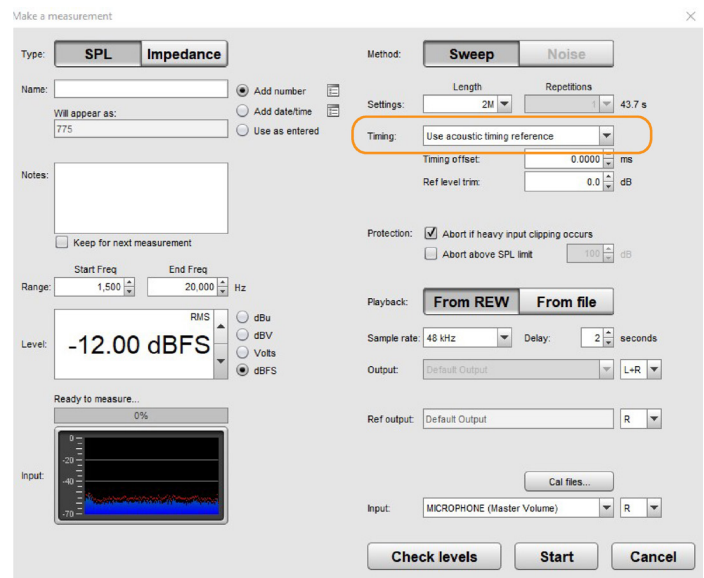
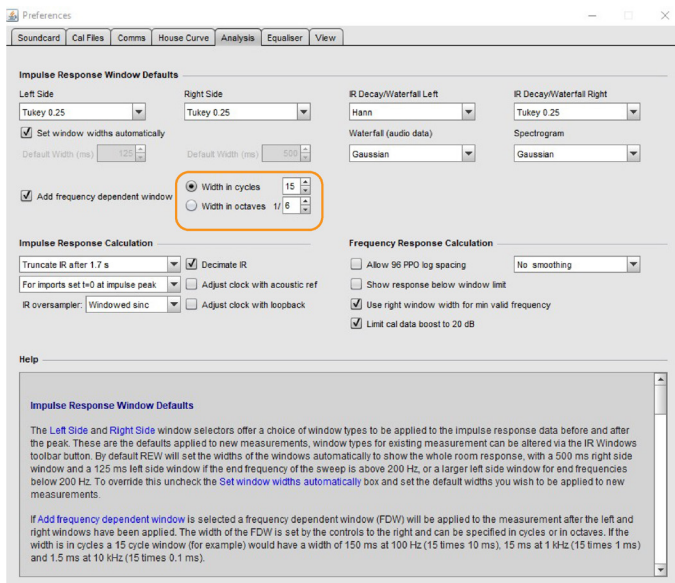


REW average measurements and impulse correction rePhase

- Tutorial is meant to be used for speakers/ drivers implementation into your listening area. Integrating impulse corrected filters for your DSP device with or without "room curve". "Room curve" bases is Bruel and Kjer work in room acoustics: <https://www.bksv.com/media/doc/17-197.pdf> Goal is to produce the best possibly sounding speakers /drivers for your room. Take a measurement point microfon`s tip vertically ,towards sealing and use a 90° calibration file for your mikrofon, resulting in much better measuring results for REW.

1.REW Preferences



1/6th octave smoothing and 15 cycles FDW to generate the correction filters and avoid 'micro-managing' the amplitude and phase corrections.

Timing reference activated.

2. Room curve

Data used to make a "room curve" can be imported to REW as a text file or using "target settings" in REW.

"EQ" -> "target settings" -> activate "Add room curve".

It could take to do the same tuning to reach the desired result and this task is up to you. Settings in a picture is just a guideline/starting point.

Assuming you follow this tutorial with a goal is to implement a "room curve" with impulse corrected filters for your DSP device.

Then, "Equalizer" chooses "rePhase", because we will end up with producing FIR filters in rePhase software for your device.



Importing text file into the REW software.

First you must make a text file. Copy digits and paste in to your text editor and save it as txt file with a name (free choice) :

25.198 0.000
31.748 -0.001
40.000 -0.005
50.397 -0.016
63.496 -0.039
80.000 -0.079
100.794 -0.134
126.992 -0.203
160.000 -0.290
201.587 -0.397
253.984 -0.528
320.000 -0.683
403.175 -0.866
507.968 -1.087
640.000 -1.351
806.349 -1.651
1015.937 -1.972
1280.000 -2.302
1612.699 -2.634
2031.873 -2.967
2560.000 -3.300
3225.398 -3.634
4063.747 -3.967
5120.000 -4.300
6450.796 -4.634
8127.493 -4.967
10240.000 -5.301
12901.592 -5.634
16254.986 -5.967
20480.000 -6.301
22050.000 -6.301

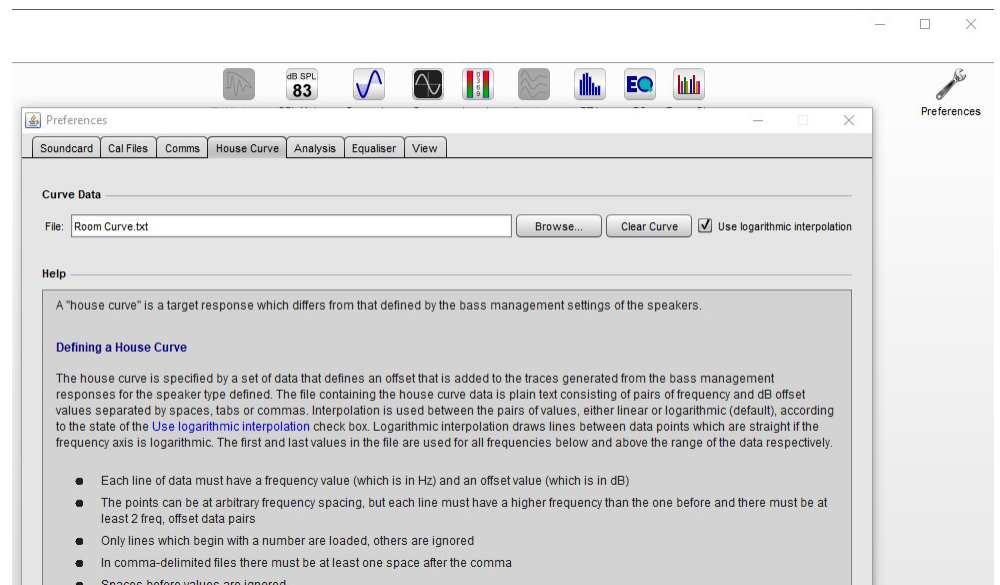
This room curve based on Bob Katz' recommendations, diyAudio forum.

0 -90.0
5 -6
6 -3
7.5 -1
10 -0.1
20 0.0
50 0.0
100 -1
200 -2
400 -3.0
800 -4.0
1000 -4.2
2000 -4.7
3200 -5.3
6400 -6.5
12800 -7.5
19200 -8
22050 -10

This room curve based on "fluid" recommendations, diyAudio forum.

Then saved file import to REW.

"Preferences" ->"House Curve"->"Browser" find your saved file and import it.



How to Integrate impulse corrected filters and "room curve" for your DSP device ,will be explained in detail in a step 7. Combined filter

Each and every person on this earth is equipped with a singular hearing organ belonging only to him and no one else has it the same sound interception experience. "Room curve" is how most people would love sound to be in their listening room for music. It is up to you to use " room curve" or don't.

If you are using it:

"room curve" should be activated for all times in REW software. From step one of this tutorial up to the end.

If you are not using it:

Just don't activate .

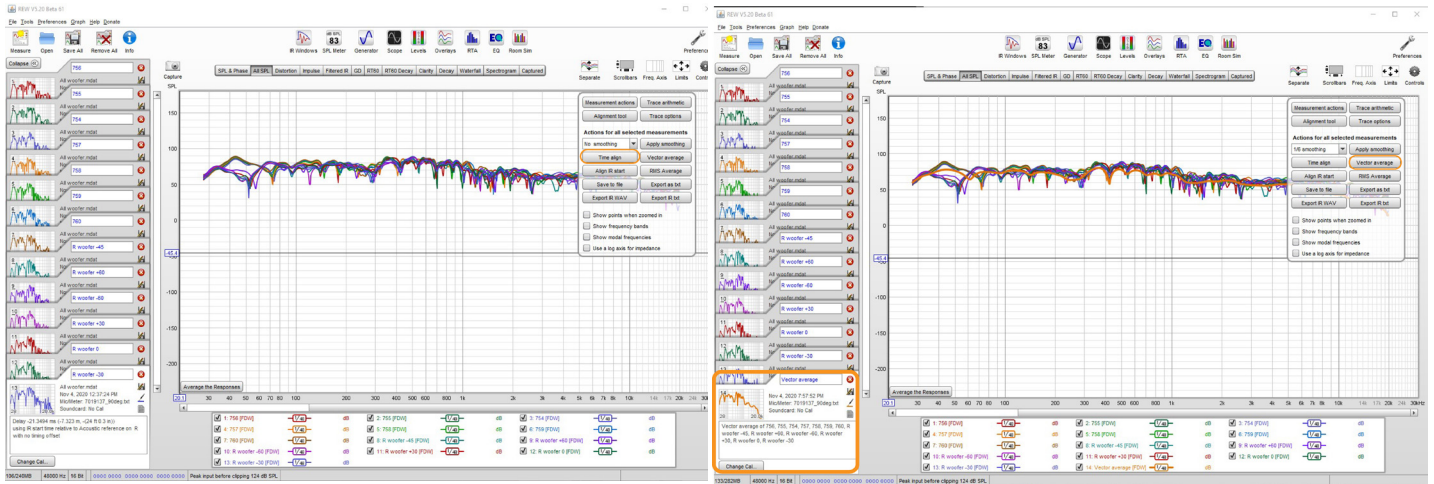
How you will use the tutorial depends on your DSP hardware/software set up. Following up to the end of tutorial step 7. Combined filter,we will end up with having needed information for rePhase software to produce file for FIR filter for your hardware/software.

Theoretically it would be only one filter that you need,minimizing use of PEQ filters.

If your set up doesn't have FIR in it,tutorial for you it is usefully up to step 4.PEQ generated filter.

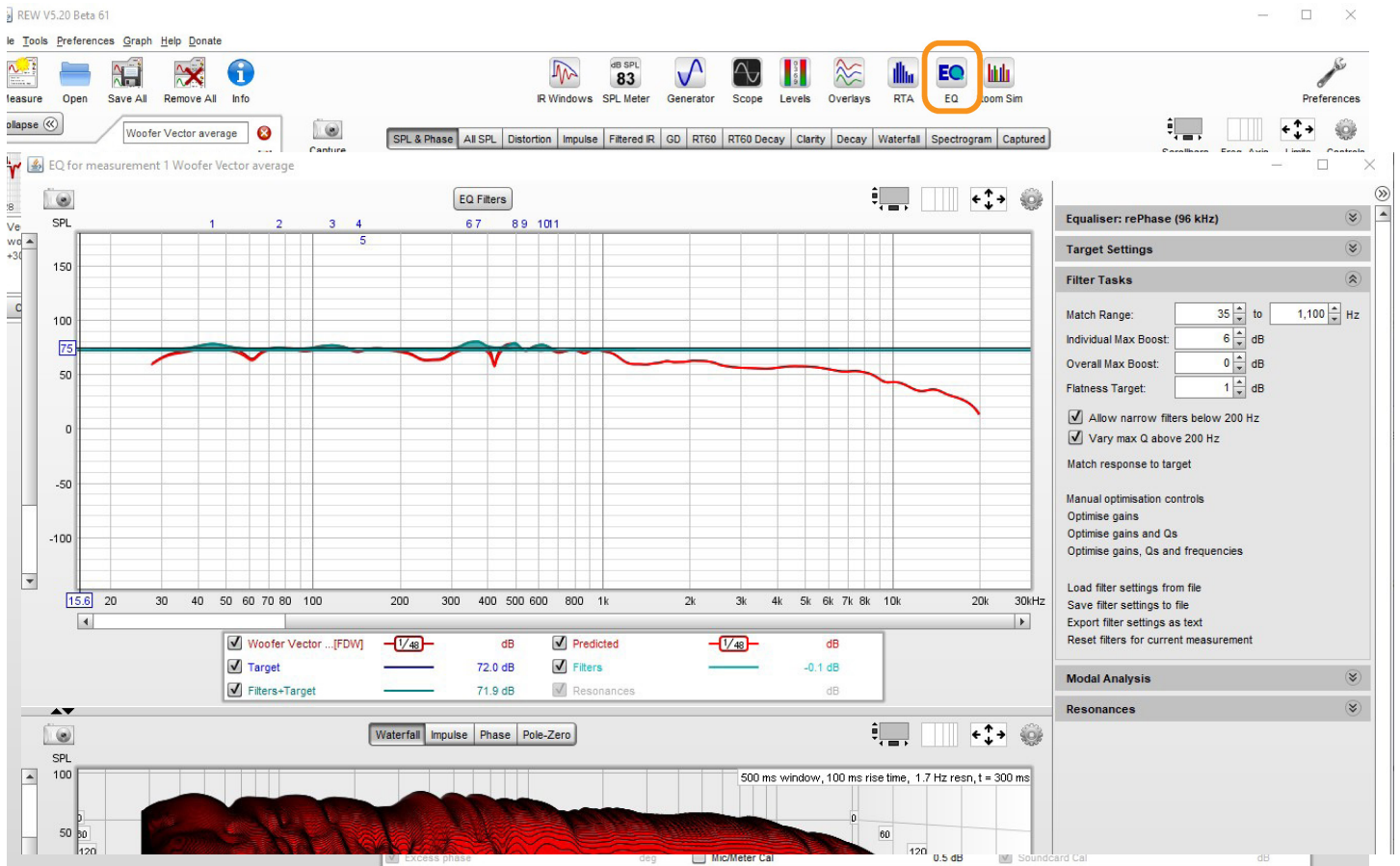
3. Averaged measurements.

Take several meaningful measurements representing your listening area . Import all of them in to the REW .
"All SPL" -> "Control" -> "Time Align"->"Vector average".

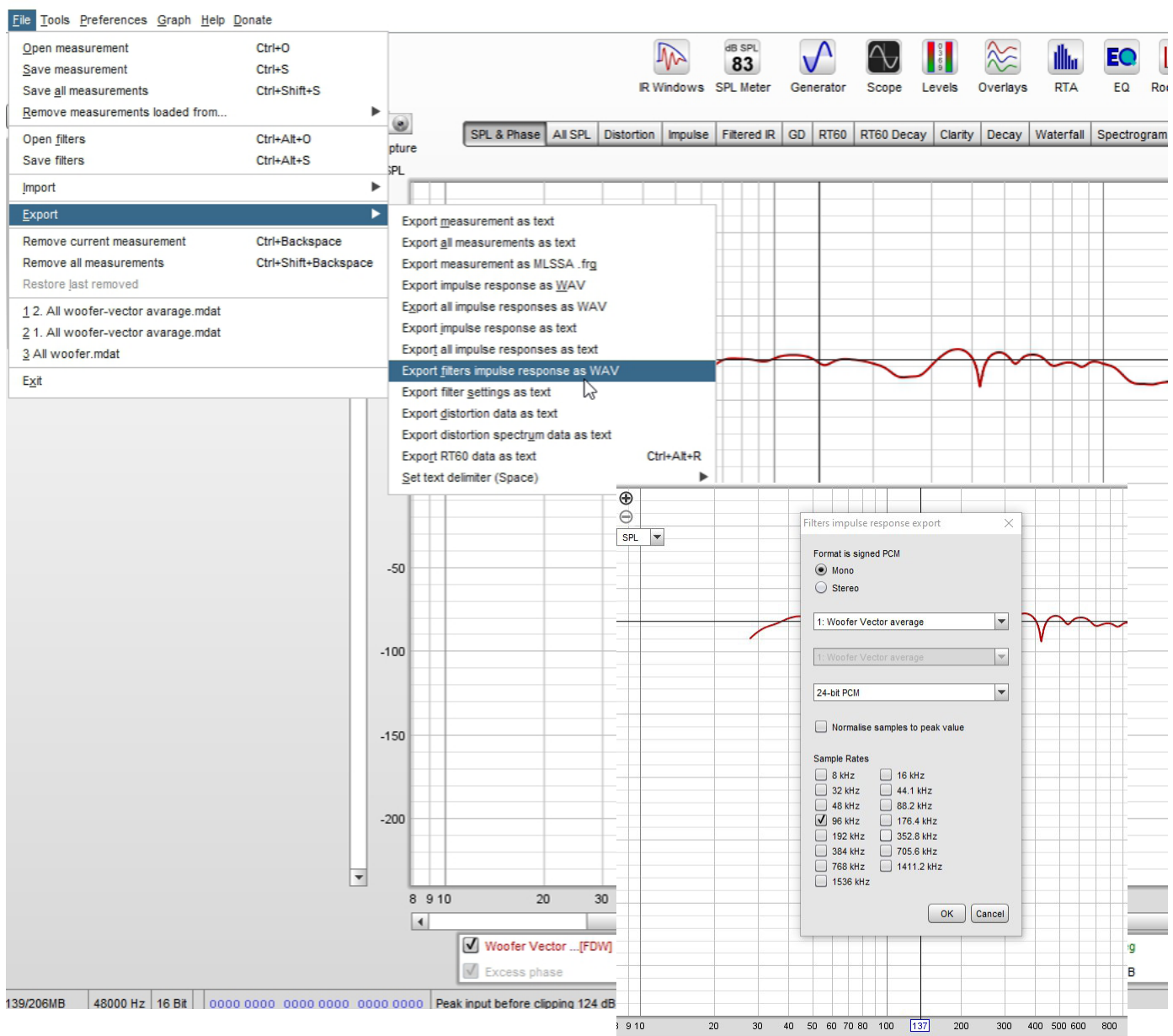


4. PEQ generated filter

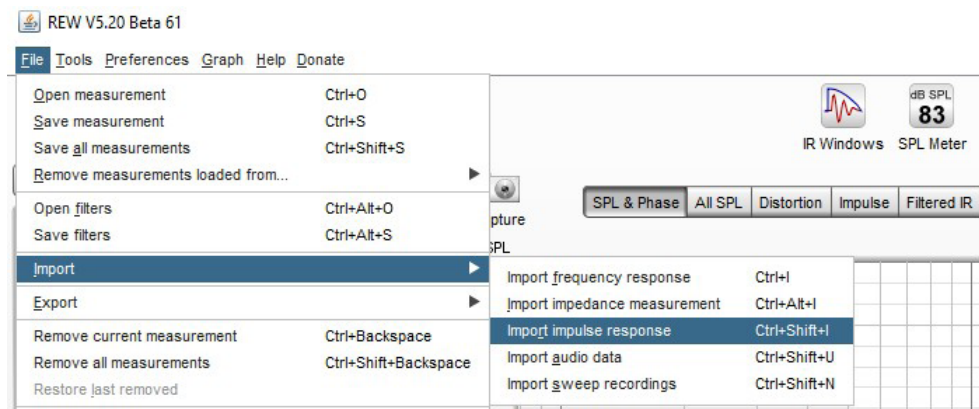
We have ended up with a generated "Vector average" in REW measurement. Next is to make a PEQ filter for "Vector average" measurement. At this point you would like to save the PEQ filter to be used in your DSP.
Or exported PEQ filter before as an xml for RePhase to create a combined filter -step 6. Combined filter



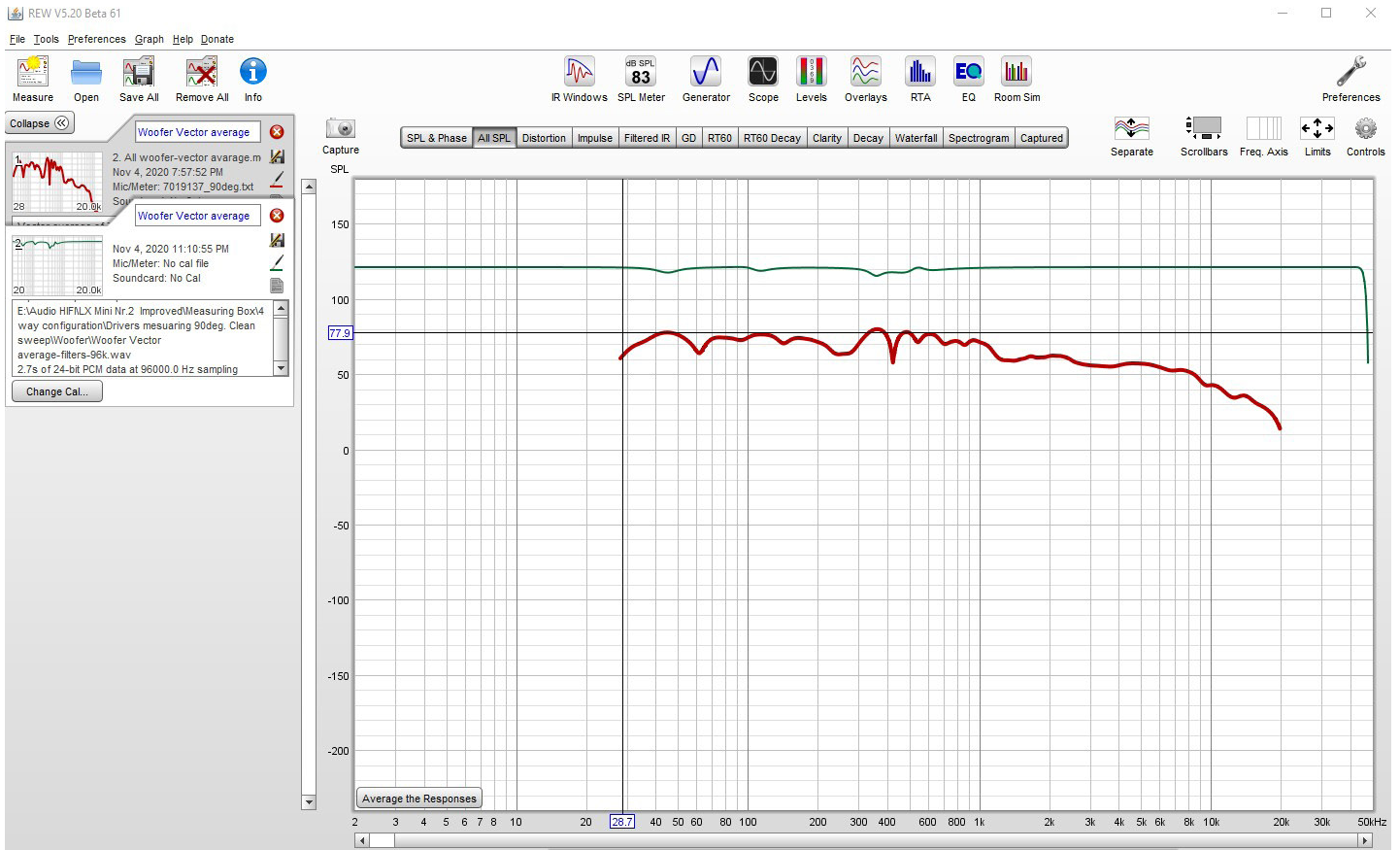
Go back to main REW window and File-> Export -> Export filters impulse response as a wav file and save it .



Import wav file. Main REW window " All SPL". Controls--> File--> Import --> Import impulse response.



Result should be “ Vector average ” and “ Vector average .wav ” measurements with in a main REW window “All SPS” tab activated.



5.Trace Arithmetic.

Controls -> Trace arithmetic -> Choose both measurements in windows A and B -> Choose A*B -> Generate

Decay Clarity Decay Waterfall Spectrogram Captured

Separate Scrollbars Freq. Axis Limits **Controls**

Trace arithmetic

A: 1: Woofer Vector average

B: 2: W. Vector average wav

A * B Generate

Measurement actions **Trace arithmetic**

Alignment tool Trace options

Actions for all selected measurements

No smoothing Apply smoothing

Time align Vector average

Align IR start RMS Average

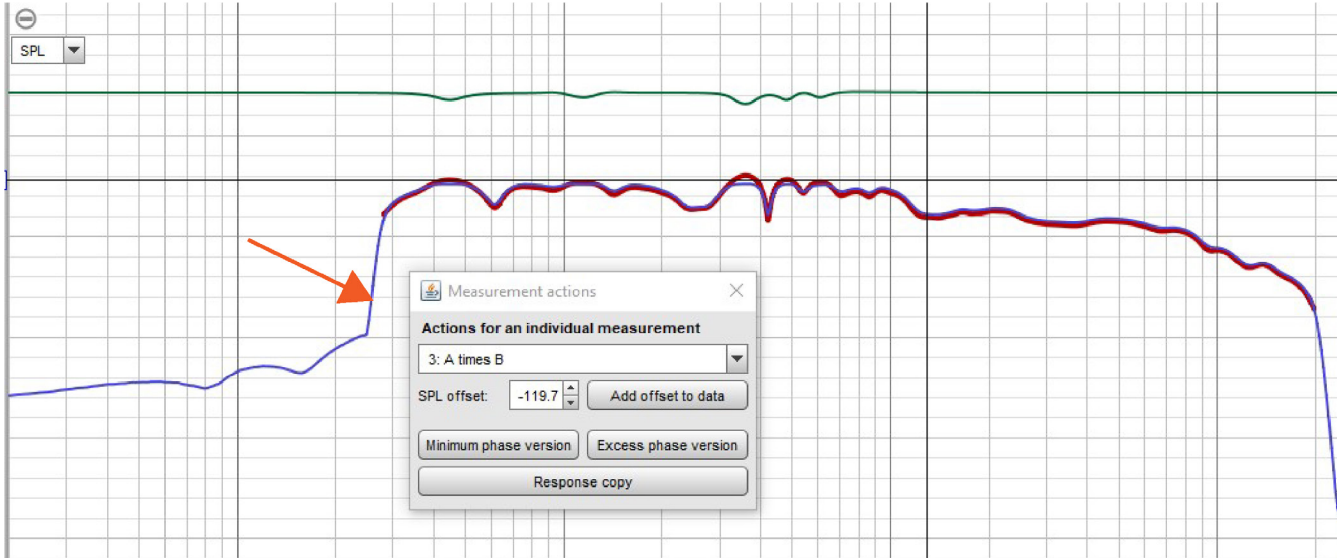
Save to file Export as txt

Export IR WAV Export IR txt

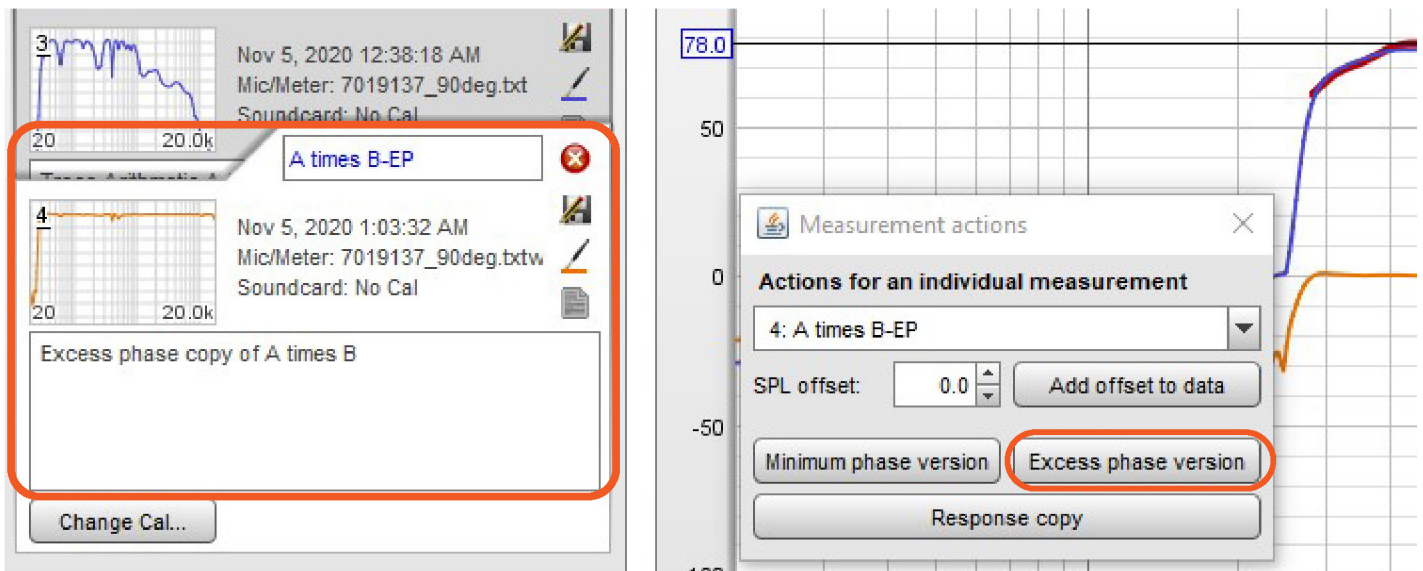
Show points when zoomed in

Controls -> Measurement actions -> A times B -> then enter a negative "SPL offset" to match "Vector average" value -> when you are happy press "Add offset to data".

The ultimate level does not matter for this only relative level so use the same amount for each channel. If the level not be reduced the measurement will end up at 150dB or more.



When press " Excess phase version". That will result to "A times B-EP" measurement .



6. Finalization

Main REW window " All SPL"File>Export>Export measurement as text.

Import saved txt file to rePhase. Then from rePhase generate *.bin file be used in MiniDsp FIR filter.

7. Combined filter

Everything the same as in the first part of step **“3.PEQ generated filter”**.

“Vector average ” measurement. Next is to make PEQ filter for “Vector average “. Choose “Equalizer Rephase” -Filter Tasks-Save filter settings to file. Filtar will be saved as *.xml file

The screenshot shows the rePhase software interface. On the left is a frequency response graph with a red curve and a green reference line. In the foreground, a 'Select speaker' dialog box is open, showing 'Left Subwoofer' selected in a dropdown menu, with 'OK' and 'Cancel' buttons. On the right, the 'Target Settings' panel is visible, with the 'Filter Tasks' section highlighted. The 'Filter Tasks' section includes: Match Range: 30 to 800 Hz; Individual Max Boost: 6 dB; Overall Max Boost: 0 dB; Flatness Target: 1 dB; and two checked options: 'Allow narrow filters below 200 Hz' and 'Vary max Q above 200 Hz'. Below these are 'Manual optimisation controls' (Optimise gains, Optimise gains and Qs, Optimise gains, Qs and frequencies), 'Load filter settings from file' (with 'Save filter settings to file' highlighted), 'Export filter settings as text', and 'Reset filters for current measurement'.

Open Rephase import *.txt file made in step **6.Finalization**. Then : Paragraphic Gain EQ -> Tools-> Import REW filter settings and import saved *.xml file

The screenshot shows the rePhase software interface. On the left, the 'File' menu is open, showing options: Load Settings..., Load Recent Settings, Load Settings From Clipboard, Save Settings, Save Settings As..., Save Settings To Clipboard, Reset Settings, Import Measurement... (highlighted), Import Measurement From Clipboard, Clear Measurement, and Clear Result. On the right, the 'Paragraphic Gain EQ' window is open, showing a frequency response graph with a range of ±12dB. The 'Tools' dropdown menu is open, showing options: load EQ settings..., load EQ settings from clipboard, save EQ settings as..., save EQ settings to clipboard, import REW filter settings... (highlighted), toggle constant Q / proportional Q, invert, bypass all, and activate all.

Finally "Generated" file in RePhase for your device.

Start Woofer - rePhase 1.4.3

File View Help

140
120
100
80
60
40
20
0
-20
-40
-60
-80
-100

10Hz 20Hz 50Hz 100Hz 200Hz 500Hz 1kHz 2kHz 5kHz 10kHz 20kHz

180°
157.5°
135°
112.5°
90°
67.5°
45°
22.5°
0°
-22.5°
-45°
-67.5°
-90°
-112.5°
-135°
-157.5°
-180°

General Filters Linearization Linear-Phase Filters Minimum-Phase Filters Paragraphic Phase EQ Paragraphic Gain EQ

Bank 01 EQ type constant Q minimum-phase Range ± 12dB Tools Presets

+12
0
-12

dB	-5.80	-1.70	-4.10	+5.80	-4.40	-4.80	-5.20	-7.40	+6.00	-9.30	+5.80	0.00	0.00	0.00	0.00	0.00	0.00
Q	4.64	9.17	4.76	6.22	3.22	9.55	11.77	12.71	10.54	8.55	5.09	10	10	10	10	10	10
Hz	44.9	75.9	116	143.5	148.5	344	372	498	534	624	674	100	100	100	100	100	100
	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass	bypass

Impulse Settings

taps 1024 samples

FFT length 16384 samples

centering 0%
use closest perfect impulse

windowing blackman

optimization none to -100 dB

rate 96000 Hz

format 32 bits IEEE-754 (.bin)

filename woofercombinedfilter

directory E:\Audio HIFi\X Mini Nr.2 Improv

generate

impulse delay: 0.001 samples, 0 ms
max response: 0.21 dB, max impulse: -0.05 dB

Ranges Measurement

import from file import from clipboard
bypass clear

gain offset 0 dB

time offset 0 usec

hide magnitude hide phase
invert response invert polarity

name: to rephase
data: 557 mag points : 557 phase points
ranges: [20Hz, 20kHz] : [-37.5dB, 0.84dB]

My settings for MiniDsp FIR filter

Special thanks "fluid" from diyAudio without this involvement this tutorial was not to be born in to existence.
Thank you "fluid"