

VituixCAD

The Loudspeaker Simulation Tool

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Online Manual

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VituixCAD

General information

VituixCAD is loudspeaker simulation software. Design philosophy is to simulate loudspeaker behavior in full space. Even though emphasis is on power response, polar responses and directivity index, it is possible to design a loudspeaker without comprehensive angled measurements. This is not encouraged though.

VituixCAD offers possibility to design and simulate up to 6-way loudspeaker. Each way can have up to 4 drivers in different configurations.

Software package includes everything one needs for simulating and designing a loudspeaker. Additional to simulator itself, there is enclosure simulator, response merger and advanced calculator tool included.

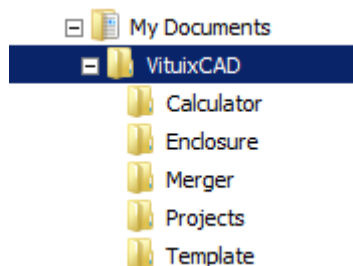
This document is divided into three sections; general information about the software, quick user guide and detailed descriptions of tools, views and theory behind the software.

System requirements

- VituixCAD is tested on Windows XP, 7, 8 and 10.
- [Microsoft .NET Framework 4](#) or newer. Tested up to 4.7.1.
- Minimum screen resolution is 1024x768 (4:3) or 1280x720 (16:9), but 1600x900 or more is recommended.

Installation and Upgrade

Folders of User files after installation:



Command line arguments (optional)

VituixCAD.exe <i>"path\filename"</i>	Opens VituixCAD project file ending with ".vxp"
VituixCAD.exe <i>Calculator</i>	Starts main program and Calculator tool
VituixCAD.exe <i>Enclosure</i>	Starts main program and Enclosure tool
VituixCAD.exe <i>Merger</i>	Starts main program and Merger tool

Support

Press F1 for help. The latest [online manual](#) is opened to your default browser from internet. Add bookmark to the browser for fast access to the latest revision.

You can save pdf document in (My)Documents\VituixCAD folder to speed up loading or if internet connection is not continuously available. Notice that possible updates won't reach you via F1 key as long as local manual is found in the folder.

Quick User Guide

Preface

This user guide is a chronological walkthrough on how to design a loudspeaker with VituixCAD. Commonly, design process starts with deciding enclosure size, drivers, radiator type, alignment etc. Enclosure tool is used for simulating enclosures, different radiator types and alignments. Next step is to have comprehensive set of acoustic and electrical measurements of the construction. Merger tool is used for combining far field and near field responses. After these prerequisites are met, simulation phase can be started. Whether Your goal is to design a speaker with or without interim listening tests, You'll need quality control of some sort. Some prefer their ears, some prefer measurements. If a loudspeaker measures perfectly, but sounds worse than anything You've ever heard, something is terribly wrong. This guide will not teach You how to listen a loudspeaker, but will cover basic QC steps and VituixCAD Calculator, which can be used for various calculations and manipulations for measured data.

Building prototypes and crossovers are not covered in this guide. This guide will also assume You have suitable measurement gear and software and understanding about how to measure loudspeaker drivers for design purposes. More detailed description of tools and views are provided in [Detailed descriptions](#) part of this document.

Checklist for designing a loudspeaker

Investigate acoustic parameters, dimensions, materials and speaker placement possibilities of the listening room. It is wise to fix issues of bad environment (the room) first rather than trying to handle everything with massive and complex speaker design.

Basic engineering

- Decide acoustic design; type and amount of directivity, radiator types, ways, driver size and count.
- Estimate possible sensitivity range and crossover frequency ranges.
- Select initial drivers and directivity components to reach previous targets
- Simulate low frequency radiators with [Enclosure tool](#)
- Simulate [baffle diffraction](#) and export cabinet impact response
- Design the cabinet.

Construction

- Build flexible prototype or final cabinet depending on uncertainties in the design
- Connect temporary cables to individual drivers or driver groups for acoustical and electrical measurements.

Measurements

- Prepare turning table for polar response measurements. Manual turning table is easy to make and fast to use for example with Clio 10/11 or ARTA 1.9.
- Choose directions for off-axis measurements. You are not forced to measure full or half circle around the speaker with constant 5 or 10 deg steps, but it is highly recommended. Simulation is possible with less than 10 directions per axis. Don't waste Your chance to get all measured data at once. There is no need to measure vertical axis if vertical measurement would be "equal enough" to horizontal measurement. For example full range driver in the center of square box is symmetrical in both directions.
- Measure polar response of each driver or driver group as far field measurement. Use equal off-axis angles for all drivers. Don't let measuring program to corrupt timing: use semi-dual (or full dual) channel measurement to lock time reference to mic capsule. Measure time-windowed responses from the same or at least known/measured distance to the reference point to maintain common time reference with different ways & drivers.
- Measure near field responses of midrange and woofer cone(s) and port(s) if anechoic environment is not available. Arrange radiator to half space to avoid baffle loss. Some amount of baffle loss exists with small cabinets in full space even if measurement distance is less than 8 mm. Use same output voltage with far field measurement (if possible without clipping or excessive distortion) to help merging of near field and far field measurements.
- Measure impedance responses of each driver or driver group.

Merge and manipulate response data

- Merge far field and near field responses with [Merger tool](#) if you didn't measure low frequency radiators (<300 Hz) from far field in anechoic environment.
- Include cabinet impact response (from diffraction simulator) in near field responses.
- [Export merged responses](#) as separate txt-files (or as extended data file if smoothing is not needed).
- Smooth responses with [Calculator tool](#) if necessary.

Simulate loudspeaker with VituixCAD

- [Create new empty project](#) and enter *Description*
- [Insert number of ways](#)
- [Select number of drivers and electrical connection](#)
- [Enter driver names, locations and possible rotation or inclination](#)
- [Insert frequency responses](#)
- [Insert impedance response](#)
- Outline rough targets for [axial response](#) and [power response](#)
- Outline rough targets for [axial responses per way](#)
- Design the crossover
 - Insert [filter blocks](#) manually or by the [Wizard](#)
 - Adjust [parameter/component values](#) of filter blocks manually or by the [Wizard](#)
 - Play with circuit topologies and parameter/component values until [axial response](#), [power response](#), [directivity index](#), [polar responses](#) and [impedance response](#) meet Your targets.
 - [Save](#) project periodically. [Save as...](#) most promising intermediate results.

Built and install crossover

Quality control

- Mandatory QC-measurements
 - Angled measurements in horizontal and vertical planes, at least 30 deg steps
 - Impedance response
 - Listen to Your favorite tracks

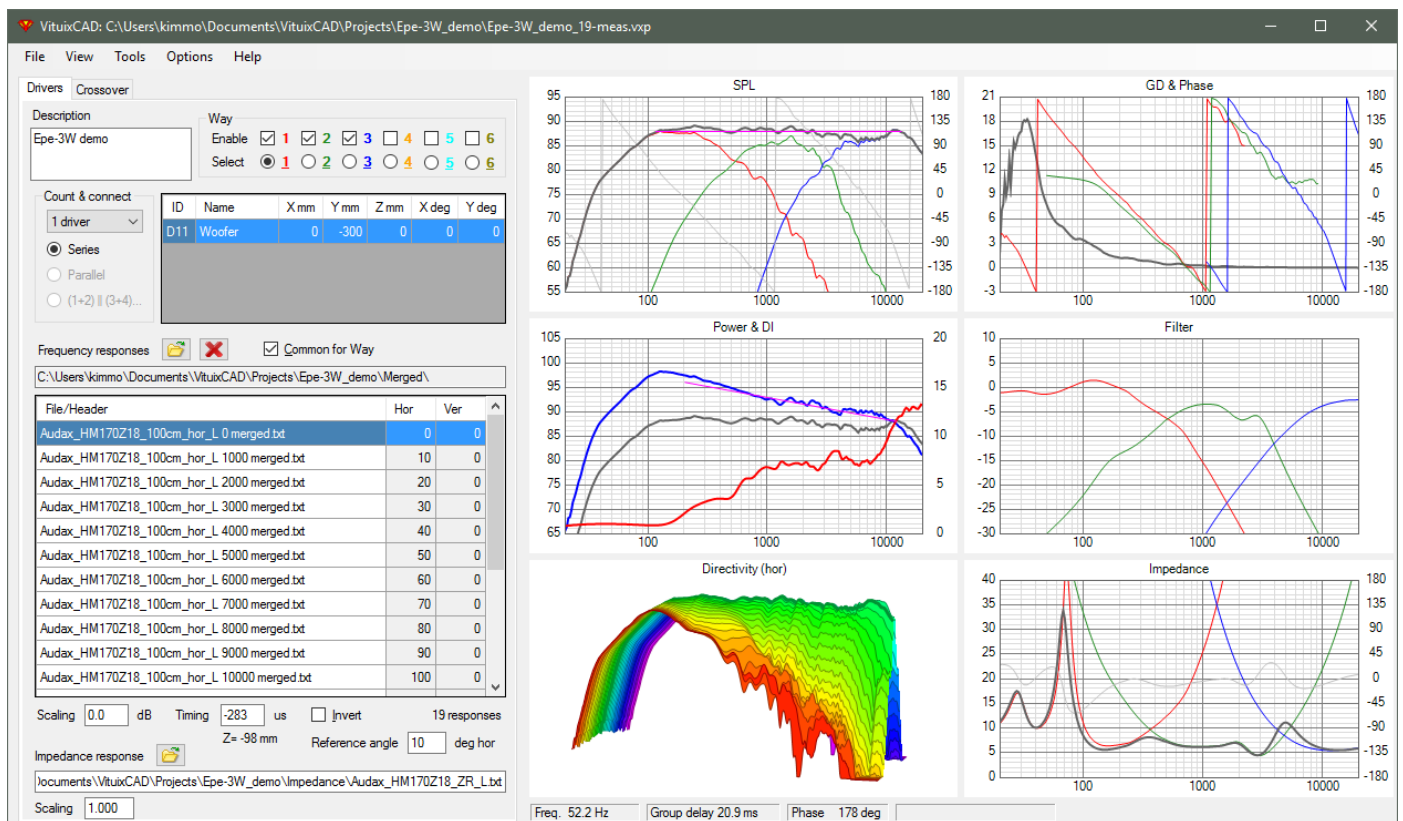
If you are not satisfied -> back to drawing board

Additional QC-measurements

- Excess group delay
- Harmonic distortion
- Intermodulation distortion
- Compression
- Acoustic compatibility to your listening room; room response, clarity parameters.

Detailed descriptions

Main window



Main window is divided into two sections, control section on the left and dashboard (graphs) on the right.

Menus

File

New	(Ctrl+N)	Create new empty project
Open	(Ctrl+O)	Open existing project
Recent >		Open previously opened project
Save	(Ctrl+S)	Save current project
Save as...		Save current project with a different filename
Export >		Export of Frequency response, Frequency response of Way, Power response, Directivity Index, Impedance response, Group delay, Filter gain of Driver, Impedance response of Way, Polar frequency responses, Biquad coefficients of Way
Exit		Exit VituixCAD

View

Crossover schema	(Ctrl+F)	Open crossover schematic window (preview only)
Optimize	(Ctrl+T)	Open SPL Target setting and Optimizer window
Parts list	(Ctrl+L)	Open Parts list window
Impulse response	(Ctrl+I)	Open Impulse response window for preview and export
Power dissipation	(Ctrl+P)	Open Power dissipation window
Units		Show engineering units (Ohm, F, H) in schema windows
Part #		Show part numbers instead of values in schema windows

Tools

Enclosure	(F3)	Open Enclosure tool
Merger	(F4)	Open Merger tool
Calculator	(F5)	Open Calculator tool
Diffraction	(F6)	Open Diffraction simulator
SPL Trace	(F7)	Open SPL Tracer tool

Auxiliary	(F8)	Open Auxiliary calculator
Options		
	(Alt+O)	Open Options window
Help		
Online Manual EN	(F1)	Open user manual in English
Online Manual DE		Open user manual in German
About		Information about VituixCAD. View changelog.

Drivers tab

Ways

Way

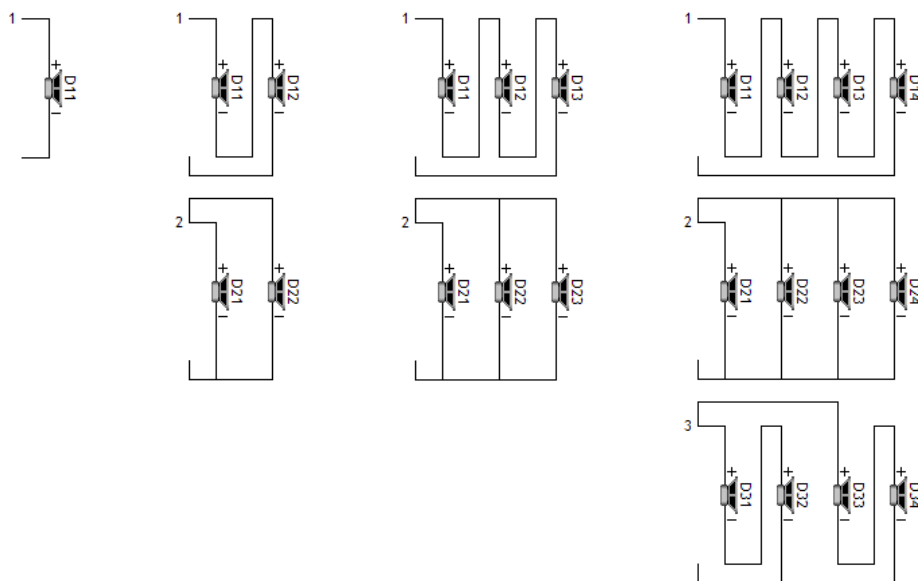
Enable ☒ 1 ☒ 2 ☒ 3 ☐ 4 ☐ 5 ☐ 6

Select ☒ 1 ☐ 2 ☐ 3 ☐ 4 ☐ 5 ☐ 6

Enable checkboxes control how many ways is in the project. If You are building a two way loudspeaker, check 1 and 2. First way is usually the lowest frequency range. Ways are possible disable while design process without losing entered data. This enables focusing to a single way after whole project is built up. Unused ways could also save alternative designs which can be swapped with primary design for comparison.

Select radio buttons control which way You are currently working on.

Count & connect



VituixCAD supports up to 4 drivers per way. Available configurations are series (1), parallel (2) and series-parallel (3).

Driver list

Each driver can be provided with name and location relative to “design origin”. Design origin is typically endpoint of listening axis on front baffle surface. $X [mm]$ is horizontal coordinate of center point; negative to left and positive to right. $Y [mm]$ is vertical coordinate; negative down and positive up. $Z [mm]$ is horizontal distance coordinate; negative closer to mic and positive further from mic.

Horizontal rotation $X [deg]$ or vertical inclination $Y [deg]$ of drivers is also supported, but angle should follow measured directions because VituixCAD does not interpolate frequency responses between off-axis angles.


Rotation $X [deg]$ is positive to right, and inclination $Y [deg]$ is positive to up.

Multiple drivers should be entered as a single driver if they are measured in the prototype cabinet as a package; all connected to power amplifier at the same time. Location is entered as a difference between measurement and design origins.

Example 1: Location (X,Y,Z) = (0,0,0) if multiple driver package is measured on design (listening) axis.

Example 2: Location (X,Y,Z) = (0,-400,0) if multiple driver package is measured 400 mm below design (listening) axis.

Frequency responses

Add driver's frequency responses by clicking folder button or dropping files into response list. You can use individual off-axis responses or LspCAD extended format file created with Merger tool. Delete button  clears response list. Drivers #2-4 are able to use all measurements of driver #1 by checking *Common for Way*.

Maximum of 74 frequency response measurements per driver is supported. Loaded responses are verified against other ways & drivers. **Directions which are common for all enabled ways and drivers are included in simulation.**

C:\Users\Kimmo\Documents\VituixCAD\Projects\Epe-3W_demo\Far\

File/Header	Hor	Ver
Vifa_M10MD-39-08_100cm_hor_L 0.txt	0	0
Vifa_M10MD-39-08_100cm_hor_L 500.txt	5	0
Vifa_M10MD-39-08_100cm_hor_L 1000.txt	10	0
Vifa_M10MD-39-08_100cm_hor_L 1500.txt	15	0
Vifa_M10MD-39-08_100cm_hor_L 2000.txt	20	0

C:\Users\Kimmo\Documents\LspCAD\data\Epe\20140113\Extender\Vifa_M10MD-39-08_100cm_hor_L 0.txt

File/Header	Hor	Ver
Axial response	0	0
## EXTENDED DATA ## OFF AXIS 5,0	5	0
## EXTENDED DATA ## OFF AXIS 10,0	10	0
## EXTENDED DATA ## OFF AXIS 15,0	15	0
## EXTENDED DATA ## OFF AXIS 20,0	20	0

Scaling dB Delay us ☐ Invert 19 responses
Z= 0 mm Reference angle deg hor

Frequency responses can be scaled (dB), delayed ($\pm \mu\text{s}$) and polarity can be inverted with controls below frequency response list. *Hor* and *Ver* angle can be modified by entering new value to the field if program fails to parse value from the filename or measurements are swapped intentionally.

Reference angle is direction in horizontal plane which is shown as axial response in SPL, Power & DI and Phase response graphs. Also directivity index calculation is using *Reference angle* as main axis. Optimizing to single off-axis direction is useful if axial response is bad or not representative or measurement data is poor and accurate power response approximation is not available. Default value is 0 deg hor.

Impedance response

Select impedance response file for a driver(s) by clicking folder button or dropping file into text box.

Impedance response can be scaled as well with a multiplier. If multiple drivers are entered as a single driver, scaled impedance response should represent total impedance of the driver package.

Supported frequency and impedance response file types

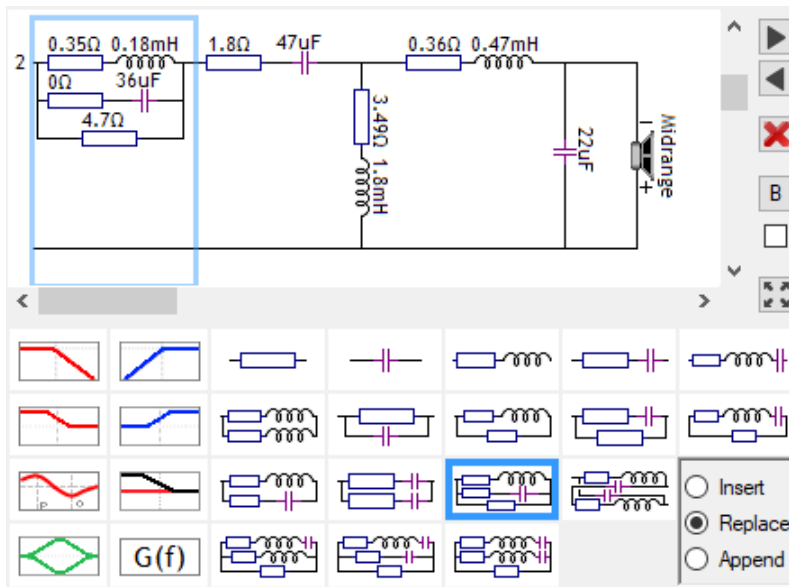
VituixCAD supports tab, space or semicolon delimited .txt or .frd or .zma (for impedance). Following software exports are supported:

- AudioTools
- ARTA, LIMP
- Clio
- Edge
- FRD tools

- HOLM Impulse
- justMLS, LspCAD 6 extended format
- Klippel
- LspLAB
- REW
- SoundEasy
- XSim

Crossover tab


Filter schema



Maximum of 15 filter blocks can be assigned for each way via *Crossover block menu* or a wizard. *Drag & drop* from menu to schema is available.

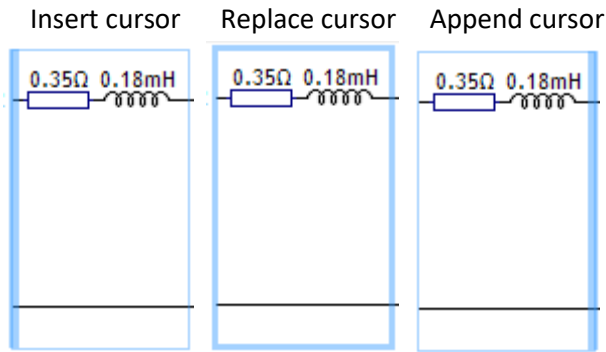
Filter blocks can be moved by *Drag & drop*, or forward and backwards within the same net by clicking arrow buttons on the right. Single block can be copied by *Ctrl + Drag & drop*. Selected block can be removed by clicking delete button **X**. Filter network can be deleted by pop-up menu (right click). Network can be Copy - Pasted to another way or project by pop-up menu. Selected block can be bypassed by checkbox below *B* button on the right. Bypass status of all blocks on the selected way is inverted by clicking *B* button. Bypassed blocks are grayed in schema view.

Up to ten most recent crossover changes and situations before parameter adjustment can be restored with **Undo** (*Ctrl+Z*) command in context menu of filter schema. Way settings such as gain, delay and invert, and every single parameter change is not saved to undo buffer.

Schema view can be expanded over Block menu, Connection menu and Wizard by clicking expand button . Schema window showing total crossover without bypassed blocks and disabled ways can be opened by selecting *View→Crossover schema (Ctrl+F)*.

Insert/Replace/Append mode

Filter blocks are placed in the schema according mode selection. In *Insert mode* new blocks are placed under the cursor without overwriting existing blocks. *Replace mode* overwrites the block under the cursor. *Replace mode* activates automatically when any filter block in the schema is clicked. In *Append mode* new blocks are added right after currently selected block. You can toggle between modes by radio buttons or pressing *Insert* key when mouse cursor is over schema.



Filter block types

Active Low pass and High pass filters (f). Shape and Order are selected from list boxes:

- 1st order
- 2nd order (Q)
- Linkwitz-Riley 12...48 dB/oct.
- Bessel 12...48 dB/oct.
- Butterworth 12...48 dB/oct.
- Chebyshev 0.5dB 12...48 dB/oct.
- 1st order [linear-phase](#)
- 2nd order [linear-phase](#) (Q)
- Linkwitz-Riley [linear-phase](#) 12...48 dB/oct.
- Horbach-Keele [linear-phase](#) (R)
- Horbach-Keele "MTM" [linear-phase](#) (R)
- Brickwall [linear-phase](#)

Active Shelving Low pass and High pass filters (f, gain). Shape is selected from list box:

- 1st order
- 2nd order (Q)
- 1st order [linear-phase](#)
- 2nd order [linear-phase](#) (Q)

Active Linkwitz-Transform (fo, Qo, fp, Qp)

Active All-pass filters (f). Shape and Order are selected from list boxes:

- 1st order
- 2nd order (Q)
- 1st order [phase linearization](#)
- 2nd order [phase linearization](#) (Q)
- Linkwitz-Riley [phase linearization](#) 12...48 dB/oct.
- Bessel HP [phase linearization](#) 12...48 dB/oct.
- Butterworth HP [phase linearization](#) 12...48 dB/oct.

Active Peak/Notch filters. Shape is selected from list box:

- Parametric EQ (f, Q, gain)
- [Gain EQ](#) (f, Q, gain)
- [Phase EQ](#) (f, Q, phase)

Transfer function file, any supported response file type.

Passive R1

Passive C1

Passive R1+L1

Passive R1+C1

Passive R1 || C2

Passive (R1+C1) || R2

Passive (R1+L1) || R2

Passive (R1+C1) || (R2+C2)

Passive (R1+L1) || (R2+L2)

Passive (R1+L1) || (R2+C2)

Passive (R1+L1) || (R2+C2) || R3 (*Parallel notch*)

Passive R1+L1+C1 (*Series notch*)
 Passive (R1+L1+C1) || R2
 Passive Lattice all-pass
 Passive (R1+L1+C1) || (R2+L2) || R3
 Passive (R1+L1+C1) || (R2+C2) || R3
 Passive (R1+L1+C1) || (R2+L2+C2) || R3

Important! Active filters having [blue text](#) are NOT minimum-phase. Blocks in the schema views have 'FIR' text for information. Convolver plugin or DSP device with FIR features is needed for real life application. Transfer function of active filters per way can be exported as impulse response in wav or txt file format. See section [Impulse response](#).

Biquad coefficients

Active IIR blocks can be exported or copied to clipboard as digital biquad filter coefficients b0, b1, b2, a1, a2:

$$H(z) = \frac{b_0 + b_1 z^{-1} + b_2 z^{-2}}{1 + a_1 z^{-1} + a_2 z^{-2}}$$

Format is compatible with miniDSP Xover/PEQ Advanced view. For example 3rd order Butterworth LP 1000 Hz:

```
biquad1,
b0=0.004015505022858,
b1=0.008031010045716,
b2=0.004015505022858,
a1=1.86140844453211,
a2=-0.877470464623539,
biquad2,
b0=0.061511768503622,
b1=0.061511768503622,
b2=0,
a1=0.876976462992757,
a2=0,
```

Biquads of Way can be exported as text file with *File -> Export -> Biquad coefficients of Way*.

Coefficients can be copied to clipboard with context menu of filter schema (right click):

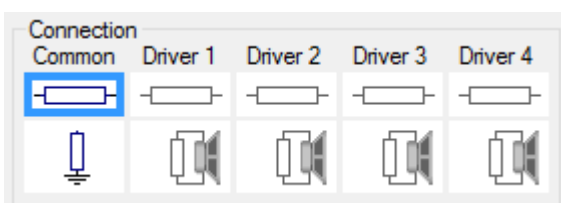
Copy Biquad coefficients of selected Block (Ctrl+B)
Copy Biquad coefficients of Way (Ctrl+W)

Note! Select correct sample rate from [Impulse response](#) window before copying/exporting biquads. FIR, transfer function file, passive and by-passed blocks are ignored. Stability of biquad filters is not checked.

Connection

Active blocks are always common for all drivers as they are placed before power amplifier. The amplifier is connected to passive network and drivers. Location of passive block is selectable via *Connection* menu:

- series in common (way) net
- shunt in common (way) net
- series in driver's net
- shunt in driver's net

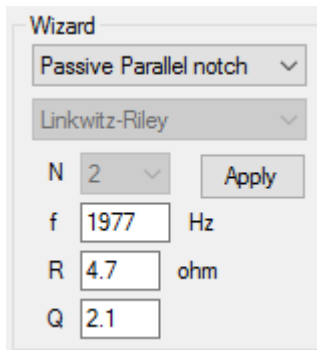


In common net, filter is connected in parallel (shunt) or in series with all drivers. In driver's net, filter is connected to the particular driver.

Passive filter block can be connected as shunt by pressing *Shift* key while dropping or adding block into schematic. This eliminates extra clicking of Shunt button (in Connection group) when adding shunt blocks.

Filter block can be dropped also into Connection group to assign block to final driver's net. This eliminates extra clicking of Series/Shunt D1..4 button when adding blocks into driver nets.

Wizard

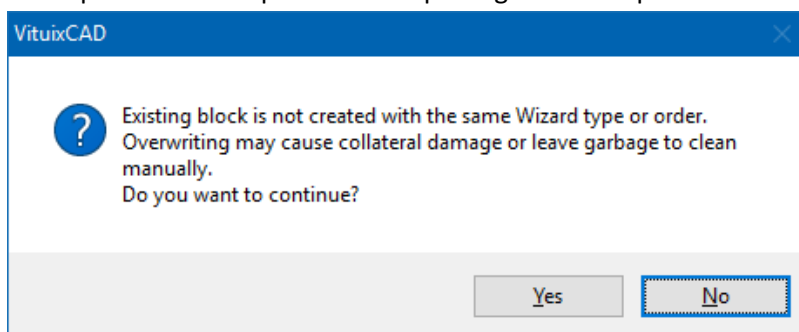


Wizard can be used to add or replace passive filter blocks or update parameters. The following filters types are available through the Wizard:

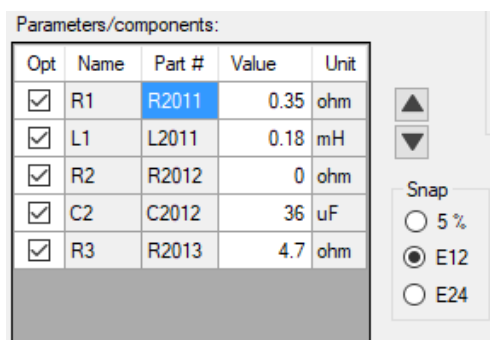
- Passive High pass; 1st...4th order Butterworth, Bessel, Chebychev 1.0dB or Linkwitz-Riley (f, ZL)
- Passive Low pass; 1st...4th order Butterworth, Bessel, Chebychev 1.0dB or Linkwitz-Riley (f, ZL)
- Passive Parallel notch (f, R, Q)
- Passive Series notch (f, R, Q)
- Passive Lattice all-pass (f, RL)
- Passive Zobel (Re, Le)
- Passive L-pad (ZL, gain)

Other types should be added manually to the block list.

Filter blocks created by the Wizard are placed according [Replace/Insert/Append mode](#). Higher order passive filters and L-pad need multiple blocks. Replacing with multiple blocks may require confirmation:



Parameters/components



Opt	Name	Part #	Value	Unit
<input checked="" type="checkbox"/>	R1	R2011	0.35	ohm
<input checked="" type="checkbox"/>	L1	L2011	0.18	mH
<input checked="" type="checkbox"/>	R2	R2012	0	ohm
<input checked="" type="checkbox"/>	C2	C2012	36	uF
<input checked="" type="checkbox"/>	R3	R2013	4.7	ohm

Selecting filter block opens corresponding parameters/components to the list. Component values can be entered directly to the *Value* field. Additionally, component value can be increased/decreased by *Alt+Up/Down* key or arrow buttons on the right or mouse wheel. Increment is defined by component *Snap* value. Available values are 5 %, E12 or E24.

Parameter will be included in frequency response optimizing if *Opt* field is checked. Otherwise parameter is excluded and existing value locked. See [Optimize](#).

Way # settings

Way 2 settings

☐ Gain
 dB

☐ Delay
 us

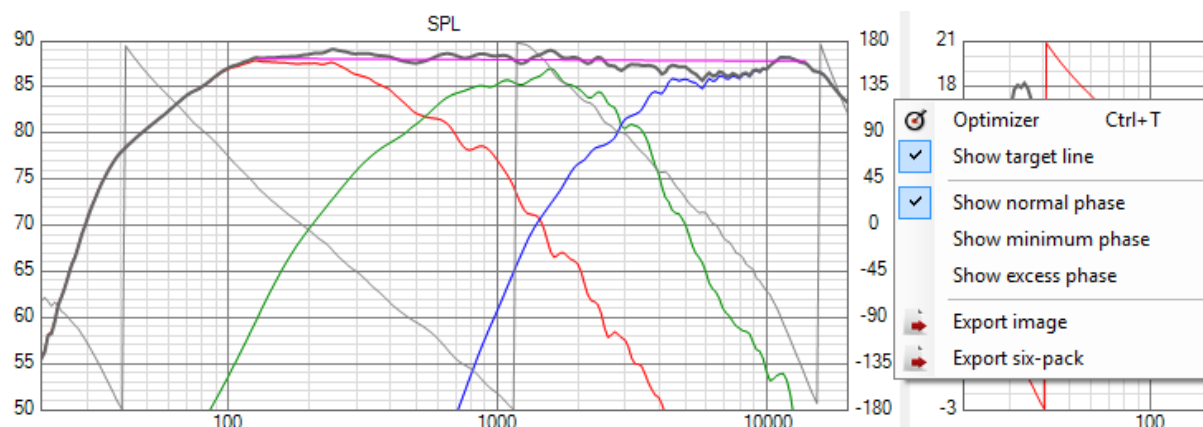
☒ Invert
 mm

☒ Enabled

Adjustable active gain [dB] and DSP delay [μ s] are available for each way. Gain and delay can be adjusted automatically with optimizer by checking the label. See [Optimize](#). Polarity of the way can be inverted by *Invert* checkbox. *Enabled* is a mirror twin with checkbox on [Drivers tab](#).

Dashboard (Graphs)

SPL



By default, SPL graph shows total SPL, total SPL target, SPL per way, SPL per individual driver and total phase. All lines are responses to *Reference angle*, see [Frequency responses](#). Color coding for traces:

Total SPL: black

Total phase: gray. Optional, disabled by unchecking Show Normal phase in context menu.

Minimum phase: lime. Optional, enabled by checking Show Minimum phase in context menu.

Excess phase: steel blue. Optional, enabled by checking Show Excess phase in context menu.

SPL per way: red, green, blue, orange, cyan, olive

SPL per individual driver: light red, light green, light blue, light orange, light cyan, light olive

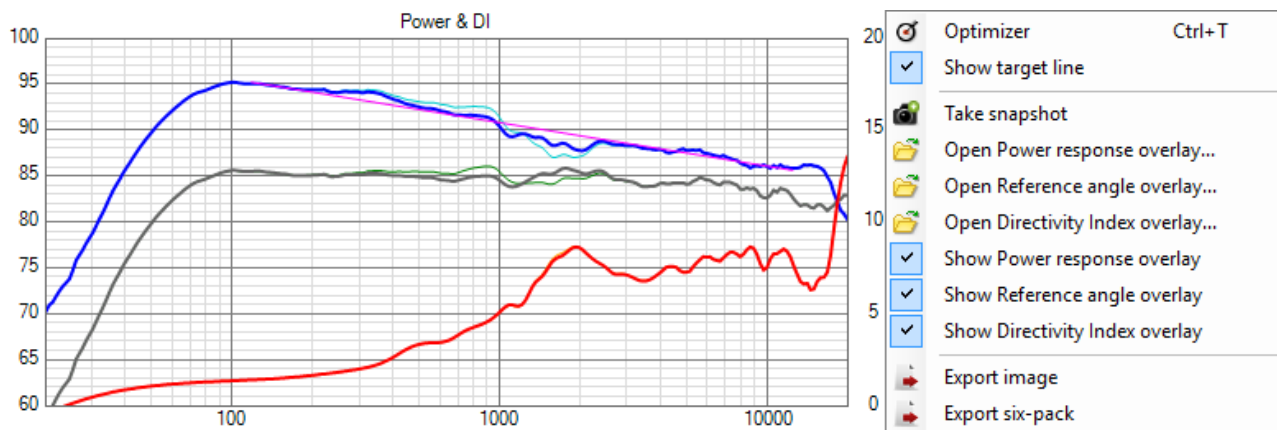
Adjustable SPL target: magenta. Optional, disabled by unchecking Show Target line in context menu.

SPL Target can be adjusted by dragging the line ends with mouse while Shift or Control key is pressed. This is target line for axial response optimizing. See [Optimize](#).

Zooming

Every graph can be zoomed to full size and back to dashboard by double clicking in middle of the chart area.

Power & DI

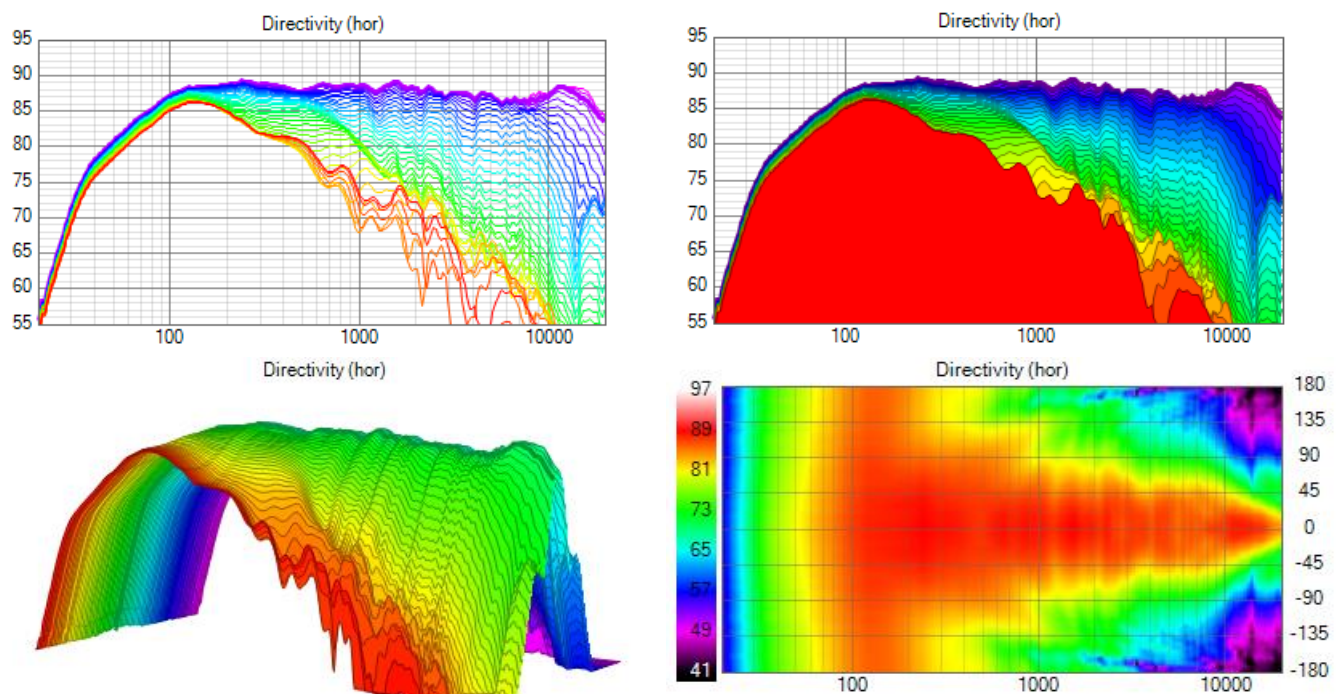


This graph shows power response approximation (blue), frequency response to *Reference angle* (black) and directivity index (red) traces. There is also adjustable target trace (magenta), normally set for power response. Target can be adjusted by dragging the line ends with mouse while Shift or Control key is pressed. This is target line for power response optimizing. See [Optimize](#).

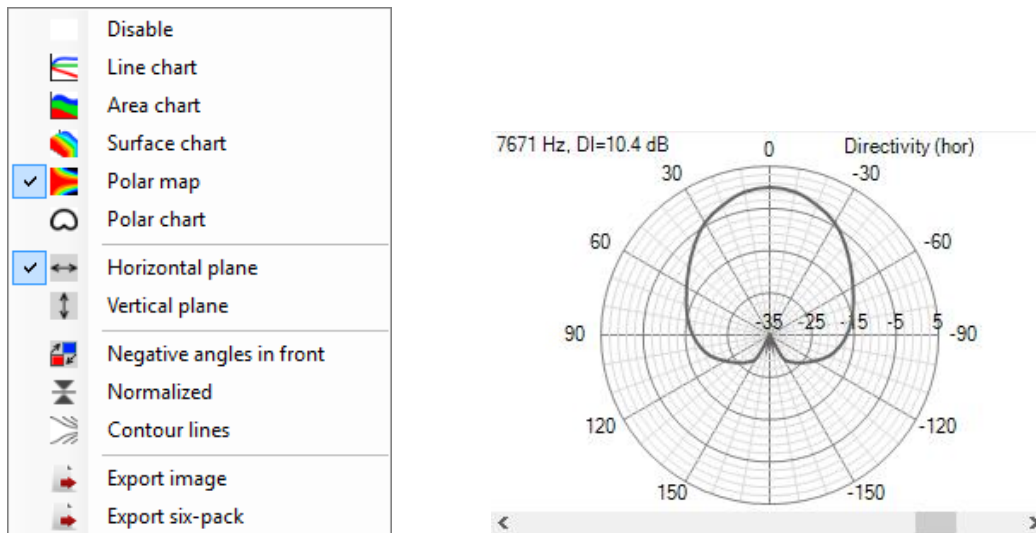
See Options: Power response & DI calculation for more information what is included in the calculation and how it is calculated.

Chart can include three overlays, typically one for each curve. Snapshot/overlay menu opens with right click. *Take snapshot* command saves current situation. Overlay responses can also be loaded from frequency response files. Power response and Reference angle overlays are scaled to left Y-scale, and Directivity Index overlay to right Y-scale. Visibility of overlays is selected individually with *Show overlay* commands.

Directivity



Directivity chart options context menu (right click on the graph):



This graph shows directivity simulation as line chart, area chart, surface chart, polar map (aka heat map) or polar chart. You must have frequency responses to all angles (common for all ways and drivers) You want to show in this graph. Response to *Reference angle* is emphasized with thick line.

Rotation, inclination, zooming and panning are available for surface chart with dragging and wheeling with a mouse. Pan graph by pressing Ctrl-key while dragging. Limits for rotation and inclination are 10...170 deg.

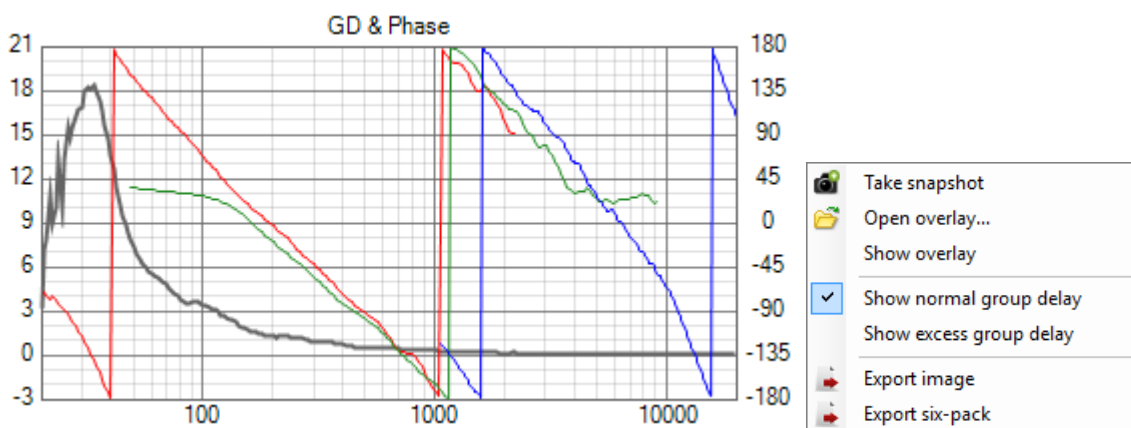
Checking *Polar chart* will show polar plot at frequency selected with horizontal scrollbar.

Checking *Negative angles in front* will invert angle-axis of the plot.

Checking *Normalized* will show flat response to *Reference angle*.

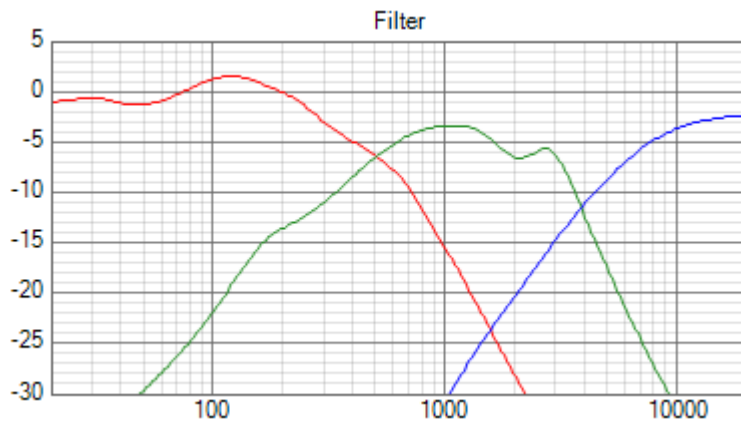
Checking *Contour lines* will show edges of level ranges with Polar map. Level steps are initially 3 dB.

Group delay & Phase



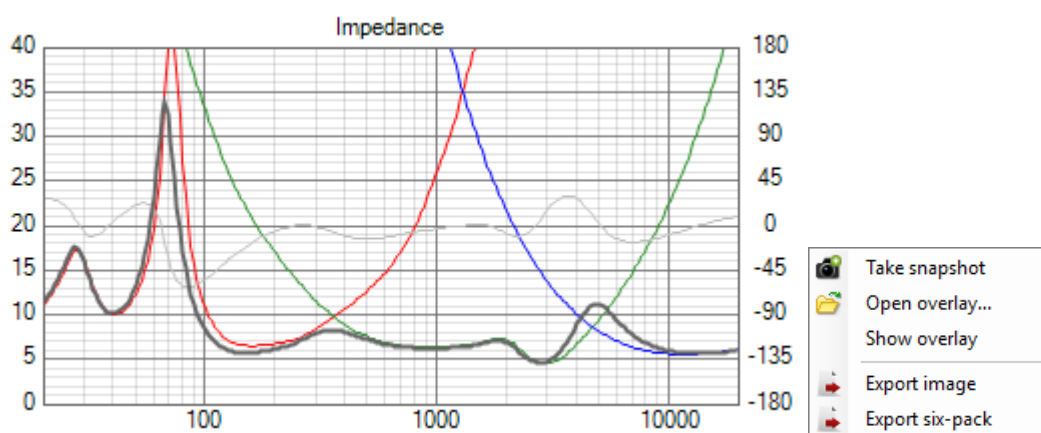
This graph shows total group delay (black) and phase response of individual drivers to *Reference angle*. Color coding follows the same rules as defined in Ways section. Optional Excess group delay (steel blue) is enabled by checking Show Excess group delay in context menu. Group delay can be hidden by unchecking Show Normal group delay.

Filter



This graph shows filter gain of individual drivers. Color coding follows the same rules as defined in Ways section.

Impedance

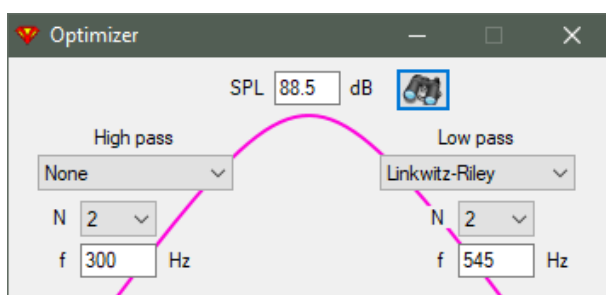


This graph shows total impedance magnitude (black), total impedance phase (gray) and impedance response magnitude per way. Color coding follows the same rules as defined in Ways section.

Phase, Filter and Impedance charts can show one overlay curve. Snapshot/overlay menu opens with right click. *Take snapshot* command saves current situation. Overlay can also be loaded from frequency/impedance response file.

Optimizer

SPL Target curve of Way



Each way can be assigned with an individual target curve. Optimizer window can be activated with context menu of SPL chart. Select Axial response, Way # to show target curve. High pass, low pass and band pass curves are available. VituixCAD supports the following target slopes: 1st...8th order Butterworth, Bessel, Chebychev 0.5 dB and Linkwitz-Riley.

Target SPL level can be adjusted manually using the text field or read automatically from total SPL target by clicking binocular button . Filter design can be selected from the first dropdown menu (High pass or Low pass). Second dropdown menu (N) controls order of the slope. Corner frequency of the slope is adjusted in the last text field (f).

Optimize

Axial response of currently selected way can be adjusted automatically to the target by checking **Axial response, Way #**. Create target curve in this window and specify frequency range with two text boxes where optimizer is allowed to calculate squared error to target. Select parameters to be optimized from Parameters/components grid in Crossover tab. Start solver with *Optimize* button.

Filter response (magnitude only) of currently selected driver can be adjusted automatically to the target by checking **Filter gain, Driver #**. Open target response to overlay of Filter chart and specify optimized frequency range with text boxes. Select parameters to be optimized from Parameters/components grid in Crossover tab. Start solver with *Optimize* button.

Total axial and power responses can be adjusted automatically to the target lines by checking **Axial response and Power response**. Weighting between responses is controlled with percent values. Higher value produces smaller difference between target and response. For example Axial=60%, Power=40% allows more error in power response than axial response which could be better if speaker is very directive or intended listening distance is very short. Response is optimized to both slope and level of target line by checking **Seek level**. If unchecked, optimizer does not care about level - just slope within range of target line.

Target lines are set in SPL and Power & DI charts. Optimizer calculates squared error within frequency range of each target line. Select parameters to be optimized from Parameters/components grid in Crossover tab. Include ways with checkboxes (1...6). Start solver with *Optimize* button. Check **Minimum impedance** and enter preferred minimum value to text box if you like to control impedance response. Squared error increases if minimum impedance drops below the setting. Minimum is detected from total impedance while optimizing axial and power responses. Minimum is detected from impedance of selected way while optimizing axial response of way or filter gain.

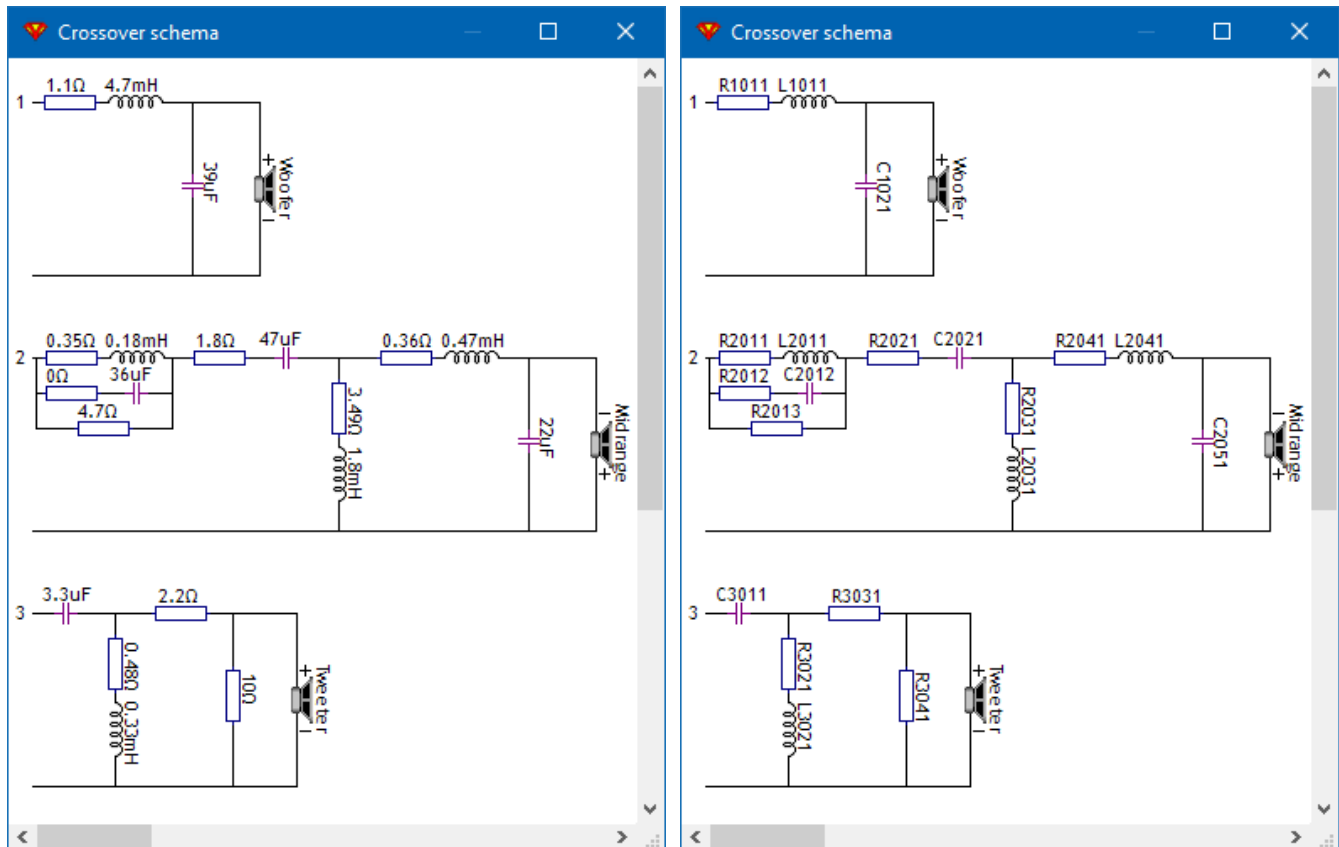
Passive crossover components can be rounded to the closest value in standard E-series by selecting E12, E24 or E48. Note! Values are rounded after optimization which will increase squared error in the end.

Optimization could end up to bad result if initial parameter values are too far from good solution and method finds wrong local minimum. Adjust parameters manually closer to acceptable solution and restart solver with *Optimize* button. Result can be rejected with **Undo** button. Undo is able to restore up to ten most recent changes.

Optimizer stops when error is zero (rarely) or *Stop* button is pressed or maximum evaluations is reached. Initial maximum is 300 evaluations. Simple problems with only few parameters to optimize could be solved with less than 100 evaluations.

Optimizing can be accelerated by disabling Directivity chart, especially Polar map and Surface chart.

Schema window



Units

Part #

Schema window shows the total crossover network without disabled ways and bypassed filter blocks. Visibility of values and units or component names is controlled by *Units* and *Part #* in *View* menu.

Parts list

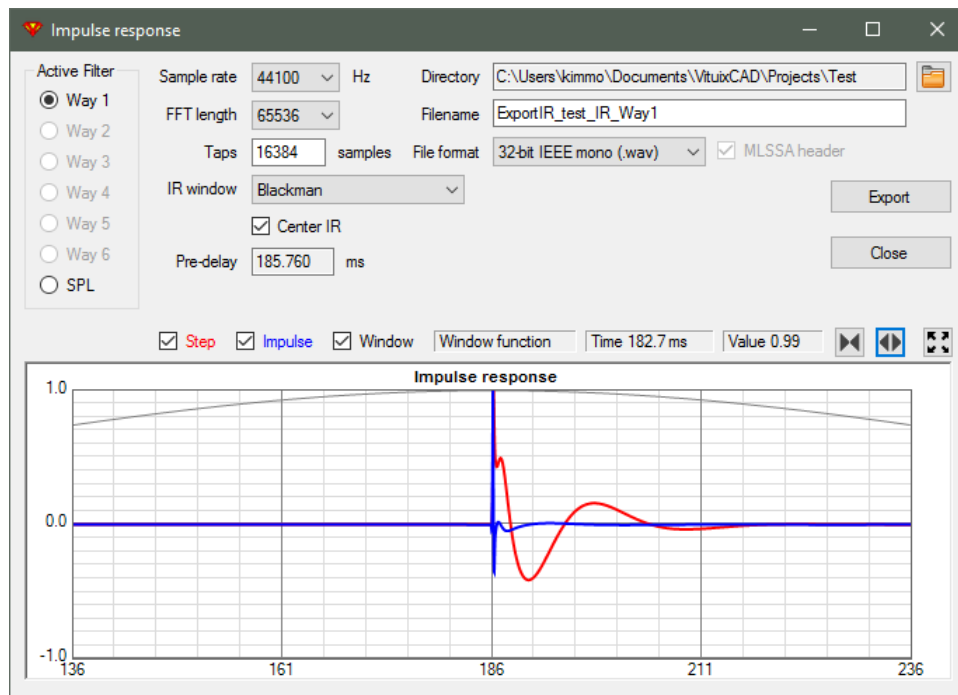
Parts list			
Way 1			
R1011	1.1	ohm	
L1011	4.7	mH	
C1021	39	uF	
Way 2			
R2011	0.35	ohm	
L2011	0.18	mH	
C2012	36	uF	
R2013	4.7	ohm	
R2021	1.8	ohm	
C2021	47	uF	
R2031	3.49	ohm	
L2031	1.8	mH	
R2041	0.36	ohm	
L2041	0.47	mH	
C2051	22	uF	
Way 3			
C3011	3.3	uF	
R3021	0.48	ohm	
L3021	0.33	mH	
R3031	2.2	ohm	
R3041	10	ohm	

Parts list shows passive filter components and values as text. List can be copied (right click, Select All, Copy) and pasted to Your favorite spreadsheet or text editor. Please be careful when copy - pasting, resistance of coils are shown as individual resistors or included in value of actual resistor in series with the coil.

Impulse response

Transfer function of active filters per way or total/axial SPL can be exported to impulse response in time domain. Typical application is to produce impulse response as wav-file for speaker controller with DSP FIR support or convolver plugin. Passive filters (per driver) cannot be exported directly to impulse response but it is possible with workaround: replace driver's acoustical frequency response with synthetic flat response and export total/axial SPL response while other ways disabled.

Frequency response is extrapolated to cover band from 0 Hz to Sample rate / 2 Hz (Nyquist) before inverse Fast Fourier Transform. After inverse FFT, impulse response is multiplied by window function to reduce artifact errors due to spectral leakage.



Sample rate: 44100, 48000, 96000, 192000 Hz.

Initial value is 44100 Hz, equal to CD audio.

FFT length: 8192, 16384, 32768, 65536, 131072, 262144 bins.

Initial value is 65536. Frequency resolution of FFT is calculated $\text{Sample rate} / \text{FFT length}$.

For example $48000 \text{ Hz} / 65536 = 0.732 \text{ Hz}$.

Taps: 512...131072 samples.

Maximum taps is limited up to $\text{FFT length} / 2$. Up/down arrow keys step value to the next 2^N .

IR window function: Rectangular, Bartlett, Hanning, Hamming, Blackman, Blackman-Harris, Nuttall, Blackman-Nuttall, Cosine.

Initial window is Blackman. Check shape and centering via graph with adequate time span.

See wikipedia: [Window function](#). Note: Advanced window functions are designed for spectrum analyzing with high S/N where silent side lobes are required. IR export works fine with classic Blackman, Hanning etc.

Center IR shifts timing of source frequency response before inverse FFT by increasing phase angle (by $\text{delay} * \text{frequency} * 2\pi \text{ rad}$). Added delay is half of the impulse response length. Impulse peak is not necessarily located exactly at the middle point. Added delay is shown in **Pre-delay** text box. That value can be entered or copied to analyzing application to restore timing/phase of original frequency response.

Center IR can be unchecked if centering is not needed, typically with IIR filters. Delay of 1.0 millisecond is added to allow minor pre-ringing and timing tolerance without trimming first samples of the impulse peak.

File format: 16-bit PCM mono (.wav), 16-bit PCM stereo (.wav), 32-bit IEEE mono (.wav), 32-bit IEEE stereo (.wav), 64-bit IEEE mono (.wav), 64-bit IEEE stereo (.wav), 32/64-bit mono (.txt), miniDSP binary file (.bin), miniDSP manual mode (.txt, copied also to clipboard).

Signal in 16-bit PCM wav is scaled to ± 32760 , and IEEE wav to ± 0.999 to avoid notification of possibly clipped values. Stereo wav has the same signal in both channels.

Value scaling in text file is equal to source frequency response. Text file has single column from 0.0 s with step of 1/Sample rate [s]:

```
8.49378929663085E-25
1.64194430575251E-15
-3.17746589425256E-15 ...
```

MLSSA header can be added in text file to help reading impulse response with applications having *ASCII MLSSA file* (.txt) import - like ARTA:

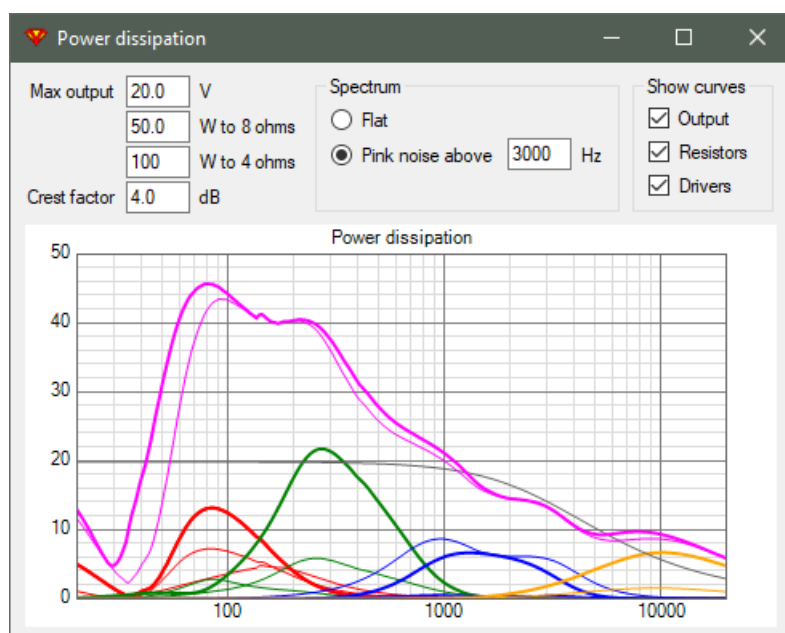
```
0                start character
0.0226757369614512 1000/Sample rate
16384            taps
```

Impulse and step **curves** are updated automatically with selected IFFT parameters when crossover is changed. Time scale can be expanded and compressed with arrow buttons. Graph can be zoomed to full window for design-time preview.

Power dissipation

Graph shows electrical power spectrum of amplifier output and drivers and resistors in passive crossover networks. Cursor label shows driver ID and name or part number and resistance value. Visible curves can be selected with checkboxes. Adjustable parameters for output signal from power amplifier:

- Maximum voltage or power to constant 8 or 4 ohms load relative to gain of way.
- Crest factor. Ratio of maximum output to the effective RMS value in dB, representing compression/loudness of music signal. Value should be short-term minimum to ensure that resistors are able store and dissipate energy. Typical crest factor for whole music track or album could be too high.
- Spectrum: Flat or Pink noise above specified frequency. Pink noise represent effective RMS spectrum of music signal. Typical corner frequency of pink spectrum is about 2 kHz, but significant variation exists within music genres and recordings. Initial value is 3 kHz.



Options

Angle parsing from filename

To allow VituixCAD to parse measurement angles and axis from frequency response files, You have to define file naming format/syntax. *Keywords* define how to distinguish between horizontal and vertical axis. Horizontal axis is selected if keyword of vertical is not found in the filename. *Search direction* defines whether VituixCAD should start parsing angle value from the beginning of the filename (hor_+150_myproject_mydriver.txt) or the end (myproject_mydriver_hor_+150.txt). *Number format* defines how angle value is formatted in the filename. For example if You use 1500 to present 15 degrees, use *Integer multiplied by* 100. Test tool is provided for testing Your syntax.

Frequency responses

Mirror missing defines if VituixCAD should mirror missing measurement data:

- Measurement to equal positive angle selected if negative angle is not available (or vice versa)
- Measurement to equal horizontal angle selected if vertical angle is not available (or vice versa)
- Measurement to equal absolute horizontal angle selected if vertical angle is not available (or vice versa).

Active PEQ response as analog RLC. Q-value and frequency response of Active peak/notch (IIR) filters are compatible with analog RLC circuits: series $Q = \sqrt{L/C}/R$, parallel $Q = \sqrt{C/L} \cdot R$. Reference level of BW is about -3dB with high peaks.

If not checked, Q-value and frequency response are compatible with many digital devices such as DCX2496 or miniDSP. Response peak is narrow because reference level of BW is $A[\text{dB}]/2$.

Setting is saved in project file (vxp).

Active IIR response as digital biquad. If checked, frequency response of active IIR filters is compatible with digital biquad; not ideal at high frequencies due to finite sample rate. Setting is saved to project file (vxp). Sample rate is selected in Impulse response window.

Frequency axis of charts is limited up to Nyquist frequency (sample rate / 2) if digital biquad option is checked.

Note! Internal frequency range is full 10-40k which causes error to minimum phase calculation if sample rate is less than 96 kHz.

Bessel phase normalized. Phase angle at nominal frequency is normalized according filter order: 90° (2nd), 135° (3rd), ... 360° (8th). Level at nominal frequency varies: -4.8 dB (2nd), -6.3 dB (3rd), ... -13.5 dB (8th). This is popular Bessel function with many digital devices such as Xilica or miniDSP.

If not checked, level at nominal frequency is normalized to -3.0 dB. This is common in design tables of linear circuit manufactures such as Analog Devices and Texas Instruments.

Before designing crossover, select Bessel response normalization which is compatible with your target system.

Setting is saved in project file (vxp). *Listening distance* is virtual distance from loudspeaker to listener or microphone, needed to calculate phase differences and amplitude relations between drivers in different locations. Enter typical listening distance in mm. Default value is 2500 mm.

Display

Font for crossover schematic and visibility of tooltips are selectable.

Power response & DI calculation

Intensity on spherical surface is normally selected for common sized single or multiway speakers. Intensity on spherical surface around speaker is calculated from radial measurements in horizontal and vertical planes.

$$Q(f) = \frac{2N}{\pi \sum_{n=1}^N \left| \frac{p(\theta_n)}{p(0)} \right|^2 \sin \theta_n}$$

$$DI(f) = 10 \log_{10} Q(f)$$

Intensity on cylinder surface is practical selection for long line sources, or if either horizontal or vertical directivity is temporarily interesting - not accurate power response & DI result. Intensity on cylinder surface around speaker is calculated as average pressure from radial measurements, typically in a single (horizontal) plane.

Checkboxes control which planes are included in power response and directivity index calculations; horizontal, vertical or both.

If *Half space* is checked, angles >90 deg are excluded from power response and DI calculation. Directivity chart shows angles -90...+90 deg only. This setting is meant for flush mounted or other clearly uni-directional speakers. Common box speakers and dipoles with DI <10 dB should be measured and simulated to full space.

Image export

Single W x H is size of exported chart image. Default size is 480x270 px.

Six-pack W x H is size of one exported chart in group of all six charts in main program or Enclosure tool. Default size is 400x225 px.

Default size can be set by double-clicking the label.

Graph scales

Frequency axis

Internal frequency range is fixed 10...39794 Hz with density of 48 points/octave, but you can limit visible scale. Options are fixed 20...20000 Hz or custom range with minimum 10...400 Hz and maximum 1000...40000 Hz.

SPL, Directivity

Span controls vertical scale of SPL graphs. SPL, Power & DI and Directivity waterfall span: 20, 30, 40 or 60 dB.

Group delay

Span controls vertical scale of GD & Phase graph. GD span: 2, 4, 8, 16, 24 or 40 ms.

Filter gain

Max defines upper limit and *Span* controls vertical scale of filter gain graph. Filter gain span: 30, 35, 40, 45, 50 or 60 dB.

Impedance

Max defines upper limit of impedance graph. Impedance maximum: 20, 30, 40, 60 or 80 ohm.

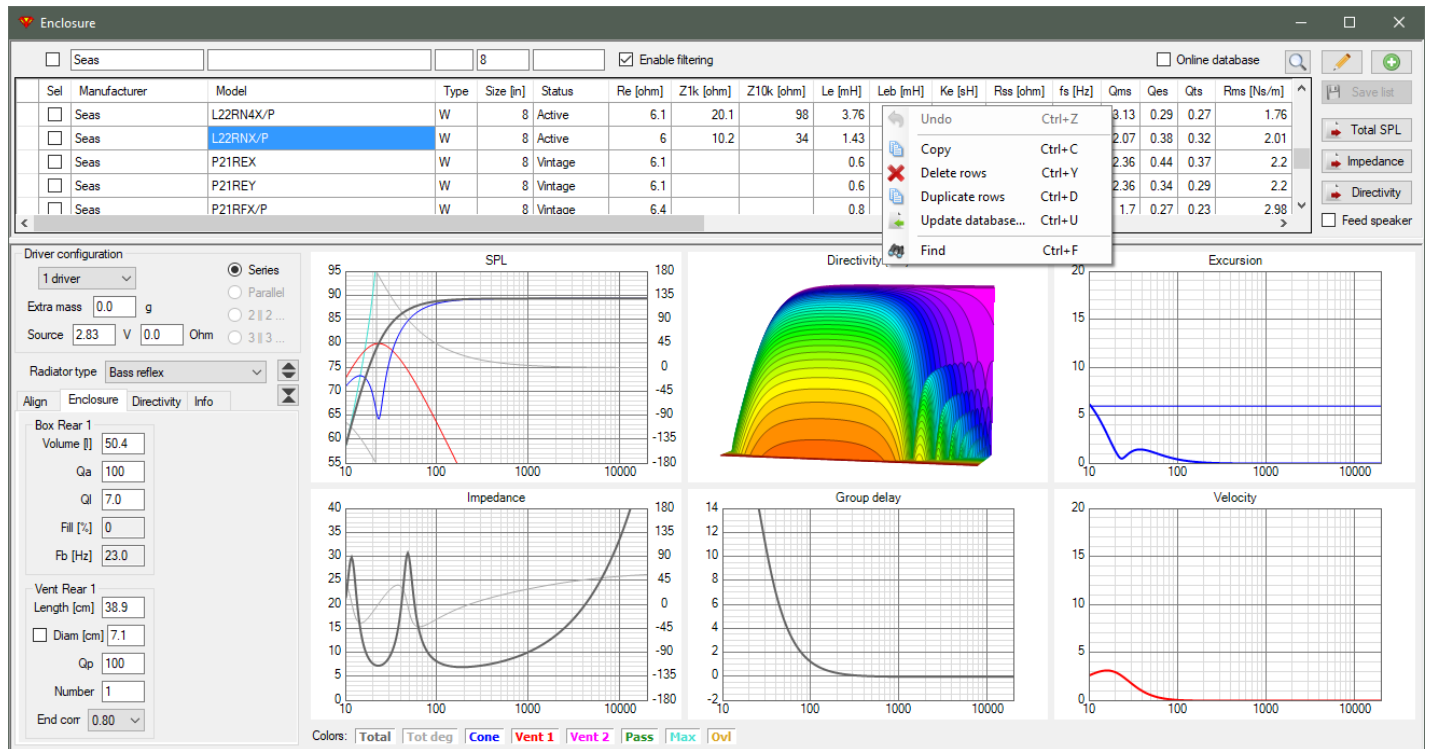
External tools

Paths in the text fields define applications VituixCAD should open when pressing corresponding buttons/menu items. Select application by clicking folder button or dropping file into text box.

Baffle diffraction	Executable of external diffraction simulator, for example Tolvan Edge
Spice	Executable of LTspice IV or compatible circuit simulator
Web search	Search command for drivers in Enclosure tool

Baffle diffraction text box should be empty for activating internal diffraction simulator.

Enclosure tool



Driver database

VituixCAD reads online database from <https://kimmosaunisto.net/> when *Online database* is checked. Adding, modifying and removing drivers are possible only with local database.

Filtering is enabled by checking *Enable filtering*. You can filter driver list by user selection (checkbox in *Sel* column), Manufacturer, Model, Type, Size and Status. Filter is updated by pressing Enter or by moving cursor to another text box. Criteria in multiple fields is logical AND. Single field can contain several criterion separated with semicolon (;) or comma (,). Criteria in a single field is logical OR. For example:

Seas and BMS drivers are visible when Manufacturer=Seas;BMS

8" Scan-Speak and Seas drivers are visible when Manufacturer=Scan;Seas and Size=8

Driver database includes the following fields:

Manufacturer	
Model	
Type	S, W, M, F, C or PR (passive radiator)
Size	Nominal diameter [inches]
Status	Active, Discontinued, Preliminary or Vintage
Re	DC resistance [Ohm]
Z1k	Impedance at 1 kHz [Ohm]
Z10k	Impedance at 10 kHz [Ohm]
Le	Voice coil inductance [mH] or Bound inductance [mH], see Impedance models
Leb	Free inductance [mH], see Impedance models
Ke	Semi-inductance [sH], see Impedance models
Rss	Shunt resistance [Ohm], see Impedance models
fs	Free air resonance [Hz]
Qms	Mechanical Q factor
Qes	Electrical Q factor
Qts	Total Q factor
Rms	Mechanical resistance [Ns/m, kg/s]
Mms	Moving mass with air load [g]
Cms	Suspension compliance [mm/N]
Vas	Equivalent volume [liters]

Sd	Effective cone area [cm ²]
Bl	Force factor [N/A, Tm]
Pmax	Maximum long term input power [W]
Xmax	Maximum linear excursion, one way peak [mm]
Revision	Datasheet revision or date by manufacturer
Updated	Date/Name in format yyyy-mm-dd/First name Last name

Driver list can be sorted by clicking column header. Right click in driver row opens context menu with more options to search and modify driver list. Context menu options:

Undo (all changes)
Copy
Delete rows
Duplicate rows
[Update database](#)
Find



button searches for selected driver from web (Google search with Your default browser).

Parameters of selected driver are modified with button. New driver is added with button. Parameters are accepted and window closed with OK button.

Edit parameters

Manufacturer Scan-Speak
Model 26W/8861T00
Type W
Size 10 inches
Re 6.56 Ohm
fs 19 Hz
Qms 5.2
Qes 0.33
Qts 0.31
Rms 0.630 Ns/m, kg/s
Mms 44.7 g
Cms 1.27 mm/N
Vas 231 liters
Sd 320 cm²
BL 9.25 NA, Tm
Pmax 170 W
Xmax 9 mm
Status Active
Revision February 22, 2011.
Updated 2016-02-27/Datasheet Import

Basic Z model
Z1k 0 Ohm
Z10k 0 Ohm
Extended Z model ?
Le 0.795 mH
Leb 0.148 mH
Ke 0.0277 sH
Rss 2290 Ohm
Crosscalc
n0 0.466 %
SPL 88.8 dB/W
USPL 89.6 dB/2.83V
EBP 57.6
Vd 288 cm³

OK Cancel

Add new driver

Manufacturer
Model
Type
Size 0 inches
Re 0 Ohm
fs 0 Hz
Qms 0
Qes 0
Qts 0
Rms 0 Ns/m, kg/s
Mms 0 g
Cms 0 mm/N
Vas 0 liters
Sd 0 cm²
BL 0 NA, Tm
Pmax 0 W
Xmax 0 mm
Status
Revision
Updated

Basic Z model
Z1k 0 Ohm
Z10k 0 Ohm
Extended Z model ?
Le 0 mH
Leb 0 mH
Ke 0 sH
Rss 0 Ohm
Crosscalc
n0 0 %
SPL 0 dB/W
USPL 0 dB/2.83V
EBP 0
Vd 0 cm³

OK Cancel

Check *Crosscalc* to update other parameters while one parameter is adjusted.

n0, SPL, USPL, EBP and Vd are calculated for information only.

Driver database is simple tab-delimited text file: *(My)Documents\VituixCAD\Enclosure\VituixCAD_Drivers.txt*. Database can be maintained with external tool like Microsoft Excel, provided that original file format including column headers, measurement units and decimal symbol are not modified.

Impedance models

Enclosure tool can use two different impedance models:

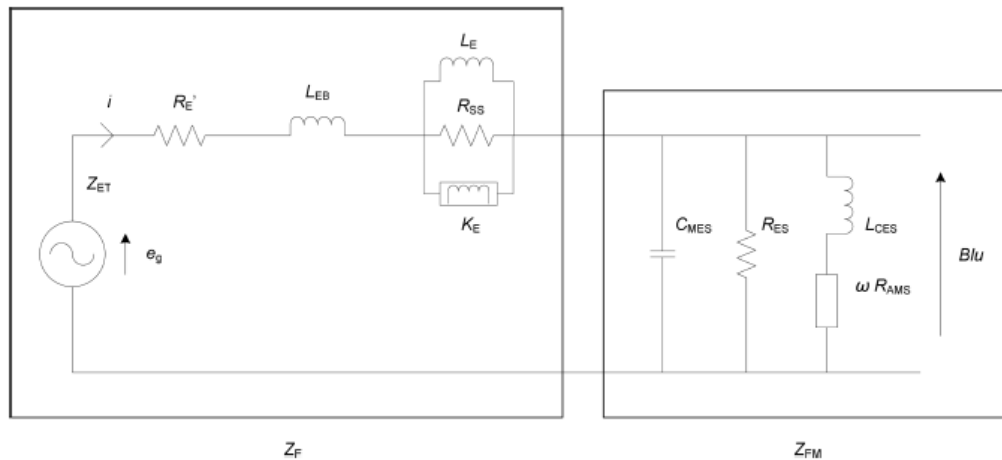
1. **Basic impedance model.** Exponential curve fitted with two impedance points entered by user: Z1k and Z10k.

$$Z_L = K f^n \quad n = \log \left(\frac{Z_{L10k}}{Z_{L1k}} \right)$$

This is primary model used for impedance response calculation if both Z1k and Z10k are greater than zero. Entered values should be greater or equal to Re.

2. **Extended impedance model** described in detail in the paper [Frequency Dependence of Damping and Compliance in Loudspeaker Suspensions](#) by Knud Thorborg, Carsten Tinggaard, Finn Agerkvist and Claus Futtrup, published in [JAES Volume 58 Issue 6 pp. 472-486; June 2010](#).

Loudspeaker equivalent circuit (seen from electrical side):



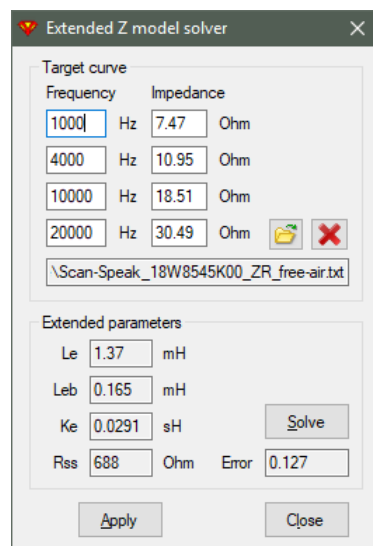
Calculation rules if all parameters of extended impedance model are not applied:

- Semi-inductance K_e is used if bound inductance L_e is blank or zero
- Bound inductance L_e is used if semi-inductance K_e is blank or zero
- Shunt resistance R_{ss} is ignored (=infinite) if blank or zero.

This is secondary model used for impedance response calculation if Z_{1k} or Z_{10k} or both are blank or zero.

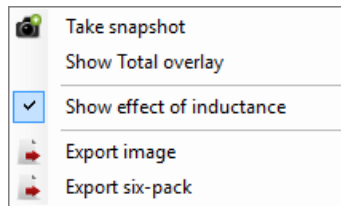
Extracting of L_e , L_{eb} , K_e and R_{ss} from impedance curve on manufacturer's data sheet or impedance measurement is possible with **Extended Z model solver** which opens with **?** button after Extended Z model text in Edit parameters window.

Note! R_e , f_s , R_{ms} , M_{ms} , V_{as} , S_d and Bl should be entered and >0 before extraction.



Enter four points within 1-20 kHz from impedance curve, or open impedance response file (txt,zma) and adjust frequencies if not optimal. Press **Solve** button to start. Optimizer searches L_e , L_{eb} , K_e and R_{ss} giving close to target. Parameters can be accepted for driver with **Apply** button if sum of squared error is less than 0.5. Repeat solving or adjust initial values of L_e , L_{eb} , K_e and R_{ss} in *Edit parameters* window if solver didn't find acceptable solution (Error field is **red**) and it's certain that target points are not impossible for extended Z model.

Voice coil inductance (with losses) is included in frequency response calculation by checking *Show effect of inductance* in context menu of SPL chart.

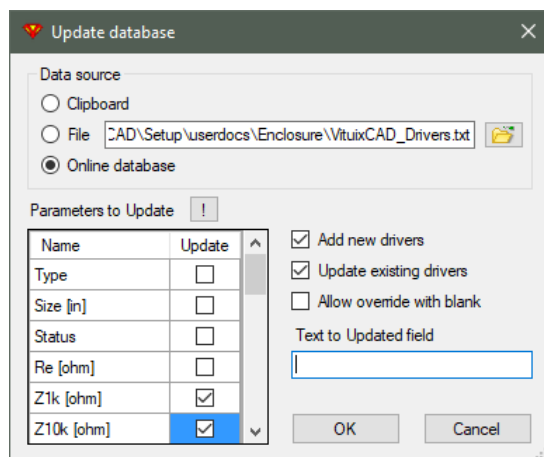


Update database

Local driver database can be updated with data on the clipboard or local text file or online database @kimmosaunisto.net.


Order and amount of columns in the source data and destination (local) database must match if the source data does not have header line with parameter names in the first row. Column match should be verified especially with data on the clipboard. Local text files and online database usually have the header row i.e. the values are copied to the correct fields regardless of the column order.

Update database window opens via context menu of driver grid.



Select data source: clipboard, text file or online database.

Drivers whose Manufacturer and Model are not found in the local database are added by checking *Add new drivers*.

Drivers whose Manufacturer and Model are found in the local database are updated by checking *Update existing drivers*. Select parameters to update from the list. Selections are inverted with  button.

Allow override with blank enables replacing of existing value with blank. Otherwise blank value in the source data won't override existing non-blank value.

Text to Updated field overrides existing text. Text in the source data is copied if text box is left empty.

Driver configuration

Drop down menu provides selection for amount of drivers. If more than one driver is selected, radio buttons on the right control how drivers are connected.

Number of drivers; 1, 2, 3, 4, 6, 8, 9 or 12 pcs

Electrical connection; series, parallel, 2 || 2 ..., 3 || 3 ...

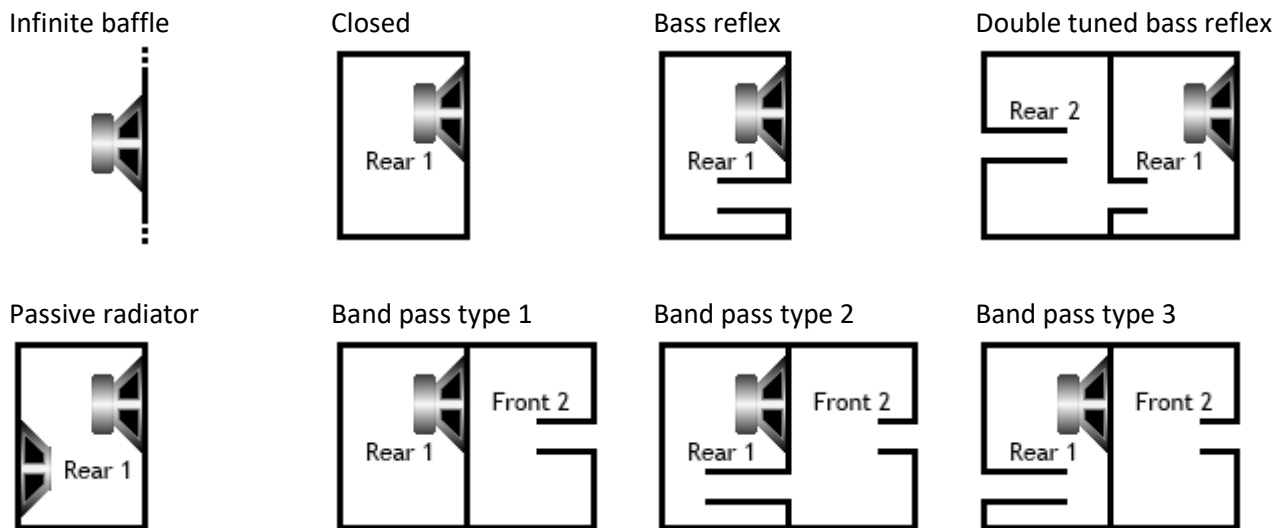
Extra mass per driver [g] can be used to simulate increase of M_{ms}, which reduces f_s and sensitivity of the driver. Mass is reset to zero grams by double-clicking 'Extra mass' label.

Amplifier's output voltage [V] and output resistance [Ohm] are common for all drivers. Voltage can be changed from default 2.83 V to simulate cone excursion and vent air velocity with higher electrical power. Default voltage is restored by double-clicking 'Source' label. Output resistance emulates output impedance of power amplifier or

cable resistance. Actual series resistance is quick and dirty way to increase electrical Q factor and decrease sensitivity.

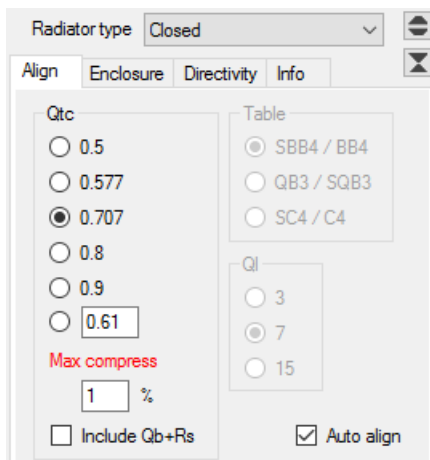
Radiator type

Radiator types supported by Enclosure tool:



Tabs

Align –Closed and Bass reflex radiator alignment



Closed box is aligned by selecting or entering Qtc. Optional high alarm limit for non-linearity [% on Xmax] due to air compression is available. Box volume is limited and requested Qtc is not produced if alarm limit is exceeded (red text). Increase % value until red color disappears to get requested Qtc if you don't care about compression distortion. Box Q entered on Enclosure tab and series resistance are included in alignment by checking *Include Qb+Rs*. Otherwise alignment is done by basic formula: $V_b = V_{as} / ((Q_{tc}/Q_{ts})^2 - 1)$. Both options are approximations, but normally including Qb+Rs is giving results closer to effective Qtc around system resonance.

Bass reflex is aligned by Thiele/Hoge/Bullock -tables: SBB4/BB4, QB3/SQB3 and SC4/C4 with Ql 3, 7 or 15.

Auto align executes alignment with entered parameters when driver or driver configuration or radiator type is changed.

☐ Crossover of current driver

Filter transfer function of currently selected driver in the main program can be applied for driver in Enclosure tool. Linking is done with *Crossover of current driver* checkbox in *Align* tab. Responses in Enclosure tool are refreshed on the fly while adjusting the filter. Response is shown in *Group delay* chart with dB scale (right).

Enclosure – Parameters for different radiator types

Adjustable box and vent parameters vary depending on chosen radiator type.

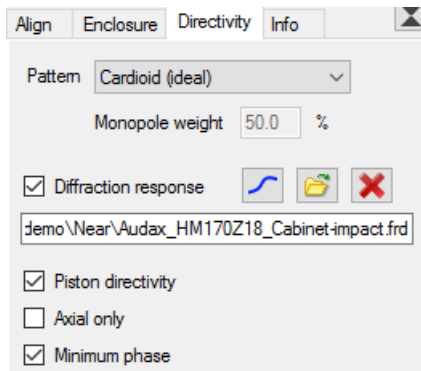
Adjustable box parameters: Volume [l], Qa, Ql. Box resonance frequency Fb [Hz] is calculated.

Fill [%] is not implemented in this version.

Adjustable vent parameters: Length [cm], Diameter [cm], Qp, Number, Total end correction; 0.614, 0.732, 0.80, 0.850, 0.90, 0.95 or 1.00

Get from table button reads passive radiator parameters from current row of driver database. Regular driver with motor is also possible passive radiator.


Directivity – Baffle step and directivity




Directivity tab contains selection of textbook radiation pattern:

- Omni (ideal)
- Cardioid (ideal)
- Super-cardioid (ideal)
- Hyper-cardioid (ideal)
- Dipole (ideal)
- Gradient (ideal). Weight of monopole part is adjustable 0...100 %.

This enables creation of ideal off-axis responses for the simulated direct radiator. Select *Omni (ideal)* to disable textbook pattern.

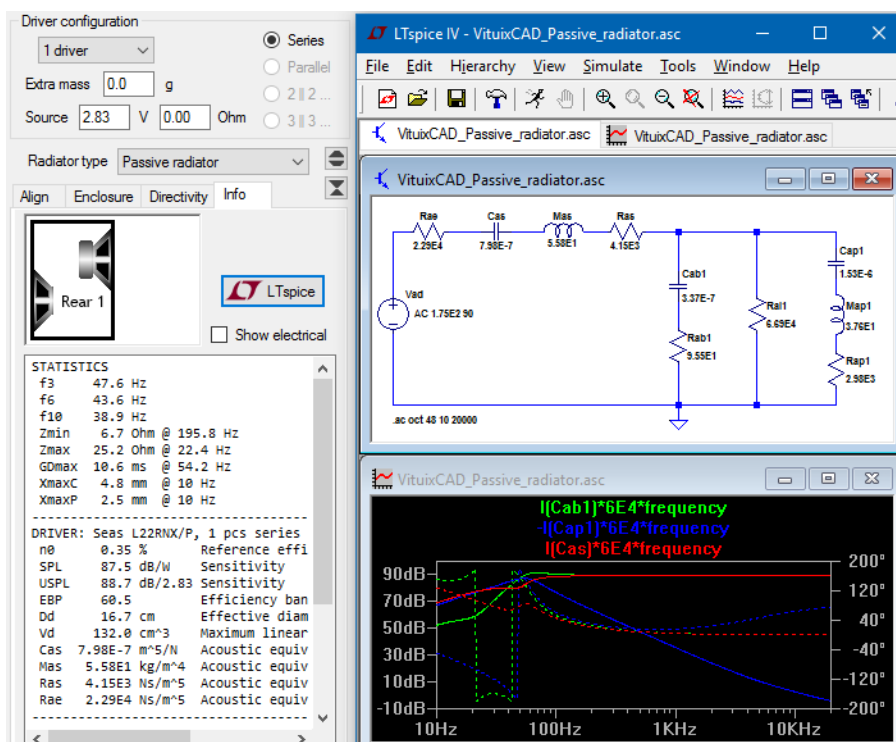
Single cabinet impact response is possible to include in full space simulation shown in Directivity graph. Load simulated baffle response by clicking *Open* button , and enable it by checking *Diffraction response*.

Check also *Axial only* to prevent processing, displaying and exporting of off-axis responses which are obsolete when baffle simulation contains only axial response. Diffraction simulator is executed by clicking *Baffle step* button .

Directivity by piston radius is included in simulation by checking *Piston directivity*. Off-axis angle is limited in calculation within 0...90 deg if *Pattern* selection is not *Dipole (ideal)*.

Responses of *Directivity export* can be calculated as minimum-phase by checking *Minimum phase*. That is default due to better accuracy.

Info – Acoustical and electrical equivalent parameters



Info tab shows enclosure image, calculated acoustical parameters of driver, boxes and ports. Optional display of electrical equivalent values is available. Info tab also lists statistics: f_{-3dB} , f_{-6dB} , f_{-10dB} , min. impedance, max. impedance, max. group delay, max. excursion of cone and passive radiator and max. air velocity of vents.

Optional execution of external LTspice IV circuit simulator is available. Acoustical parameters of driver and enclosure are passed into LTspice. Path for LTspice executable needs to be defined in [Options window](#).

Note! Effect of voice coil inductance is not shown in SPL curve.

Export functions

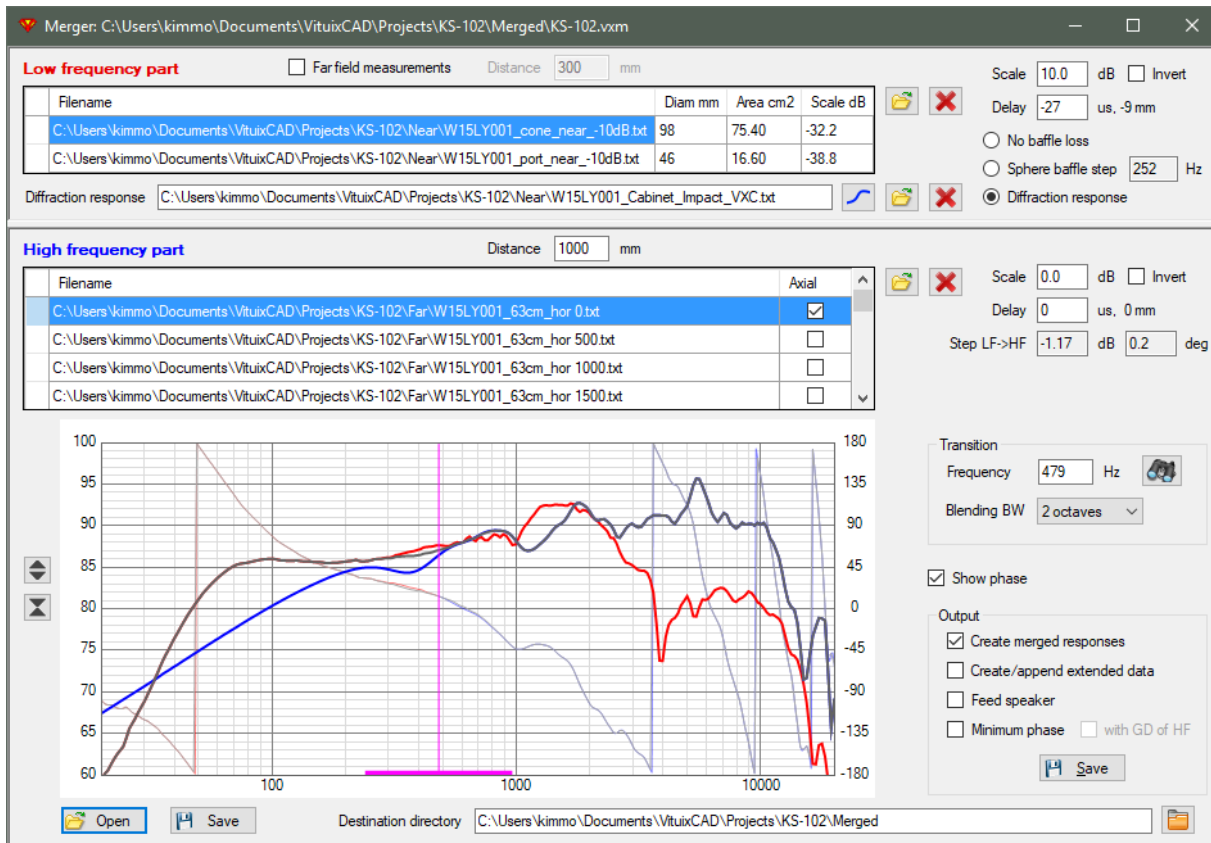
Total SPL, impedance response and directivity can be exported if You choose to use them in simulating loudspeaker driver instead of acoustic and electrical measurements. Directivity (off-axis) responses and impedance response are routed for currently selected driver in the main program by checking *Feed speaker* before exporting.

Dashboard (Graphs)

Every graph can be zoomed to full size and back to dashboard by double clicking in middle area of particular graph. Color coding of traces is defined below dashboard. Dashboard of enclosure tool includes following graphs:

- Total SPL [dB] and phase, SPL of cones and ports, maximum SPL and Total SPL overlay. Snapshot, frequency response file and visibility of overlay are controlled with context menu (right click).
- Total impedance [Ohm] and phase, and Impedance overlay. Snapshot, impedance response file and visibility of overlay are controlled with context menu (right click).
- Directivity: waterfall, surface chart, polar map or polar plot. Positive horizontal angles only.
- Group delay [ms]. Optional frequency response [dB] of crossover.
- Peak excursion and Xmax [mm] of cone and passive radiator.
- Air velocity [m/s] of vents or input Power: volt-ampere [VA], real power [W] and Pmax [W]. Visible chart is selected from context menu (right click). Power is forced visible for Infinite baffle, Closed and Passive radiator due to missing vents.

Merger tool




Merger tool merges frequency responses using two separate sections: Low frequency part and High frequency part. Low frequency responses can be either far field measurements or near field measurements with baffle simulation. Graph shows preview of low frequency part (red), high frequency part (blue), merged response (black) and optionally phase. Graph cursor (magenta) controls transition frequency and shows also blending range.

Low frequency part with Near field measurements

Merger tool merges near field measurements + diffraction simulation or 2π + diffraction simulation or 4π simulation to time-windowed far field measurements. Merged off-axis responses contain directivity information below transition frequency based on time-windowed axial response divided by time-windowed off-axis response. Reliable directivity information at low frequencies requires long time window.

Uncheck *Far field measurements*. Near field frequency responses are added by clicking folder button on the right or dropping files into list. You can have multiple responses, usually near field for cone(s) and near field for port(s). Enter diameter [mm] or area [cm²] of particular radiator and Merger tool will calculate scaling to far field automatically. Scaling can be adjusted via *Diam [mm]*, *Area [cm²]* or *Scale [dB]* in response rows and by total low frequency part via *Scale* or *Distance* text box by entering a value or by Up/Down keys.

For baffle diffraction there are three options available. *No baffle loss*, *Sphere baffle step* (adjustable nominal frequency) or *Diffraction response* simulated with internal or external baffle simulator; Edge, Leap, LspCAD, SoundEasy etc. Diffraction response can be selected by clicking folder button or dropping file into text box. Diffraction simulator is executed by clicking *Baffle step* button .

Low frequency part with Far field measurements


Low frequency response list must have equal off-axis angles (with same angle coding in file names) to high frequency response list. Otherwise LF and HF responses cannot be paired.

Check *Far field measurements*. Far field frequency responses are added by clicking folder button on the right or dropping files into list. Enter *Distance* of low frequency and high frequency measurements and Merger tool will calculate scaling of LF responses automatically. Scaling can be adjusted via *Scale* text box by entering a value or by Up/Down keys.

High frequency part

Far field measurements can be added by clicking folder button or dropping files into list. Scaling can be adjusted manually via *Scale* text box on the right. Axial response is selected by checking *Axial* column in response file list. Default axial response is 0 degrees in horizontal plane. Merged responses (graph below High frequency part) for particular angle can be previewed by clicking corresponding response from file list.

Transition

Transition from low frequency to high frequency part can be set manually via *Frequency* text box, graph cursor or Up/Down keys or automatically by clicking binocular button . Automatic option searches for lowest magnitude crossing point of low and high frequency curves.

Warning is given with red background color if transition frequency exceeds maximum near field frequency of the largest low frequency radiator. $f_{NFmax} = c/\pi/Dd$ ($c = 344.0$ m/s).

Magnitude and phase blending range between low and high frequency parts can be selected from drop down list: none, 1, 2, 3 or 4 octaves.

Delay of low frequency part is calculated automatically on transition frequency change but can be adjusted manually.

Output

Choose which items You want to output. *Create merged responses* will combine low frequency and high frequency responses into individual response files. *Create/append extended data* will combine LF and HF responses into a single file, having LspCAD 6 extended data format. Merged responses are routed for currently selected driver in the main program by checking *Feed speaker* before saving.

Merged responses are exported as minimum-phase by checking *Minimum phase*. Measured and entered delays are lost and all responses at all frequencies are normalized to the same acoustic center = 0 mm. Color of merged phase response is lime in the chart.

Excess group delay of HF response at transition frequency x 1.4 is added to merged minimum phase response by checking *with GD of HF*. This option saves measured delay (at transition frequency x 1.4) and delay adjusted by user.

Minimum phase options may be needed if measured far field HF responses are not minimum-phase at transition frequency, though radiator is actually minimum-phase. Significant error is possible with some measurement programs if IR time window is short. Forcing to calculated minimum phase is not recommended if responses are measured with dual channel gear and phase error at transition frequency is only few degrees.

Destination Directory

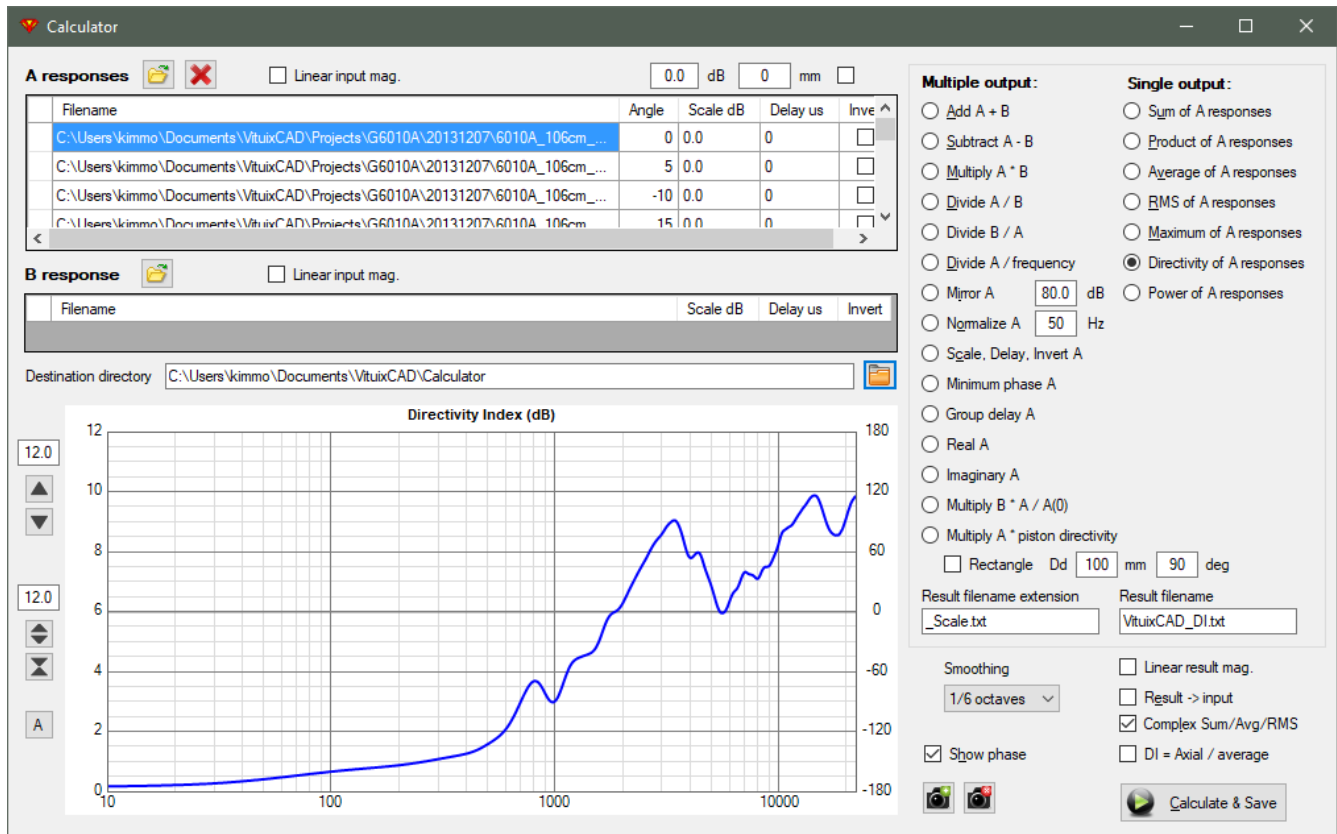
Choose work directory where You want to save output files.

Save and Open

Merger project can be saved with *Save* button in the bottom left corner. File extension is vxm, internally XML.

Saved merger project can be opened with *Open* button or dropping vxm-file into Merger tool window.

Calculator tool



Calculator tool is a generic math tool for response file processing. Most of the math skills required for speaker simulation are included in Enclosure tool, Merger tool and speaker simulator. Calculator tool provides some additional functions and flexibility to process calibration files, preprocess measurements before simulation or analyze complete speaker or a single driver.

Responses

Primary response list is 'A responses'. It accepts multiple responses. 'B response' is single response.

Add response files by clicking folder button or dropping files into list. Multiple file selection with Open dialog is enabled. Delete button clears whole response list. Uncheck *Linear input mag* if magnitude is in dB value in the response files. Check *Linear input mag* for impedance, pressure or voltage responses where magnitude is not in dB value. Phase angle unit should be degrees. Calculator wraps phase automatically within -180...+180 deg. All loaded files are processed.

Manual scaling, timing and polarity inversion is available for each response. Simultaneous scaling/timing/inversion of all responses is available above response list. Phase angle in the filename is detected and visible after filename. It is needed for directivity and power response calculation. See [Options](#) for more information.

Multiple output

Functions for each input response A, producing multiple result responses:

Add A + B

Responses A added by response B.

Default result filename extension is _Add.txt.

Add subwoofer response (B) to responses of main speaker (A).

Subtract A – B

Responses A subtracted by response B.

Default result filename extension is _Subtract.txt.

Multiply A * B

Responses A are multiplied by response B.

Default result filename extension is _Multiply.txt.

Create far field response by multiplying near field measurements (A) with cabinet diffraction simulation (B).

Test equalizer, high pass or low pass filter by multiplying raw responses (A) with filter transfer function (B).

Divide A / B

Responses A are divided by response B.

Default result filename extension is _DivideAB.txt.

Normalize polar measurements by dividing off-axis responses (A) with axial response (B).

Correct punch of uncalibrated measurements (A) by multiplying with calibration file, representing total frequency response of your measurement system (B).

Divide B / A

Response B is divided by responses A.

Default result filename extension is _DivideBA.txt.

Divide A / frequency

Responses A magnitude is divided by frequency.

Default result filename extension is _DivideByFreq.txt.

Calculate cone excursion response from near field measurement (A) by dividing each magnitude value with associate frequency.

Mirror A

Responses A mirroring aka vertical flipping over entered dB value.

Default result filename extension is _Mirror.txt.

Create equalizer target response by mirroring raw response over entered level.

Create correction response by mirroring total frequency response of your measurement system.

Normalize A

Responses A normalizing to magnitude of the first response A at entered frequency.

Default result filename extension is _Normalize.txt.

Reduce excess directivity of time-windowed off-axis responses by normalizing responses at 40 Hz of axial response in case you are sure that radiator is perfect omni until 40 Hz.

Scale, Delay, Invert A

No calculation - just responses A magnitude scaling, time shifting and polarity inversion.

Default result filename extension is _Scale.txt.

Scale measurements (A) to estimated or known SPL [dB/2.83V/1m].

Smooth measurements (A) without any other manipulation.

Resample measurements (A) from linear to logarithmic frequency increment; from response export of REW to 24...48 points/octave. This may require appropriate time shifting to maintain correct phase information.

Time shifting of measurements (A) if time reference (0 s) point is at the mic capsule or starting point of IR time window was too much before impulse peak.

Invert measurements (A) if mic polarity was inverted while measurement or your mic & preamp combination is constantly inverting.

Minimum phase A

Responses A converted to minimum-phase. Response tails below 10 Hz and above 22 kHz are estimated by the first and last 1/2 octaves. Default result filename extension is _MinPhase.txt.

Group delay A

Responses A group delay in milliseconds.

Default result filename extension is _GroupDelay.txt.

Real A

Responses A converted to real: phase angle is set to 0 deg or -180 deg if *Invert* is checked.

Default result filename extension is _Real.txt.

Create correction file for magnitude only. Normally this corrupts minimum phase features, but may be useful if phase information is irrelevant or harmful.

Create full impedance response for ideal resistive component from plain magnitude response.

Imaginary A

Responses A converted to imaginary: phase angle is set to 90 deg or -90 deg if *Invert* is checked.

Default result filename extension is _Imaginary.txt.

Create full impedance response for ideal reactive component from plain magnitude response.

Multiply B * A / A(0)

Creates off-axis responses for measured or captured axial response B with directivity information in responses A. Directivity data can be simulated with Diffraction tool or compatible set of far field measurements. Response A to 0 degrees is reference in directivity calculation.

Default result filename extension is _MultiplyBdA.txt.

Multiply A * piston directivity

Responses A multiplied by piston directivity. Calculation parameters are piston diameter for circular or width for rectangular radiator, and off-axis angle in degrees. Off-axis angle coded in filename in response list A is applied if 0 degrees is entered. Directivity function for circular radiator is $2 * J_1(k * a * \sin(\text{angle})) / (k * a * \sin(\text{angle}))$, where $J_1(x)$ is 1st order Bessel function of first kind, k =wave number and a =radius. Directivity function for rectangular radiator is $\text{Sinc}(k * x * \sin(\text{angle}))$, where x is width. Phase shift is approximated with $-k * x * \sin(\text{angle})$.

Default result filename extension is _PistonDir.txt.

- Create off-axis response including piston directivity from single axial or near field response.

Single output

Functions for multiple input responses, producing single result response:

Sum of A responses

$$= A_0 + A_1 + A_2 + \dots$$

Default result filename is VituixCAD_Sum.txt.

Create total response of multiple radiators

Sum near field measurements of all cones and ports. Each response A can be scaled for different radiating area.

Create total response of multiple ways/bands.

Simulate comb-filtering effects by summing non-delayed and delayed responses. See Complex calculation.

Product of A responses

$$= A_0 * A_1 * A_2 * \dots$$

Product calculation produces overflow error quite soon if several files is loaded. Typically product is needed for maximum two...three responses.

Default result filename is VituixCAD_Product.txt.

Average of A responses

$$= (A_0 + A_1 + A_2 + \dots) / N$$

Default result filename is VituixCAD_Average.txt.

Create listening window response by averaging ± 30 deg hor and ± 5 deg ver responses. See Complex calculation.

RMS of A responses

$$= \text{SQRT}((A_0^2 + A_1^2 + A_2^2 + \dots) / N)$$

RMS is alternative for simple average (arithmetic mean). Square scales single magnitude value for area or power, for example from sound pressure to intensity. See Complex calculation.

Default result filename is VituixCAD_RMS.txt.

Maximum of A responses

Searches maximum magnitude from responses (A) for each frequency point. Phase angle of result response is taken from selected row.

Default result filename is VituixCAD_Maximum.txt.

Create reference response for manual Directivity Index calculation if preferred reference response is maximum pressure within listening window instead of single axial response (which could contain diffraction dips).

Directivity of A responses

Calculates Directivity Factor $Q(f)$ from radial measurements (A) if *Linear result mag* is checked and *DI=Axial/average* is not checked.

Calculates Directivity Index $DI(f)$ from radial measurements (A) if *Linear result mag* and *DI=Axial/average* are not checked. Unit of result is dB.

Phase angle should be included in response filenames in order to calculate intensity on spherical surface from radial measurements. Angle step must be constant.

'Horizontal 0 deg' response is automatically selected as directivity reference. Content of that file should be modified in order to use some other measurement or calculated result as a reference.

Default result filename is VituixCAD_DI.txt.

Directivity can be calculated as axial to average pressure ratio by checking *DI=Axial/average*. This option is valid if polar response set is real 3D containing equally spaced measurements on full spherical surface around the radiator. Another application is to calculate either horizontal or vertical directivity, without requirement of correct result for full space.

Power of A responses

Power response approximation is calculated as Reference response magnitude + Directivity Index + $10 \cdot \log_{10}(4\pi)$.

This method requires valid responses for Directivity Index calculation, specified in the previous section.

Default result filename is VituixCAD_Power.txt.

Additional options

Complex Sum/Avg/RMS should be checked in order to calculate complex vector sum, average or RMS with phase angle information. Complex calculation is sensitive to phase angle; sum of two equal magnitudes with opposite polarity = 0. This is default option giving correct results with frequency responses.

Absolute magnitudes are summed if *Complex calculation* is not checked. This option is useful if phase information is too random or nonsymmetrical (like with multiple room responses) causing steep magnitude dips in result response. Phase angle of result response is calculated with complex numbers anyway, but minimum phase features are not completely maintained.

Smoothing options are 1/1, 1/2, 1/3, 1/6, 1/12 octaves or none.



Result files can be recycled to input responses by checking *Result -> input*. Multiple result files are recycled to responses A, and single result file to response B. This enables calculation sequences without manual loading of result files to input. Result files are saved in *Destination directory*, which can be changed via folder button.

Result files are created by clicking *Calculate & Save* button. Calculation to graph without result file creation is executed when response files are loaded or calculation formula is selected or smoothing or any other additional option is changed.

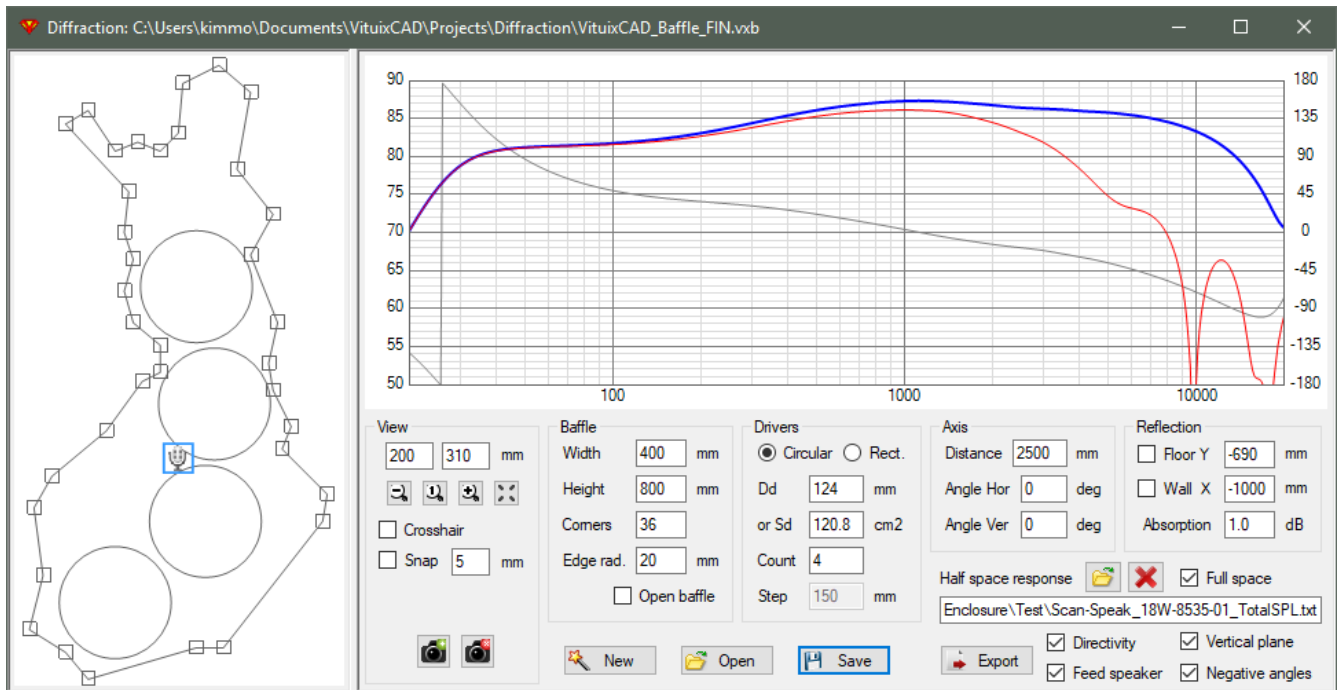
Graph

Enter title directly into graph for publishing of captured image.

Maximum and span of magnitude axis are adjusted by arrow buttons or entering value or Up/Down keys in the text boxes. Magnitude can be auto scaled by clicking A button. Scale of phase axis is constant -180...+180 deg.

Max. 10 overlays can be added into graph with *Add overlay* button . *Clear overlay*  deletes the latest visible overlay.

Diffraction tool



Diffraction simulator calculates cabinet impact or full space frequency response of driver(s) in a baffle. External diffraction simulator should not be selected in [Options window](#) in order to use this internal simulator. Simulation is based on simple ray theory: each driver is a point source sending 72 rays towards baffle edges with fixed 5 deg steps. Path lengths of shortest 1st order diffractions are calculated and summed with delay i.e. phase information in a listening point. Magnitude of diffracted rays is frequency dependent. Weighting factors are calculated from radiator dimensions and edge radius. Maximum amount of corners is 36, and minimum is 3. Baffle can contain up to 50 drivers with equal dimensions. Simplified room response with floor and side wall reflections is available.

Initializing

Start baffle designing by entering main dimensions; width, height and number of corners. Select driver shape; circular or rectangular. Enter effective diameter Dd or area Sd for circular or width and height for rectangular, number of drivers and Step which is vertical distance between drivers (if more than one). Press **New** button to create initial shape with drivers and mic in the drawing area.

Adding and deleting of corners and drivers is available via pop-up menu. Corner is added in the cursor position, but you may need to rotate two corners to get new one to correct position. Driver is added in the cursor position.

Zooming Out, 1, In and Fit are available via zoom buttons . You can pan baffle image by dragging when nothing is selected (blank area clicked). Unit of location coordinates is millimeter. Zoom 1 equals 1px=1mm. Origin (0, 0) is initially in the bottom left corner. Origin can be moved to cursor position by clicking left button when Ctrl key is pressed. If corner, driver or microphone is selected, origin moves to the center point of selection.

Half space frequency response of driver can be loaded for full space response simulation. Enable full space view and export by checking **Full space**. Y-scale is adjusted automatically.

Check **Open baffle** for dipole speaker with thin I-baffle.

Adjusting

Individual objects; corners, drivers and microphone are selected by clicking left button above the image. Selected object is highlighted and location (X, Y) is visible in **View** group. Location can be adjusted by dragging or entering coordinates or pressing arrow keys in the text boxes. Snap setting restricts object movement to intervals specified with the text box. Snap mode is activated by a checkbox. Object is moving orthogonally if Shift key is pressed while dragging. Crosshair cursor helps object aligning. Cabinet impact response graph is updated while adjusting.

Remove selection (deselect) by clicking blank area. Coordinates of mouse cursor are shown as long as nothing is selected.

Edge radius and *Driver Dd* or *Sd*, *Width* and *Height* can be adjusted after project initialization by entering value or pressing up/down keys in the text boxes. Drawing and cabinet impact response graph are updated while adjusting.

Mic/listening point

Microphone symbol represents listening point at *Axis Distance*. Baffle can be rotated horizontally and vertically by entering *Axis Angle Hor* or *Ver*, or pressing up/down keys in the text boxes. Rotation origin is perpendicular point of microphone on (unrotated) baffle surface.

Reflections

Check *Floor* and enter floor elevation in mm to include floor bounce in frequency response. Side wall reflection is included by checking *Wall*. Negative X coordinate locates wall on the left side. Enter absorption within 0...20 dB to reduce amplitude of boundary reflections.

File functions



Save baffle project by pressing *Save* button. File extension is .vxb, but it's human-readable xml file containing main dimensions and location of corners and drivers. You can print the file for helping final enclosure design.

Open existing baffle project by pressing *Open* button.

Export simulated Cabinet impact response or full space frequency response shown in the graph by pressing *Export* button.

You can also export off-axis responses with 10 deg steps by checking *Directivity*. Responses in vertical plane 10...170 deg are exported by checking *Vertical plane*. Responses to negative angles -10...-170 deg are exported by checking *Negative angles*. Responses are routed for currently selected driver in the main program by checking *Feed speaker* before exporting. Enter root file name for responses. Program extends file name with horizontal/vertical keyword and angle, using coding defined in [Options window](#).

Graph

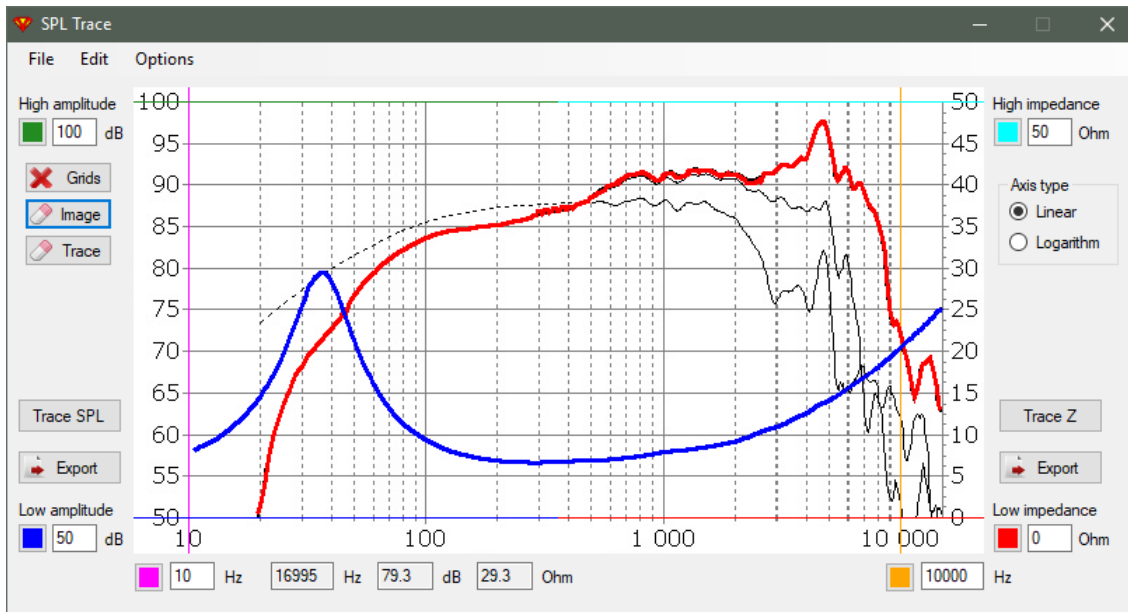
Max. 10 overlays can be added into response graph with *Add overlay* button . *Clear overlay*  deletes the latest visible overlay.

Note!

1) *Bevel* can be entered as *radius*.

2) *Long planar radiator* should be constructed by stacking multiple small rectangular drivers with *Step* = *Height*.

SPL Trace



SPL Trace tool can capture frequency and impedance responses from bitmap loaded from file or pasted from the clipboard. Several image file types are supported such as png, jpg, bmp and gif. Some external tool or print screen is needed for copying bitmap to clipboard from datasheet (pdf) or web page. *Snipping Tool* installed in Windows 10 works fine.

Recommended size of bitmap is 800x400px or more to enable adequate resolution and accuracy. Bitmap is shown in *SPL Trace* window with original size or fitted smaller if needed.

Note! Background color of bitmap should be white or other very light with luminance (HSL) >90 %. Dark background should be replaced with image editing software such as paint.net.

Extracting of responses:

1. Open image file (File->Open image or Ctrl+O) or paste from the clipboard (Edit -> Paste image or Ctrl+V).
2. Set frequency boundaries: Point and drag the lowest labeled frequency in the graph with magenta cursor. Enter frequency to the text box, or adjust value with up/down arrow keys or mouse wheel. Point and drag the highest labeled frequency with orange cursor. Enter or adjust frequency value in the text box.
3. Set SPL boundaries: Point and drag the lowest labeled SPL with blue cursor. Enter or adjust SPL value in the text box. Point and drag the highest labeled SPL with green cursor. Enter or adjust SPL value in the text box.
4. Set impedance boundaries: Point and drag the lowest labeled impedance with red cursor. Enter or adjust impedance value in the text box. Point and drag the highest labeled impedance with cyan cursor. Enter or adjust impedance value in the text box.
5. Select impedance axis type: Linear or Logarithm. Linear scale has absolute increments and typically starts from zero Ohms. For example from 0 to 60 Ohms with steps of 10 Ohms. Intervals of logarithmic scale are ratios, and scale starts above zero Ohms. For example 2, 4, 8, 16, 32, 64 Ohms.
6. Move mouse cursor inside the graph area and check that Hz, dB and Ohm values in the status line are calculated correctly.

Warning: *Paste* and *File open* commands reset previous settings if image size changes.

7. Erase gridlines within chart area if luminance of line color is <90 % and color difference to response traces is small. No need to erase if gridlines are very light gray (or dotted line) and response traces are thick with saturated color (red, green, blue, magenta, cyan).

Click **X Grids** button to start erasing. Point with crosshair cursor the lines where all horizontal and vertical

gridlines are found, but the cursor lines crosses response traces as little as possible. Accept with left click. Gridlines (with luminance <90 %) should be gone.

8. You can use *Image rubber* for cleaning vicinity of response curves. Crossing off-axis responses could interfere tracing of axial and impedance responses.
Warning! Do not cut axial or impedance responses too much because *Undo* is not available.
9. Trace amplitude response. Click **Trace SPL** button and then starting point of frequency response at the lowest visible frequency. Click again few pixels further if tracing was not started. Traced path is marked with thick red overlay line. If the tracing is interrupted or distracted in an intersection, you can correct the path by clicking a point few pixels after the problem. You can also restart from earlier position or beginning as many times you like.
10. You can use *Trace rubber* for cleaning bad points of traced curve. For example short jumps above or below correct path. Erased parts will be interpolated when the response is exported to a file.

Check Options -> *Smooth* 1/12 oct. to remove small peaks and tiny steps due to pixels of original bitmap. Smoothing of 1/12 oct. is applied to exported responses.

Default density of export is 48 points/octave. Lower densities are available in Options menu, but not recommended if responses are created for VituixCAD.

11. Export amplitude response with **Export** button after tracing is completed and red overlay line looks fine. Enter filename and select extension (txt or frd). Phase response is calculated automatically as minimum phase. Program estimates slopes of frequency response extensions below and above visible range.
12. Trace impedance response. Click **Trace Z** button and then starting point of impedance response at the lowest visible frequency. And so on...
13. Export impedance response with **Export** button...

Traced sound pressure points can be copied to clipboard with Edit -> Copy raw SPL. Traced impedance points can be copied to clipboard with Edit -> Copy raw Z. Phase angle is zero and decimal symbol defined in Control panel.

Press Shift key while moving mouse above the image to open circular **Magnifier** of 400 %.

Auxiliary calculator

Time window

Diagram showing a speaker and a microphone with distances of 1000 mm. The results are:

Reflection	2236 mm
Time window	3.59 ms
Min. frequency 1/1 oct.	278 Hz
1/2 oct.	557 Hz
1/3 oct.	835 Hz

Calculates maximum length of time window (ms) without reflections. Enter shortest distance from driver to mic, from driver to floor and from mic to floor (mm). Calculates also path length of reflection (mm) and minimum frequency (Hz) with 1/1, 1/2 and 1/3 octs. resolution.

Wave length

Diagram showing a speaker and a microphone with distances of 1000 mm. The results are:

Frequency	1000 Hz
Full wave length	344.0 mm
1/2 wave length	172.0 mm
1/4 wave length	86.00 mm
Travel time	1.000 ms
	0.5000 ms
	0.2500 ms

Calculates full or 1/2 or 1/4 wave length (mm) and sound travel time (ms) in air by frequency (Hz). Enter full or 1/2 or 1/4 wave length to calculate frequency and travel time. Enter travel time of full wave or fraction to calculate frequency and wave length.

Near field

Diagram showing a speaker and a microphone with distances of 1000 mm. The results are:

Piston diameter Dd	100 mm
Effective area Sd	78.54 cm ²
Max. near field frequency	1095 Hz
Max. distance to mic	5.5 mm

Calculates maximum near field frequency (Hz) and maximum distance (mm) to microphone by effective piston diameter (Dd mm) or area (Sd cm²). This gives approximately -0.5 dB error at f_{NFmax} in theory if mic is in the center point of ideal piston. In practice maximum applicable frequency is usually lower due to non-centered mic location and mechanical properties of the cone.

$$f_{NFmax} = c/\pi/Dd \quad (c = 344.0 \text{ m/s})$$

Max. distance to mic is 5.5% of Dd.

Sample time

	Time	Distance in air
1 sample	20.83 us	7.167 mm
2 samples	41.67 us	14.33 mm
139 samples	2896 us	996.2 mm

Calculates time (μs) and distance in air (mm) for 1 and 2 and any number of samples by sample rate (Hz). You can also calculate number of samples and distance by time, or number of samples and time by distance.