours mini	us i second	1	Measure- ment

Measure- nents	 SPL actual, Leq, Lmin, Lmax, LCpeak Gliding LAeq with t = 5", 10', 15' and 60' All measurement results simultaneously available Correction value measurement wizard Noise curves NC, NR, PNC, RC with post-processing Noise exposure level LEX with post-processing Logging all data or subsets in selectable intervals Recording of wav-files (ADPCM), a new wav-file starts every 12 hours (max. wav-file size 512 MB) Recording of sound levels that exceed limits Digital I/O interface for external peripherals control
Real-Time Analyzer RTA	 Frequency resolution: 1/3 octave, 1/1 octave, wide band Frequency range: 6.3 Hz to 20 kHz, level resolution: 0.1 dB Band pass filters conform with IEC 61260 class 0, ANSI S1.11-2004, class 1 Frequency weighting: X-Curve in accordance with ISO 2969 Capturing of a single reading into the internal memory for comparative measurements
Remote Vleasure- nent optional)	• Querying measurement data online via the USB interface



Functions of Extended Acoustic Pack (optional)	 SLMeter/RTA function Recording of linear wav-files (24 bit, 48 kHz) Percentiles for wide band and 1/1 and 1/3 octave spectrum 1%, 5%, 10%, 50%, 90%, 95%, 99% Sampling: every 1.3 ms Wide band: in 0.1 dB wide classes, based on sampling LAF 1/1 and 1/3 octave spectrum: in 1.0 dB wide classes, based on Lxy (x= A, C or Z, y= F or S) Dynamic range: 140 dB * Sound Exposure Level LAE 100ms logging Event-triggered audio and data recording Time weighting: Impulse (Lxl, Lxleq with x= A, C, Z) True peak level in 1/1 and 1/3 octave resolution Clock-Impulse Maximum Level (TaktMax) and values as specified in DIN 45645-1 FFT function High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz RT60 function Reverberation time RT60 in 1/3 octave resolution 	Functions of Spectral Limits Option (optional)	 SLMeter/RTA function True peak level in 1/1 and 1/3 octave resolution FFT function High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz 1/12 octave function High resolution RTA function "1/12 Oct + Tol" Selectable 1/1, 1/3, 1/6 and 1/12 octave spectral resolution FFT and 1/12 octave function Capturing of multiple readings into the internal memory Comparing measurement results against captures with relative or absolute curve display Comprehensive tolerance handling with tolerance masks based on captures for passed/failed measurements Export and import of tolerance and capture files



Acoustic Ana	lyzer	Polar
FFT Analysis	 Real-time FFT with actual level, Leq, Lmin, Lmax Level resolution: 0.1 dB Ranges: 7 Hz - 215 Hz, 58 Hz - 1.72 kHz, 484 Hz - 20.5 kHz with 142 frequency bins shown on display Optional: High-resolution Zoom-FFT with selectable frequency ranges and resolution up to 0.4 Hz in the range of 5 Hz to 20 kHz Optional: Capture and tolerance function with multiple readings for comparative measurements and passed/failed analysis 	1/12 Analy (optic
Reverbera- tion Time RT60	 1/1 octave bands results from 63 Hz - 8 kHz, based on T20 Optional: 1/3 octave bands results from 50 Hz - 10 kHz, based on T20 Range: 10 ms - 14 seconds Measurement in accordance with ISO 3382 using Schroeder-method Test signal: Impulse source or gated pink noise generated by the MR-PRO, MR2 or the included NTi Audio Test CD 	Cinel Mete (optio
Delay Time	 Propagation delay between electrical reference signal and acoustic signal using the internal microphone Range: 0 ms - 1 second (0 m - 344 m) Resolution: 0.1 ms Test signal: NTi Audio delay test signal generated by the MR-PRO, MR2 or the included NTi Audio Test CD 	Spee Intell (optio

Polarity	 Checks polarity of speakers and line signals Positive/Negative detection of wideband and individual 1/1 octave bands through internal micro- phone or XLR/RCA connector Test signal: NTi Audio polarity test signal gener- ated by the MR-PRO, MR2 or the included NTi Audio Test CD
1/12 Octave Analysis (optional)	 Actual level, Leq, Lmin, Lmax Selectable 1/1, 1/3, 1/6 and 1/12 octave spectral resolution Capturing of multiple readings into the internal memory Comparing measurement results against captures with relative or absolute curve display Comprehensive tolerance handling Creating tolerance masks based on captures for passed/failed measurements
Cinema Meter (optional)	 Measurements in 1/3 octave resolution according SMPTE ST 202:2010 and SMPTE RP 200:2012 An interactive assistant guides the user through dedicated measurement procedures.
STIPA Speech Intelligibility (optional)	 Single value STI and CIS test result in accordance with IEC 60268-16, ISO 7240-16, ISO 7240-19, DIN VDE 0828-1, DIN VDE 0833-4 Ambient noise correction Automated averaging of measurements Modulation indices and individual band level results with error indicator Test signal: NTi Audio STIPA signal generated by the MR-PRO, NTi Audio TalkBox or the STIPA Test CD



Audio Analyz	zer
Level RMS	 True RMS detection in V, dBu, dBV and dBSPL Range XLR/RCA input: 2 µV - 25 V (-112 dBu to +30 dBu) Accuracy: ± 0.5 % @ 1 kHz, Flatness: ± 0.1 dB @ 12 Hz to 21.3 kHz Bandwidth (-3 dB): 5 Hz to 23.6 kHz Resolution: 3 digits (dB scale),5 digits (linear scale) or 6 digits (x1 scale)
Frequency	 Range: 9 Hz to 21.3 kHz Resolution: 6 digits Accuracy: < ± 0.003%
THD+N (Total Har- monic Distortion + Noise)	 Range: -100 dB to 0 dB (0.001% to 100%) Minimum level: > -90 dBu Fundamental frequency range: 10 Hz to 21.3 kHz Measurement bandwidth: 5 Hz to 23.6 kHz Resolution: 3 digits (dB scale) or 4 digits (linear scale) Residual THD+N @ XLR/RCA input: < 2 μV
Scope	Auto ranging, auto scaling
Filter	 Frequency weighting: A, C, Z Highpass 100Hz, 400 Hz, 19 kHz, Bandpass 22.4 Hz - 22.4 kHz in accordance with IEC468-4
Remote Measure- ment (optional)	• Querying measurement data online via the USB interface

Input / Outpu	ut Interfaces
Audio Inputs	 XLR balanced with input impedance = 200 kOhm, phantom power: +48 V switchable, automated sensor detection for NTi Audio's ASD measurement microphones and pre-amplifier MA220 RCA unbalanced with input impedance >30 kOhm Built-in condenser microphone for polarity testing, delay measurements and voice note recording
Audio Outputs	Built-in speakerHeadphone connector 3.5 mm Minijack Stereo
USB Inter- face	USB mini connector for data transfer to PC, XL2 Projector and/or charging of Li-Po battery
Digital I/O	Connection interface to accessories • XL2 Input Keypad • Digital I/O Adapter Box • Digital I/O Adapter PCB
TOSLink	24 bit linear PCM audio signal output (prepared for later firmware extension)
Memory	SD Card included (4 GByte), removable, storing measurement data in ASCII format, screen shots, voice notes and wav-files



_	
Power	 Rechargeable Li-Po battery included
Supply	» Type 3.7 V / 2260 mAh
cappiy	<i>,</i> ,
	» Typical battery lifetime > 4 hours
	» Range: 3.3 - 4.5 VDC
	 Dry cell batteries type AA, 4 x 1.5 V
	» Typical battery lifetime > 4 hours
	» Range: 3.7 - 6.0 VDC
	Linear external power supply 9 VDC
	» Range: 7.5 - 23.0 VDC @ minimum 6 Watt
	» Charges Li-Po battery during operation
	• USB-Power Supply charges Li-Po battery during
	operation

General		
Clock	Real-time clock with lithium backup battery	
Calibration	 Recommended calibration interval: one year Microphone calibration with external calibrator supported Optional calibration certificate for new instruments available 	
Mechanics	 Tripod or microphone stand mount Wire stand mounted on rear side Display: 160 x 160 pixels grey scale with LED back light Dimensions (L x W x H) 180 mm x 90 mm x 45 mm 7.1" x 3.5" x 1.8" Weight: 480 g (1 lb) including built-in Li-Po battery 	
Temperature	-10 °C to +50 °C (14° to 122°F)	
Humidity	5% to 90% RH, non-condensing	
Electromag- netic Compat- ibility	CE compliant: EN 61326-1 Class B, EN 55011 class B EN 61000-4-2 to -6 & -11	

All specifications are according the standard IEC61672. Other applicable standards are listed the corresponding specification position.

19. Technical Data Microphones

	M2210 Class 1 Frequency Response	M2215 Class 1 Frequency Response	M4260 Class 2
Microphone Type	Or	nnidirectional, pre-polarized condens free field microphone	er,
Capsule / Transducer	1/2" detachable capsul	e with 60UNS2 thread	1/4" permanently installed cap- sule
PreAmplfier Type	MA	220	-
Flatness	Class 1 (IEC61672-1) ±1 dB @ 5 Hz - 20 Hz ±1 dB @ 20 Hz - 4 kHz ±2 dB @ 4 kHz - 20 kHz		Class 2 (IEC61672-1) +1/-4.5 dB @ 5 Hz - 20 Hz ±1.5 dB @ 20 Hz - 4 kHz ±4.5 dB @ 4 kHz - 20 kHz
Frequency Range		5 Hz - 20 kHz	
Residual Noise Floor typical	19 dB(A)	25 dB(A)	29 dB(A)
Maximum SPL @THD 3%, 1 kHz	144 dBSPL	153 dBSPL	144 dBSPL
Sensitivity typical	-34 dBV/Pa ±3 dB (20 mV/Pa) @ 1 kHz	-42 dBV/Pa ±3 dB (8 mV/Pa) @ 1 kHz	-31.7 dBV/Pa ±3 dB (26 mV/Pa) @ 1 kHz
Temperature Coefficient	±0.015 dB / °C	±0.015 dB / °C	±0.02 dB / °C
Pressure Coefficient	-0.02 d	B / kPa	-0.04 dB / kPa
Influence of Humidity (non-condensing)	< ±0.01 dB		< ±0.4 dB
Long Term Stability	> 250 years / dB		-
Temperature Range	-10°C to +50°C (14°F to 122°F)		0°C bis +40°C (32°F bis 104°F)



	M2210 Class 1 Frequency Response	M2215 Class 1 Frequency Response	M4260 Class 2	
Temperature Range	-10°C to +50°C (14°F to 122°F)			
Humidity		5% to 90% RH, non-condensing		
Electronic Data Sheet	NTi Audio ASD in accordance with IEEE P1451.4 V1.0, Class 2, Template 27			
Output Impedance		100 Ohm balanced		
Power Supply		48 VDC phantom power, 3 mA typical		
Connector		Balanced 3-pole XLR		
Dimensions	Leng	Length 150 mm (5.9"), diameter 20.5 mm (0.8")		
Weight	100 g,	3.53 oz	83 g, 2.93 oz	
NTi Audio #	600 040 020 600 040 045 600 040 02			





(free field conditions @ reference direction 0°)



	XL2 + M2210	XL2 + M4260
Linear Measurement Range	• 28 dB(A) - 144 dB in accordance with IEC 60804	• 35 dB(A) - 144 dB in accordance with IEC 60804



Appendix 1: Standard - Optional Features

	Standard Features	Optional
Sound Level Meter Frequency Weighting	ACZ	
Sound Level Meter Time weighting	F S EQEQ PK	Extended Acoustic Pack: Impulse Sound Exposure Level Percentile Sound Pressure Levels: 1% 5% 10% 50% 90% 95% 99%
Sound Level Meter Correction	K1 K2 off	



	Standard Features	Optional
Sound Level Meter Parameter	live(max(min)Prev	 Extended Acoustic Pack: Clock-Impulse Maximum Level (Taktmaximalpegel) in ac- cordance with DIN 45645-1: T3 T3en T5 T5en Calculated levels in accordance with DIN 45645-1: LAFT5eq^{-L}Aeq LAIeq^{-L}Aeq LCeq^{-L}Aeq
Sound Level Meter RTA Spectrum Analyzer		Extended Acoustic Pack or Spectral Limits Option: True peak level
Sound Level Meter Audio Recording	Recording of wav-files (ADPCM compression)Voice note recording	• Extended Acoustic Pack: Recording of wav-files (24 bit, 48 kHz)
Sound Level Meter Triggered Event Recording		Extended Acoustic Pack:



	Standard Features	Optional
Sound Level Meter Logging		100 ms logging
FFT Analysis Measurement Range	200 1k7 20k	Extended Acoustic Pack: Usr with zoom function Spectral Limits: Usr with zoom function
FFT + Tol Capture and Tolerance	not available	• Spectral Limits: Capture and Tolerance functionality
RT60		• Extended Acoustic Pack: 1/3 octave band results
1/12 Oct + Tol	function not available	 Spectral Limits: 1/12 Oct + Tol Capture and Tolerance functionality
STIPA	function not available	STIPA Option with STIPA functionality
Querying measurement data online via the USB interface		Remote Measurement Option supporting SLMeter/RTA and RMS/THD+N function



Appendix 2: Factory Default Profiles

You may start up the XL2 with individual preset application profiles.

_	t Prof. node	I(6:			
#DIN:	5905-5	_			_
	100dB 100dB		Audi	io 🖬	
	100dB	_	Audi	08	

The profile **Full mode** is the default configuration with all available measurement functions enabled.

DIN15905-5 / DIN15905-5 & Audio (Sound Level Monitoring in accordance with DIN15905-5)

The following limits apply at all areas accessible by the visitors during the measurement period of 30 minutes:

- Maximum averaged level = 99 dB The XL2 displays this maximum level with the measurement result L_{Aeq+k1}.
- Maximum peak level L_{Cpeak} = 135 dB

SLMeter 123 X	SLMeter 123 XLRAUD) (ASD 15:35 (
LAeq5"+k1			
93.9dB			
L _{fleg} +k1	92.7dB		
L _{CPKma×} +k2	116.0dB		
III 50 RHGE 150	set 00: 30:00 फ 00:23: 34		



These levels are shown in the numeric result page:

- **LAeq5"+k1** Time-averaged sound pressure level LAeq with correction value k1 and a moving time window of 5 seconds.
- LAeq+k1 Time-averaged sound pressure level LAeq with correction value k1.
- **L**Cpeak+k2 C-weighted peak level L_{Cpeak} with correction value k2.

The following sound levels are displayed in the real-time analyzer page:

LZFhold3 For tracing of feedback frequencies. The peak hold time is set to 3 seconds.

LZFlive Actual real-time spectra.

How to Measure

- Select the page ${f KSET}$ with the rotary wheel ${igodot}$.
- Measure the correction values k1 and k2 as described in the chapter "Correction Value KSET".
- Press the start button 🖭.
- The icon **LOG** flashes during the ongoing measurement. In the case where the profile **#DIN15905-5 & Audio** has been selected the icon **AUD** flashes, indicating the additional audio wav-file logging.
- During the measurement you can monitor the actual sound level L_{Aeq5"}+k1 in order to remain below the permitted limits; alternatively you may monitor the spectrum analyzer for tracing possible feedback frequencies.
- Press stop 🗩 to finish the measurement.



In accordance with DIN15905-5, all sound levels are measured, displayed and logged to the SD Card automatically. The settings are locked, which ensures that the correct parameters are measured.

The profile **#DIN15905-5 & Audio** includes the recording of **Compressed+AGC** wav-files.



Limits

This profile is configured with the following limits, which trigger the lights on the limit button and external Digital I/O Adapter. This facilitates a quick response to sound levels that exceed these limits.

SLMeter Lim XLR 💼 🛛 🕯 850 19658 🛑				
Limit LED: On				
	LAeq5"+k1			
Red	> 99.0dB			
Orange	> 95.0dB			
Green	≤ 95 . 0dB			
I/O Box	: On	STAT: 1234		
	LAeq5"+k1	L _{Aeq5"} +k1		
0ut 1	> 99.0dB	and > 99.0dB		
Out 2	> 95.0dB	and ≤ 99 . 0dB		
Out 3	≤ 95 . 0dB	and ≤ 95 . 0dB		
0ut 4	≦ 0 . 0dB	and ≤ 0.0dB		

Measurement Results

You may import the log files into a report generator software based on MS Excel, providing automatically-completed live event reports with noise level charts. The report generator software is a free download on the support website at http://my.nti-audio.com for all registered XL2 customers. SLV 100dB / SLV 100dB & Audio (Sound Level Monitoring in accordance with SLV)

The following limits apply at all areas accessible by the visitors during the moving measurement period window of 60 minutes during the live event:

• Maximum averaged level

The XL2 displays this maximum level with the measurement result $\text{L}_{Aeq60'+k1}.$

Type of live event	Max. level	Requirements
3	100 dB	 Sound level recording Results to store for 30 days Level warning to visitors Hearing protection distributed Chill out zone < 85 dB(A)
2	96 dB	Sound level monitoringLevel warning to visitorsHearing protection distributed
1	93 dB	- Sound level monitoring

• Maximum peak level LAFmax = 125 dB





These levels are shown in the numeric result page:

- **LAeq5"+k1** Time-averaged sound pressure level L_{Aeq} with correction value k1 and a moving time window of 5 seconds.
- LAeq60'+k1 Time-averaged sound pressure level LAeq with correction value k1 and a moving time window of 60 minutes.
- **LAFmax+k1** Maximum sound pressure level L_{AFmax} with correction value k1.

The following sound levels are displayed in the real-time analyzer page:

LZFhold3For tracing of feedback frequencies. The
peak hold time is set to 3 seconds.LZFliveActual real-time spectra.

How to Measure

- Select the page KSET with the rotary wheel ${igodot}$.
- Measure the correction value k1 as described in the chapter "Correction Value KSET". (k2 is not required for SLV)
- Press the start button 🖭.
- The icon **LOG** flashes during the ongoing measurement. In the case where the profile **#SLV 100dB & Audio** has been selected the icon **AUD** flashes, indicating the additional audio wav-file logging.
- During the measurement you can monitor the actual sound level L_{Aeq5"}+k1 in order to remain below the permitted limits; alternatively you may monitor the spectrum analyzer for tracing possible feedback frequencies.
- Press stop 🗩 to finish the measurement.



(i) In m au er su Tr re

In accordance with SLV, all sound levels are measured, displayed and logged to SD Card automatically. The settings are locked, which ensures that the correct parameters are measured.

The profile **#SLV 100dB & Audio** includes the recording of **Compressed+AGC** wav-files.

Limits

This profile is configured with the following limits, which trigger the lights on the limit button and external Digital I/O Adapter. This facilitates a quick response to sound levels that exceed these limits.

(SLMeter	r Lim XLR	.]46	ISD 15:28 🗨
Limit L	ED: On		
	L _{Aeq5"*k1}		
Red	>100.0 dB		
Orange	> 96.0dB		
Green	≤ 96 . 0dB		
I/O Box	: On		STAT: 123
I/O Bo×	: On LAeq5"+k1		stat:123" LReg5"*k1
I/O Bo× Out 1		and	
	L _{Req5"*k1}	and and	L _{Aeq5"+k1}
Out 1	L _{Req5"*k1} >100.0dB		L _{Req5} "+k1 >100.0dB

Measurement Results

You may import the log files into a report generator software based on MS Excel, providing automatically-completed live event reports with noise level charts. The report generator software is a free download on the support website at http://my.nti-audio.com for all registered XL2 customers.

Sound Exposure Level

The sound exposure level L_{AE} is the L_{Aeq} normalized to 1 second. The sound exposure level is measured to compare noise events with different time durations. The L_{AE} reports the amount of noise from an event such as an individual aircraft fly-over. $L_{AE} = L_{Aeq} + 10 \times \log$ (time in seconds)



Ε

Time-Average Sound Level or Equivalent Continuous Sound Level in accordance with IEC 61672



Moving Time-Average Sound Level or Moving Equivalent Continuous Sound Level with selectable time intervals of 5 seconds, 10 minutes, 15 minutes or 60 minutes. The moving timeaverage sound level is an Leq based on a moving time window.

Sample:

10h 00min 00sec Start of XL2 measurement

10h 00min 05sec Leq5" = Leq of this 5 seconds

Appendix 3: Description Sound Levels



A-Frequency Weighting Filter in accordance with IEC 61672

Applicable for most common sound pressure level measurements, compensating for human hearing perception at lower sound pressure levels.



C-Frequency Weighting Filter in accordance with IEC 61672

Applicable for very high sound pressure levels, compensating for human hearing perception at high sound pressure levels; typically used for measurements of peak values, e.g., LCpeak



Capture

This selection offers the reference curve, which has been previously stored with **Capture** in the spectrum analyzer of the SLMeter function. Application example:

• Comparing the performance of the left and right speakers in a live event sound setup.



- 10h 00min 06sec Leq5" = Leq of the time window from 10h 00min 01sec to 10h 00min 06sec
- 10h 00min 07sec Leq5" = Leq of the time window from 10h 00min 02sec to 10h 00min 07sec

Application example:

- Display the moving Leq over 5 seconds in accordance with DIN15905
- Measure the moving Leq over 60 minutes in accordance with SLV2007

Fast Time Weighting

Time constant t = 125 ms; the time weighting is an exponential function of time, which defines how changes of the instantaneous sound pressure level are averaged for useful sound pressure level results. The measurement results are shown accurately for sound sources with a minimum length of 0.5 seconds. Fast time weighting is typically used for common sound level measurements.



Holding Maximum Level

For tracing of feedback frequencies in the RTA page. The hold time can be set to 3, 5 or 10 seconds.

I

Impulse Time Weighting

Time constant increase t = 35 ms; time constant decrease t = 1500 ms; the time weighting is an exponential function of time, which defines how changes of the instantaneous sound pressure level are averaged for useful sound pressure level results. The measurement results are shown accurately for sound sources with a minimum length of 0.1 seconds. Impulse time weighting is used for impulsive sound sources.



K2

Correction Value k1

Applies to RMS based values (L_{AF}, L_{Aeq}, ...). The correction value k1 is measured or manually set in the page **Kset** of the sound level meter function.

Correction Value k2

Applies to time-weighting setting PK (= peak). The correction value k2 is measured or manually set in the page **Kset** of the sound level meter function.





Parameter live Actual current sound pressure level.



Parameter max Maximum sound level of the measurement cycle.



Parameter min

Minimum sound level of the measurement cycle.



Correction Value off No correction value, default setting.



Peak Level Application example:

Measure L_{Cpeak} in accordance with DIN15905

The RTA spectrum analyzer measures the true peak level with the optional Extended Acoustic Pack. The peak hold time can be set to 0, 1 or 5 seconds.



Previous Leq

Applicable for timer mode Repeat or Repeat Synchronized; displays the time-averaged sound level Leq of the last measurement cycle.



Percentile Sound Level - Level Statistics

The statistical distribution of sound pressure levels is commonly used for environmental noise analysis, such as road traffic or community noise assessments. $L_{AFxX\%}$ is the noise level exceeded during xx% of the measurement period; e.g., LAF90% is the noise level exceeded during 90% of the measurement period.

Please see the specifications at chapter "Technical Data XL2 -> Functions of Extended Acoustic Pack (optional)" in this manual.



Slow Time Weighting

Time constant t = 1 second; the time-weighting is an exponential function of time, which defines how changes of the instantaneous sound pressure level are averaged for useful sound level results.

Parameter T3

Clock-Impulse Maximum Level (Taktmaximalpegel) LAFT3 in accordance with DIN 45645-1.

Maximum sound pressure level measured with frequency weighting A and time weighting F within a time interval of 3 seconds.





Parameter T3 equ ivalent

Time-average Clock-Impulse Maximum Level (Taktmaximalpegel) ${\sf L}_{AFT3eq}$ in accordance with DIN 45645-1.



Parameter T5

Clock-Impulse Maximum Level (Taktmaximalpegel) L_{AFT5} in accordance with DIN 45645-1. Maximum sound pressure level measured with frequency weighting A and time weighting F within a time interval of 5 seconds.

T5e9

Parameter T5 equivalent

Time-average Clock-Impulse Maximum Level (Taktmaximalpegel) ${\sf L}_{\rm AFT5eq}$ in accordance with DIN 45645-1.



Inverted X-Curve

In the film and recording industry an X curve is also known as a wide-range curve, e.g., used for cinema installations; conforms to ISO 2969.



Z-frequency weighting in accordance with IEC 61672 (= flat frequency response, no filter) Overall sound pressure level, all sound signal components are included; required for special applications.



Appendix 4: Common Sound Levels

Display	Application	Setting
L _{AE}	Sound Exposure Level The sound exposure level L_{AE} is the L_{Aeq} normalized to 1 second. The sound exposure level is measured to compare noise events with different time durations. The L_{AE} reports the amount of noise from an event such as an individual aircraft fly-over. $L_{AE} = L_{Aeq} + 10 \times \log$ (time in seconds)	A .⇒E
L _{AF}	Actual Sound Pressure Level SPL with A frequency weighting and F time weighting, compen- sating for human hearing perception at lower sound pressure lev- els; applicable at levels lower than 100 dB.	A-> F-> Live-> off
L _{AFmin}	Minimum Sound Pressure Level Measures the minimum environmental noise level during the measurement period.	A -> F -> min
L _{Aeq}	Time-Average Sound Level or Equivalent Continuous Sound Level Averaged sound level over time with A frequency weighting.	A -> EQ -> off



L _{Aeq} dt	Level LAeq_dt "delta t" Level of actual logging interval. e.g. Logging interval = 1 second, then LAeq_dt is the time averaged level of the past 1 second	
L _{Aeq} + k1	Time-Average Sound Level with Correction Value The actual measurement position often differs to the loudest posi- tion of the live event. Thus the correction value k1 is the correc- tion value of L_{Aeq} between the loudest position and the actual measurement position, measured using pink noise. The correc- tion value k1 is measured or manually set in the page KSET of the sound level meter function.	A -> EQ -> K 1
L _{Aeqxx}	 Moving Time-Average Sound Level Measurement of the moving sound level over time Leq with A-weighting frequency and F-weighting time responses; the parameter xx is user-defined from the available settings: 5 seconds for live sound monitoring 10 minutes (in accordance with the French standard) 15 minutes (in accordance with the British standard) 60 minutes (in accordance with SLV 2007) 	A -> EQ T -> 60' -> off



L _{Aeq60} ′ +k1	Moving Time-Average Sound Level with Correction Value k1 The actual measurement position often differs to the loudest posi- tion of the live event. Thus the correction value k1 is the correction value of $L_{Aeq60'M}$ between the loudest position and the actual measurement position, measured using pink noise. The correc- tion value k1 is measured or manually set in the page KSET of the sound level meter function. Application: Measurements in accordance with SLV 2007	A -> EQ _T -> 60' -> <u>K 1</u>
L _{Cpeak} + k2	Peak Sound Pressure Level with Correction Value The actual measurement position often differs to the loudest posi- tion of the live event. Thus the correction value k2 is the correc- tion value of L_{Cpeak} between the loudest position and the actual measurement position, measured using pink noise. The correc- tion value k2 is measured or manually set in the page KSET of the sound level meter function.	C -> PK -> K 2







LΕΧ	 quirements for the particular the risk to particular the risk to particular the risk to steady noise L_{EX}, L_{Aeq} measured to formula applies for L_{Aeq} + 10 x log (Steady noise with Measure L_{Aeq} at the corresponding sure level post-pr and displayed Varying noise level L_{EX,8h} = L_{Aeq} measure 	enter the data with JTi Audio noise expo- _{3h} will be calculated	A -> EQ -> off		
	The exposure limit	values and r	espective ad	ctions are:	
		LEX, 8h	LCpeak		Action
	Lower limit value	80 dB(A)	135 dB	The employer shall make individual hearing protec- tors available to workers	
	Upper limit value	85 dB(A)	137 dB	Hearing protectors shall be used	
	Exposure limit	87 dB(A)	140 dB		cteristics of the hearing pro- dequately reduce the levels