

Manual for the sound card oscilloscope V1.47

C. Zeitnitz

english translation by P. van Gemmeren, K. Grady and C. Zeitnitz

This Software and all previous versions are NO Freeware!
The use of the software and of the documentation is granted free of charge for private and non-commercial use in educational institutions.

Support the further development of the program by obtaining a private donation license!

Any commercial usage requires a corresponding license.

LabView and all its components
© 2014 National Instruments Corporation. All Rights Reserved.
Details about the licensing terms can be found in the file license_NI.txt in the installation directory

For retail versions contact the author.

© C. Zeitnitz 2005-2017

Web-page: http://www.zeitnitz.eu/scope_en

The sound card oscilloscope is a digital oscilloscope with an integrated signal generator, frequency analysis (FFT) and wave file recorder

1 Requirements

- Windows 2000 , XP, Vista, Windows 7 (x64), Windows 8
- A PC with a sound card installed.
- 50MB of disk space

2 Installation

Start the install package and follow the instructions. The program can be started thereafter through the program menu of the Windows operating system. A network installation is possible as well (see section 5).

3 Description

This software can be used for the display and analysis of sound waves. The data can be recorded both directly from the sound card (with a microphone or LINE input), or from a source such as a CD or Mediaplayer. The input to the oscilloscope is defined by the Windows sound mixer (see below). The software obtains its input data for the sound card via the Windows interface. It does not communicate directly with the sound card. Therefore sound card problems should be troubleshot at the operating system level.

The user interface is arranged like a conventional oscilloscope. However, in the program window, additional XY display, frequency analysis, and settings are provided.

3.1 Oscilloscope

The software shows the left and right channel of the sound card in the oscilloscope window. The left channel is represented as a green line and the right channel as a red line. In the user interface window there are knobs and input windows for the following three functions: Amplitude, Time, and Trigger. By default both channels are shown on the screen. Unchecking the corresponding box above the screen, will disable this channel.

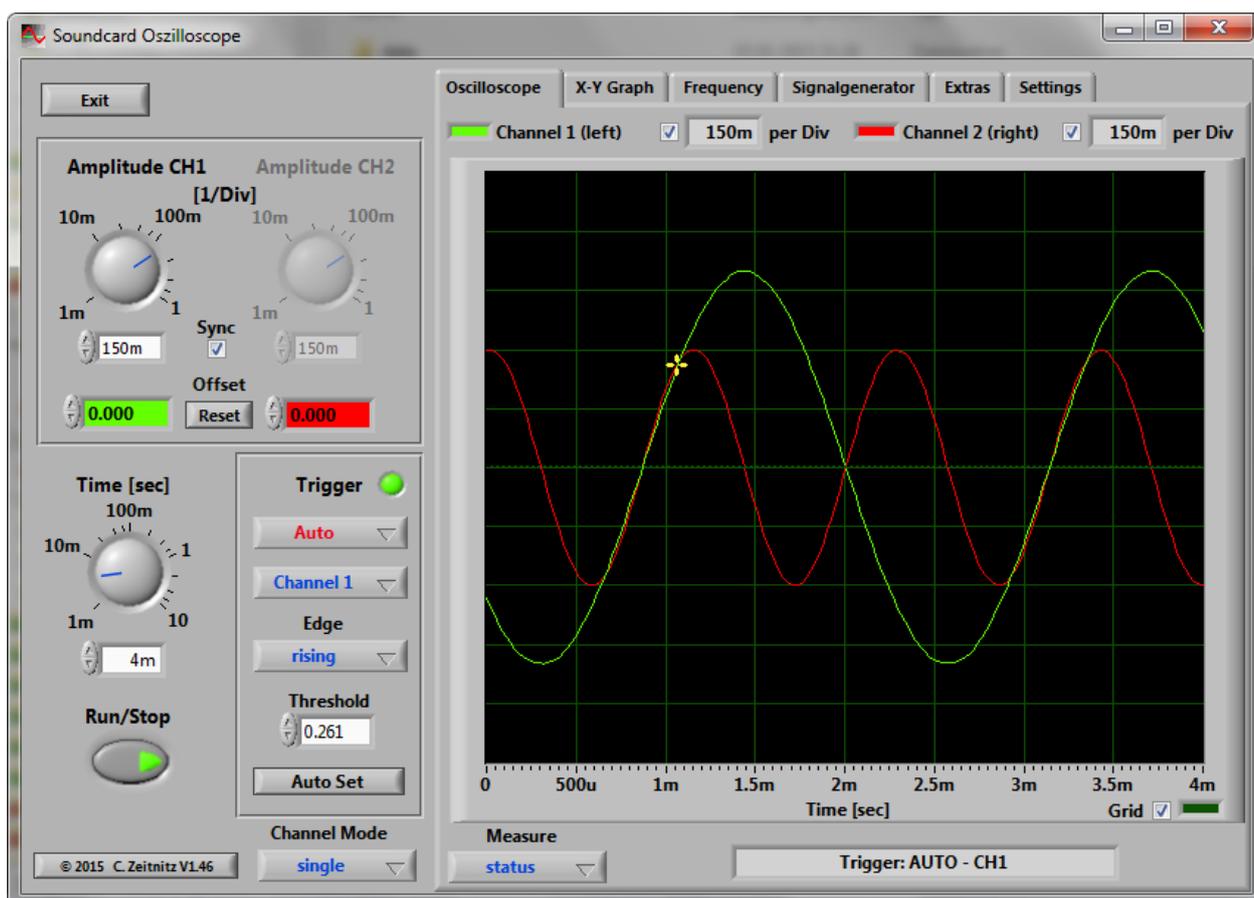


Figure 1: Soundcard oscilloscope

3.1.1 Amplitude settings

The amplitude scale of the two channels can be set independently as well as synchronized. The latter is enabled at program start-up and can be disabled by un-checking "Sync CH 1&2" on the front panel.

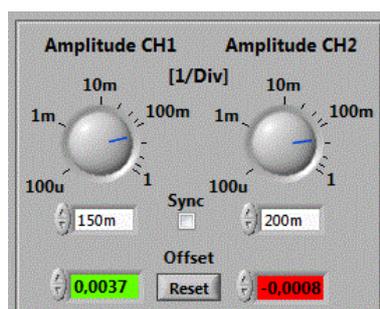


Figure 2: Amplitude settings and channel offsets

The amplitude values are given in units per division of the oscilloscope screen and are displayed for both channels above this screen. The amplitude value corresponds to the digitized sound level divided by 32768. This represents the 16Bit resolution of the data, which are taken with the sound card. Due to the different settings of the volume in Windows the absolute sound level cannot be determined directly! Therefore the presented values are to be understood in arbitrary units. The amplitude setting refers both to the oscilloscope window and to the XY graph. An offset can be assigned to each channel individually via the appropriate input window; thereby the two traces can be separated from each other. A click into one of the offset fields will result in two horizontal cursors to show up in the oscilloscope's screen. The offset can now be changed by moving these cursors with the mouse, or by entering a value into one of the fields. If the signal of the channel is outside the visible window of the screen, the cursor is shown at the upper or lower

edge of the screen (dependent where the actual signal is located). The cursors will automatically disappear from the screen after a few seconds without a change of an offset.

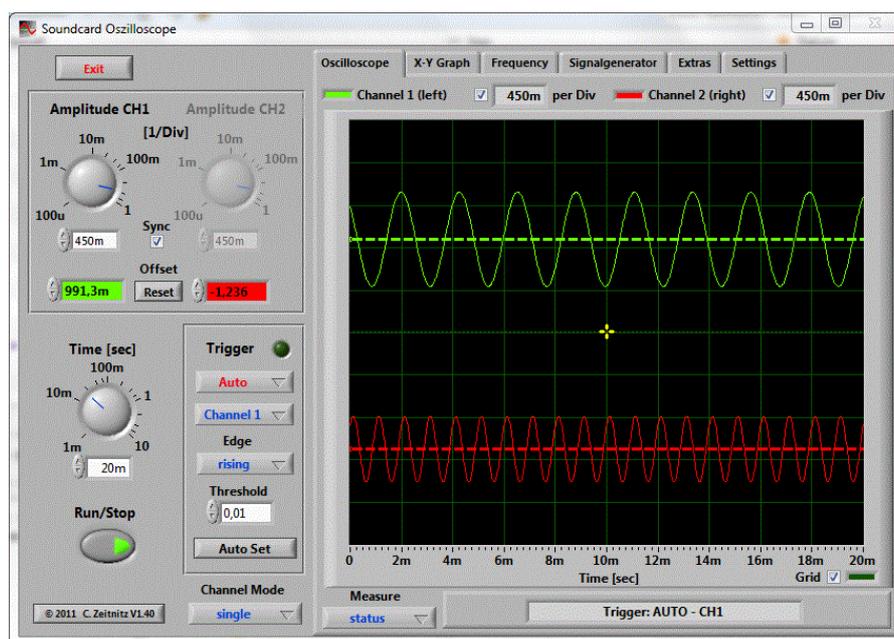


Figure 3: Offset cursors visible on screen

3.1.2 Timebase

The **Time** setting refers to the entire represented range and **NOT** to the value per unit as with a normal oscilloscope! The range goes from 1ms to 10,000ms. The larger the range, the smaller is the used scanning rate. This is unavoidable because of the extent of computer cpu use. In the trigger setting "single" the scanning rate is increased again, since computer utilization is less important here.

3.1.3 Trigger

The trigger setting modes are "off", "auto", "normal" and "single". These correspond to the standard modes of oscilloscopes. The trigger threshold can be adjusted either in the input window of the trigger selection, or by shifting the **yellow cross** in the oscilloscope window using the mouse. The trigger time can only be adjusted by shifting the cross with the mouse.

In the single SHOT mode of the trigger the RUN/stop switch is deactivated automatically and must be pressed again for a new data-taking run.

The button "Auto set" triggers the program to estimate the optimal time base and trigger level. The main frequency found in the trigger channel is used to obtain the time base. The threshold is taken from the signal amplitude. If the amplitude is too small, the button has no effect. Below approx. 20Hz the result is not reliable due to the limited time window used for the analysis.

3.1.4 Channel Mode

By default, two channels are shown in the oscilloscope window. With the mode selection switch at the bottom of the program window, the sum, difference or product of the channels can be chosen.

3.1.5 Data Analysis

The “measure” selector below the scope window allows to enable a real time analysis (frequency and amplitude (RMS and peak-to-peak value)). The corresponding check boxes are visible, when “Hz and Volts” is selected.

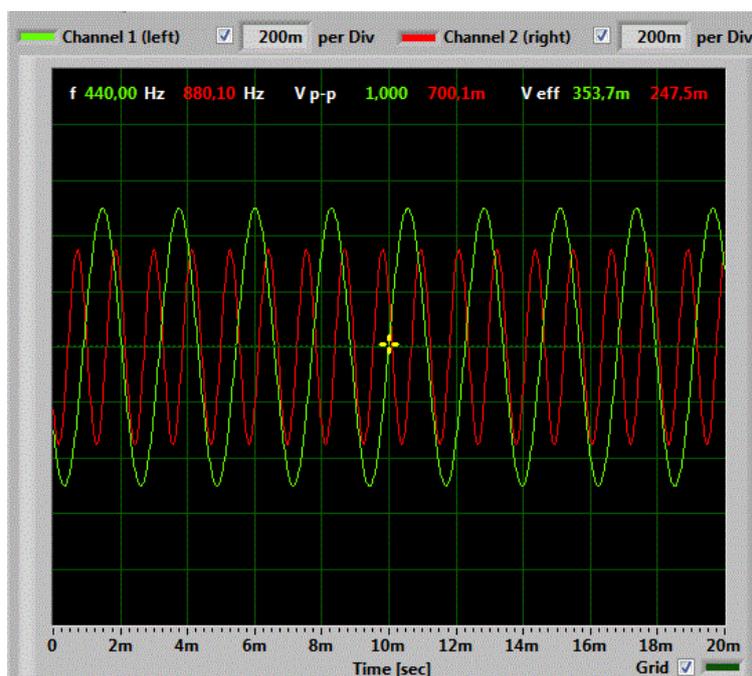


Figure 4: Automatic measurement of frequency and amplitude of signals

3.1.6 Logging Data

The measured values, which are displayed on the screen, can be written simultaneously to a log file (in CSV format). This feature is enabled by checking the “log to file” box. This will trigger a file dialog to appear in order to define the file name of the log file. The update rate of the log file is defined by the rate on the screen, which can be defined in the “Settings” tab of the scope.

The logging will only terminate, when the “log to file” box is unchecked, or when no values are displayed on the screen.



Please note: The measured values are only written to the log file, when the values are actually visible to the user. Switching to the XY-mode or frequency analysis will pause the logging of the data!

3.1.7 Measure Signal Properties

The amplitude or Time/frequency can be measured with the help of cursors in the oscilloscope window. To scan the values of the channels, right click into the screen, when the acquisition has been stopped via the Run/Stop button. You can grab the cursor with the mouse, or left click to move it to the position of the mouse. A fine adjustment of the cursor can be achieved by the left and right cursor keys of the keyboard.

Additional cursors to measure the signal properties in more details can be activated through the selector box underneath the window. The cursors can be shifted with the mouse.

In the amplitude mode the values for the two cursors as well as the amplitude difference is displayed.

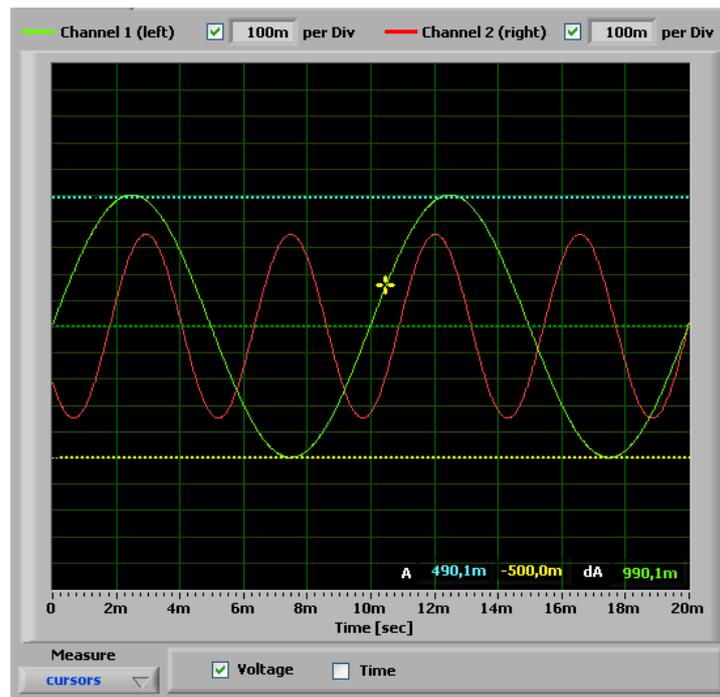


Figure 5: Amplitude analysis with the cursors. The shown values correspond to channel 1.

For the time mode the time difference and the appropriate frequency are shown directly. The data can also be examined in more detail by using the zoom (only when the data acquisition is stopped). The detail around the position of the orange cursor line is increased. By shifting the cursor the range can be changed. When the cursor is getting close to the edge of the window, the window is shifted automatically in the corresponding direction. The speed of the shift depends on the distance to the edge. Amplitude and voltage cursors can be enabled simultaneously.

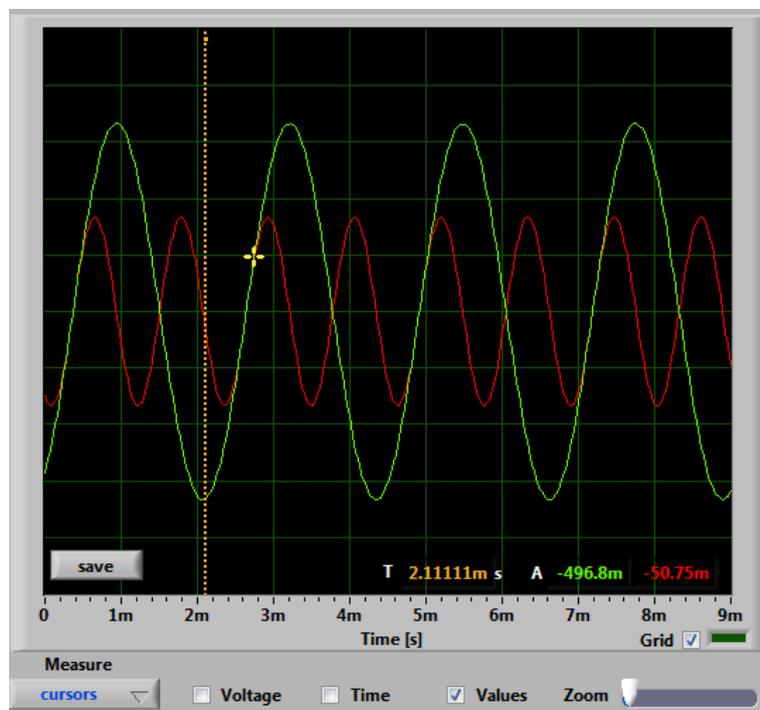


Figure 6: Zoom of the wave around the orange cursor with time and amplitude values displayed

The time position of the orange cursor and the corresponding amplitude values (green and red) are displayed in the screen as well.

3.2 Setting the Colors

The colors of the graphs and grid (oscilloscope and xy-graph) can be set by clicking on the colored legends. The color menu (see Figure 7: Changing the color of the graphs and grid Figure 7) allows to change the color immediately.

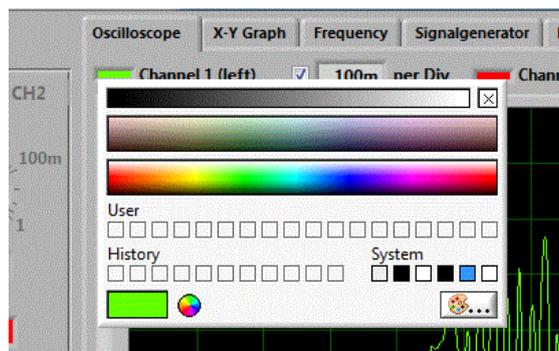


Figure 7: Changing the color of the graphs and grid

3.3 X-Y Graph

Here the two channels are displayed against each other. Thereby e.g. Lissajous figures can be produced. For this the frequencies can be adjusted in the signal generator.

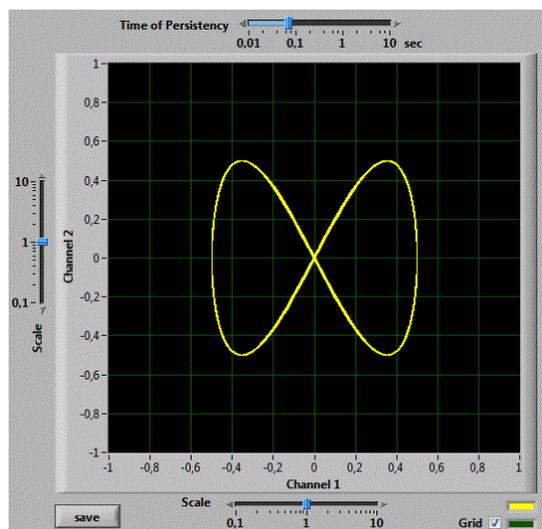


Figure 8: Lissajous Figure for $f_1 = 440\text{Hz}$, $f_2 = 880\text{Hz}$ and a phase of 0°

The slider above the graph allows to change the time of persistency of the shown data. For a longer time setting increases the time window displayed on the screen. Fast changing signals should better be displayed with a short persistency.

The controllers along the x and y axis permit a scaling of the appropriate channel (zoom in or out). The represented range is chosen by adjusting the amplitude knob in the program window.

3.4 Frequency Analysis

In the "frequency analysis" window, the display shows the result of the Fourier analysis of the selected channel. The channel can be chosen with the selection button above the grid. By default, the graph shows the amplitude of 0 - 10,000 Hz. The amplitude as well as the frequency can be displayed with a logarithmic scale or dB.

The vertical scale can automatically be adjusted by selecting the auto-scale check-box above the graph. A manual adjustment is possible by double-clicking the maximal or minimal value of the axis and entering a new value. This should be done only if auto-scale is disabled.

Below the graph is a roll bar and a zoom sliding control; they permit the indicated range to be changed. These should be only used if data taking has been stopped with the run/stop button. The zoom slider shows details of the frequency analysis: use the mouse (grab or click) to set the perpendicular yellow line to the frequency of interest (fine adjustment with the left/right cursor keys) and drive the zoom slider up to the desired detail.

When the data acquisition is stopped via Run/Stop the magnitude at the cursor position will be shown on the screen.

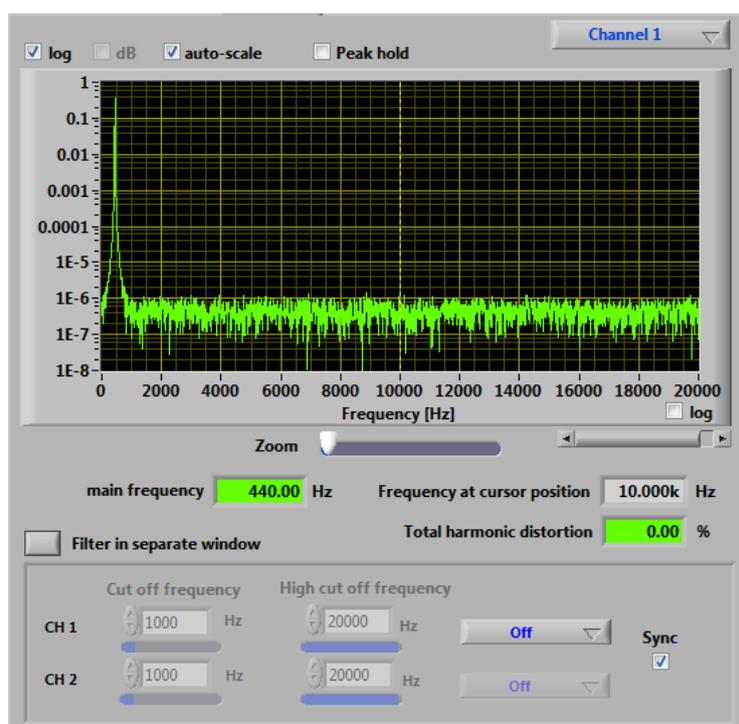


Figure 9: Frequency analysis of a 440Hz signal

The two output values underneath the sliders show the frequency at the cursor position and the value of the strongest frequency found from a harmonious analysis of the data. Note that the Fourier analysis is always based on data with the full sampling rate of 44.1kHz. Therefore, the time controller automatically jumps to a pre-defined value when this window is active.

Selecting "peak hold" allows to store the maximal amplitude values of the Fourier analysis. This allows to display the transfer function, when using the white noise generator.

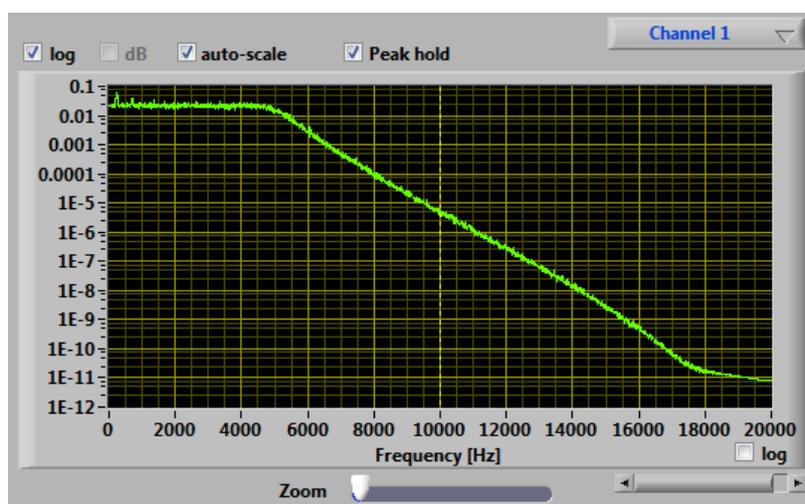


Figure 10: Transfer function utilizing the peak hold function with the white noise generator

Under the frequency analysis an adjustable frequency-selective filter (Butterworth filter 10th order) is also provided. Four kinds of filter can be selected: Low-pass, high-pass, band-pass and stop band filter. The critical frequencies can be adjusted with the sliding controls accordingly.

Above the frequency-selective filter is a button to open filter control in a separate window. This function allows one to observe the effect of the filter directly in the oscilloscope window. Double-clicking on the button or closing the window re-establishes the original settings.

3.5 Transfer Function

In addition to the frequency analysis of an individual channel it is possible to measure the transfer function. This measurement uses the ratio of Channel 1 and Channel 2 to determine the frequency dependency of the transfer characteristic. In order to obtain the transfer function one should select a noise signal or a square wave in the signal generator in order to cover the full frequency spectrum in a single measurement. Alternatively a frequency sweep can be utilized. Channel 1 should contain the original signal and the Channel 2 the filtered one.

3.6 Waterfallgraph

A waterfall graph displays the frequency analysis as function of time. For this the Fourier analysis is done by default every 120msec. The amplitude of each frequency is displayed as a color code (black: lowest amplitude, red: highest amplitude). Figure 11 shows the waterfall graph of channel 1. The period of subsequent frequency analyses can be adjusted with the parameter FourierPeriod in settings.ini. Be aware, that a small value leads to substantial increase of the load on the computer. The sensitivity of the display can be increased with the slider above the screen.

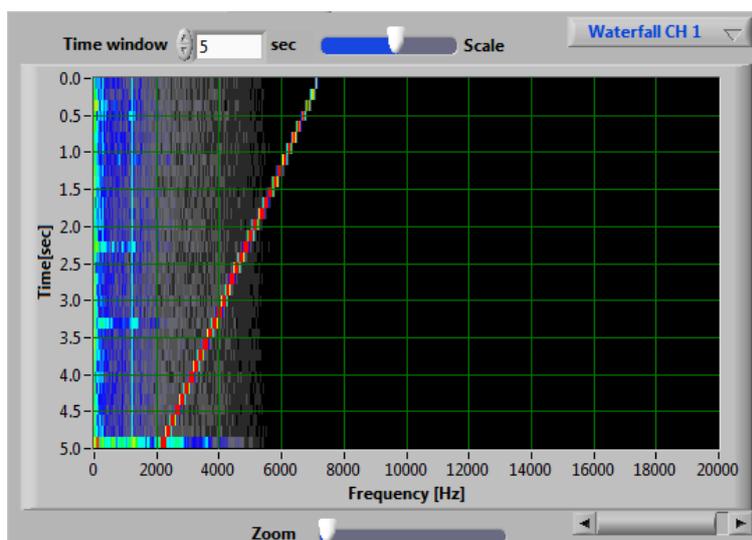


Figure 11: Waterfall graph of channel 1

3.7 Storing Display Data

The graphics visible on the display (oscilloscope screen, frequency analysis, xy-graph) can be stored, when the data acquisition has been stopped by the "RUN/STOP" button. A "save" button is displayed within the graphs area. After pressing the button a file selector box is displayed to select a file name and the preferred graphics format (BMP, JPG or PNG). Automatically the graph is saved in color and in black-white. In addition a text file (extension CSV) containing the actual data is stored with the same name. This contains the data as a Comma-separated-value list, which can be imported into Excel. Be aware, that the output to the CSV file is localized and the decimal separator (comma or dot) is selected depending on your local settings. Importing these data into Excel might lead to wrong results, if a different decimal separator is used.

3.8 Signal Generator

A 2-channel signal generator is integrated into the program. The generator can be released from the program window by pushing the button above the panel. A second click on the button will embed the generator again.

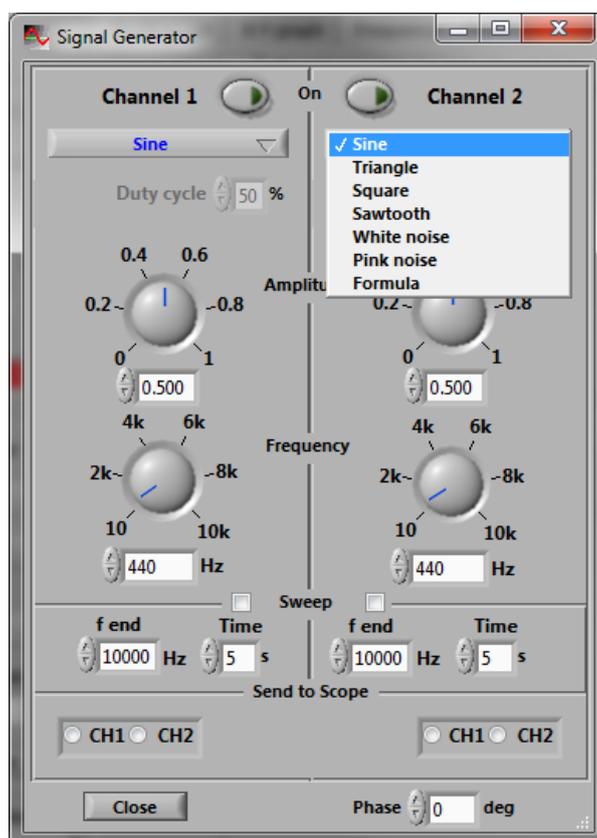


Figure 12: Signal Generator

The generator outputs sine, rectangle, triangle, and saw tooth waves with variable amplitude and frequency. A white noise generator is included as well. The phase of the signal can be adjusted. The “Sweep mode” allows to sweep the frequency from the main frequency to f(end) continuously within the specified time window.

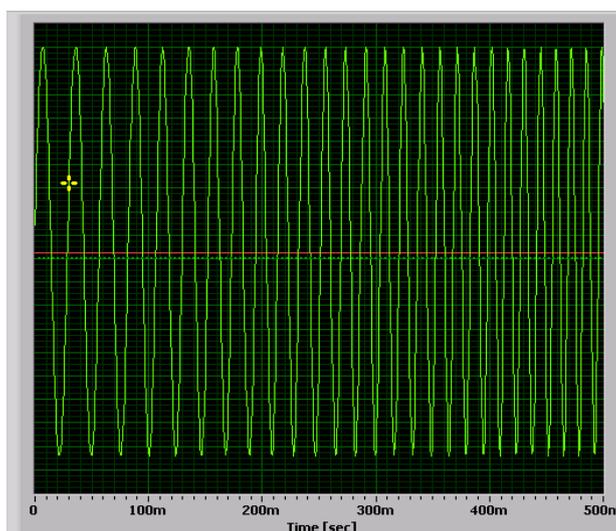


Figure 13: Automatic frequency sweep

Upon opening the signal generator, both channels are deactivated and must be switched on by a button at the bottom of the window. The frequency can be changed in steps of 0.5Hz. The generator signal can be sent directly to the sound card. This must be activated in the sound mixer of the Windows operating system (usually designated as "Wave Out"). If in addition the recording of the "Wave" source is activated, the signals are visible in the oscilloscope and can be displayed (e.g. to produce Lissajous figures). The “Send to scope” selection buttons will add the generator signals to the corresponding scope channel. This is independent of the selected recording source of the scope.

Define Signal by Formula

Selecting "Formula" from the pull down menu, will activate an input field for a mathematical formula. This formula will be evaluated by [muParser](#) to generate the signal.

The parameters of the generator are accessible via the following predefined variables:



t	time of the current sample
a	selected amplitude
f	selected frequency
w	calculated $\omega=2\pi f$
wt	pre-calculated product of ω and time t
phi	phase between the two channels (rad)
fs	sampling rate of the signal z.B. 44100Hz
ns	number of the current sample

Enter has to be pressed to activate the evaluation of the formula. If a syntax error is present, the whole formula will be marked red and has to be corrected.

Available functions: sin, cos, tan, asin, acos, atan, sinh, cosh, tanh, asinh, acosh, atanh, log2, log (log10), ln, exp, sqrt, sign, abs, min, max, mod

Operators: +, -, *, /, ^

Defined constant: pi

Noise

rand()	random distribution of the signal amplitude within ± 1
white()	white noise within ± 1
pink()	pink noise (1/f) within ± 1
brown()	Brownian noise (1/f ²) within ± 1

For the white, pink and Brownian noise, the amplitudes follow a Gaussian distribution with a RMS of approximately 0.25. Hence the amplitudes cover $\pm 4\sigma$.

Special Signals

square(t,f,d)	square wave with frequency f and duty cycle d (in %), with amplitude 1
triangle(t,f)	triangular wave with frequency f and amplitude 1
sawtooth(t,f)	sawtooth wave with frequency f and amplitude 1
pulse(t,b,T)	series of pulses with width b, distance T between the pulses and amplitude 1

Examples

1. Sine wave: $a*\sin(w*t)$ oder $a*\sin(wt)$ oder $a*\sin(2*\pi*f*t)$
2. Sine wave with noise: $a*\sin(wt)+white()*a/2$
3. Repeating damped sine wave: $a*\sin(wt)*exp(-f*\text{mod}(t,10/f))$. The damping is achieved by an exponential function, which is reset every 10 periods.
4. Lissajou-Figure: channel 1: $a*\sin(3*wt)*\sin(4*wt)$ and channel 2: $a*\sin(4*wt)$

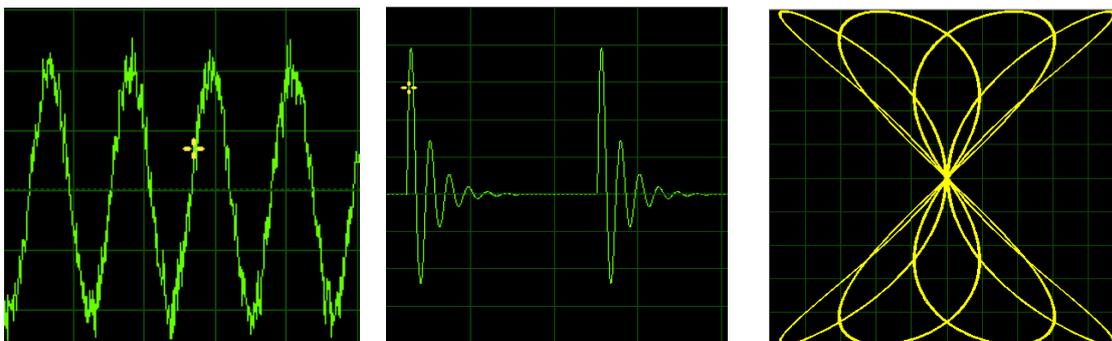


Figure 14: Signals from examples 2, 3 and 4 (XY)

3.9 Settings

In this window, there are some settings for the Windows audio devices. On the right side are the audio devices for sound input and output. If several sound systems are present, the equipment used can be selected here. Checking the “auto enable” option will unmute the selected device automatically as well as setting the volume to the given value in percent.

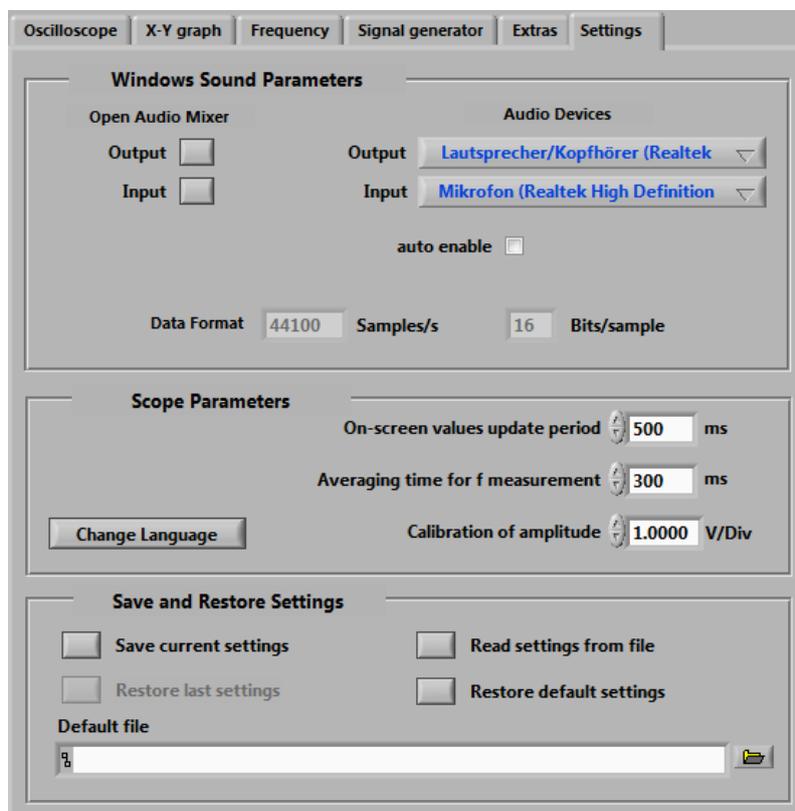


Figure 15: The settings tab contains the windows sound settings and general settings of the program

On the left side are buttons to start the Windows audio controls. Note that Windows XP/2K will open a small mixer window for each push of the button. In case of newer Windows Versions, only a single windows can be open at a given time.

The shown sampling rate and the bits per sample is given for information only. In order to change the sampling rate, refer to the expert section below.

Some attributes of the scope can be controlled by changing three values under “Scope Parameters”. The update period will change the time between consecutive on-screen values (frequency and/or amplitude values). At the same time the averaging time to calculate the on-screen value is changed as well. The latter happens for the calculation of the strongest frequency on the frequency analysis screen, when changing the second value. Calibrating the actual amplitude of the scope can be achieved by means of a sine signal with a known amplitude (measured in volts) and dividing this value by the measured amplitude.

The language of the program can be with the corresponding button. The change of the language will be applied at the next startup of the program



Figure 16: Language selection window

At the bottom of the settings window the settings of the program can be saved to a XML file, restored or reset to default values.

3.9.1 Loopback devices

A special case of listed sound devices are loopback devices. These capture the data played by an output (e.g. speaker) device. This is useful to capture the sound played by a media player. Be aware, that DRM protected sound (e.g. from a protected DVD) is NOT captured.

3.10 For Experts: changing the configuration

The behavior of the scope can be changed by modifying or adding parameters in the file *settings.ini*. The file is located in the path %appdata%\scope, which usually is C:\Users\\Appdata\Roaming\scope. If this does not exist, after the first start of the program, check the path %userprofile%\scope.

The first option in the file should be enabled (ExpertMode=TRUE) in order to have a fast access to this file via a button on the settings tab of the program. In addition this will enable the change of the sampling rate and bits sample within the program.

The standard settings for the soundcard are 44.1kHz with 16Bit resolution per sample. Higher sampling rates and sample resolutions can be set by setting the corresponding parameters "SamplingRate" and "Bits", or in expert mode within the program. Most current soundcards (even onboard versions) support up to 100kHz and 16Bit and higher. If the soundcard does not support the sampling rate and/or bit resolution, a different device, or the internal loopback, will be selected by the program. If only parameters supported by the hardware of the sound card should be used, the option "AcceptRate" should be set to the value "DEVICE". The Windows mixer provides a wider selection of sampling rates and bits per sample, but might resample the audio stream. If the mixer support is wanted, set "AcceptRate" to "MIXER".

An additional parameter in the file *settings.ini* is the "MaxFrequency", which determines the maximal value for the displayed frequency in the Fourier analysis. The default value is half the sampling rate.

The length of data which is analyzed by the Fourier analysis is by default 120 ms long. This allows to observe frequencies down to approximately 20Hz. If you want to measure lower frequencies you can add the option "FourierTimeWindow=500" into the *settings.ini* file. The number gives the sample length in milliseconds. The frequency resolution is approximately given by the inverse of the length of the analyzed time window. Be aware, that a large number slows down the update of the Fourier analysis substantially and requires more CPU cycles.

Some sound cards invert the input signals before the digitization. This can be corrected by adding the option "InvertSignalCH1=true" and "InvertSignalCH2=true" in the *settings.ini* file.

In order to have a reasonable screen resolution when zooming in by a large factor, the resolution can be increased by setting MaxSamplesScale to a value up to 100. This will increase the load on the system-

Addition information:

The list of available devices contains only devices, which support the selected sampling rate and bits/sample.

Here an example for a settings.ini file:

```
Expert_Mode=TRUE
SamplingRate=100000
Bits=16
AcceptRate=mixer
MaxFrequency=20000
InvertSignal=TRUE
FourierTimeWindow=200
MaxSamplesScale=50.0
```

3.10.1 Signal Sources for the Oscilloscope

The following inputs are usually available:

- **Line-In** Port on the PC
- **Microphone** Port on the PC, or internal (e.g. Laptop) – often only mono
- **Wave Out** internal sound, e.g. MP3 player, Media-Player; signal generator
- **CD Player** Music directly from a CD
- **Loopback of output devices** signal from playback devices (e.g. media player)
- **Scope Loopback** Pseudo device, which directly couples the generator output to the scope.

The equipment to appear on the oscilloscope must be selected from the inputs mentioned above.

The simplest case is the "Scope Loopback" device (in- and output). The signals of the generator are directly displayed on the scope screen. With some sound cards, several sources can be selected at the same time. The volume of the equipment can also be adjusted here. This has a direct effect on the amplitude of the oscilloscope! In Windows Vista/7/10 at least a single recording device has to be active. A green check mark

appears for these devices in the sound control panel. If no recording device can be found, the scope will switch to the internal “Scope Loopback” device.

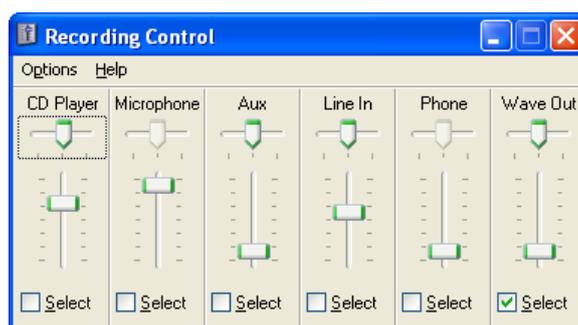


Figure 17: Windows XP - Selection of inputs in Windows Audio Mixer

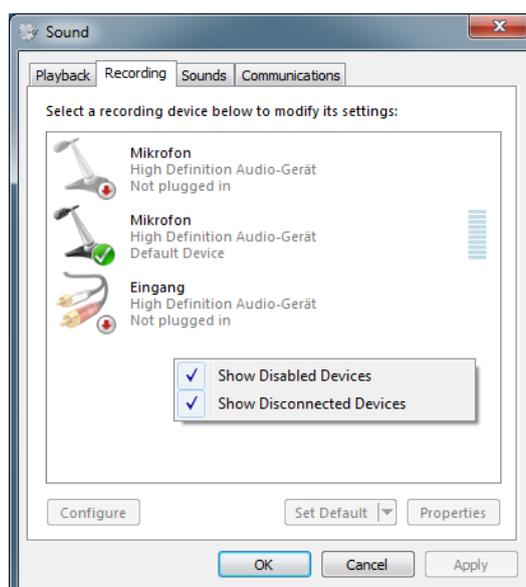


Figure 18: Windows 7 – a microphone is active (green check mark). Inactive devices are not visible, but can be included in the list by checking the corresponding context menu entry (right mouse button)

3.10.2 Signal Output via Sound Card

In order to define which sound is sent to the sound card output, the appropriate equipment must be selected in the Windows Audio Mixer. Frequently several sources are merged at the same time here.

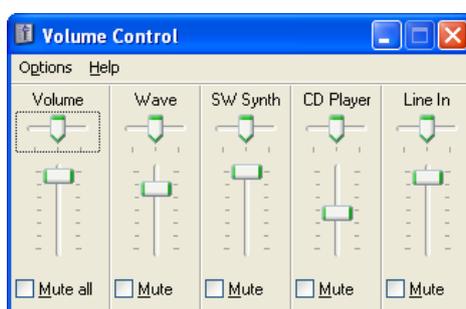


Figure 19: Selectable outputs

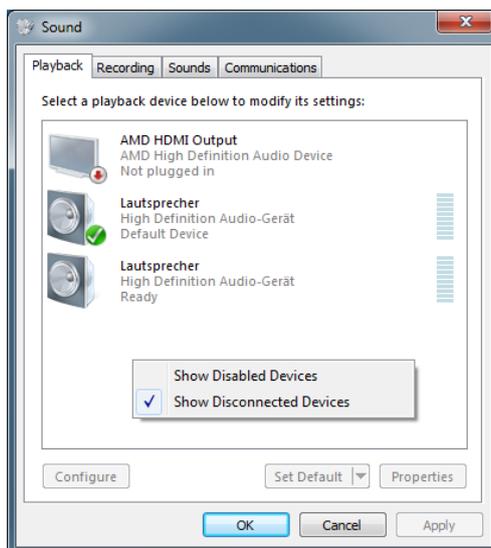


Figure 20: Windows 7 – a speaker is active (green check mark). Inactive devices are by default not visible, but can be included in the list by checking the corresponding context menu entry (right mouse button).

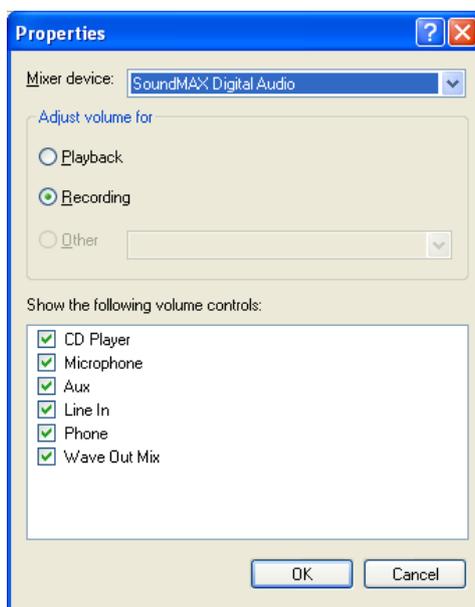


Figure 21: Properties of audio input devices

Important:

Windows XP: It can sometimes occur that an input or an output is not listed in the window. In this case it must be activated under → Options → Properties.

Windows 7: Inactive devices do not show up by default. Click on “Show Disabled Devices” in the context menu to make them visible.

3.10.3 Arduino as a signal source for the Scope

An Arduino board can be used as a signal source. This requires a special operating program of the Arduino. An example program is enclosed in the installation package of the Scope (see installation directory under Arduino/scope.ino). This program samples two analog voltages on the input pins A0 and A1 and digital values from pins D3 and D4. The sampling rate is 10kHz with a selectable resolution of 8 or 10 bits. In order to adapt the voltages to defined levels, additional hardware (operational amplifier) is required.

The communication is done via the USB connection, which shows up on the PC side as a serial port (Windows COM port).

For the automatic detection of the Arduino, an entry in the configuration file `usb_serial_devices.ini` is required. This entry states the USB vendor and product ID and the parameters of the device (sampling rate and voltage range).

A detailed description how to program and utilize an Arduino board is enclosed in the installation directory under `arduino/arduino.pdf`

3.10.4 Audio Recorder

The Audio Recorder allows to save data to a Wave file. The name of the output file has to be selected before the Pause or Record button is pressed.

Three different modes are available to store data:

1. **Trigger (auto)** Save automatically the currently triggered data
2. **Trigger (manual)** Manually save the last triggered data to the file
3. **Rec. Button** Start the writing to the file with the record button (independent of the trigger)

Independent of the mode only a limited chunk size is written to the output file. The length is defined by the corresponding selectors in the Recorders window. The length is by default defined by the oscilloscopes window, but can be set by the user to a different value (uncheck the Automatic box). In all cases the writing will stop, when Pause or Stop is pressed.

Be aware, that the selected file will be overwritten **WITHOUT** any warning! Since the current file will be closed after the Stop button has been pressed, define a new output file **BEFORE** pressing Pause or Record! The resulting Wave file will contain 100 samples of silence between the recorded data chunks. Cue points at the beginning of the Wave file mark the start of each written chunk.

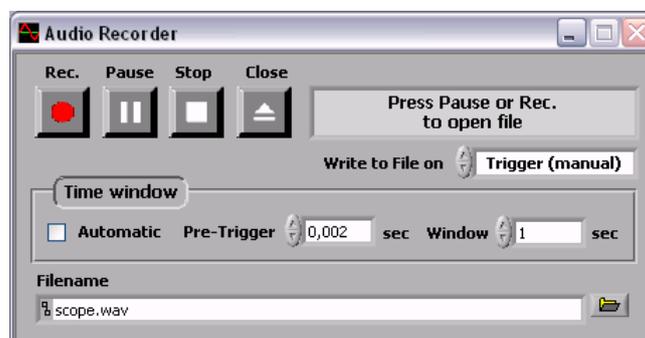


Figure 22: Audio Recorder window

4 Licensing the Program

The use of the software and of the documentation is granted free of charge for private and non-commercial use in educational institutions.

4.1 Commercial Usage

A commercial usage of the program is only allowed, if a corresponding license has been obtained. Such a license is available from <http://www.shareit.com/product.html?productid=300341159> . More informations are available on the products web-page https://www.zeitnitz.eu/scope_de .

4.2 Private Donation License

A private donation licence is available at <http://www.shareit.com/product.html?productid=300341154> . By purchasing such a license you can support the further development of the soundcard scope..

After obtaining a license, you will find the license key on the web-page as well as in the e-Mail you will receive. In order to license the program, follow these steps:

- Re-start the program
- On the windows with the copyright message press the button „License Program“

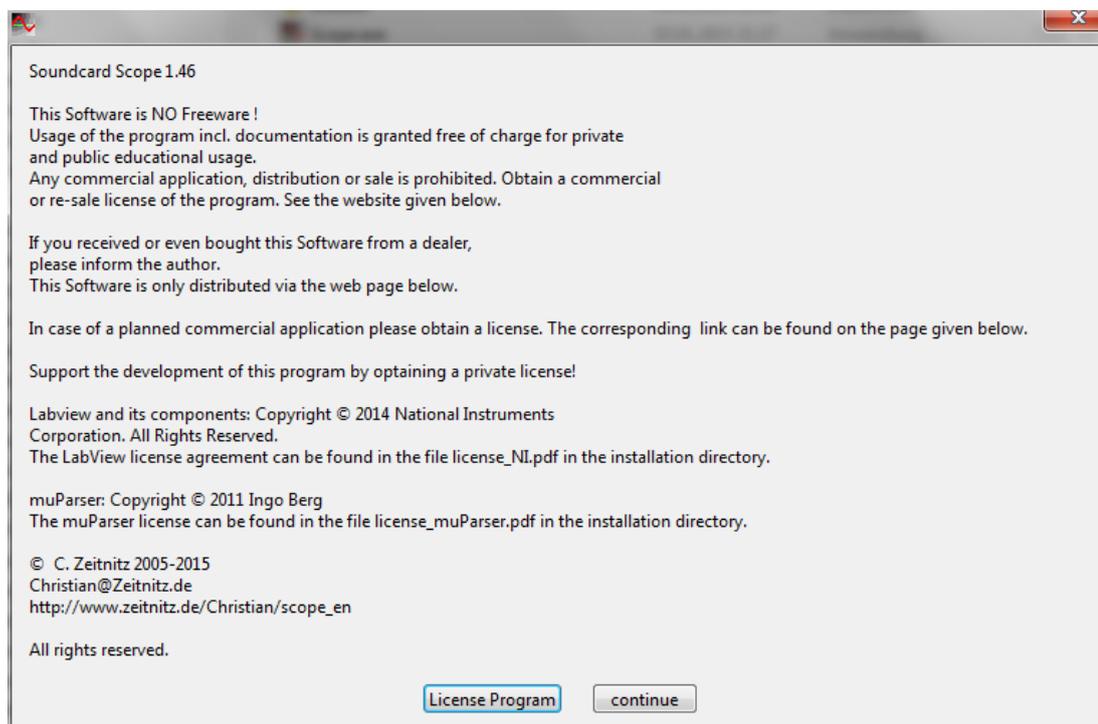


Figure 23: Press the button „License Program“

- The next window allows to enter the license code
 - copy-paste either directly from the e-mail or
 - select the file which contains the license code



Figure 24: Either paste the license code into the first fiels, or select the file with the code

- After the license process is complete the program will start regularly

After the program has been licensed, the screen with the copyright message will no longer appear when starting the program

The license can only be removed by deleting the file „license.txt“ from the settings directory C:\Users\\Appdata\Roaming\scope. This path is the standard path for Windows 7/8.

5 Unattended installation

In order to install the program over the network, the file scope.exe has to be unpacked utilizing 7-Zip. Afterwards the installation is done by the command “msiexec /i install.msi /quiet“. To uninstall use the command “msiexec /uninstall install.msi /quiet“.

6 Conclusion

I hope you will have a lot of fun with this program. If something goes wrong and you discovered a bug, please send mail to info@zeitnitz.eu.

If you use the program for a project at and university or school I would like to know about it.

This program can be used and passed on for use within the school and private sector freely. For planned commercial use please contact info@zeitnitz.eu.

7 Trouble Shooting

Certainly this program might still have some errors, however some standard problems are caused by the sound card and/or Windows.

An error message is shown when starting the program

The error message, that the installed LabView Run-Time engine (version 7.1) is incompatible with the required one (version 7.1.1), is caused by a previously installed LabView Run-Time engine. In order to get the program running, uninstall the Run-Time engine (Control panel → Add/Remove Program → National Instruments) prior to a re-installing this software.

No soundcard is found

Check in the hardware manager, that Windows actually has a soundcard correctly installed. Some soundcards recognize if speakers or a microphone is installed. This is the default behaviour under Windows Vista/Win7. In this case you have to check, that at least one input/output device is enabled in the sound settings (green check mark). If no output device is enabled, the program will complain about it and terminate immediately.

The oscilloscope shows no signal and the display is frozen

Unfortunately it sometimes happens that communication with Windows breaks down. Here only terminating and re-starting the program helps!

No signal on the oscilloscope

If the signal generator is used and a channel is also SWITCHED ON, the user must select "Wave Out" for the audio mixer equipment .

No sound audible

In order that a signal on the speaker is audible, the appropriate equipment must not be deactivated. In this case check the audio mixer and enable the appropriate device. When using the signal generator, "Wave" must be selected.

Strange jumps in the signal

A large signal can overdrive the input. The maximum possible value should actually be sent to the output. With some sound cards, however, this leads to an overflow and instead of a large *positive* value, a large *negative* value is sent, which leads to a complete distortion of the signal. If such jumps are observed, the input signal should be attenuated

Program reacts very slowly

The CPU load on a slow computer (less than 1GHz) can go up to 100%, especially in the frequency analysis mode. The program will only react slowly. A solution is to reduce the amount data to be processed by changing the sampling rate in the file *settings.ini*. For this uncomment the line with the key word SamplingRate and put in a value of 22050. This corresponds to a reduction of the amount of data by a factor of 4.

In the XY-view the persistence setting has a strong impact on the system load. You might have to reduce the persistence time to obtain again a more responsive system.

Where to find *settings.ini* ?

The default path on Windows 7/8 is `C:\Users\\Appdata\Roaming\scope`

To verify the path on your machine, either type in a cmd window `dir %appdata%\scope`, or type `%appdata%\scope` into the header line of the Windows Explorer. In both cases the content of the directory is displayed, if existing. If the above path does not exist, try `%userprofile%\scope` instead.

Changes to *settings.ini* has no effect

Only the configuration file `settings.ini` in the directory `C:\Users\\Appdata\Roaming\scope` is actually used by the program. Delete the folder in order to reset to the files in the installation directory of the scope.