

MDX 0408

FIR Filters Creation User Manual

User Manual – Version 1.0



FIR filter and applications

When user uses PEQ to adjust audio signal and set a linear magnitude, he can find the phase of signal changed, due to IIR filter. However, DSP products provide user a useful tool FIR filter to adjust audio signal with a linear phase.



Some calculation:

Frequency resolution = Sampling/Taps

Available min. frequency \approx Frequency resolution*3

Means when use adjust audio signal with 48kHz, 1024 taps, FIR filters will take effect in frequency above 141Hz of audio signal. The taps value more high, the FIR filter curve more steep.

FIR filter processing audio signal will produce a certain delay:

$$\text{Delay} = (1/\text{Sampling Hz}) * \text{Taps} / 2$$

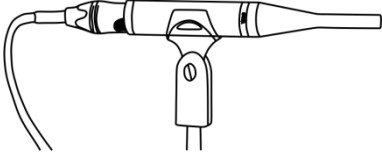
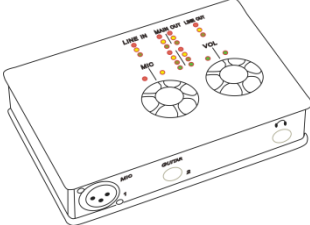

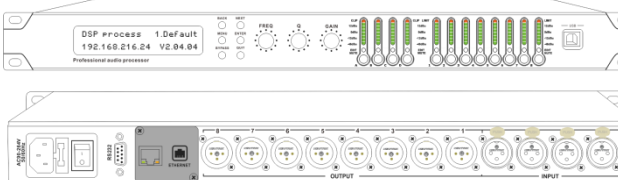
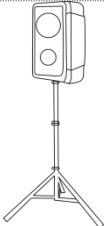
***Outputs 7 and 8 do not support 512 taps**

Taps	Sampling	48kHz	96kHz
256		2.67ms, LF 563Hz	1.33ms, LF 1125Hz
512		5.33ms, LF 279Hz	2.67ms, LF 558Hz
768		7.99ms, LF 188Hz	4.00ms, LF 375Hz
1024		10.67ms, LF 141Hz	5.33ms, LF 281Hz
2048		21.33ms, LF 70Hz	10.67ms, LF 141Hz

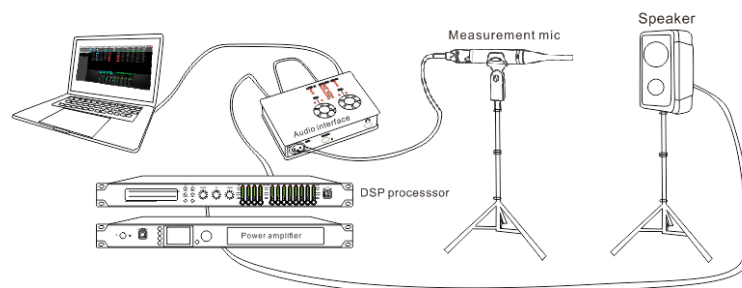
Applications:

- Linear of the phase curve of the speaker;
- Match the phase and magnitude of different speaker models within the same product line, as well as different speaker models in the installation project to make it easier to debug speaker groups and arrays;
- Dealing with linear array systems (for audience area coverage optimization);
- Frequency division optimization to improve the consistency of frequency response of multi-division speakers over their coverage Angle range.

Devices required:

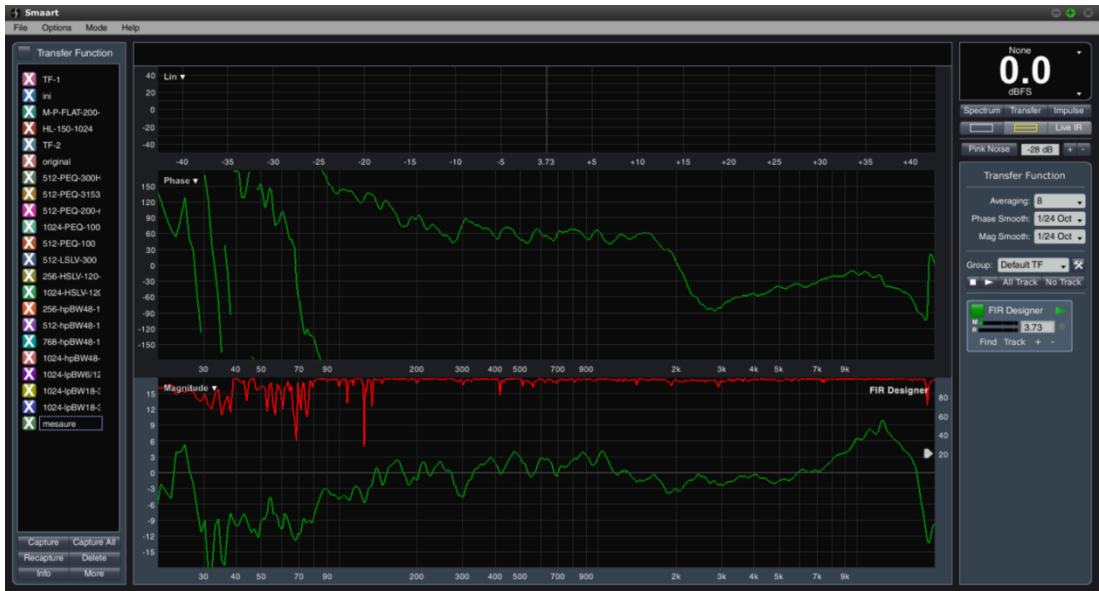
Measurement Microphone	x1	
Audio Interface	x1	
Windows PC (installed software including Smaart, rePhase or FIR Designer, Mconsole)	x1	
FIR audio processor or DSP network power amplifier	x1	
Speaker	x1	

Connection schematic diagram:

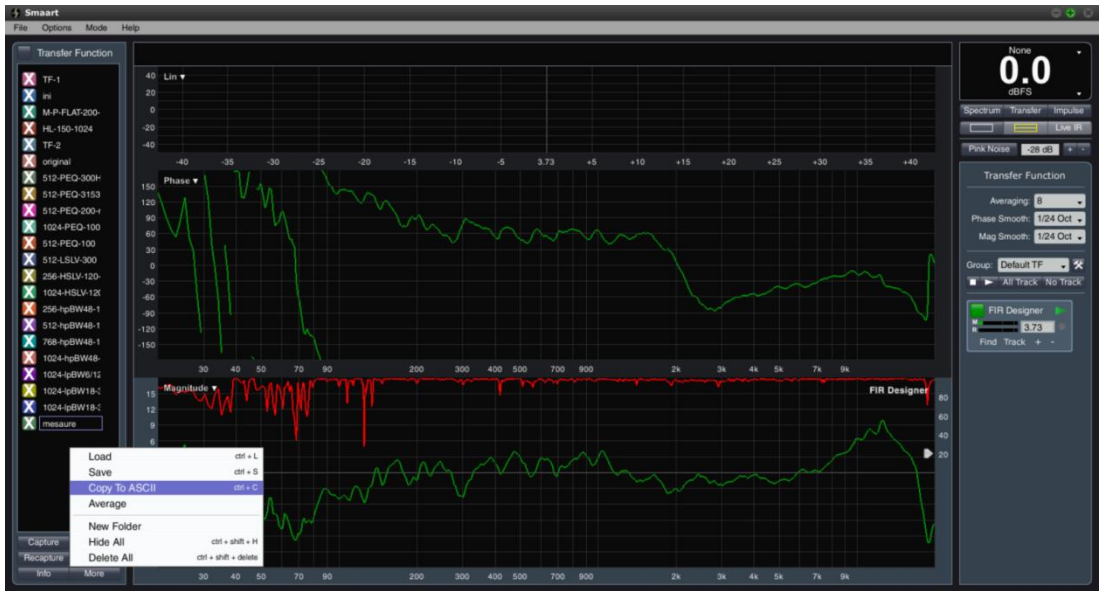


Using third party software to set FIR magnitude and phase

Step 1: measure phase curve of speaker in Smaart V7

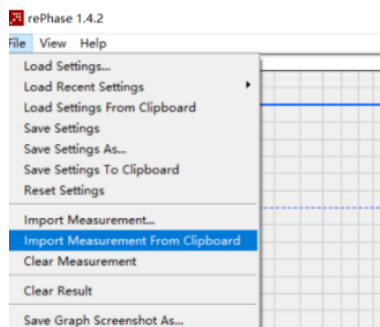


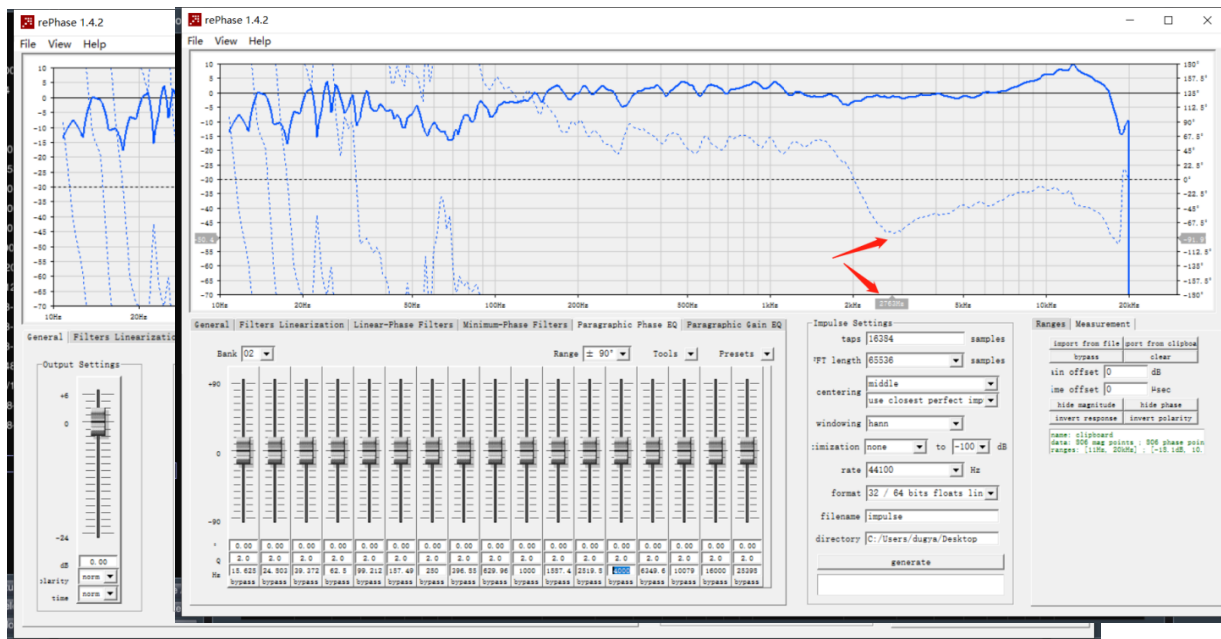
Step 2: copy curve to ASCII in Smaart V7



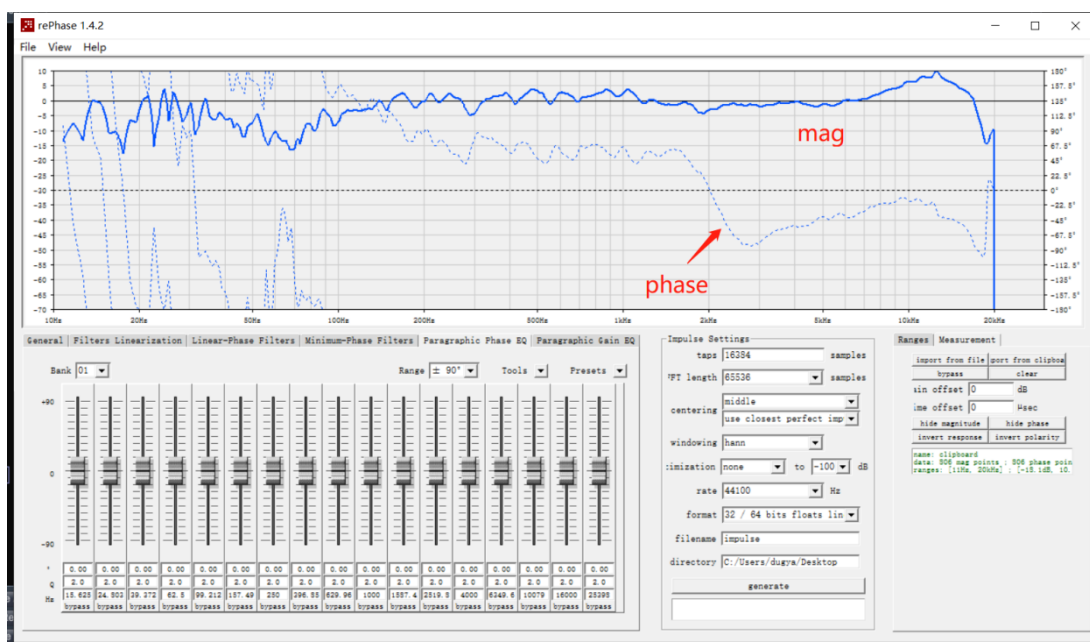
Step 3: copy curve to software rePhase

“Import Measurement From Clipboard”

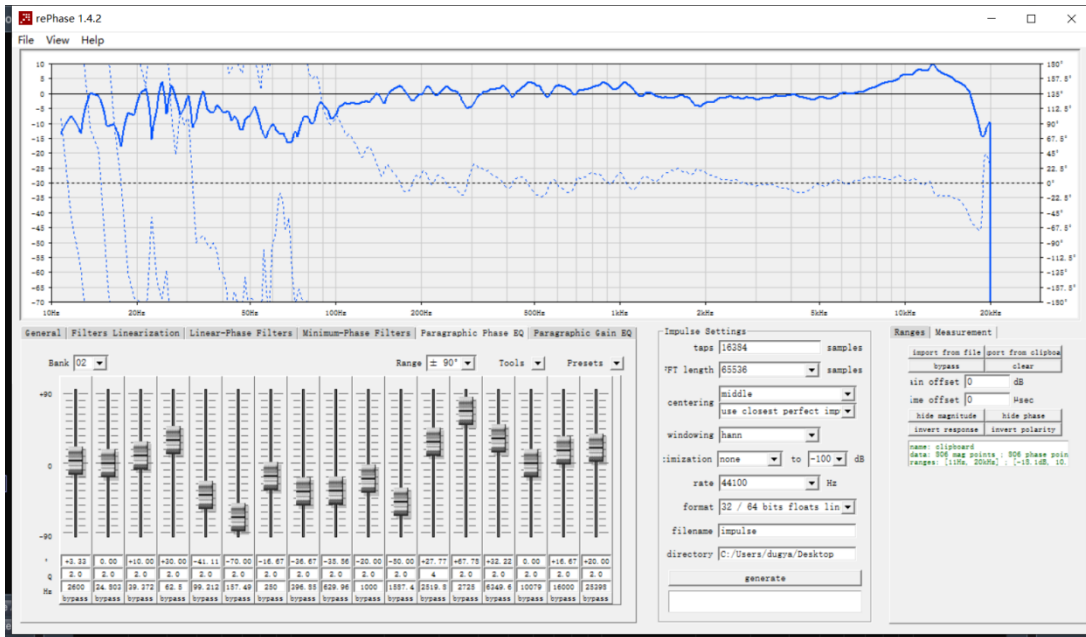




Step 4: adjust phase EQ or any other parameter in software, to match a linear phase for speaker



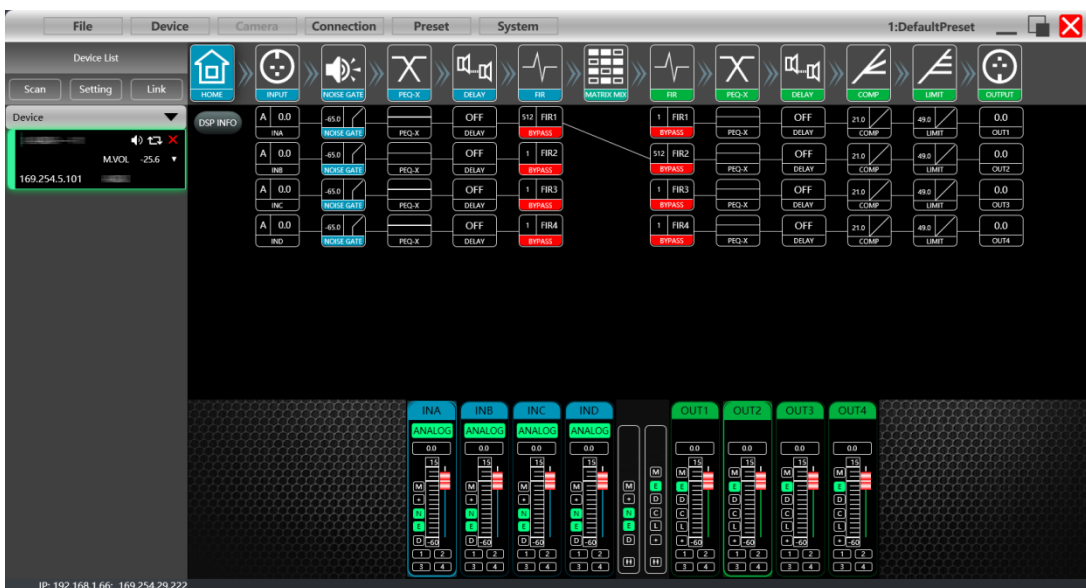
Step 5: export .txt file after setting



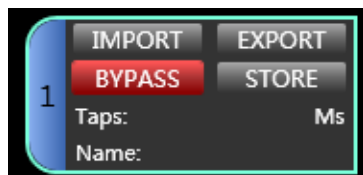
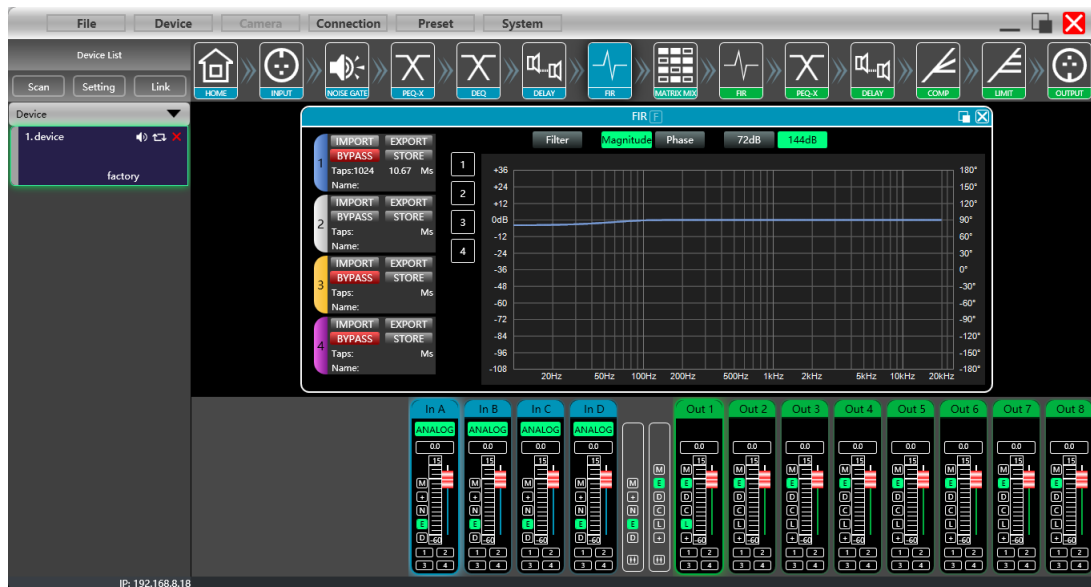
Marks:

1. Set taps in 2048/1024/768/512/256, here we set in 512.
2. Set rate in 48000Hz.
3. User can rename this file and find it easily.
4. Set directory for exporting file, such as C:/Users/User/Desktop.
5. Click “generate” to export a FIR .txt file.

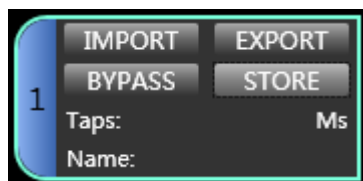
Step 6: import FIR .txt file in FIR audio processor or DSP network power amplifier



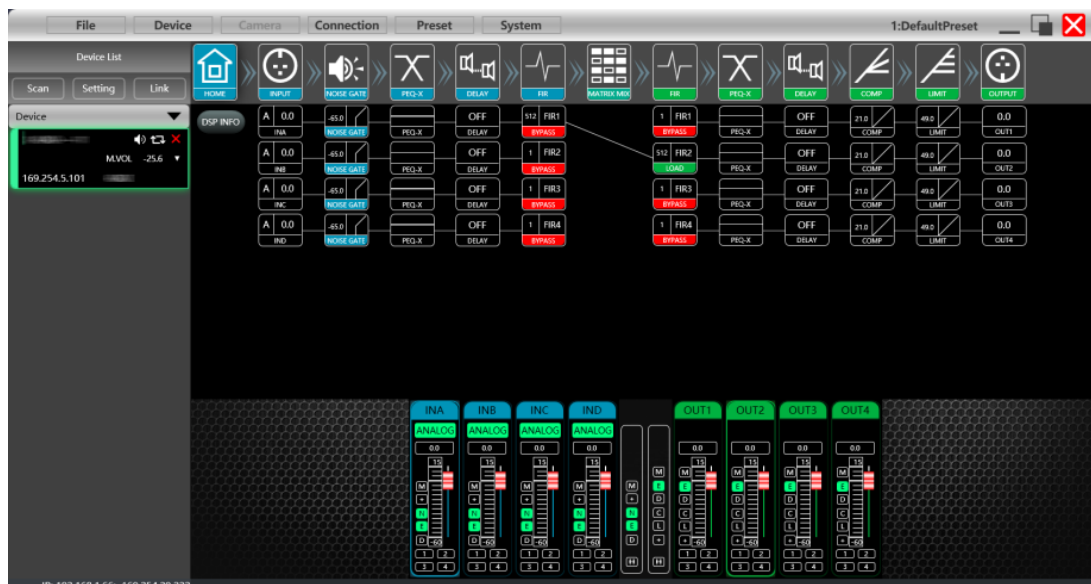
Open Mconsole software, user can choose an input channel or output channel as needed, such as FIR in output channel, it will show a FIR function window.



press **IMPORT** to import txt. file, than press **STORE** to effect this importing.



remember to cancel **BYPASS**.



Step 8: measure the curve of speaker again, use can find it become more linear.



After all setting, please remember to save a preset for your hard working in the speaker.





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