

Accuracy Versus Realism

Simulating the “human” side of measurement

By Pat Brown

The human auditory system is equipped with two input ports – left and right ears. This “binaural” processing system provides us with the ability to localize *where* sound is coming from, something that a one-eared listener would have difficulty in doing.

Playback systems may utilize any number of channels to surround the listener with sound, but two channels is always enough to simulate the human listener. Recording enthusiasts have long discovered the benefits of stereo microphones. While not necessarily “human-like,” they can produce recordings that add spaciousness and realism to the recorded material.

Two-channel acoustic measurements are important for the same reason – they add a human characteristic to the data. For our discussion here, I’ll use the term “binaural” to describe recording processes that provide data for two ears – there is no need to distinguish between making a recording and making a measurement, as either or both might be the motive of the investigator.

Let’s look at some of the ways to get binaural

data. Many modern measurement platforms support two-channel recording. We will assume that one of them is being used, allowing our discussion to be confined to microphone techniques.

One of the first decisions that must be made by the data gatherer is whether accuracy or realism is more important. After a little consideration, it becomes apparent that one cannot have both. Setup parameters that provide a more accurate view of the loudspeaker’s response will require that the effects of the environment be minimized.

On the other hand, if the effect of the room is to be considered, then accuracy will need to be sacrificed to include it. The question becomes, “Do I want to know what is *actually* happening, or do I want to know what is *perceived* to be happening?” The answer to this question will fundamentally affect the method used to collect the data.

It’s important to note that at least three responses are being gathered in the recording – the loudspeaker, the listener and the room. The listener’s response is a constant. The ear/brain system is assumed to be processing sound the same way at every seat.

The loudspeaker’s response can be dramatically position dependent, but it does not have to be.

Loudspeakers that are designed for covering an audience evenly can have a similar response over a large area.

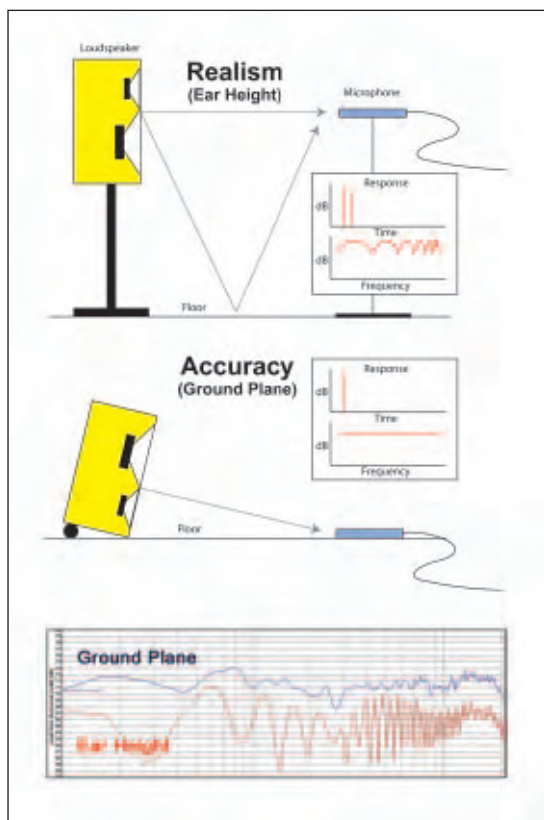


Figure 1: Responses of ear height and ground plane microphone placements.

The room also has a response, but it is unique for each listening position. This is one of the reasons why we can't correct room acoustic problems with electronics.

Is the goal of the measurement accuracy or realism? If the purpose of the measurement is to calibrate an equalizer or crossover network, then accuracy should be considered first. It is desirable to know the true acoustic response of a transducer at a point in space, usually for the purpose of improving this response through signal processing.

See **Figure 1**. A stereo microphone on a stand at ear height might convey what a listener will hear, but this response will include seat-dependent artifacts, such as a strong reflection from the floor or other nearby objects. The resultant comb filters will make it impossible to observe the response that is due to the loudspeaker alone. (For more on this subject, check out *Tech Talk* by John Murray, beginning on page 70 of this issue.)

If one were to attempt to compensate for the effect of the floor reflection, the compensation would not be correct for a closer or more distant listener seat. As such, it is best to ignore the floor reflection altogether when "tuning" the system.

Also, such a "seat dependent" response would average out if a large number of measurements were averaged across an auditorium. This is why near-field and ground-plane measurement techniques play an important role in sound system tuning.



Figure 4: The Countryman B6 lapel mic (above) makes an excellent "At-The-Ear" microphone. The foam insert is from a Shure E1 ear bud (below).

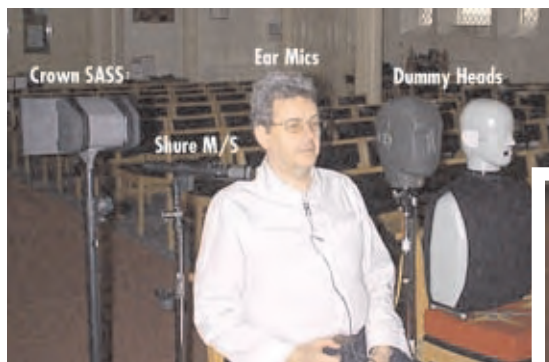
(This article is about neither – we'll table that discussion for the future.)

CASE FOR REALISM

If the measurer wants to know what a sound system/room sounds like, then accuracy must give way to realism. Realism requires a binaural recording technique, and it must include the same effects from the room that might affect a live listener.

Mic placement is actually much easier than when considering accuracy, as the measurer simply listens to the system wherever he/she likes and then replaces his/her head with the microphone. See **Figure 2** and **Figure 3** for mic choices, which include:

Stereo. A simple



Figures 2 and 3: Dummy heads, while expensive, provide stability and repeatability. They are ideal for research projects. Peter Mapp displays his arsenal of two-channel mics. (He's the one in the middle.)



force

[n. energy exerted or brought to bear]



Geo

GED T technology has been extensively and successfully festival-proven for crowds of 100,000+ worldwide. To do so, GED T delivers maximum force, but not brute force. What makes GED T's force unique is coherent, precise, full-bandwidth pattern control that includes carded bass/energy from T2885 and T2815 array modules, and unvelled upper-cardinal control to 3282, from the smallest CD18 subbass.

The CD18 directs subbass energy at the audience and away from open microphones and reverberant surfaces through innovative signal processing algorithms, intelligently applied to each of the dual-parted 18-inch woofers. The results produce substantial forward gain of an exceptionally balanced 1.0 output and 12dB+ rear attenuation.

Visit us at www.nexo-us.com for more details, or write us to prove it. We are happy to provide system demos and full documentation at your request.

NEXO

I N N O V A T E

<p>Europe, Asia, Middle East & Africa NEXO SA Tel: +33 1 48 03 19 14 e-mail: info@nexo.fr</p> <p>Singapore NEXO Far East Pte Ltd Tel: +65 742 0660</p>	<p>www.nexo-us.com</p> <p>North America NEXO US Tel: +1 815 487 6000 e-mail: info@nexo.us</p> <p>South America NEXO Tel: +54 9 11 5388 3470</p>
---	---

stereo mic can yield left/right information. Two cardioid mics in an X/Y configuration can yield convincing stereo. Spaced omnidirectional mics are another popular method. This is art, not science, so there really aren't any rules to break. If you like what

Frequency-dependent directivity is achieved by boundary-loading the mics on small, flat panels.

Head/Torso/Pinnae Simulation.

Perhaps the best binaural mic is the dummy head. This includes the effect of the head, torso, and even the ear

One way to make a "poor man's" dummy head is to utilize your own

you hear, then it's O.K.

Head Simulation. An added element of realism can be achieved by simulating the presence of a human head. The "head effect" is called the Head-Related-Transfer-Function (HRTF.) The Crown SASS uses omni mics spaced at human dimensions with an absorptive mass in between.

structure. The major benefits of this technique are customization and repeatability.

The response can be modified electronically and physically to whatever is desired, and setups can be recalled in the future if needed. Digital signal processing provides a low-cost, powerful way to modify the response.

Dummy heads can cost many thousands of dollars, but the cost is easily justified for researchers that need the benefits.

Human Mics. One way to make a "poor man's" dummy head is to utilize your own (no offense intended). Everything is already in place except the microphones. I've seen numerous mic placement mechanisms over the years, including eyeglass mounts, wires, and even earrings.

Possibly the most clever and realistic approach to date is the In-The-Ear (ITE) recording technique pioneered by Don and Carolyn Davis in the late 1980s. This involved placing probe mics at the surface of the ear drum. This technique captured the outer ear response, including the ear canal resonance. The resonance was removed with an inverse filter during playback.

A variation on this technique that sacrifices some accuracy for practicality is to place small mics at the entrance to the ear canal. I will call this "At-The-Ear" to distinguish it from the previous technique.

The mics are held in place by some foam inserts (Figure 4, previous page). The two mics have XL male connectors that can connect directly to my data recorder. I normally survey an auditorium without wearing the mics to determine the measurement positions, and then return to the seats with mics in place to gather data.

Figure 5 shows a comparison between a free-field measurement and the "At-The-Ear" placement in both the time and frequency domains. The responses have been overlaid for comparison.

The methods used to gather data are determined by the intended use of the data. This often requires more than one technique, each preserving or enhancing the information in a way that yields more insight into the particular problem being solved.

When making measurements, arrive equipped to acquire both accurate data and realistic data, and then let the question being pondered determine the preferred perspective. ■

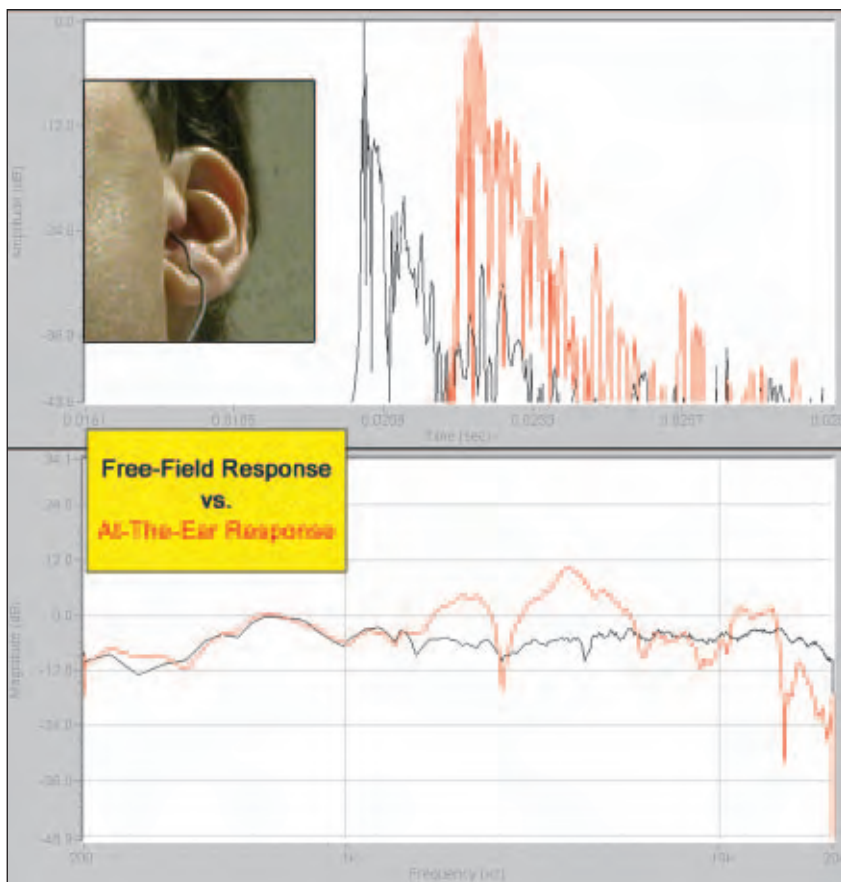


Figure 5: The impulse response and frequency magnitude of the B6 mic placed in a free-field and At-The-Ear. The impulse response of the At-The-Ear placement has been offset for clarity. Note the stark contrast between accuracy and realism in gathering data.

Pat and Brenda Brown own and operate Syn-Aud-Con, conducting training seminars around the world. For more info go to www.synaudcon.com