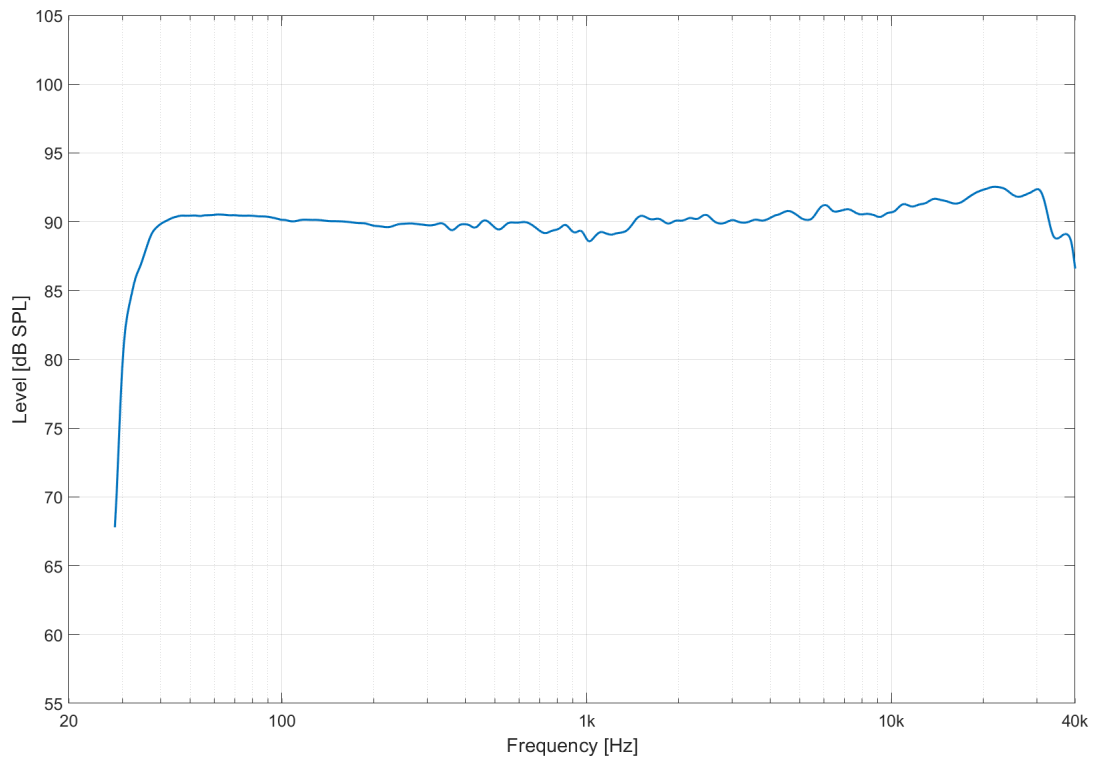


# A8H

Midfield Monitor



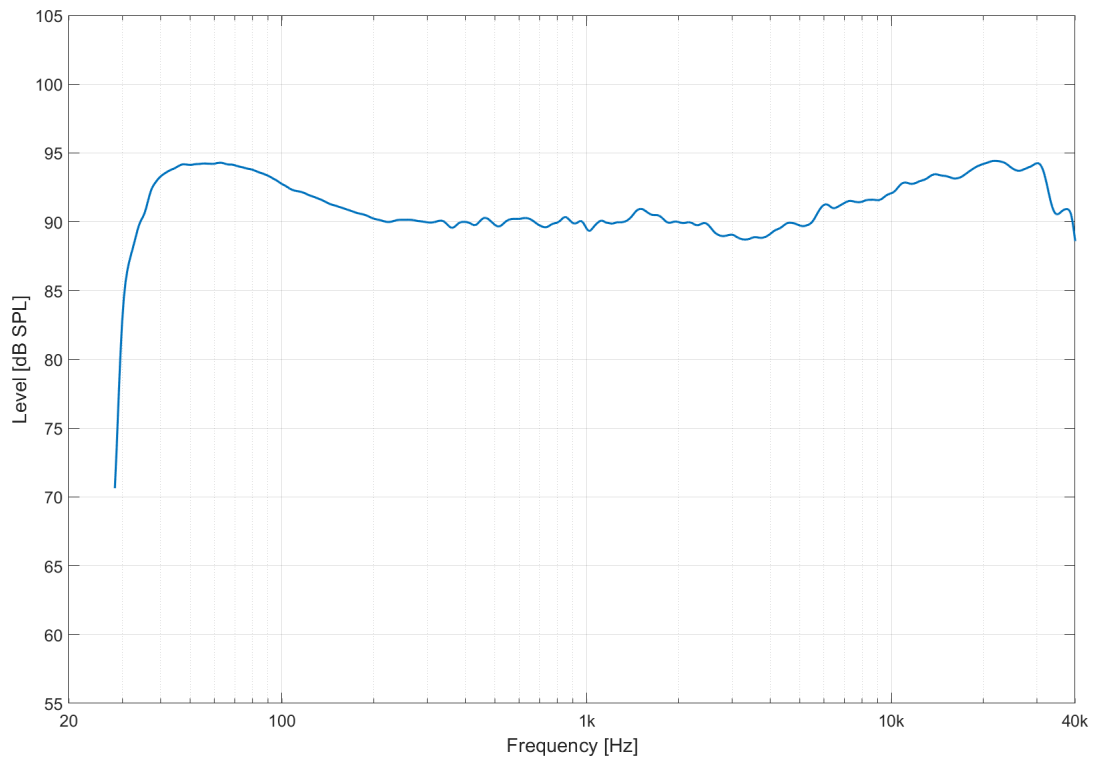
Measurement Report → English



## PURE Frequency Response

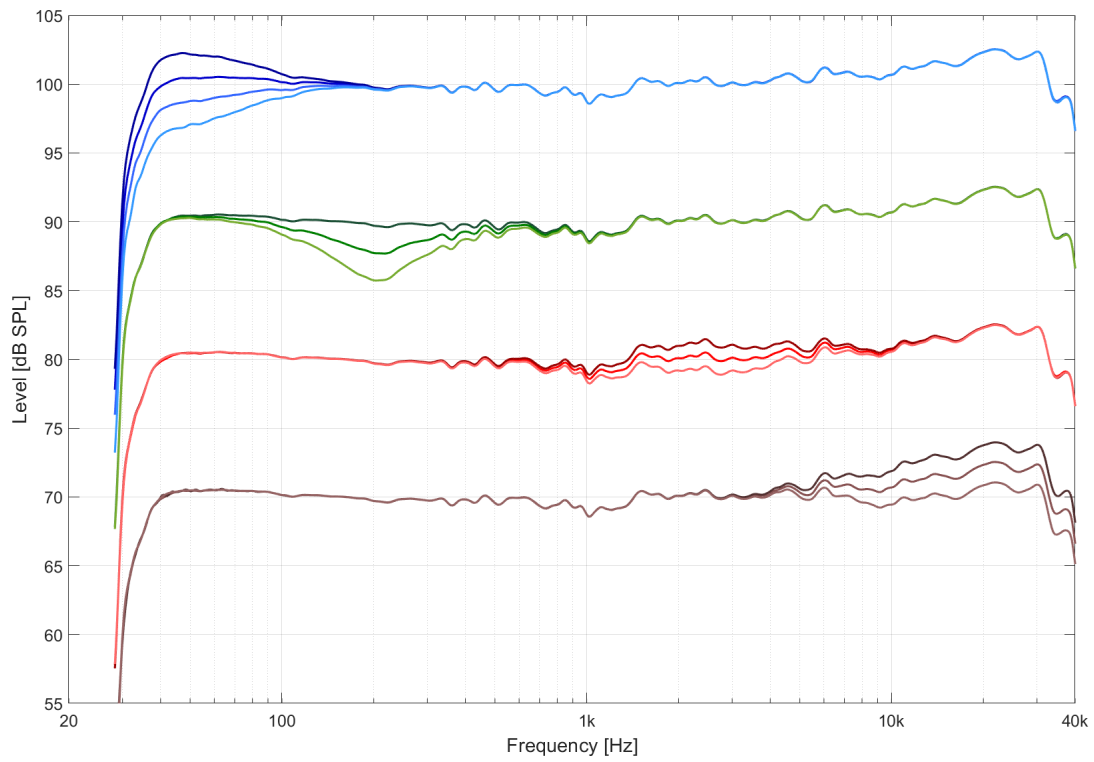
The on-axis frequency response shows the sound radiating from the loudspeaker along the acoustical axis (a position on the front panel between the tweeter and woofer). All loudspeakers are band-limited devices so there is a low frequency roll-off and a high frequency roll-off. The low frequencies should extend as deep as possible for the size of the loudspeaker. The high frequency response should extend beyond what one can hear. Between these two frequencies, the line should be as flat as possible, but resonances, edge diffraction and system tuning can stop this from happening. Using the PURE voicing setting, the A8H has a reasonably flat response but there is one feature to note:

The gentle rise toward the very high frequencies is deliberate as it compensates for the narrowing directivity of the tweeter. The result is a flatter in-room response.



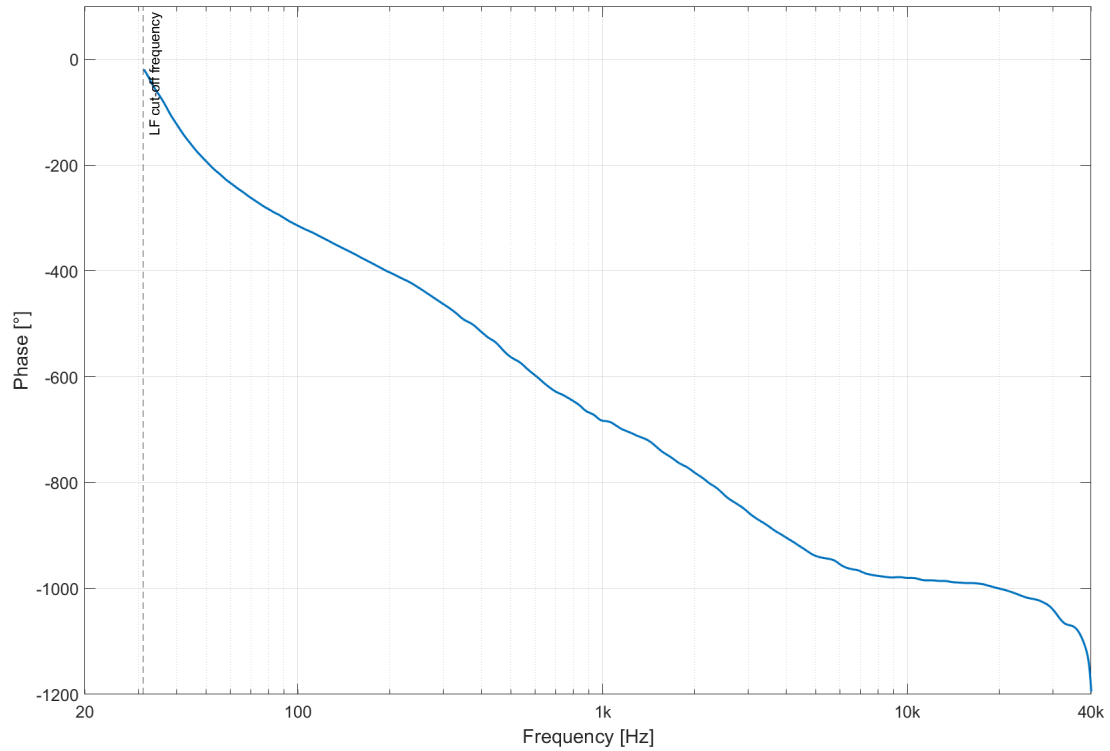
## UNR Frequency Response

The UNR voicing is typical of the historical ADAM Audio sound and is also found in the AX Series. The bass and treble are significantly boosted, and the 2-5 kHz region has been suppressed slightly. The UNR setting is useful when listening for pleasure and is not recommended for critical listening where accuracy is the highest priority.



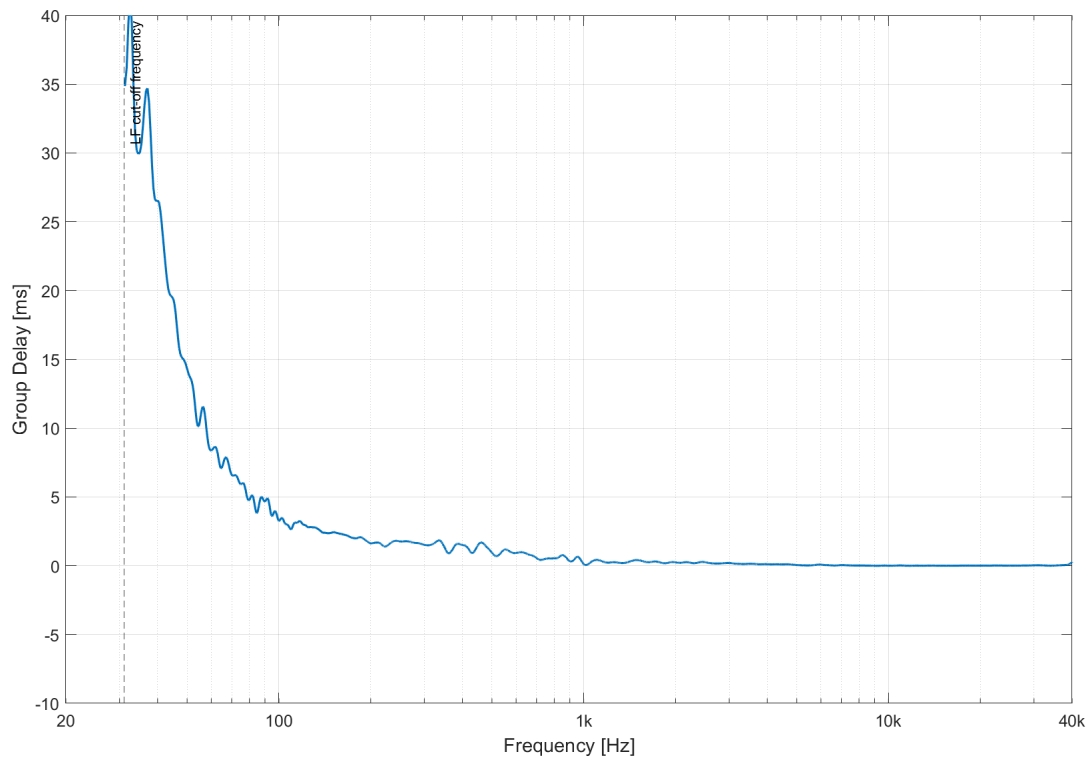
### Room Adaptation Graphs

Measured using the PURE voicing, this graph shows the frequency response of the various backplate equalization controls for Bass, Desk, Presence and Treble.



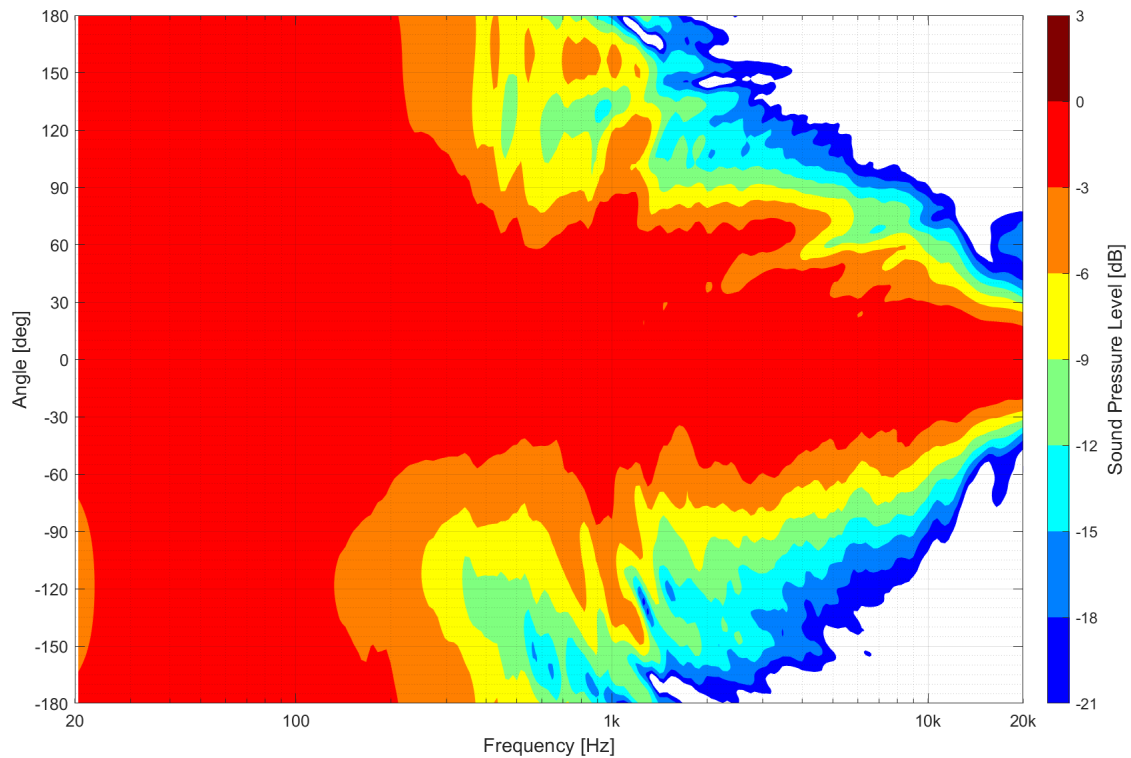
## Phase Response

The phase response shows how the phase of the loudspeaker varies with frequency. To keep latency to a minimum (1.7 ms), we have decided not to equalize the phase. This is the same in almost all analog loudspeakers and many DSP loudspeakers. Unlike other tweeters, we see that the ADAM Audio AMT tweeter has a naturally flat phase response which leads to a very open and transparent sound as all frequencies from the tweeter arrive to the listener at the same time.



## Group Delay

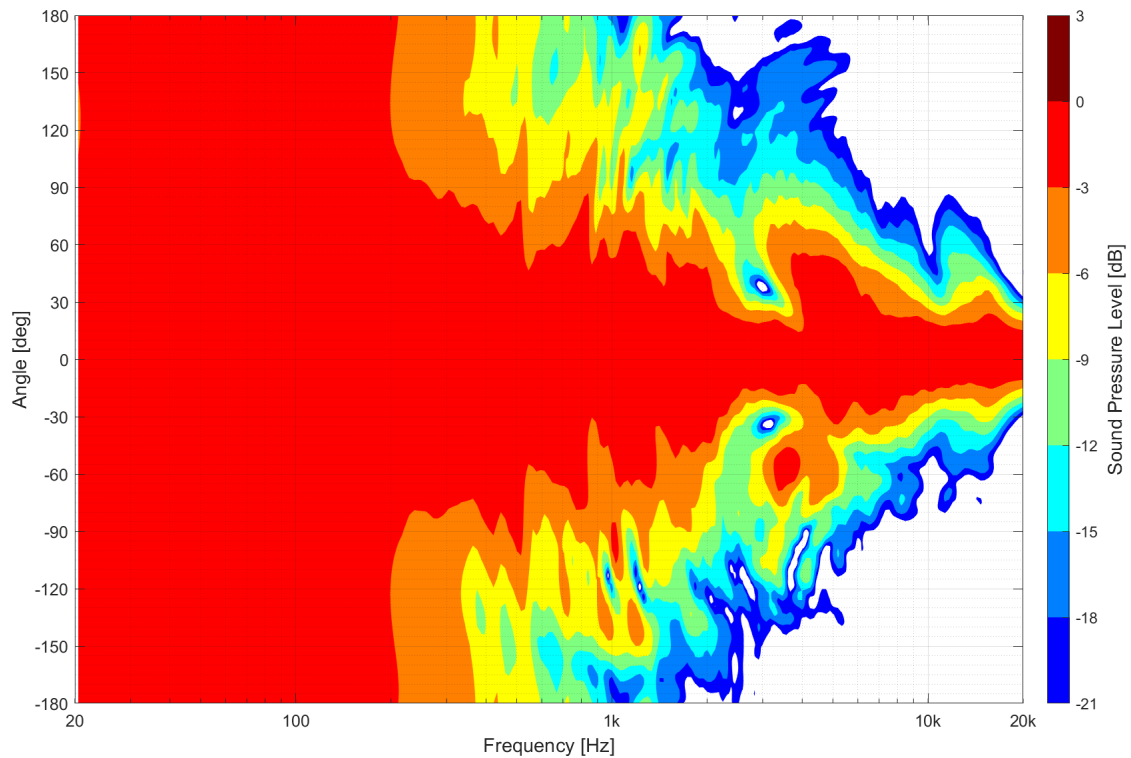
Group delay is a measure of the time it takes for the sound to travel through the loudspeaker system: from input signal to sound out of the drivers. Mathematically it is the negative slope of the phase response. Mid and high frequencies can be seen to pass through the loudspeaker system quickly whereas bass takes longer as the frequency decreases. If the roll-off slope is steeper (e.g. in a vented design vs. a sealed design) and/or at a lower frequency, the group delay peak will be higher. To keep the latency to a minimum (1.7 ms), we have decided not to equalize the phase and thus flatten the group delay. By minimizing any excess increase in the group delay, the loudspeaker sounds fast and tight which is not a common attribute in ported designs.



## Horizontal Isobars

The horizontal isobar plot shows how sound is dispersed into space from the loudspeaker in the horizontal plane. Even when moving around, the listening area in a studio typically falls within 30 degrees of the on-axis direction, so this area should have the same color (red) indicating the sound is evenly reproduced throughout the listening area. More extreme angles (represented by the area above +30 degrees and below -30 degrees on y-axis) are important for reflections. These reflections should have the same sound quality as the on-axis sound but at a lower level, so they have reduced effect on the sound at the listening position and thereby introducing minimal coloration. It is normal for the high frequencies to become quite narrowly dispersed (a function of the size of the driver relative to the wavelength and indicated by a reduced red area width) and for the bass frequencies to be very widely dispersed (red at all angles), with smaller loudspeakers being omnidirectional to higher frequencies than larger loudspeakers.

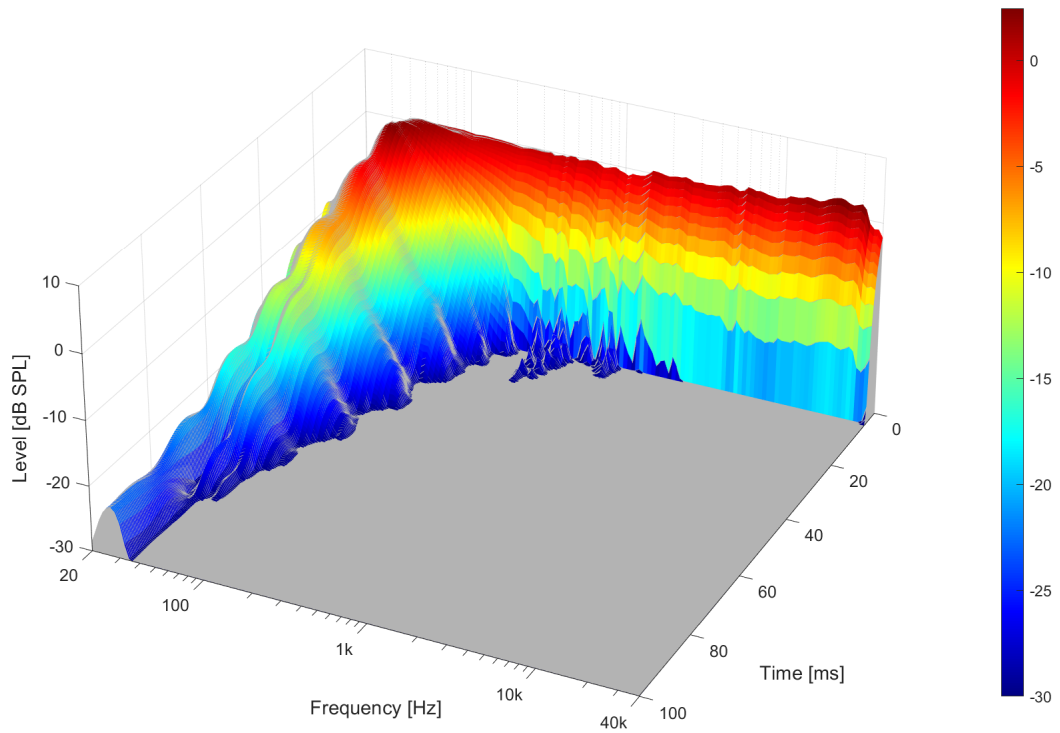
In the A8H we see the red region is quite wide, so a good quality room is preferred. The up/down symmetry on the graph is not as consistent as seen in the other products in the A Series. This translates to small differences in the left/right symmetry in the room and is to be expected given the asymmetrical placement of the drivers on the front panel. This is not a problem as long as mirror image pairs are used as left/right pairs in a system (L/R, LS/RS, etc.)



## Vertical Isobars

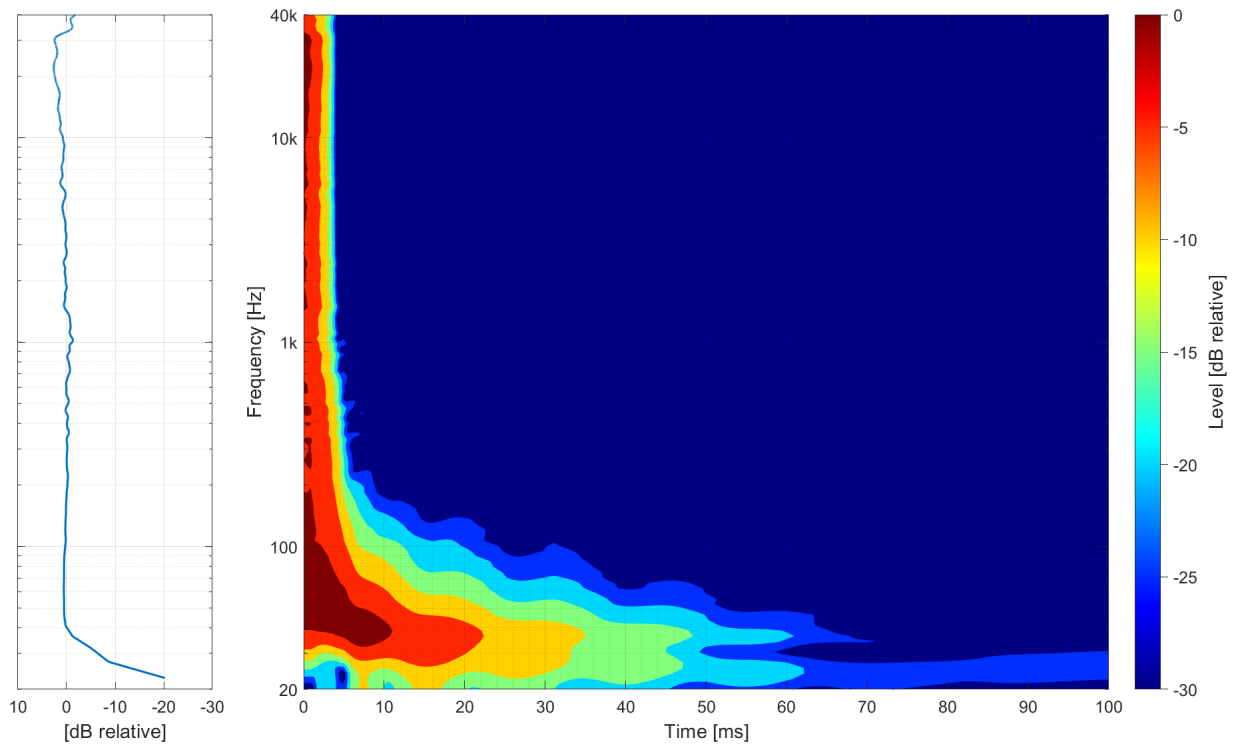
The vertical isobar plot shows how sound is dispersed into space from the loudspeaker in the vertical plane. It is not so likely that one moves up and down when listening, so this direction is somewhat less important than the horizontal plane. However, there can be strong reflections off the ceiling, floor, or desktop placed in front of the loudspeaker, so the off-axis sound quality still has some influence on coloration at the listening position. Similar to the horizontal plane, the  $\pm 20$  degree region is important and should have the same color (red) indicating the sound is evenly reproduced throughout the typical listening heights. More extreme angles (represented by the area above +20 degrees and below -20 degrees on y-axis) are important for reflections, with the desktop reflection being the strongest one. Normally one sees a significant narrowing of the directivity around the crossover in vertically placed drivers, in this case around 3 kHz. The dispersion from bass to treble will be similar to what is seen in the horizontal plane (omnidirectional in the bass, tending towards a narrowing at high frequencies). It is quite typical that the vertical directivity is not as smooth as the horizontal directivity.





## Waterfall Plot

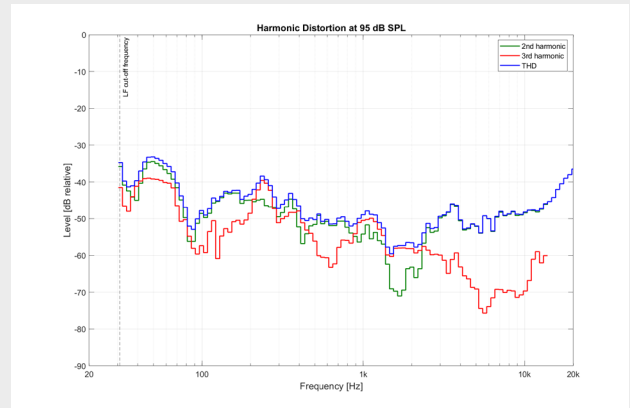
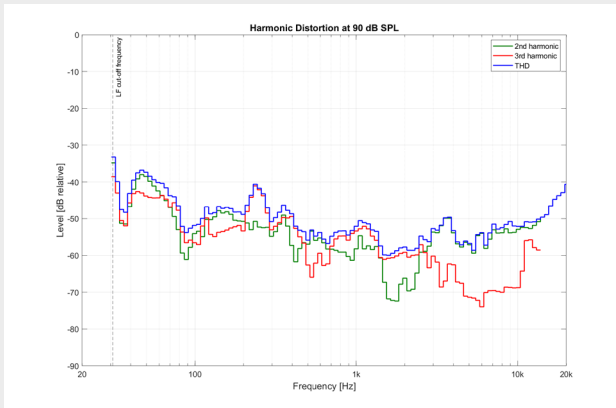
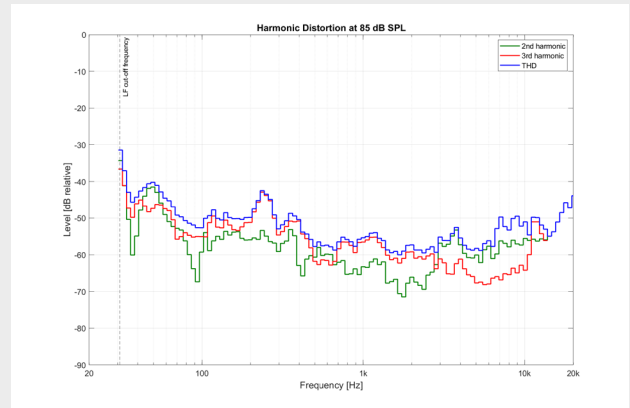
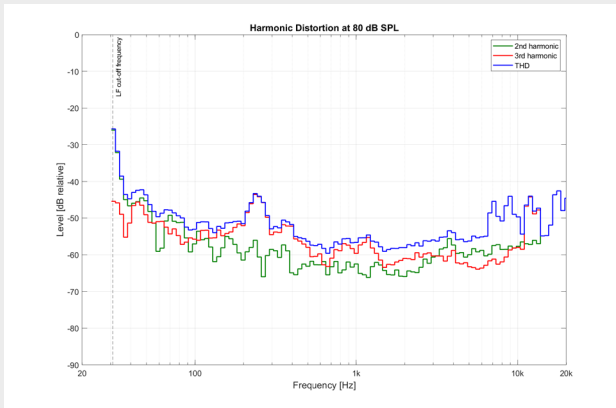
The Cumulative Spectral Decay, more commonly known as a waterfall plot, shows a series of frequency responses of the loudspeaker that are taken later and later in time as one moves from the back of the plot to the front. It is used to visualize resonances. The response at the back is the frequency response shown in the first graph above. It is typical that bass takes longer to decay, hence the decaying ridge is seen near the bottom end of the loudspeaker response. Ported loudspeakers have longer decays than sealed designs, and loudspeakers with a deeper bass response also have longer decays. Other resonances should be very short in comparison and ideally not present at all. In the A8H we see no strong resonances.



## Spectrogram

A spectrogram is a 2D version of the waterfall plot. Resonances can be seen as splashes of colour extending from the left to the right of the plot. Ideally there should be a red bar only at the left of the plot and blue everywhere else.

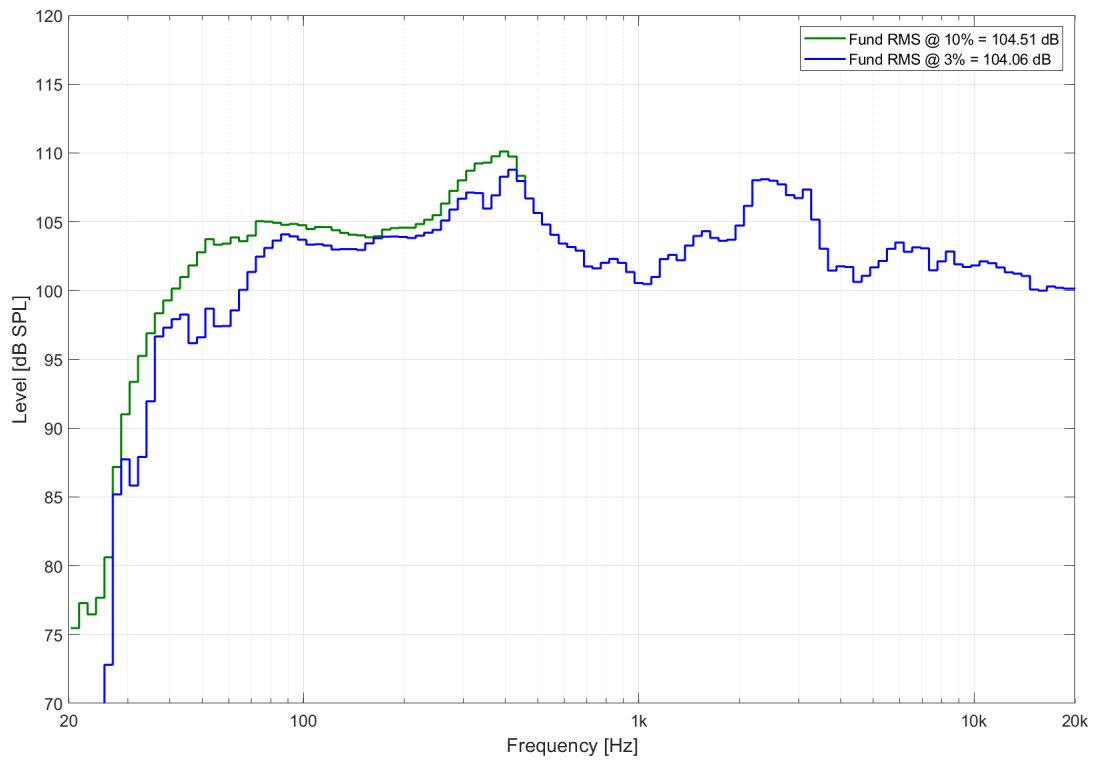
In the A8H we see the bass resonance, which is normal, and no other significant activity.



## Distortions

Distortion is sound output produced by the loudspeaker that is not present at the input. Harmonic distortion is distortion that is a multiple frequency of the original signal, for example if the input signal is 100 Hz, the second harmonic is at 200 Hz, the third harmonic is at 300 Hz, etc. Higher order harmonics are typically produced at a lower level, so if the 2nd and 3rd order harmonics are low, the higher order harmonics will be increasingly inaudible. If all the harmonics are added together, we have the total harmonic distortion (THD) which is, by definition, the highest line on the graph. Second order harmonic distortion is caused by asymmetries in the system and can sound quite warm and pleasing. Vinyl records and valve amplifiers generate a lot of this. Third order harmonic distortion is caused by clipping in the system and never sounds good. Amplifiers played too loud or drivers reaching their maximum excursion suffer from this. Clearly all forms of distortion should be minimized in a studio monitor which has the task to accurately reproduce the input signal and add no extra content to that signal. It is possible to have a high THD and have a nice sounding speaker if the THD is dominated by second order harmonic distortion, therefore THD on its own is not a good indicator of audio quality. As the output level is increased the distortion also increases and eventually, if loud enough, all the audio from the loudspeaker can be distortion (0 dB = 100% on the graph) but the loudspeaker should already be strongly limited by the protection system well before this. Note that -20 dB = 10%, -30 dB = 3%, -40 dB = 1%, -50 dB = 0.3%, etc.

The A8H is a three-way loudspeaker so mid frequency distortion is significantly lower than in a two-way loudspeaker. The large woofer also lowers distortion below 400 Hz.



## SPL

The max SPL curve shows how loud the loudspeaker can play at each frequency. The goal is to have the highest values possible, not to have a flat curve. Smaller loudspeakers will have lower max SPL at low frequencies. Loudspeakers with small amplifiers and/or drivers with a low sensitivity will also have a lower max SPL. The curve is measured using very short sine bursts that are increased in level until a defined THD (in this case 3% and 10%) is achieved.

The A8H is a larger loudspeaker with an 8" woofer, so the max SPL is high all the way down to 40 Hz.

**ADAM AUDIO GMBH**  
BERLIN, GERMANY

T +49 30-863 00 97-0  
F +49 30-863 00 97-7

INFO@ADAM-AUDIO.COM  
WWW.ADAM-AUDIO.COM

© ADAM Audio GmbH 2022

V12/22

