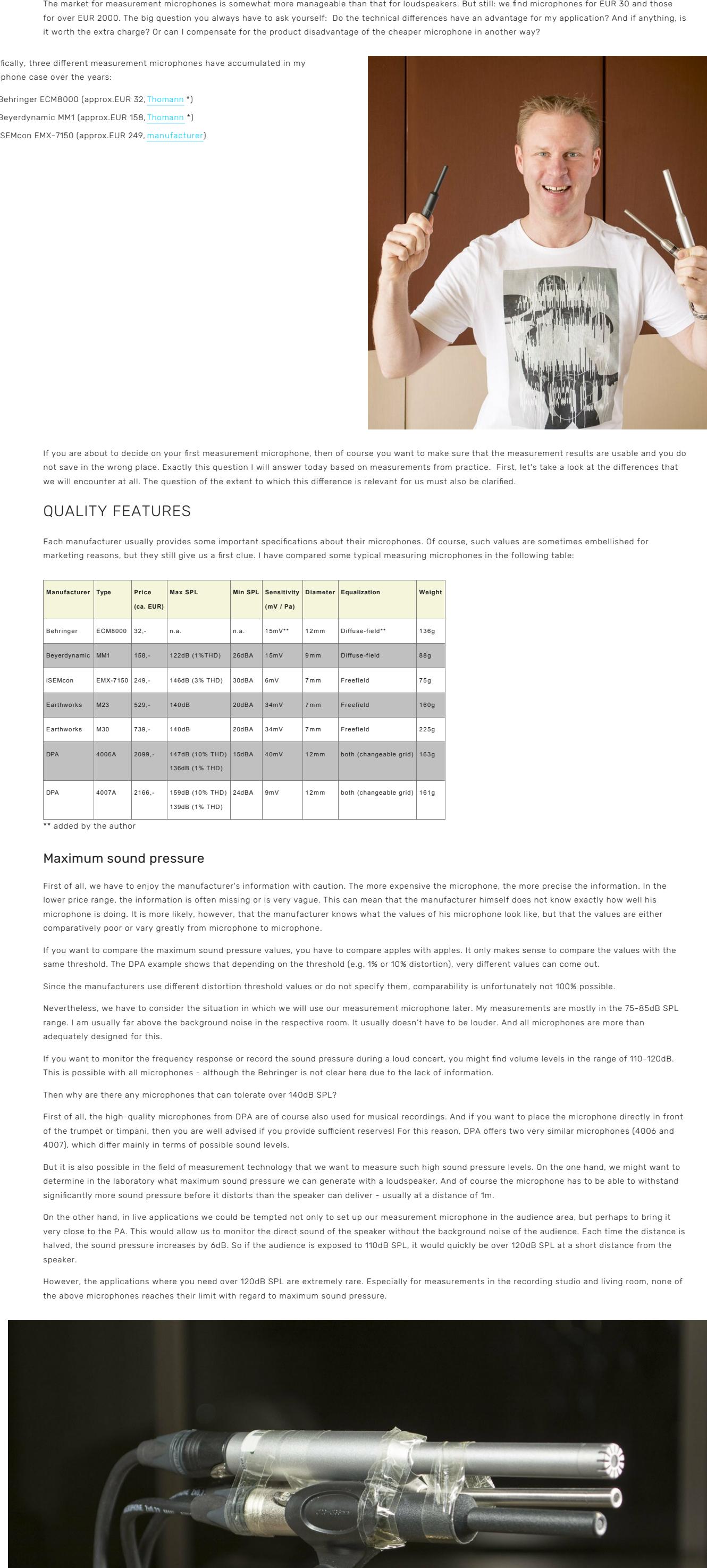


23. FEBRUAR 2020

Measurement Microphone Comparison

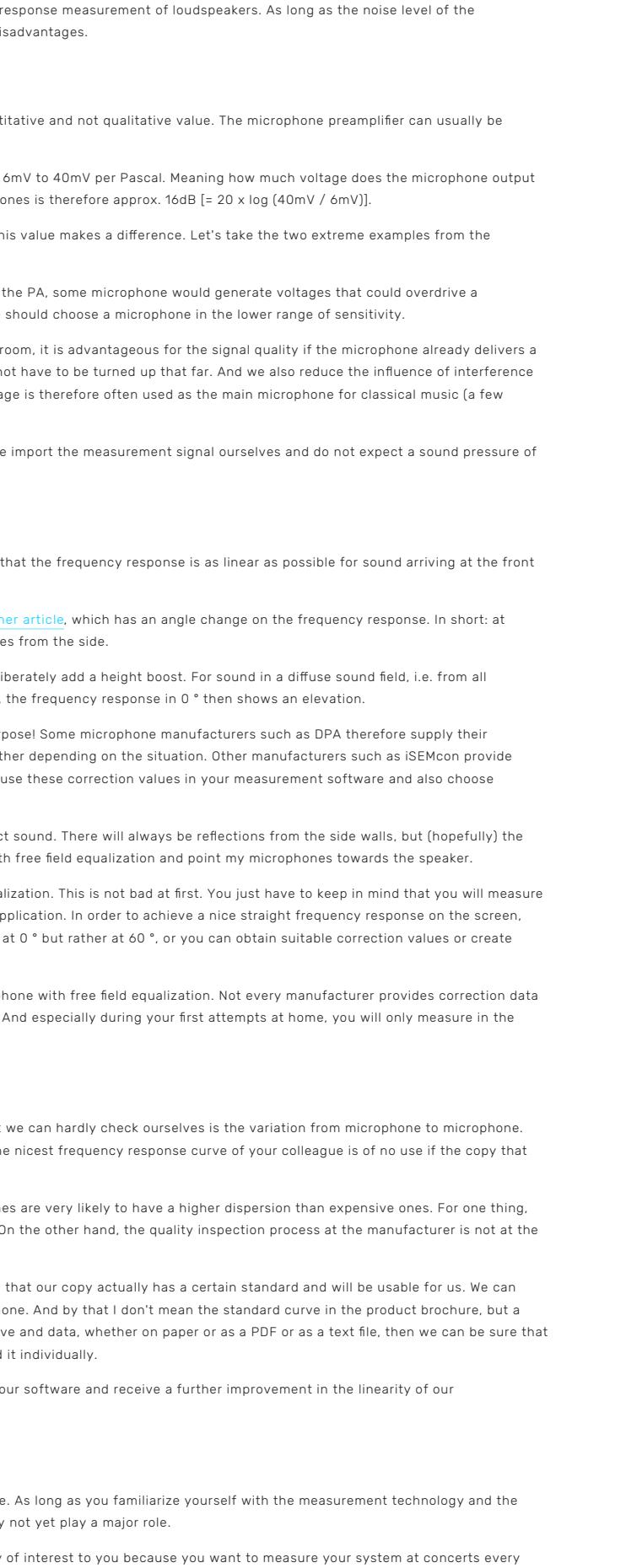
BEHRINGER ECM8000 VS. BEYERDYNAMIC MM1 VS. ISEMCON EMX-7150

It is not uncommon for music lovers to ask questions such as "which speaker is the best?", "which amplifier is the best?" and "which measurement microphone is the best?". And no matter which product you choose, if you are willing to spend more money, then there are guaranteed to get a product that is "better" than your own. A speaker for EUR 2000 will usually be better than one for EUR 200. And a speaker for EUR 20,000 will probably be even better than the speaker for EUR 2,000.

The market for measurement microphones is somewhat more manageable than that for loudspeakers. But still: we find microphones for EUR 30 and those for over EUR 2000. The big question is always have to ask yourself: Do the technical differences have an advantage for my application? And if anything, is it worth the extra charge? Or can I compensate for the product disadvantage of the cheaper microphone in another way?

Specifically, three different measurement microphones have accumulated in my microphone case over the years:

- Behringer ECM8000 (approx.EUR 32, [Thomann](#)*)
- Beyerdynamic MM1 (approx.EUR 65, [Thomann](#)*)
- iSEMcon EMX-7150 (approx.EUR 249, [manufacturer](#))



If you are about to decide on your first measurement microphone, then of course you want to make sure that the measurement results are usable and you do not save in the wrong place. Exactly this question I will answer today based on measurements from practice. First, let's take a look at the differences that we will encounter at all. The question of the extent to which this difference is relevant for us must also be clarified.

QUALITY FEATURES

Each manufacturer usually provides some important specifications about their microphones. Of course, such values are sometimes embellished for marketing reasons, but they still give us a first clue. I have compared some typical measuring microphones in the following table:

Manufacturer	Type	Price (ca. EUR)	Max SPL	Min SPL (Sensitivity)	Diameter	Equalization	Weight
Behringer	ECM8000	32,-	n.a.	n.a.	15mV**	12mm Diffuse-field**	136g
Beyerdynamic	MM1	158,-	122dB (1%THD)	26dBa	15mV	9mm Diffuse-field	88g
iSEMcon	EMX-7150	249,-	146dB (3% THD)	30dBa	6mV	7mm Freefield	75g
Earthworks	M23	529,-	146dB	20dBa	34mV	7mm Freefield	140g
Earthworks	M30	739,-	146dB	20dBa	34mV	7mm Freefield	225g
DPA	4006A	2099,-	147dB (10% THD)	15dBa	40mV	12mm both (changeable grid)	163g
DPA	4007A	2166,-	156dB (10% THD)	24dBa	9mV	12mm both (changeable grid)	161g

* added by the author

Maximum sound pressure

First of all, we have to enjoy the manufacturer's information with caution. The more expensive the microphone, the more precise the information. In the lower price range, the information is often missing or is very vague. This can mean that the manufacturer himself does not know exactly how well his microphone is doing. It is more likely, however, that the manufacturer knows what the values of his microphone look like, but that the values are either comparatively poor or vary greatly from microphone to microphone.

If you want to compare the maximum sound pressure values, you have to compare apples with apples. It makes sense to compare the values with the same threshold. The DPA example shows that depending on the threshold (e.g. 1% or 10% distortion), very different values can come out.

Since the manufacturers use different distortion threshold values or do not specify them, comparability is unfortunately not 100% possible.

Nevertheless, we have to consider the situation in which we will use our measurement microphone later. My measurements are mostly in the 75-85dB SPL range. I am usually far above the background noise in the respective room. It usually doesn't have to be louder. All microphones are more than adequately designed for this.

If you want to monitor the frequency response or record the sound pressure during a loud concert, you might find volume levels in the range of 110-120dB. This is possible with all microphones - although the Behringer is not clear here due to the lack of information.

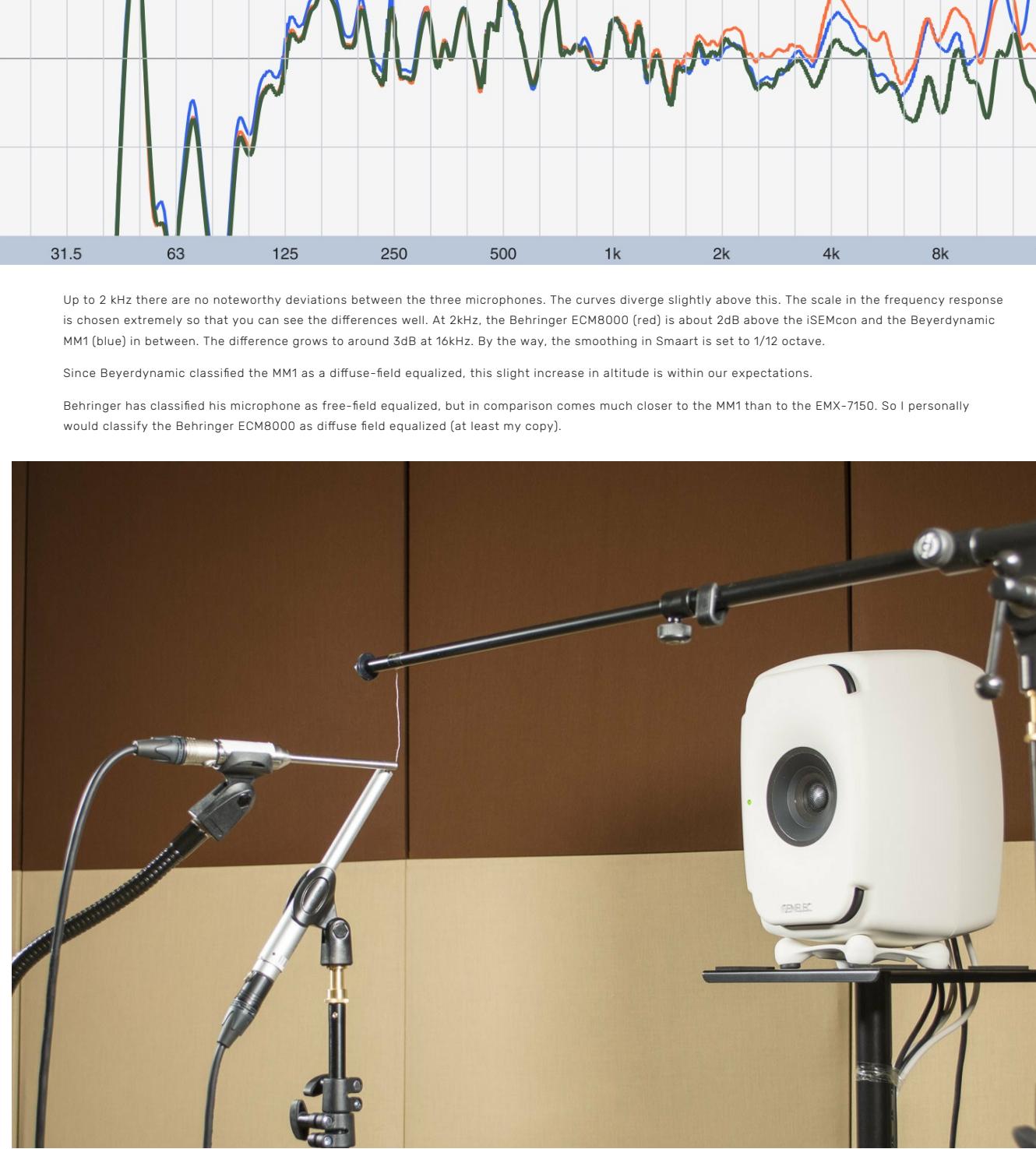
Then why are there many microphones that can tolerate over 140dB SPL?

First of all, the high-quality microphones from DPA are of course used for musical recordings. And if you want to place the microphone directly in front of the trumpet or timpani, then you are well advised if you provide sufficient reserves. For this reason, DPA offers two very similar microphones (4006 and 4007), which differ mainly in terms of possible sound levels.

But it is also possible in the field of measurement technology that we want to measure such high sound pressure levels. On the one hand, we might want to determine in the laboratory what maximum sound pressure we can generate with a loudspeaker. And of course the microphone has to be able to withstand significantly more sound pressure before it distorts than the speaker can deliver - usually at a distance of 1m.

On the other hand, in live applications we could be tempted not only to set up our measurement microphone in the audience area, but perhaps to bring it very close to the PA. This would allow us to monitor the direct sound of the speaker without the background noise of the audience. Each time the distance is halved, the sound pressure increases by 6dB. So if the audience is exposed to 110dB SPL, it would quickly be over 120dB SPL at a short distance from the speaker.

However, the applications where you need over 120dB SPL are extremely rare. Especially for measurements in the recording studio and living room, none of the above microphones reaches their limit with regard to maximum sound pressure.



Noise Level

As an opponent of the maximum sound pressure, there is of course also a minimum sound pressure. This point is reached when the microphone's own noise is louder than the ambient noise.

Now we have to keep in mind the environment in which we are likely to use our measurement. Recording studios have to struggle to ensure that their insulation is sufficiently thick and conscientious, that they can even manage a room with a quiet level of 30dB. If you want to carry out measurements that confirm exactly this quiet level, then your measuring microphone should of course be below it. And this usually means a corresponding price for the microphone and also for the preamplifier.

For everyone else (including me) who does not measure the noise level of the room, but a loudspeaker, can safely neglect this point. The ambient noise is usually in the 40-50dB range anyway. And with our measurement signal, we will usually be above it, around 70-80dB. Whether the microphone is noisy at 30dB or at 20dB, we will never be able to see that in any measurement curve outside of a laboratory.

For this reason, all the listed microphones are suitable for the frequency response measurement of loudspeakers. As long as the noise level of the microphone is over 130dB or below, there are no advantages or disadvantages.

Sensitivity

The output voltage that a microphone delivers is first of all a purely quantitative and not qualitative value. The microphone preamplifier can usually be adjusted to adapt the signal to the actual sound pressure.

When it comes to sensitivity, we talk about values in the range from 0mV to 40mV per Pascal. Meaning how much voltage does the microphone output with a defined acoustic sound signal. The difference between the microphones is therefore approx. 16dB [$= 20 \times \log(40mV / 0mV)$].

Nevertheless, as is so often the case, there are extreme cases in which this value makes a difference. Let's take the two extreme examples from the previous points:

If we dive into the range of over 120dB in front of the trumpet or close to the PA, some microphone would generate voltages that could overdrive a microphone preamplifier. If we expect this sound pressure range, then we should choose a microphone in the lower range of sensitivity.

If, on the other hand, we measure very quietly in a recording studio, it is advantageous for the signal quality if the microphone already delivers a decent signal. As a result, the subsequent microphone preamplifier does not have to be turned up that far. And we also reduce the influence of interference signals on the cable route. The DPA 4006 with its rather high output voltage is therefore often used as the main microphone for classical music (a few meters away from the orchestra).

For the usual areas of application of a measurement microphone, i.e. if we import the measurement signal ourselves and do not expect a sound pressure of over 120dB, all the microphones listed are still in the race.

Diffuse Field vs Free Field Equalization

Most measurement microphones have free field equalization. This means that the frequency response is as linear as possible for sound arriving at the front (at 0°).

I have already examined the differences for my three microphones [in another article](#), which has an angle change on the frequency response. In short: at high frequencies, the sensitivity decreases by a few dB when sound arrives from the side.

There are therefore microphones with a diffuse-field equalization that deliberately add a height boost. For sound in a diffuse sound field, i.e. from all directions, this results in an average linear frequency response. However, the frequency response in 0° then shows an elevation.

The equalization curve is not a quality feature. It only depends on the purpose. Some microphone manufacturers such as DPA therefore supply their microphones with interchangeable grids so that you can use one or the other depending on the situation. Other manufacturers such as SEMcon provide their microphones with correction curves on a USB stick so that you can use these correction values in your measurement software and also choose between the two equalizations depending on the situation.

The majority of our loudspeaker measurements will be carried out in direct sound. There will always be reflections from the side walls, but (hopefully) the main energy will come from the speaker. So I always use microphones with free field equalization. And of course the microphone has to be able to withstand significantly more sound pressure before it distorts than the speaker can deliver - usually at a distance of 1m.

Beyerdynamic writes openly about the MM1 that it has a diffuse-field equalization. This is not bad at first. You just have to keep in mind that you will measure an increase in the high frequencies for the 0° direction in an open field application. In order to achieve a nice straight frequency response on the screen, you can either turn the microphone so that the speaker no longer arrives at 0° but rather at 60°, so you can obtain suitable correction values or create them yourself for the measurement software.

From my personal experience, I would always buy a measurement microphone with free field equalization. Not every manufacturer provides correction data and not every software offers the possibility to use such correction data. And especially during your first attempts at home, you will only measure in the open field - as long as you are not starting in a reverberant bathroom!

A point that we unfortunately are not given by the manufacturer and that we can hardly check ourselves is the variation from microphone to microphone. How big are the deviations between the completely different microphones. The nicest frequency response curve of your colleague is of no use if the copy that you buy tomorrow might show a completely different curve.

One can actually produce a rough estimate with common sense that cheap microphones are very likely to have a higher dispersion than expensive ones. For one thing, the actual production quality is probably lower for low-budget products. On the other hand, the quality inspection process at the manufacturer is not the level of a high-end manufacturer.

However, there is a way how a customer can be reasonably certain that our copy actually has a certain standard and will be useful for us. We can make sure that we get an individual measurement curve for each microphone. And by that I don't mean the standard curve in the product brochure, but a very individual one from this one microphone. Because if there is this curve and data, whether on paper or as a PDF or as a text file, then we can be sure that someone at the manufacturer has unpacked our microphone and checked it individually.

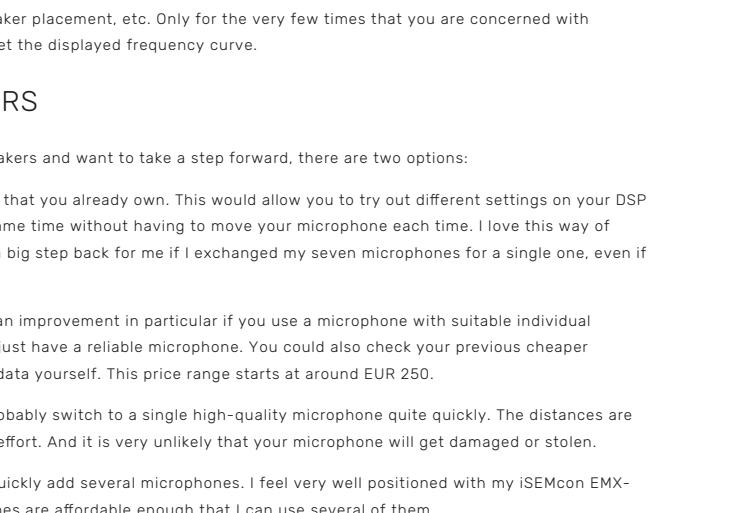
And in addition, we can sometimes even import this correction data into our software and receive a further improvement in the linearity of our measurement setup.

Robustness

The best microphone is of no use if it gives up its mind within a short time. As long as you familiarize yourself with the measurement technology and the microphone is only in your living room next to your system, this point may not yet play a major role.

But if you already know that the topic of measurement technology is only of interest to you because you want to measure your system at concerts every week in the future, then this point should not be underestimated.

Fortunately, all measurement microphones that I have picked up so far are at a good mechanical level. In addition to the actual microphone, the longevity of your microphones naturally depends not only on the microphone itself, but also on the accessories. Is the tripod stable and heavy enough that it doesn't tip over? Do you have a stable case for your microphone so that they remain intact during transport? I can recommend this small and light case so that the last point, in which I transport my 7 microphones together with the calibrator: [Thomann](#)*



Capsule diameter

I don't want to bore you with a theoretical treatise on the advantages and disadvantages of the membrane diameter. Of course, it is advantageous to use a membrane that is as small as possible, because the omni-directional characteristic then theoretically remains even at high frequencies and the drop in height at lateral angles is not too great. And of course it is advantageous to use the largest possible membrane, because then the signal-to-noise ratio is as high as possible.

If we look at the various measuring microphones on the market, one thing should be clear: the capsule diameter is not important! There are good and bad microphones with a small diameter. And there are good and bad microphones with a large membrane.

Nevertheless, the size is important for a very practical reason: you may want to calibrate the measuring system to absolute dB SPL later. Be it because you are just curious how loud you generally listen to your music. Or how loudly you mix your band. Or also because you (or your client) are legally obliged to keep a record of every event. Then at the latest you will think about buying a calibrator.

And then at the latest the question of the capsule diameter will come up! Because calibrator and measuring microphone should fit together mechanically.



You can always add adapters to the standard calibrators for the standard microphone sizes: 1/2", 1/4", 1/8", i.e. 25mm, 12.5mm and 7mm. If you buy a microphone that adheres to these standards, you will find it easy to find any calibrator with a suitable adapter later on. For example, the iSEMcon EMX-7150 fits exactly into the 1/4" adapter. The Behringer ECM8000 fits exactly into the 1/2" adapter. The Beyerdynamic is in between and can therefore not be used exactly.

Temperature range

What environments do you think of when you grab your measurement microphone and want to optimize the sound of your speakers? Do you have a winter jacket and thick gloves with you?

When I think of my measurements over the past 20 years, I've never had a winter jacket on! Nor have I been in the desert until now. Nevertheless, I would like to mention this point, of course, because every microphone certainly has a certain temperature range in which it delivers reliable results. In most cases, the microphone should still work even if you have already gone up! For the usual measurements at home or at concerts, we will probably be in the range of 15 to 30 degrees Celsius. In my view, we can therefore ignore this criterion for our decision. But if you have a job in Antarctica soon, don't claim I didn't mention it!

The first finding: the difference in sensitivity roughly matches the manufacturer's specifications. I increased the iSEMcon by 0.5dB for the further measurements so that all three curves are equally loud ($15mV / 0mV = 7.95dB$). Incidentally, the Behringer ECM8000 is exactly the same as the Beyerdynamic MM1, so I added its sensitivity 15mV / Pa in the table for the Behringer.

THE TEST SETUP

But now finally to practice! In the first step, I simply glued my three different measurement microphones as close together as possible and placed them 50 cm in front of my Genelec 8331 studio monitor ([Thomann](#)). I used Smarta V8 ([Focal Acoustics](#)) for this comparison.

With now finally to practice! In the first step, I simply glued my three different measurement microphones as close together as possible and placed them 50 cm in front of my Genelec 8331 studio monitor ([Thomann](#)). I used Smarta V8 ([Focal Acoustics](#)) for this comparison.

But now finally to practice! In the first step, I simply glued my three different measurement microphones as close together as possible and placed them 50 cm in front of my Genelec 8331 studio monitor ([Thomann](#)). I used Smarta V8 ([Focal Acoustics](#)) for this comparison.

DIFFERENCE MEASUREMENT

As a last test, I had fun setting up the EMX-7150 and the ECM8000 side by side. As a representative of free field equalization, the EMX-7150 is oriented at 0° degrees to the loudspeaker. In contrast, I set up the Behringer at 60° degrees. Both capsules are positioned approximately at the string marking.

Smart has a very useful feature for this and allows the difference measurement between two inputs. So I chose Behringer as the measurement signal and the EMX-7150 as the reference signal. And lo and behold: the curves are almost identical!

Nevertheless, the curve of the iSEMcon (green) roughly shows my expectations - more or less linear up to the highest frequencies.