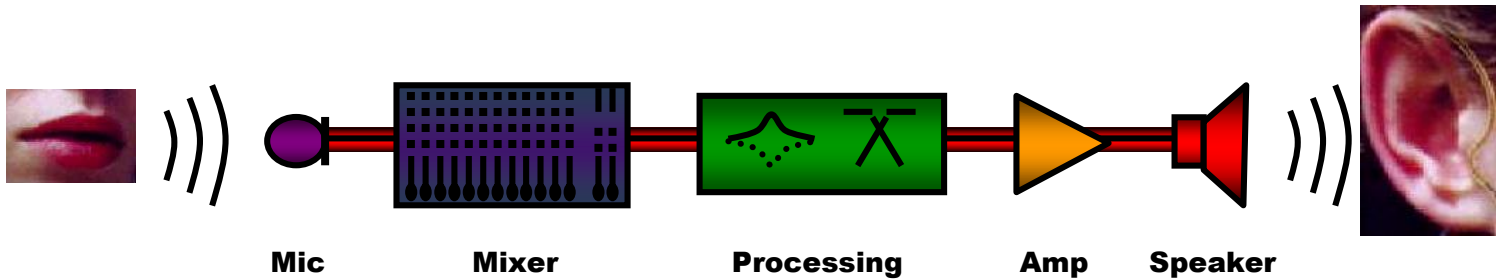


# ACOUSTIC BASICS

SIGNAL CHAIN

# The Signal chain

## Part I



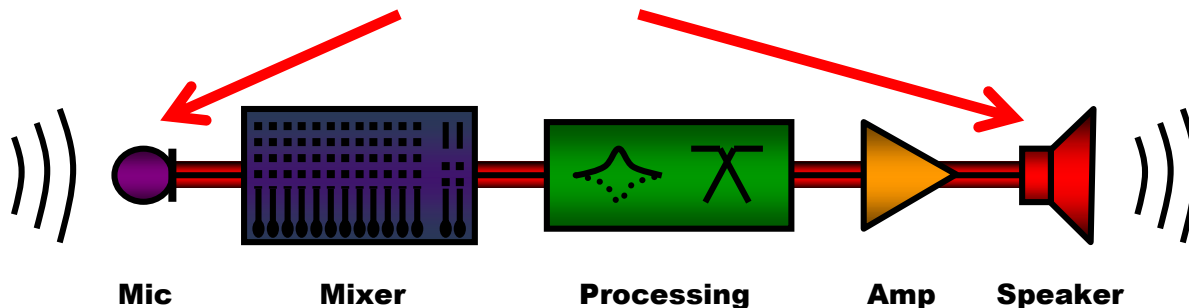
# Signal Chain

## The Signal Chain in Total

### *Microphones*

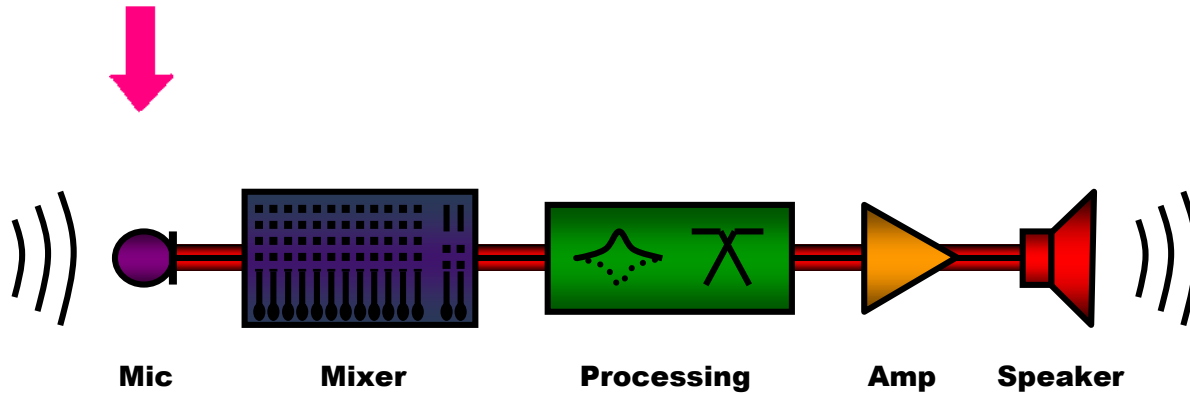
- Mixing Consoles
- Processing
- Amplifiers
- *Loudspeaker*

→ **Part I includes the in- and output side**



# Signal Chain

## Microphones



# Microphones

## Overview

### → Electroacoustical transducers:

- Dynamic microphones
- Condenser microphones

There are still some more types of microphones, but we will have a closer look to the above mentioned ones.

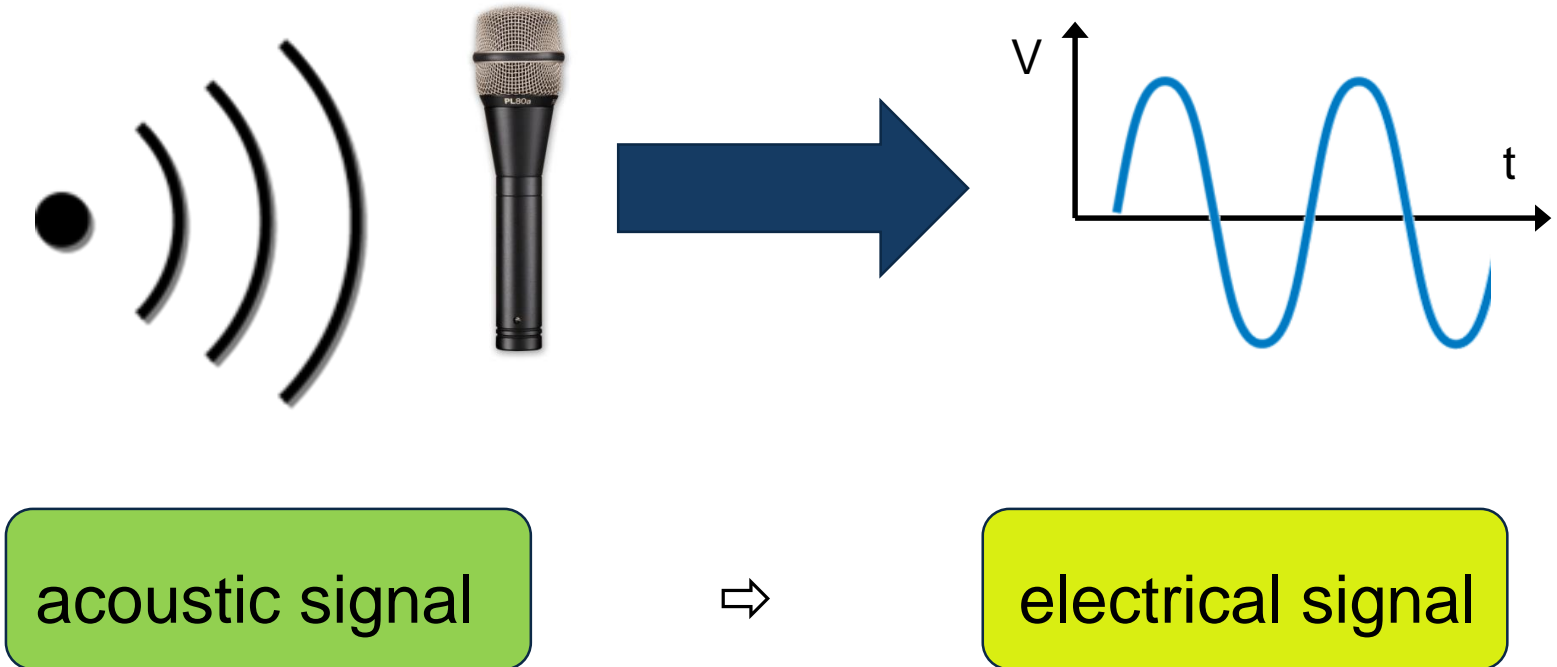
### → General terms

- Polar Pattern
- Proximity-effect
- Frequency response

# Microphones

## Electroacoustical Transducer

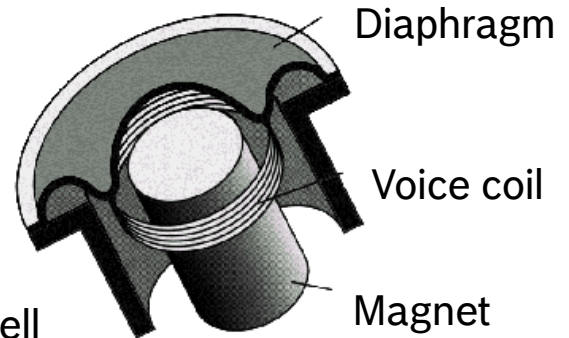
→ How the microphone converts sound into an electrical signal



# Microphones

## Microphone transducer technologies

- Dynamic microphone:
- Pro's and Con's vs. Condenser microphone
  - Simple Design, tend to be more affordable
  - Large Size – Rugged
  - Limited Frequency Response (Sound Quality)
  - Handle extremes of temperature and humidity very well



### → How is it working?

Sound waves set the diaphragm vibrating. The voice coil, which is connected directly with the diaphragm, moves in the magnetic field with a certain speed and an alternating voltage is inducted into the voice coil -> the acoustic signal is converted into a small electrical signal.

# Microphones

## Microphone transducer technologies

- Condenser (=Capacitor) microphone:
- Pro's and Con's vs. Dynamic microphone
  - Requires Power (phantom or bias voltage)
  - High sensitivity (more natural/clearer sound)
  - Extended frequency response
  - Better transient response
  - More expensive
  - An additional microphone preamp is needed to match the output signal to the following signal processing. The preamp has two functions - convert impedance from about  $1\text{ M}\Omega$  to  $150\ \Omega$  and bring the signal to "standard" - which means higher - mic level. The above mentioned phantom power supply is necessary to power the preamp circuit. If a low quality preamp is used, signal to noise ratio and dynamic problems can occur.
  - Flat frequency response and wide bandwidth are easier to achieve in comparison to a dynamic microphone.
  - The microphone cartridge can be built in smaller dimensions. It is the only choice if it comes to a subminiature lavalier microphone.



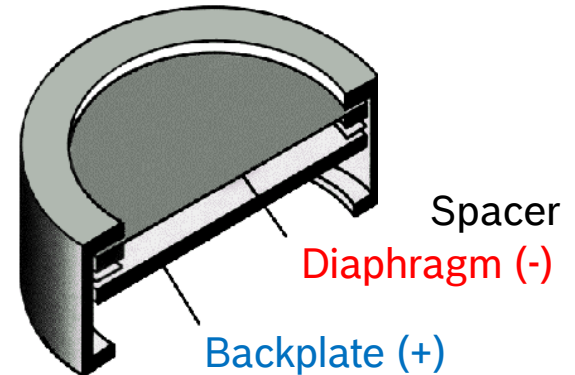
# Microphones

## Microphone transducer technologies

→ Condenser (=Capacitor) microphone:

→ **How is it working?**

The plates (diaphragm / Backplate) are electrically charged, one positive and the other negative -> that means a capacitance is built up between the plates. If sound waves set the very thin gold-coated-plastic diaphragm, which is mounted just in front of the rigid backplate, in vibration, a change of the “basic” capacitance happens and in turn this results in a variable electric current flow -> the acoustic signal is converted into a small electrical signal.



# Microphones

## Phantom Power

- Phantom power is a DC voltage (11 - 52 volts) which powers the preamplifier of a condenser microphone.
- Phantom power is supplied by
  - microphone mixer
    - If a mixer supplies 48 volts of phantom, XLR pins 2 and 3 of the microphone cable carry 48 volts DC relative to pin 1. Mixers that supply phantom power contain current limiting resistors which act as control valves. If the microphone or cable is improperly wired, these resistors limit the flow of current to the microphone and thereby prevent damage to the phantom supply circuit.
  - separate phantom power supply

# Microphones

## Phantom Power

- Phantom requires balanced circuit in which XLR pins 2 and 3 carry the same DC voltage relative to pin 1.
- The microphone cable carries the audio signal as well as the phantom voltage.
- A balanced dynamic microphone is not affected by phantom power.
- However, an unbalanced dynamic microphone (not common) will be affected. Although the microphone will probably not be damaged, it will not work properly.

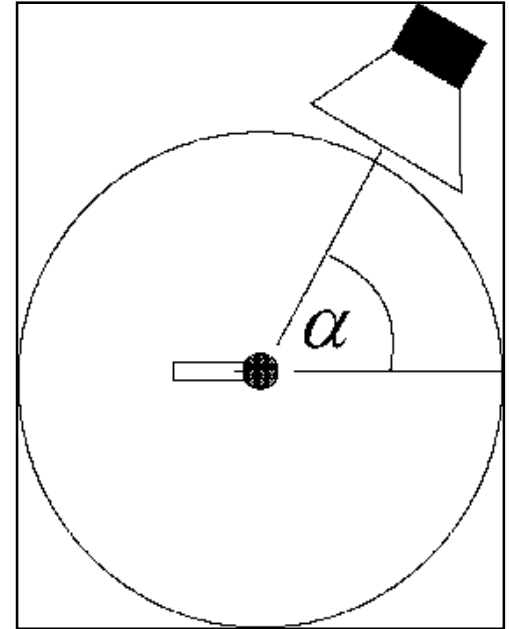
# Dynamic vs. Condenser Microphones

	Dynamic Microphones	Condenser Microphones
Construction	Simple	More complex
Cost	Less expensive	More expensive
Handling	Tolerates very rough handling	Requires careful handling
Sound Quality	Excellent over a wide frequency range	Very sensitive, smooth, natural sound even at the highest frequencies
Power Source	Does not require a separate power source	Requires phantom power or batteries
Environment	Good for live performance and some recording applications	Good for controlled environments, recording and some live applications

# Microphones

## Polar Pattern and Polar diagram

- A **polar pattern** is three dimensional. It demonstrates, how sensitive a microphone is to sound waves arriving from different directions. To measure the polar pattern, the microphone is rotated in front of a loudspeaker. The output signal is recorded in its relation to the angle between the microphone and the loudspeaker.
- A **polar diagram** is two dimensional and shows is a cross-section of the three dimensional polar pattern. That is the most common way in data sheets. The polar diagram provides information about the sensitivity of the microphone in relation to the angle between the sound source and the diaphragm / capsule of the microphone.



# Microphones

## Directionality

- The closer you get to a sound source, the higher is the sound pressure level. This local change in sound pressure level is physically defined as a pressure gradient field. If a microphone points in the direction of a sound source, the sound pressure level in front of the microphone element is higher than behind of it.
- Pressure Gradient Microphones
  - have openings at the rear side of the microphone head. Through these openings sound waves can enter the cartridge rear.
  - The difference between the pressure upon the diaphragm from the front and the rear determines the output signal of the microphone.
  - In addition both components have different phase positions, which leads either to a positive or destructive interference.

# Microphones

## Directionality

→ The overall size of the rear side openings has a strong influence upon the resulting polar pattern. It determines the acoustical impedance of the openings.

- no openings: omnidirectional
- midsize openings: unidirectional
- big openings: bi-directional

# Microphones

## Proximity-Effect

→ In easy words the proximity effect increases low frequencies if a sound source is close or too close to a directional microphone. This effect does not show up with a non-directional microphone. The short-distance sound zone is responsible for this effect at pressure gradient microphones.

→ **Reason for this effect:**

If microphones are positioned very close to sound sources the bending of the wave front (spherical wave) compared to the wavelength gains more importance. Therefore the pressure gradient increases disproportional compared to the sound pressure level. There is an phase shift component caused by the spherical wave form additive to the existing phase shift of the rear sound. The result is:

⇒ higher drive of the microphone diaphragm

⇒ higher output signal



# Microphones

## Proximity-Effect in practice

- The closer the microphone is moved to the sound source, the more low frequencies are boosted.
- At bigger distances (~more than 1 m) the voice sounds thin and without any bass.
- In „close talking“ mode the voice sounds full and deep.
- The proximity-effect can be used on purpose to let voices and instruments sound lower and fuller.
- On the other side the artist, e.g. singer has to show a high discipline in use of such microphones. For constant sound the distance between singer and microphone must be kept constant.

# Microphones

## Frequency Response of Microphones

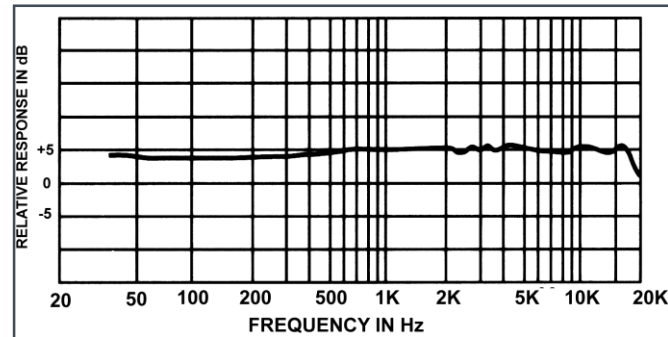
- The frequency response of a microphone is defined by:
  - the sound spectrum it can reproduce and
  - the variations of the output level within the frequency range.
- The frequency response is a major factor on the sound quality of a microphone.

# Microphones

## Frequency Response of Microphones

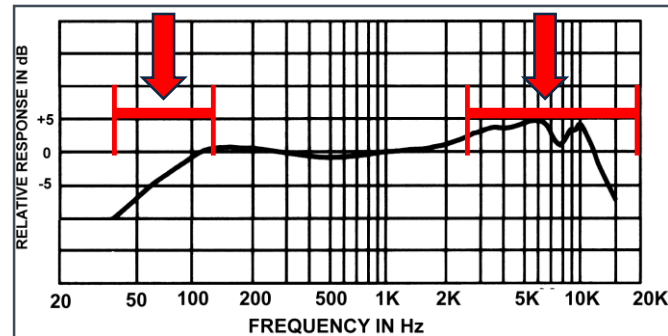
### → Flat Frequency Response

- Uniform output at all frequencies
- Extended range typical
- Smooth overall shape







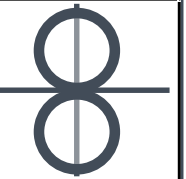
### → Shaped Frequency Response

- Output varies with frequency
- Reduced range typical
- Some peaks or dips in response



# Microphones

## Typical Polar Patterns - Overview

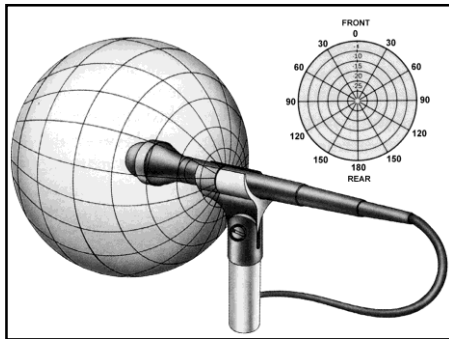
CHARACTERISTIC	OMNI-DIRECTIONAL	CARDIOID	SUPER-CARDIOID	HYPER-CARDIOID	BI-DIRECTIONAL
POLAR RESPONSE PATTERN					
COVERAGE ANGLE	<b>360°</b>	<b>131°</b>	<b>115°</b>	<b>105°</b>	<b>90°</b>
ANGLE OF MAXIMUM REJECTION (null angle)	—	<b>180°</b>	<b>126°</b>	<b>110°</b>	<b>90°</b>
AMBIENT SOUND SENSITIVITY (relative to omni)	<b>100%</b>	<b>33%</b>	<b>27%</b>	<b>25%</b>	<b>33%</b>
DISTANCE FACTOR (relative to omni)	<b>1</b>	<b>1.7</b>	<b>1.9</b>	<b>2</b>	<b>1.7</b>

# Microphones

## Polar Pattern

→ Omnidirectional:

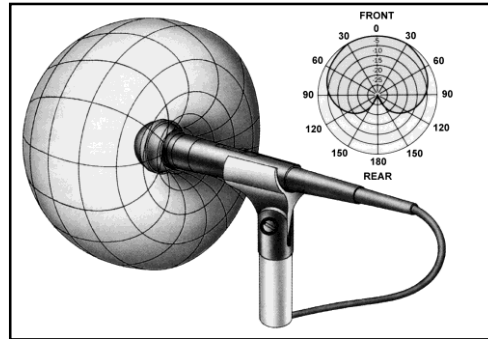
Equal Response at all angles



→ Cardioid:

Most sensitive at 0 degrees (on axis)

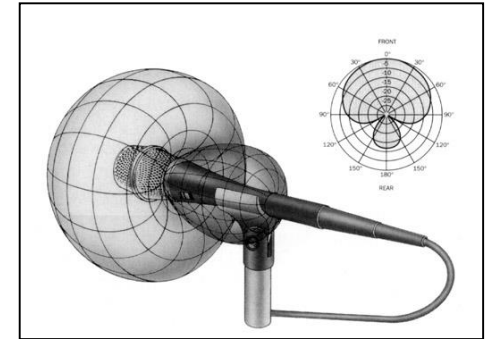
Least sensitive at 180 degrees (off Axis)



→ Supercardioid:

Most sensitive at 0 degrees (on axis)

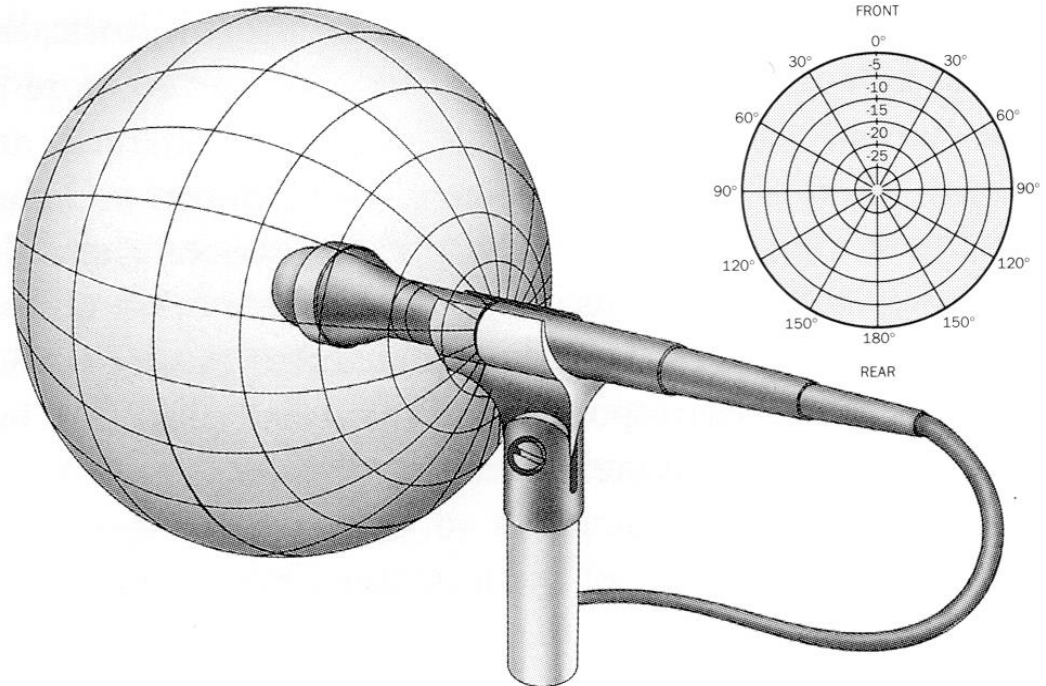
Least sensitive at 126 degrees



# Microphones

## Polar Pattern

- **Omnidirectional** Microphone is equally sensitive for sound sources out of every direction. There is no preference to any direction.
- Pressure Transducer
- No proximity effect



**OMNIDIRECTIONAL MICROPHONE**

# Microphones

## Polar Pattern

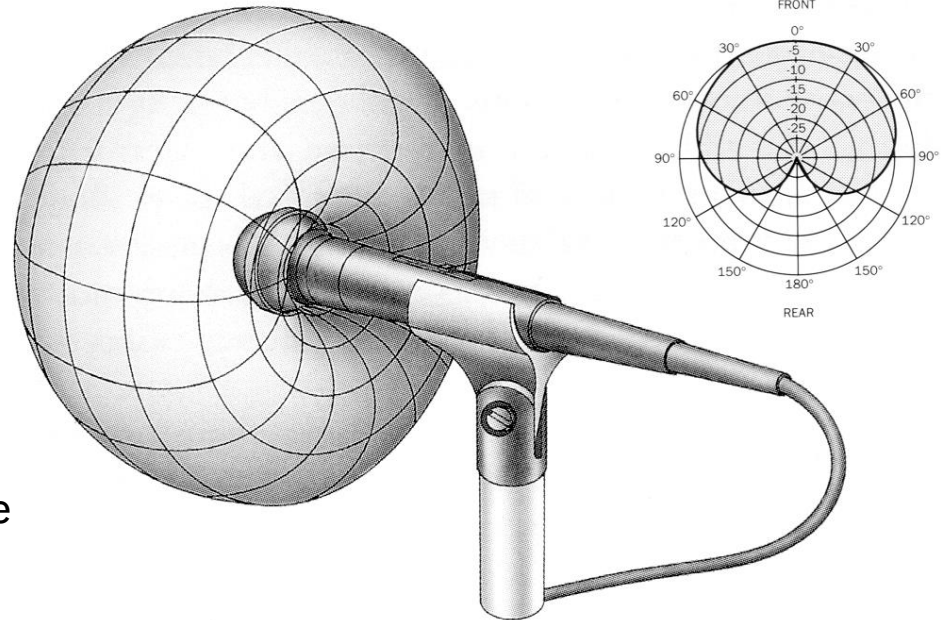
### → **Cardioid**

Microphone is most sensitive for signals in front of the capsule ( $0^\circ$ ), signals coming from the rear side are picked up at a reversed level.

Because of this directional preference, microphones of this type are used to pick up desired sound sources only.

→ Pressure-gradient microphone

→ Proximity effect



**CARDIOID (UNIDIRECTIONAL) MICROPHONE**

# Microphones

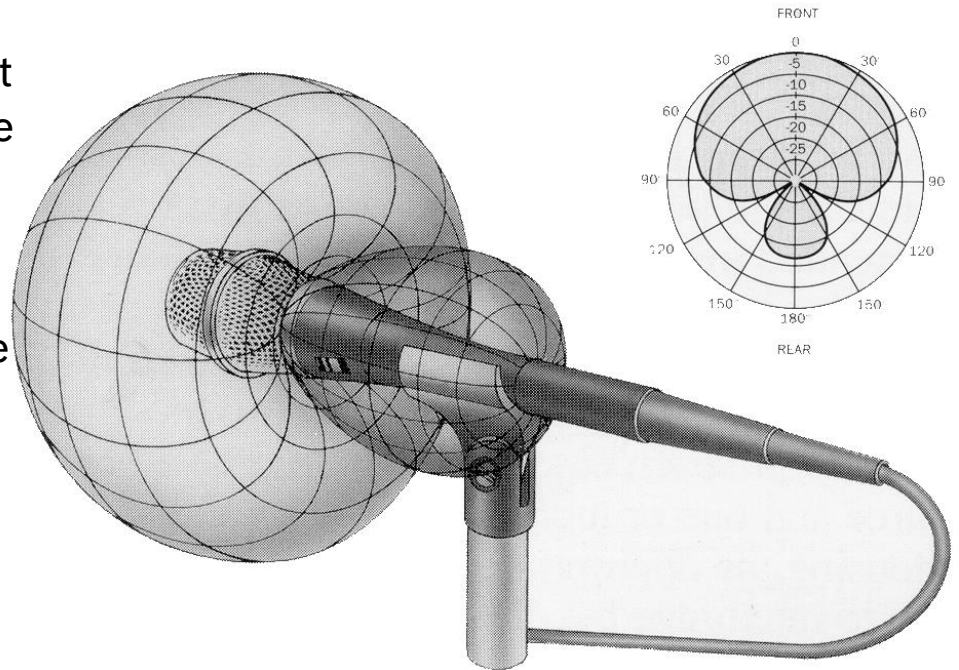
## Polar Pattern

### → Supercardioid

Microphone is most sensitive for signals in front of the capsule ( $0^\circ$ ) and least sensitive for sounds arriving at  $126^\circ$ . It is further less sensitive to ambience noise than an omnidirectional mic .

### → Pressure-gradient microphone

### → Proximity effect

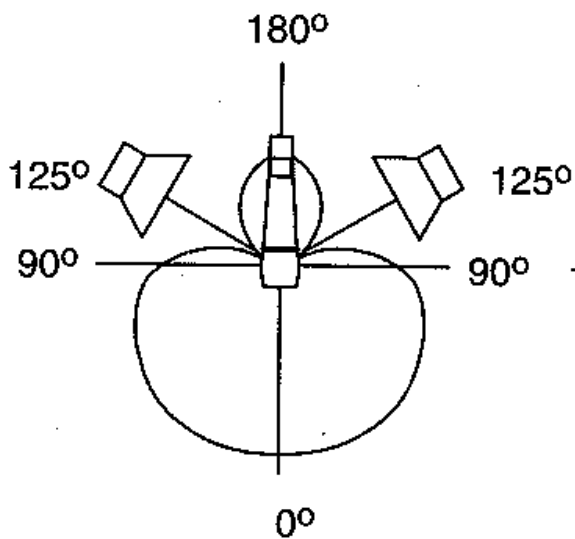




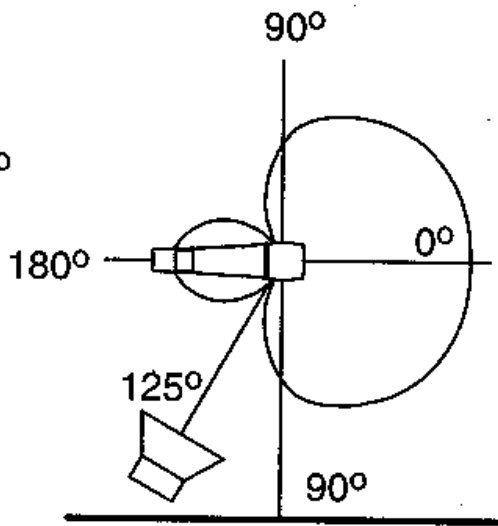
# Microphones

## Polar Pattern

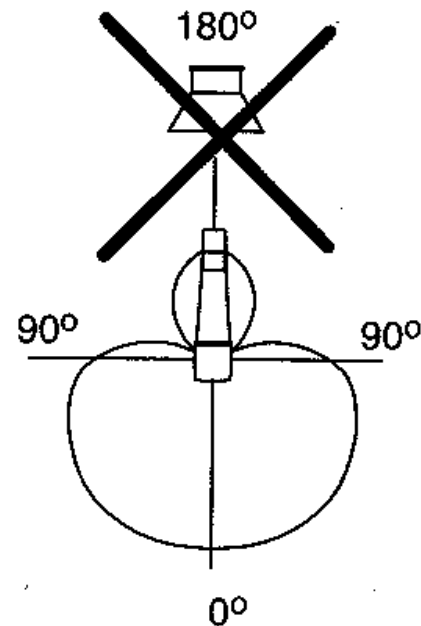
- Practical example if it comes to monitor loudspeaker placement on stages with supercardioid microphones. With correct placement of the monitor higher levels are possible without getting a feedback.



→ CORRECT



→ CORRECT



→ INCORRECT

# Microphones

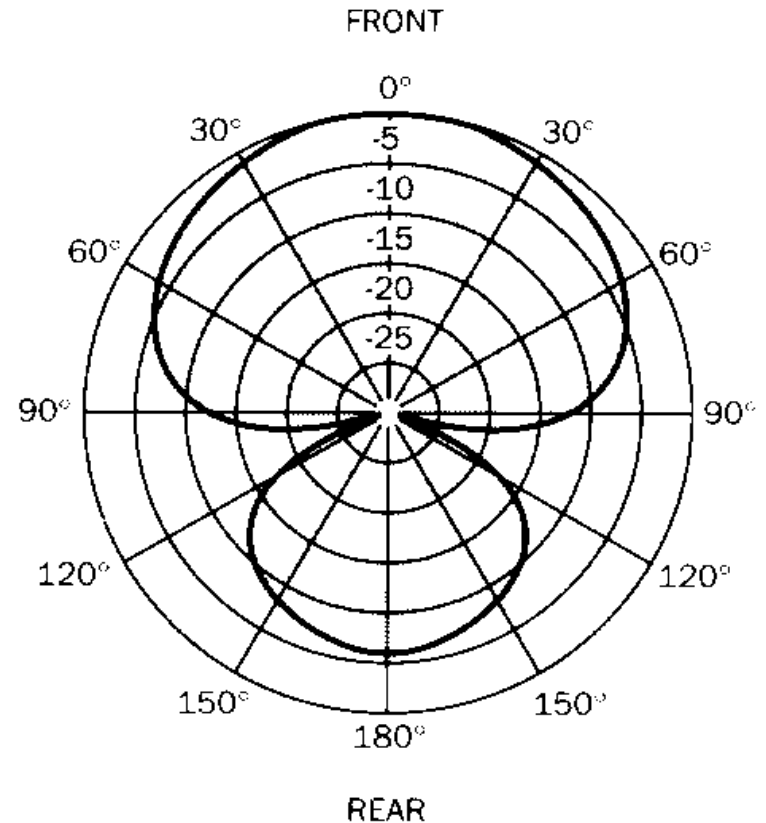
## Polar Pattern

### → **Hypercardioid**

Microphone is most sensitive for signals in front of the capsule (0°) and least sensitive for sounds arriving at 110°. Less sensitive to ambience noise than an omnidirectional microphone.

→ Pressure-gradient microphone

→ Strong proximity effect

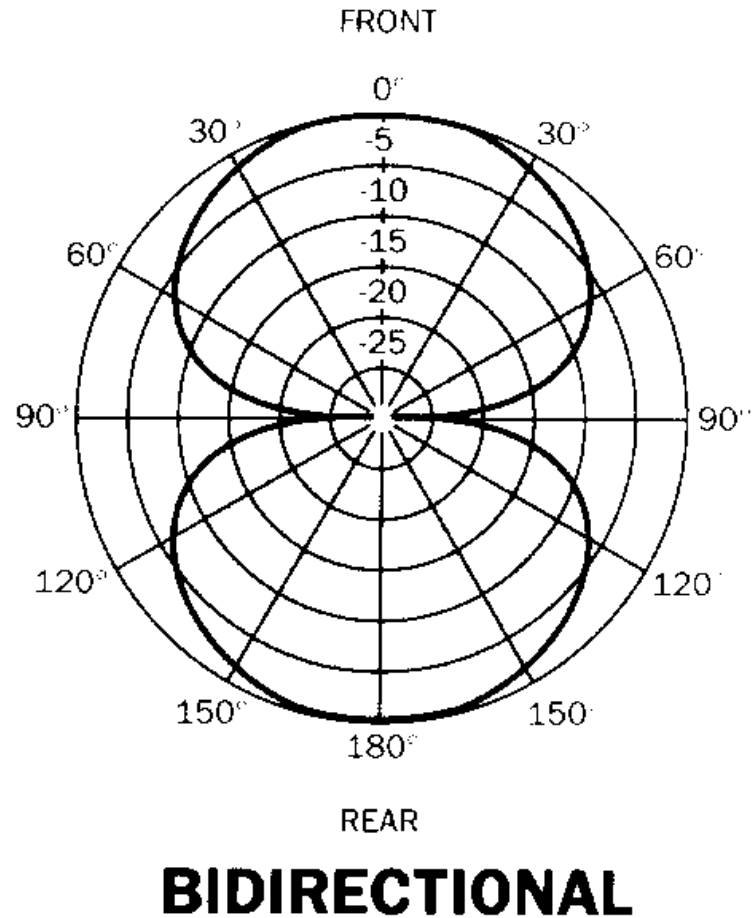


## **HYPERCARDIOID**

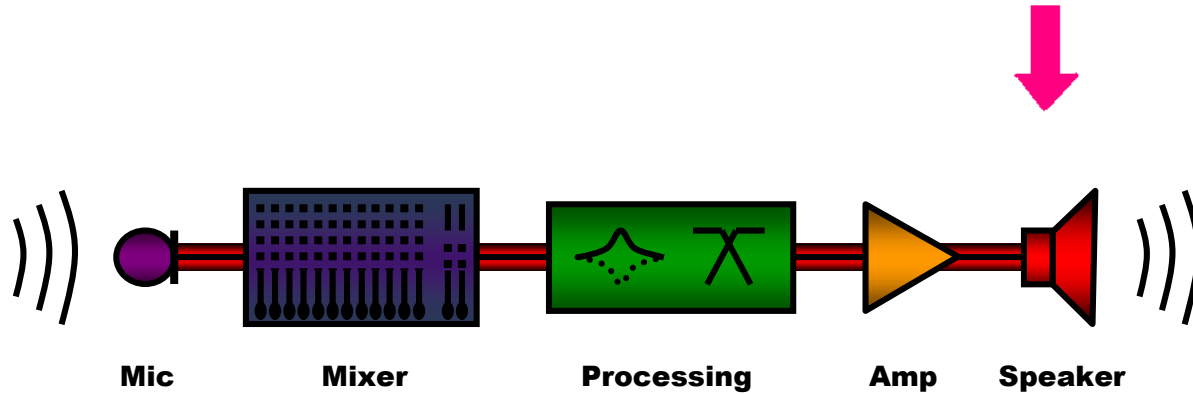
# Microphones

## Polar Pattern

- **Bi-directional / Figure of Eight** Microphone is very sensitive for sound sources in front of and behind the capsule ( $0^\circ$  and  $180^\circ$ ) and is least sensitive to sounds arriving at  $90^\circ$ . Less sensitive to ambience noise than an omnidirectional microphone.
- Pressure-gradient microphone
- Very strong proximity effect



# Loudspeaker



# Loudspeaker

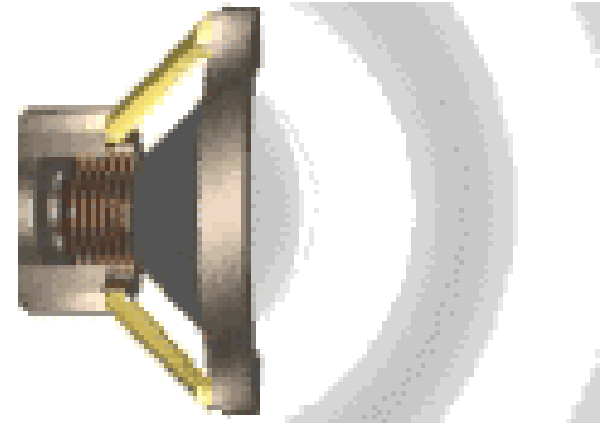
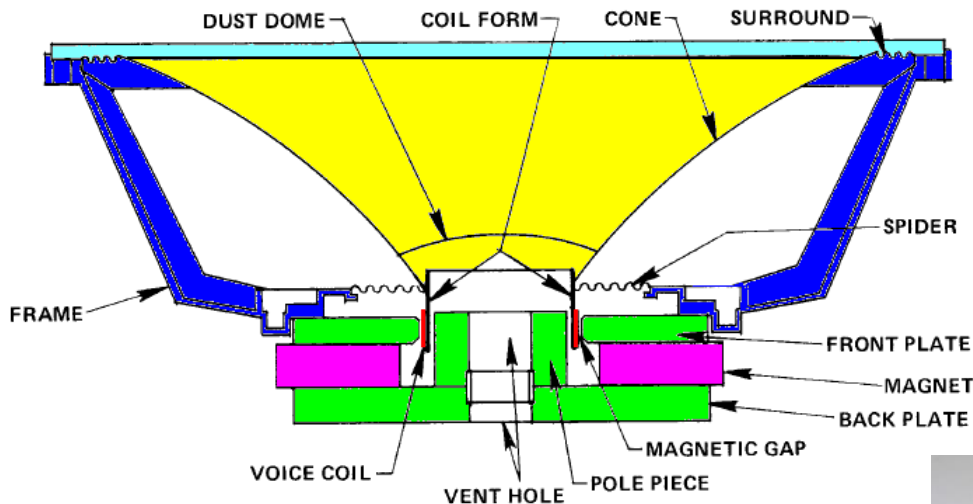
## General

- A loudspeaker is a converter from electrical energy into acoustical energy. By moving the cone synchronous to the electrical signal the electrical energy is transduced to the air by compression & decompression.
- **Typical conversion types are:**
  - **Electromagnetic :**  
Alternating current moves coil in a static magnetic field - most common in Pro Audio applications
  - **Piezo :**  
Piezoelectric material expands/contracts according to applied voltages
  - **Electrostatic :**  
Two plates attract or distract each other depending on electrical charge relations

# Loudspeaker

## Cross section of a Cone Transducer

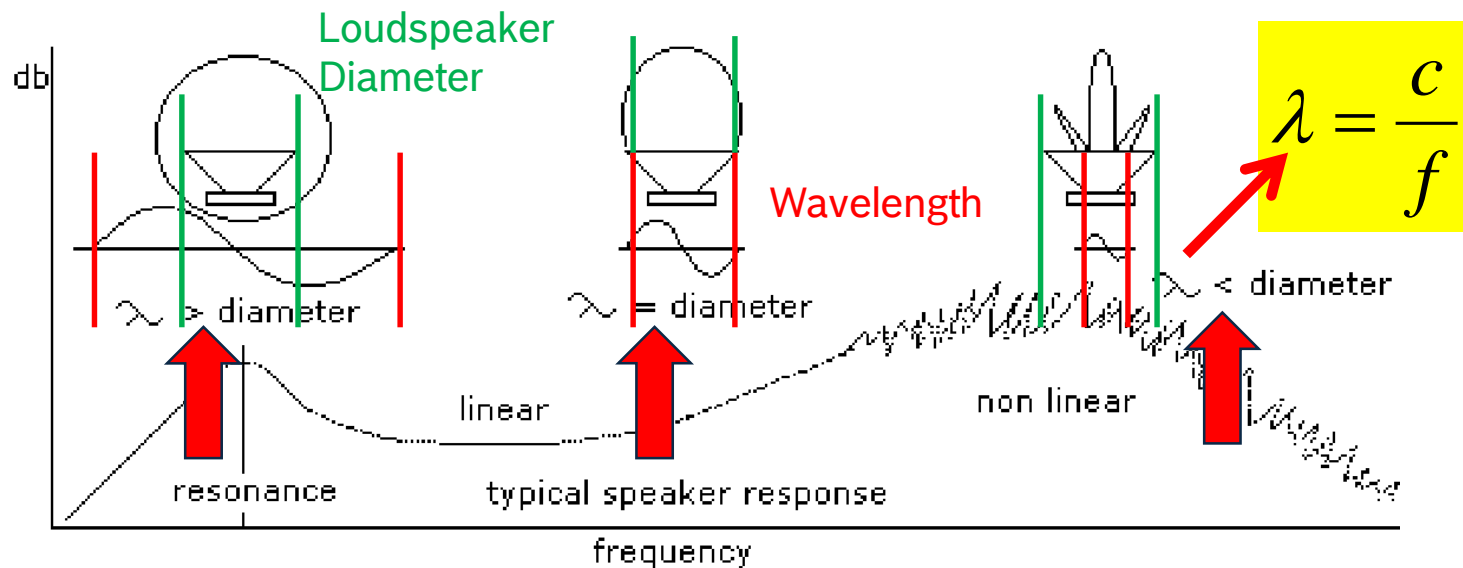
→ Electromagnetic / „Electrodynamic“ Principle



# Loudspeaker

## Bandwidth of loudspeaker chassis

- A single professional speaker chassis cannot reproduce the complete human hearing window with properties needed for professional concert sound applications. It refers to the relation of wavelength to speaker diameter. That is the reason for multiple ways / loudspeakers in professional applications to reproduce the audible frequency range.



# Loudspeaker

## Frequency and wavelength

$$\lambda = \frac{c}{f}$$

c = speed of sound in air  
[m/s]  
(344 m/s @20°C)  
f = frequency [Hz]

The relationship between wavelength and frequency for a range of frequencies is given in tabular form below

Frequency [Hz]	Wavelength [ft]	Wavelength [m]
20	56.50	17.20
31.5	35.87	10.92
40	28.25	8.60
50	22.60	6.88
63	17.94	5.46
80	14.30	4.30
100	11.30	3.44
125	9.04	2.75
160	7.06	2.15
200	5.65	1.72
250	4.52	1.38
315	3.59	1.09
400	2.83	0.86
500	2.26	0.69
630	1.79	0.55
800	1.413	0.430
1,000	0.130	0.344
1,250	0.904	0.275
1,600	0.706	0.215
2,000	0.565	0.172
2,500	0.452	0.138
3,150	0.359	0.109
4,000	0.283	0.086
5,000	0.226	0.069
6,300	0.179	0.055
8,000	0.141	0.043
12,500	0.090	0.028
16,000	0.071	0.022
20,000	0.057	0.017

→ Examples:

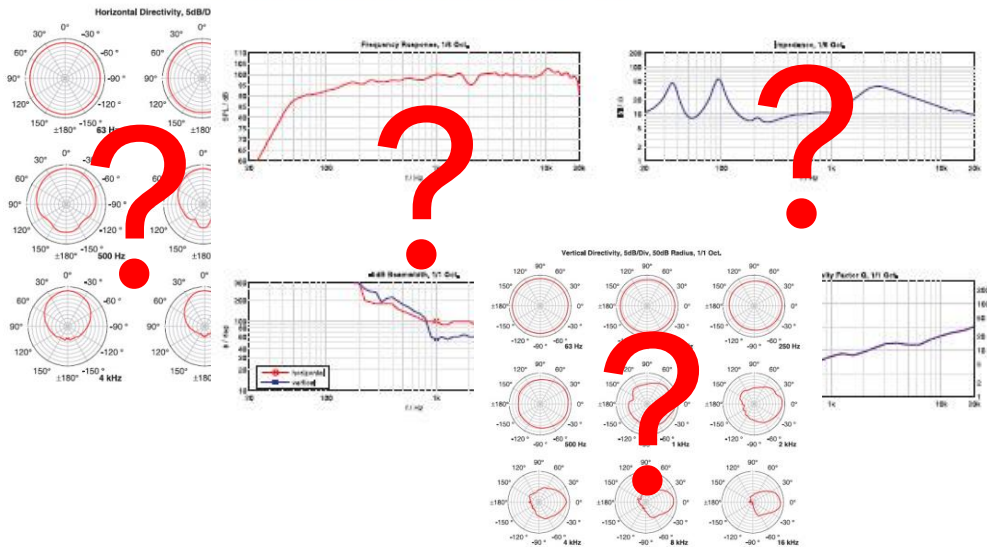
Frequency [Hz]	Wavelength [m]
100	→ 3,44
1.000	→ 0,344
10.000	→ 0,0344



# Loudspeaker Properties - Specifications

→ When we have a closer look into a datasheet of a professional loudspeaker we find a lot of information. What does it mean?

VL 122 Frequency Response, Impedance, Beamwidth, Directivity



## Technische Informationen

Architects and engineers specifications

**VL 122**  
12" / 1.4"  
2-Way Professional  
Speaker System

### Features:

- High-Power Multi-Functional 2-Way Cabinet
- 400W 8 Ohms
- Rotatable CAD-optimized 90° x 45° Constant Directivity Horn
- Switchable to Biamped Mode
- Neodymium ND Driver
- DND 12350 Woofer
- Voice Coil Tracking Protection



Wir weisen ausdrücklich darauf hin, dass bei gehängten Lautsprecherboxen die einschlägigen Sicherheitsbestimmungen zu beachten sind. Es muss unbedingt qualifizierte Fachberatung in Anspruch genommen werden!

We expressly point out that the relevant safety regulations must be observed for flying loudspeaker systems. It is imperative that qualified expert advice is sought.

<b>WARNING:</b>	
THIS LOUSPEAKER SYSTEM CAN PERMANENTLY DAMAGE HEARING! USE EXTREME CARE WHEN SETTING SOUND PRESSURE LEVELS!	
<b>DANGER:</b>	
CONSULT A CERTIFIED STRUCTURAL ENGINEER BEFORE ANY ATTEMPT TO SUSPEND THIS LOUSPEAKER! LOUSPEAKER CAN FALL FROM IMPROPER SUSPENSION, RESULTING IN SEVERE INJURY AND DAMAGE OF PROPERTY. DO NOT SUSPEND OR MOUNT ANY OTHER PRODUCT OR DEVICE FROM THIS LOUSPEAKER ENCLOSURE! USE GRADE 8 HARDWARE OR BETTER ONLY!	

Specifications	
TYPE / NAME	VL122
Order No.	05 115 000
Color	01-Black
Configuration	Passive 2-Way
Nominal impedance	8 Ohms/16 Ohms/32 Ohms
Rated power (1000 Hz)	400 Watts
Program power	500 Watts
Max. power	1000 Watts
Max. SPL	128 dB
SPL	128 dB (1m) / 118 dB (5m)
Max. SPL	128 dB (1m) (continuous with 1000 Hz sine wave)
Frequency range (10dB)	63 Hz - 20 kHz
Full range	63 Hz - 20 kHz
Frequency range (5dB)	63 Hz - 20 kHz
Measured with 1000 Hz (DIN EN 61672)	90° x 45°
Coverage range	115° x 45°
Active crossover frequency	1.2 kHz
Active crossover frequency	1200 Hz (PASSIVE) / 1200 Hz (ACTIVE)
Switch and tracking protection	Yes
Component 1 (Horn)	Electro-Voice NCR 7 Neodymium
Order No.	30 000
Component 2 (Woofer)	DND 12350
Order No.	30000
Construction	1.8" (47 mm)
Mount through	2 - 3
Construction Material	18 G. S. 1.23 x 0.2
Max. Weight	15.1 lb. (6.8 kg)
Connector model	Speaker 32 Pin 4-Wire
Dimensions (WxDxH)	(15.1" x 18.5" x 14.7")
Net weight	28kg (62.2 lb)
Packing weight	29kg (64.1 lb)
Enclosure material	Steel (painted)
Items	Check Component list
Order	Product code and serial number
Warranty	2
Warranty	30 Months
Optional Accessories	
05 115 222 Cast Cover	05 115 18
05 115 244 Mounting Bracket	05 115 14
05 115 245 Mounting Bracket	05 115 07
05 115 246 Mounting Bracket	05 115 08
05 115 247 Mounting Bracket	05 115 09
05 115 248 Mounting Bracket	05 115 10
05 115 249 Mounting Bracket	05 115 11
05 115 250 Mounting Bracket	05 115 12
05 115 251 Mounting Bracket	05 115 13
05 115 252 Mounting Bracket	05 115 14

# Loudspeaker Properties - Specifications

- Efficiency
- Sensitivity
- Frequency Response
- Impedance
- Coverage Angle
- Directivity Index DI,Q
- Power Handling

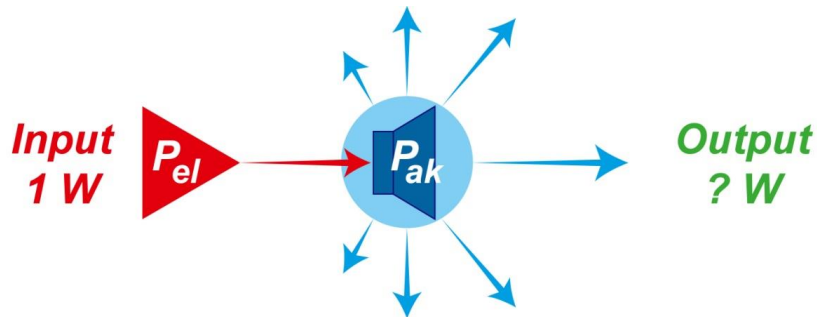
<b>Specifications</b>	
<b>TYPE / NAME</b>	<b>VL122</b>
Order No.	D 113 103
Cabinet	Full-Range
Configuration	Passive 2-Way Switchable to biamped mode
Nominal impedance	8 Ohms
Rated power RMS	400 Watts
Program power	800 Watts
Peak power	1600 Watts
SPL 2,83 V / 1m 1W / 8 Ohm	99 dB
Max. SPL 1m (calculated with peak power)	131dB
Frequency range ( -10dB )	
Full range	60Hz - 20kHz
Frequency range ( -10dB ) measured with H5000-RCM26 preset	50Hz - 20kHz
Coverage angle 1 kHz -6dB	90° * 45°
Passive crossover frequency	1.5kHz
Active crossover frequency	1.5kHz 24dB/Oct. Linkwitz-Riley, 1,5kHz Linear-Phase FIR
Voice coil tracking protection	Yes
Component HIGH	Electro-Voice ND6-8 Neodymium
Order No.	361493
Component LOW	DND 12350
Order No.	366559

# Loudspeaker

## Efficiency $\eta$ [eta]

→ Efficiency:

The efficiency factor describes the ratio between electrical to acoustic energy, which is usually not mentioned in actual datasheets. Common loudspeakers have an efficiency of only a few percent (from about 0,5% up to 20%).



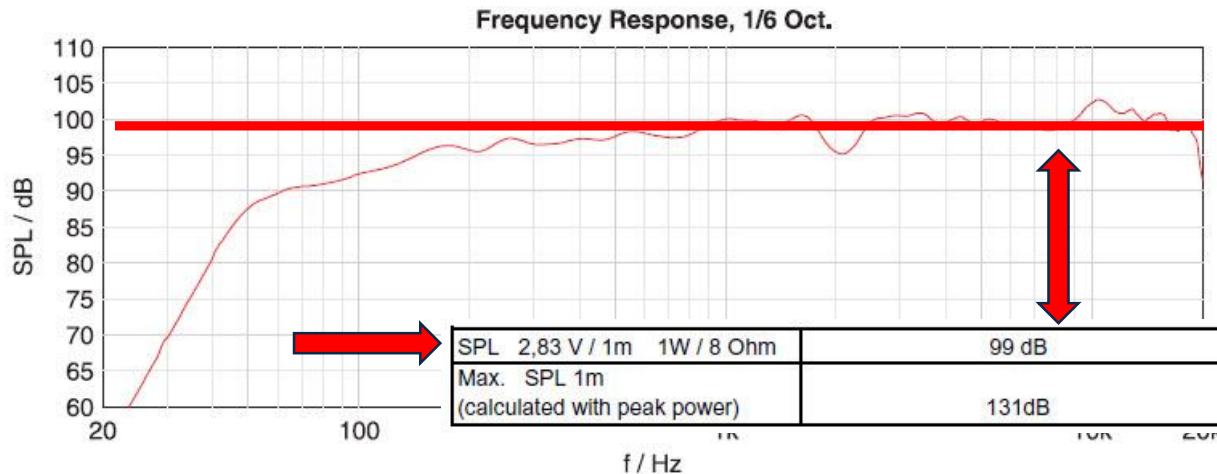
$$\eta = \frac{P_{el}}{P_{ak}}$$

Important : Don't mix the **efficiency** with On-axis **sensitivity**!

# Loudspeaker

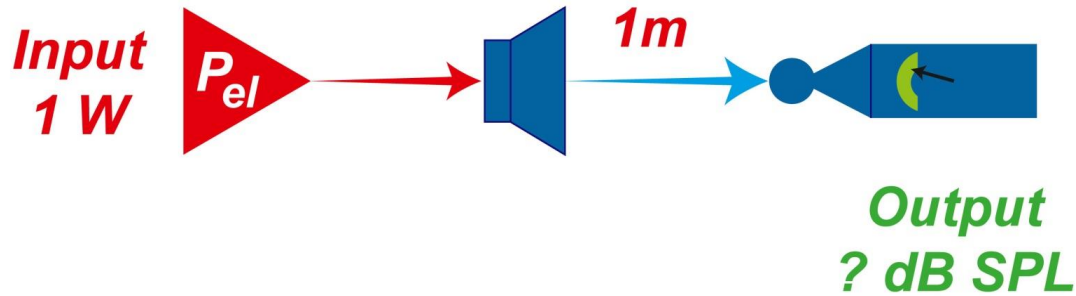
## Sensitivity

- Loudspeaker sensitivity is typically rated at 1W/1m. 1W at 8 Ohms is the equivalent of 2,83V. In order to maintain 1W of power at 4 Ohms, the voltage must change to 2V. In a few cases the sensitivity is specified at other distances, so please read the specifications carefully, when you are comparing loudspeakers by datasheet.



# Loudspeaker Sensitivity

→ Loudspeaker sensitivity is measured on axis only!

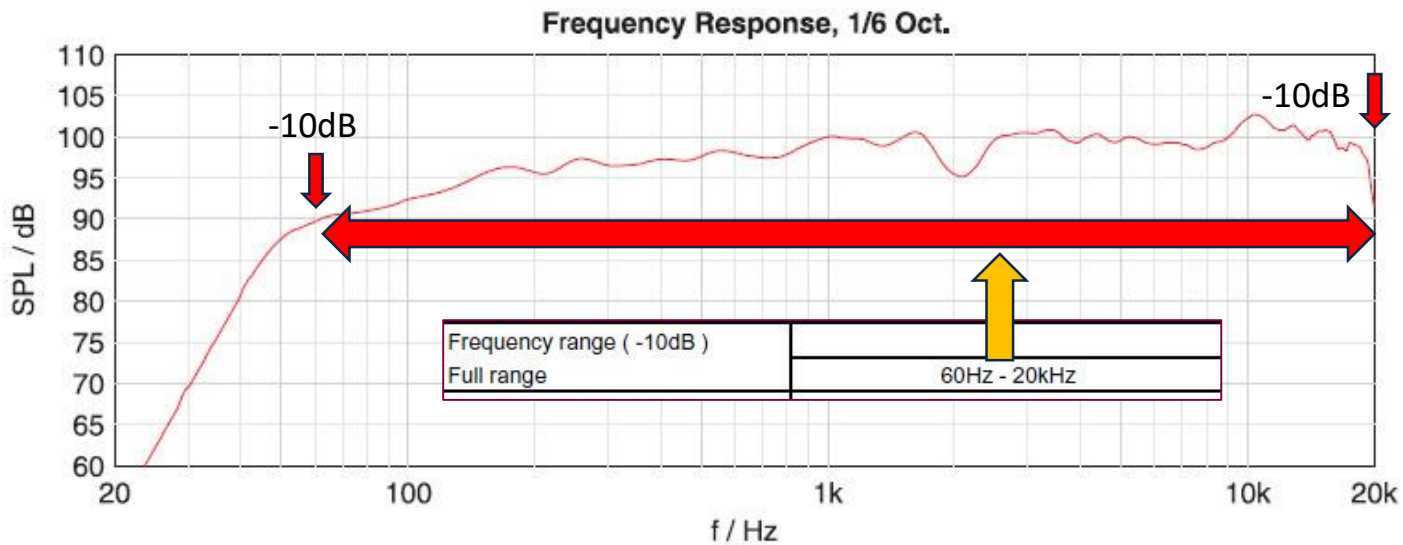


$$L_{sens} = ? \text{ dB SPL} / 1W / 1m$$

# Loudspeaker

## Frequency Response

- There are different kinds and possibilities to specify the frequency response or also called frequency range. If we look at our datasheet example it is specified within -10dB points, i.e. it shows the response of the loudspeaker, based on the sensitivity, where the level decreased by 10dB.

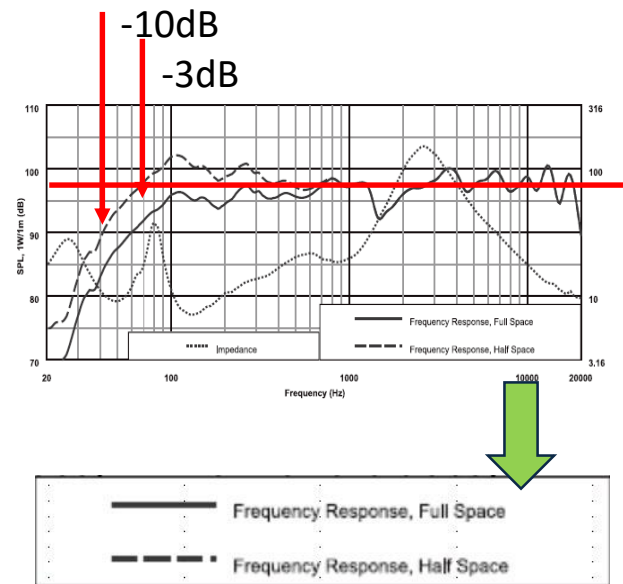


# Loudspeaker

## Frequency Response

- Another example, where -3dB and -10dB values are specified, but in this case at half space measurement.
- Boundary loading a loudspeaker has a frequency dependent effect on the response of the system. Up to 6dB of boost in the low frequency range are possible.

Freq. Response <sup>1</sup> (-3 dB):	58 Hz - 18 kHz
Freq. Range <sup>1</sup> (-10 dB):	39 Hz - 20 kHz
Rec. Hipass Frequency:	36 Hz
Axial Sensitivity:	98 dB (1W/1m)

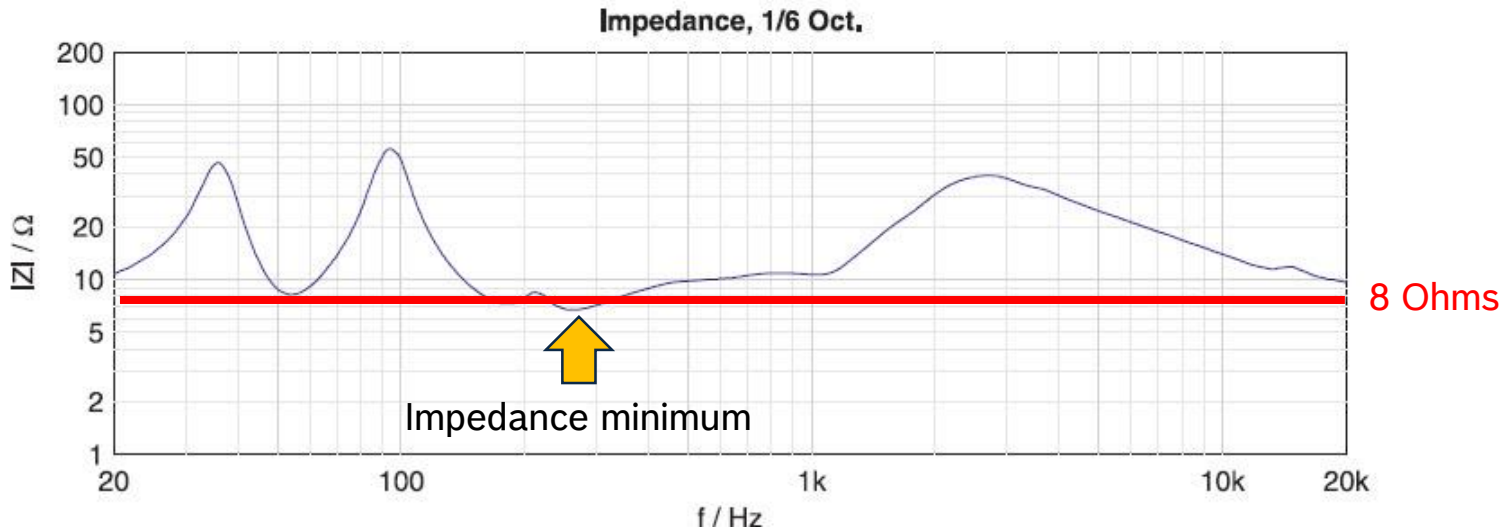


- **Full space** – Loudspeaker in free space with no boundary nearby
- **Half space** – loudspeaker on one boundary

# Loudspeaker

## Impedance

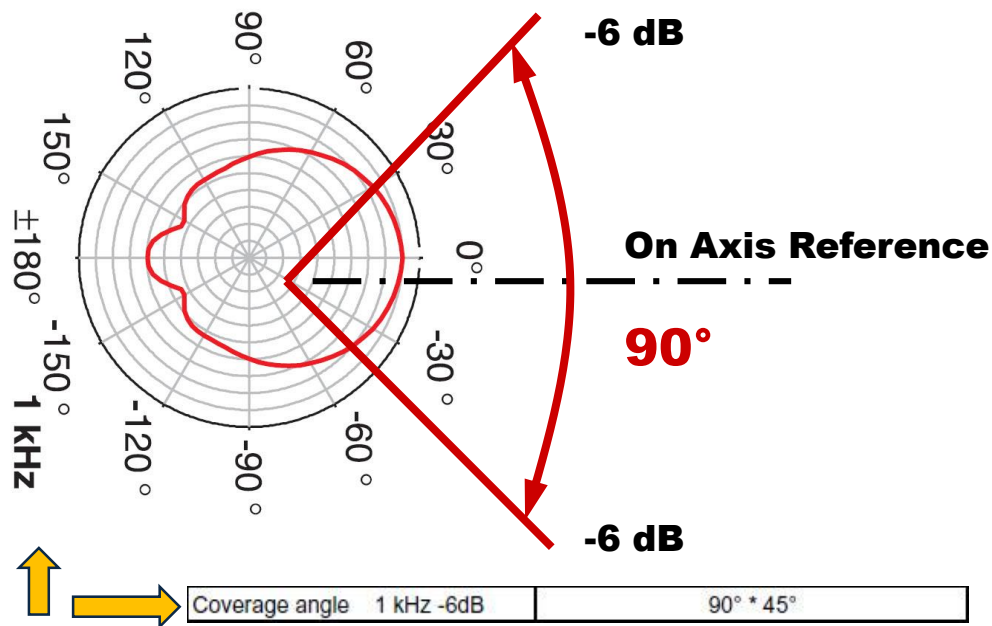
- You can see that the impedance of a nominal 8 Ohm speaker is anything except constant 8 Ohms...
- The minimal impedance should not be less than 20% beyond the nominal impedance. It is a very important value when speakers are connected in parallel to an amplifier (minimum load capability).





# Loudspeaker Coverage Angle

- The nominal coverage angle of a loudspeaker describes the angle of -6 dB SPL at a certain frequency compared to on axis level vertical x horizontal.



# Loudspeaker

## Directivity Q & Directivity Index DI

### → **Directivity factor Q:**

The Directivity factor Q is a numerical value depending on frequency, indicating the rate of acoustic power over all room directions.

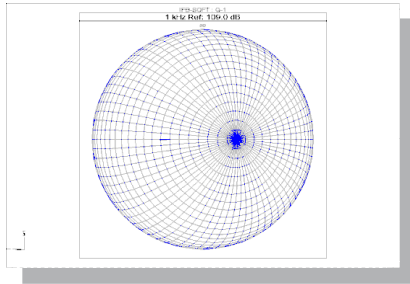
### → **Directivity Index DI:**

The Directivity Index DI is the Directivity Factor Q expressed in decibel (dB).

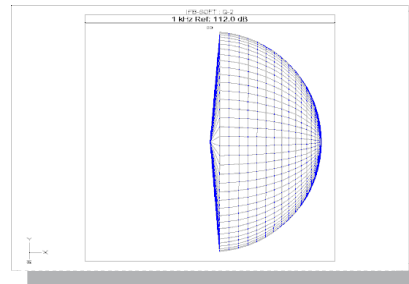
- The increase of the directivity in one area of spherical angle goes always with the decrease of the directivity in another area of spherical angle. Thus the emitted energy is more concentrated to the front.

# Loudspeaker

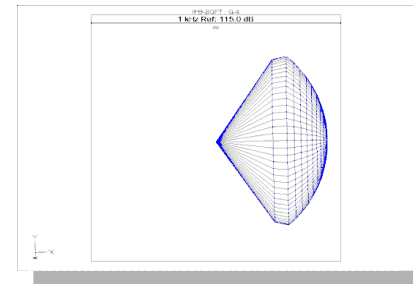
## Directivity Q & Directivity Index DI



**$Q=1$**   
 **$DI=0 \text{ dB}$**



**$Q=2$**   
 **$DI=3 \text{ dB}$**



**$Q=4$**   
 **$DI=6 \text{ dB}$**

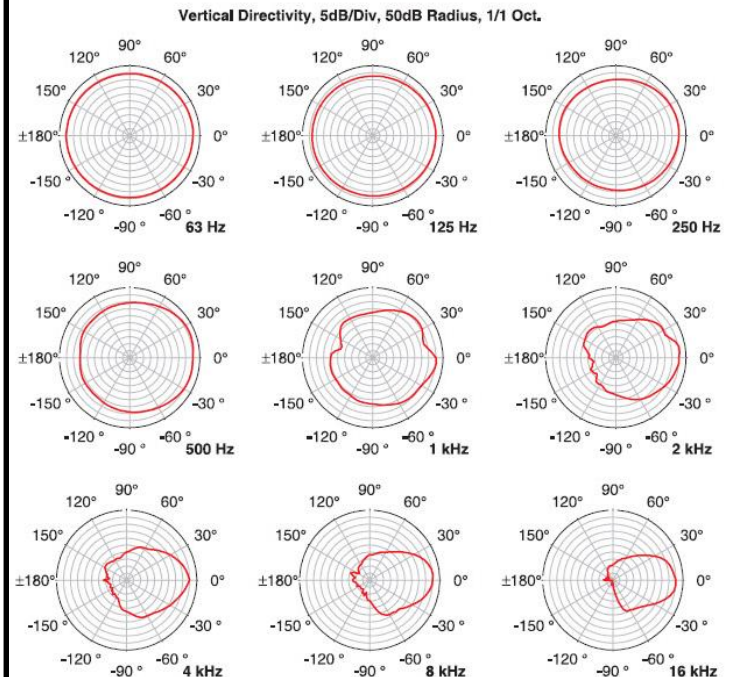
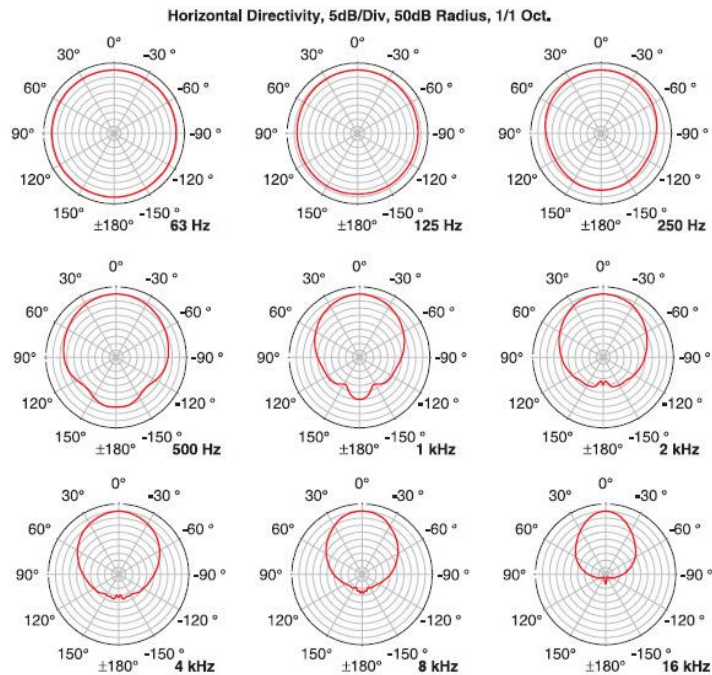
$$DI = 10 \log Q$$

→ For equal speaker efficiency : The higher directivity, the higher sensitivity

# Loudspeaker

## Polar Plots – Horizontal and vertical Directivity

- The polar plots represent a vertical and a horizontal slice of the entire spherical SPL data.



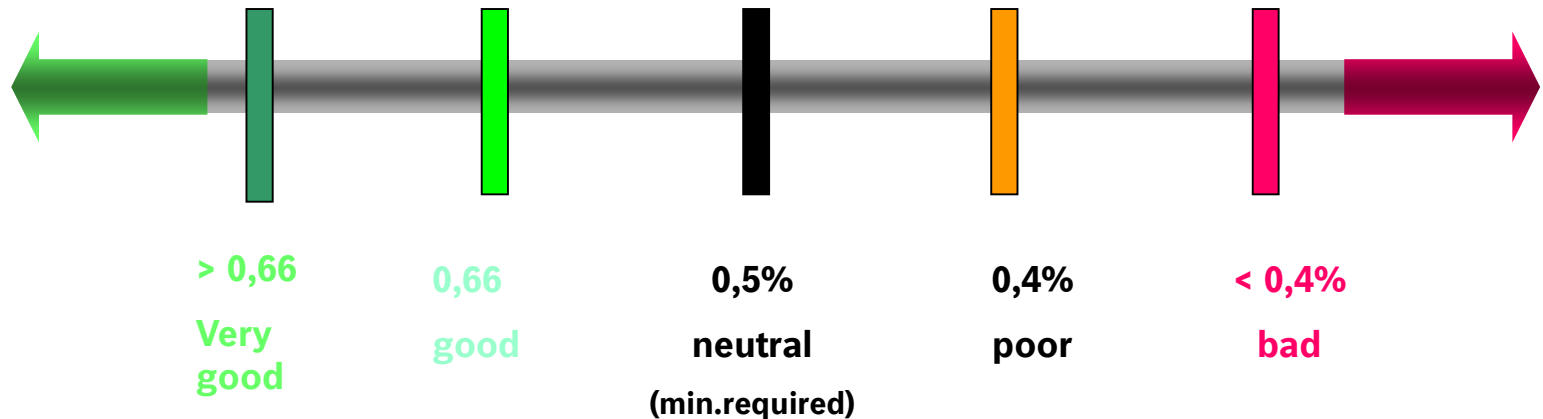
# Loudspeaker

## Directivity Q

- Why is the Directivity of a loudspeaker such an important factor for planning sound systems?

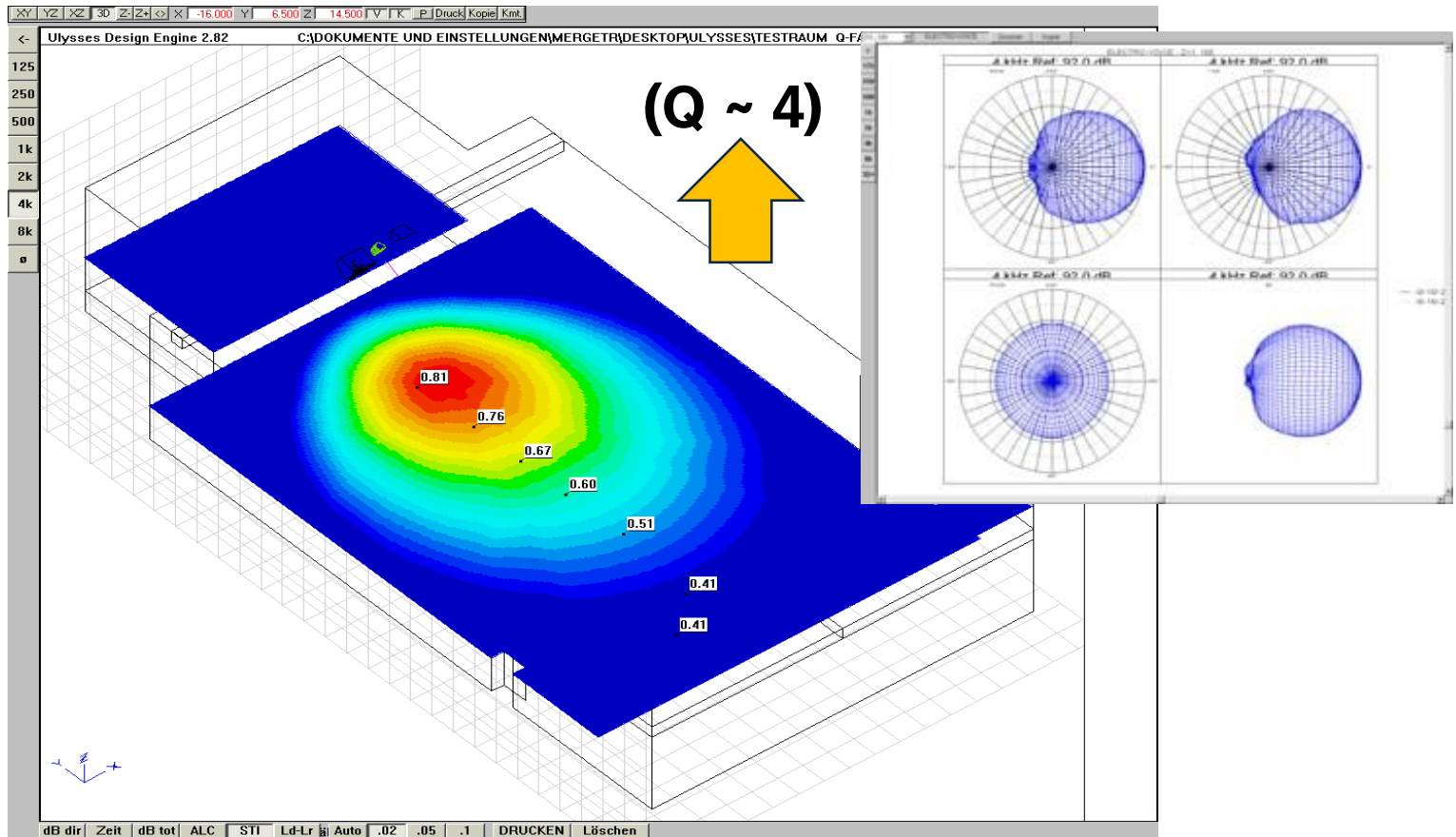
The following examples show the relationship between speech intelligibility (STI) and the directivity factor Q in the same room with equal conditions.

- STI values are in between 0 and 1:



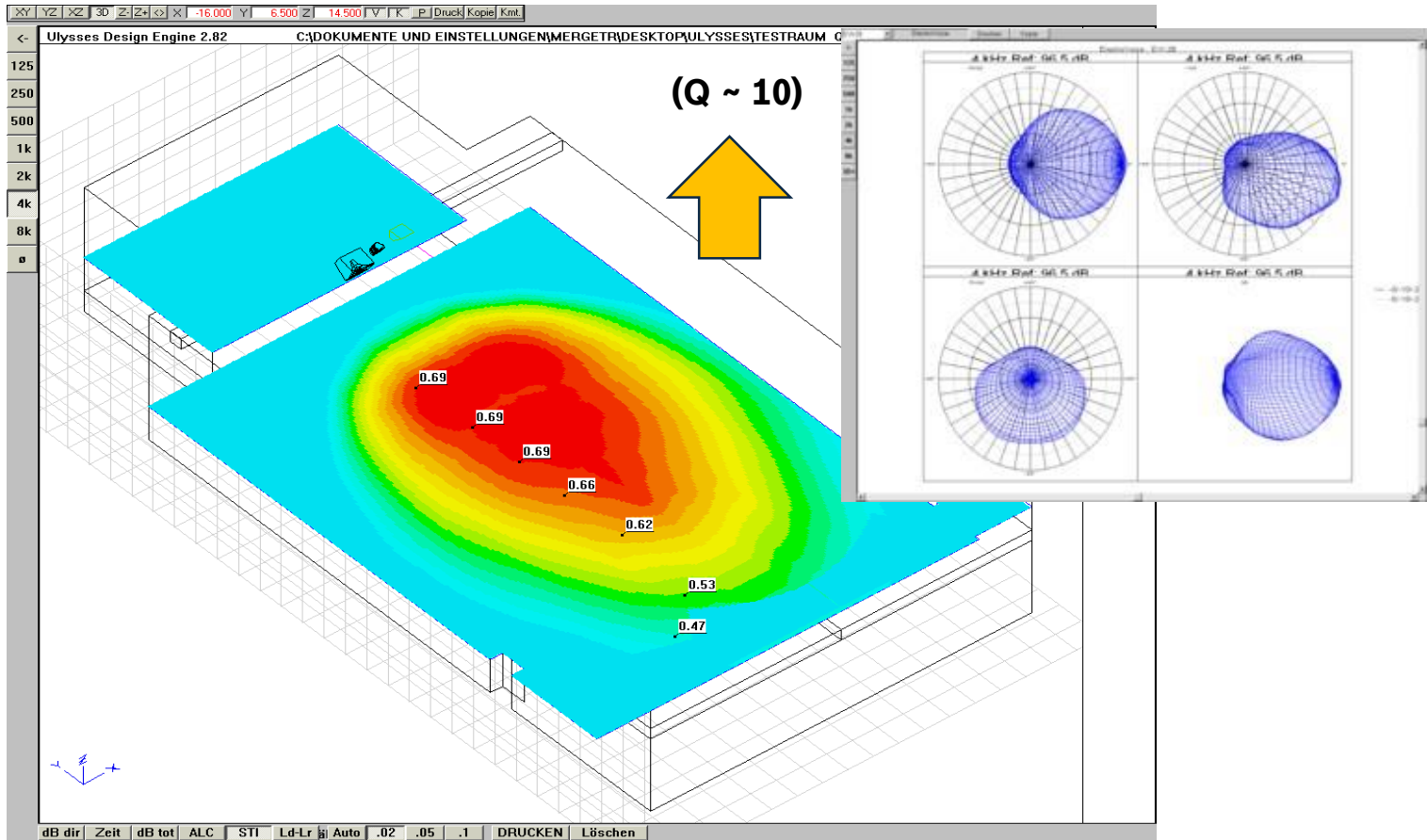
# Loudspeaker

## Directivity Q



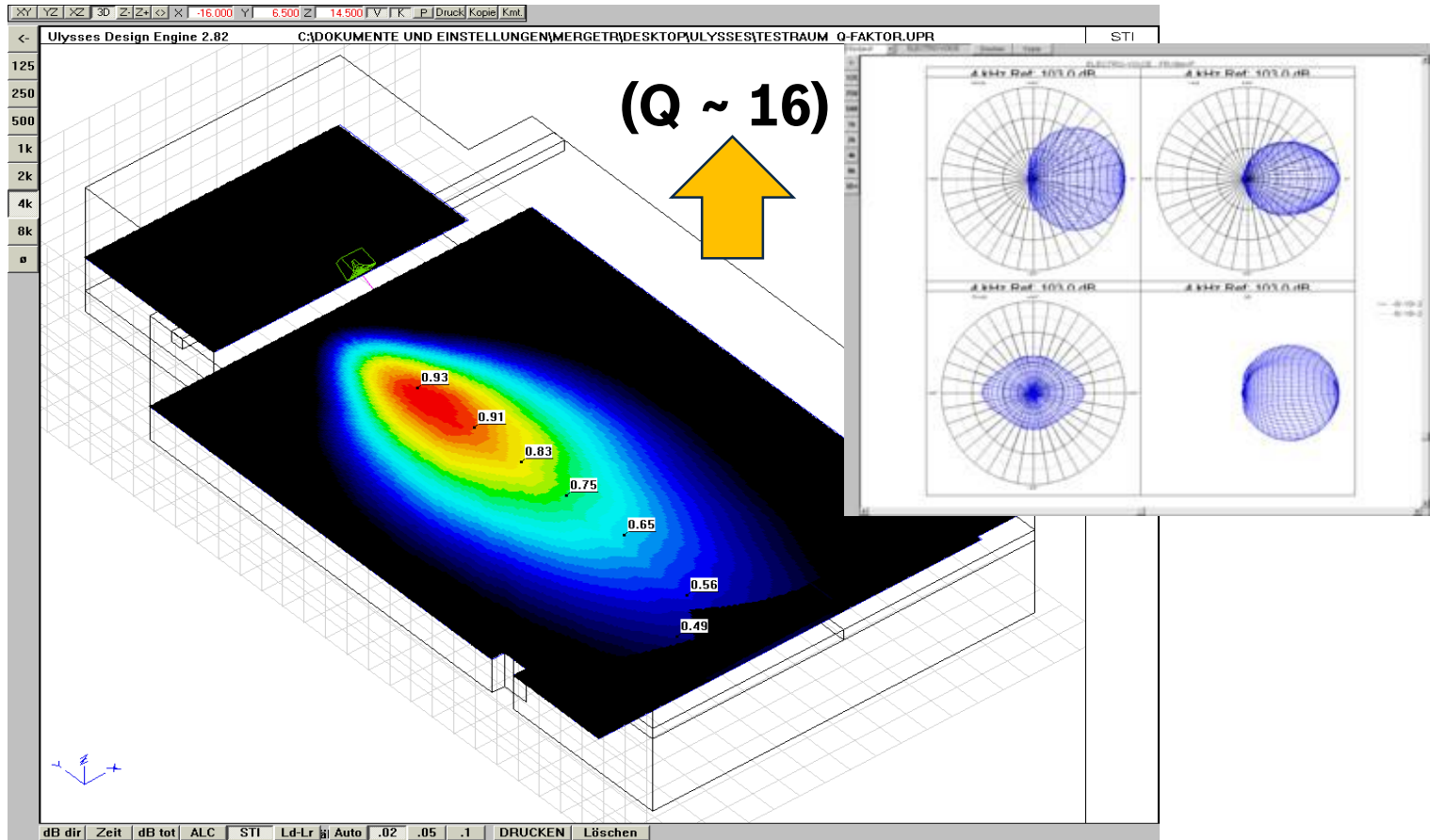
# Loudspeaker

## Directivity Q



# Loudspeaker

## Directivity Q





# Loudspeaker

## Power Handling

- There are several methods to specify the power handling of loudspeakers. Following two of the most important/common ones:

**AES2-1984** used for individual transducers (woofers and compression drivers). Uses filtered pink noise one decade apart, specific to the device under test. Two hour test procedure duration.

**EIA RS-426a** used for loudspeaker systems. Uses filtered white noise to resemble typical program content in music. Eight hour test procedure duration.

- But as mentioned there are some more...On the next page a comparison of different standards, just to show how different these methods can be.

# Loudspeaker

## Power Handling – Overview

### AES2-1984

#### *Audio Engineering Society*

- Pink Noise  
    BW: 1 Decade
- 6 dB Crest Factor
- 2 h

### EIA RS-426-A (1980)

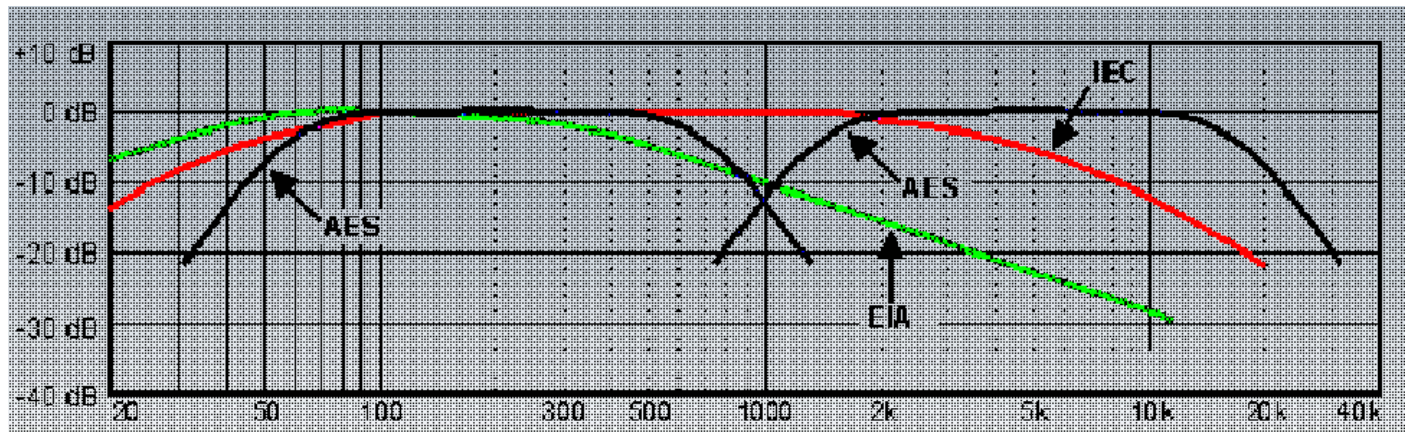
#### *Electronic Industries Association*

- Pink Noise  
    LP-Filter
- 6 dB Crest Factor
- 8 h

### IEC268-1 (1985)

#### *International Electro-technical Commission*

- Pink Noise  
    IEC-Filter (Music)
- 6 dB Crest Factor
- 100 h



# Loudspeaker

## Power Handling

→ Datasheet specifications:

Rated power RMS	400 Watts
Program power	800 Watts
Peak power	1600 Watts

→ Average Power or **RMS Power (long term)**

Calculation with RMS voltage

→ **Program Power (mid term)**

Commonly used as doubled average power

→ **Peak Power (short term)**

Calculation with peak voltage (e.g. 6 dB Crest Factor -> Peak = 4 x RMS)

Maximum SPL values of loudspeakers in datasheets are usually only calculated values with Peak power.

Max. SPL 1m (calculated with peak power)	131dB
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# Loudspeaker

## Crest Factor

- The Crest factor describes the ratio of peak vs. long term RMS value (root mean square) of a signal.

