

Sound technology

TNGD10 - Moving media

The hearing ability...

- ✦ 20-20000 Hz
 - 3000 & 4000 Hz
 - octave = doubling of the frequency
 - the frequency range of a CD?
- ✦ 0-120+ dB
 - the decibel scale is logarithmic
 - the dynamic range of a CD?



Digital sound

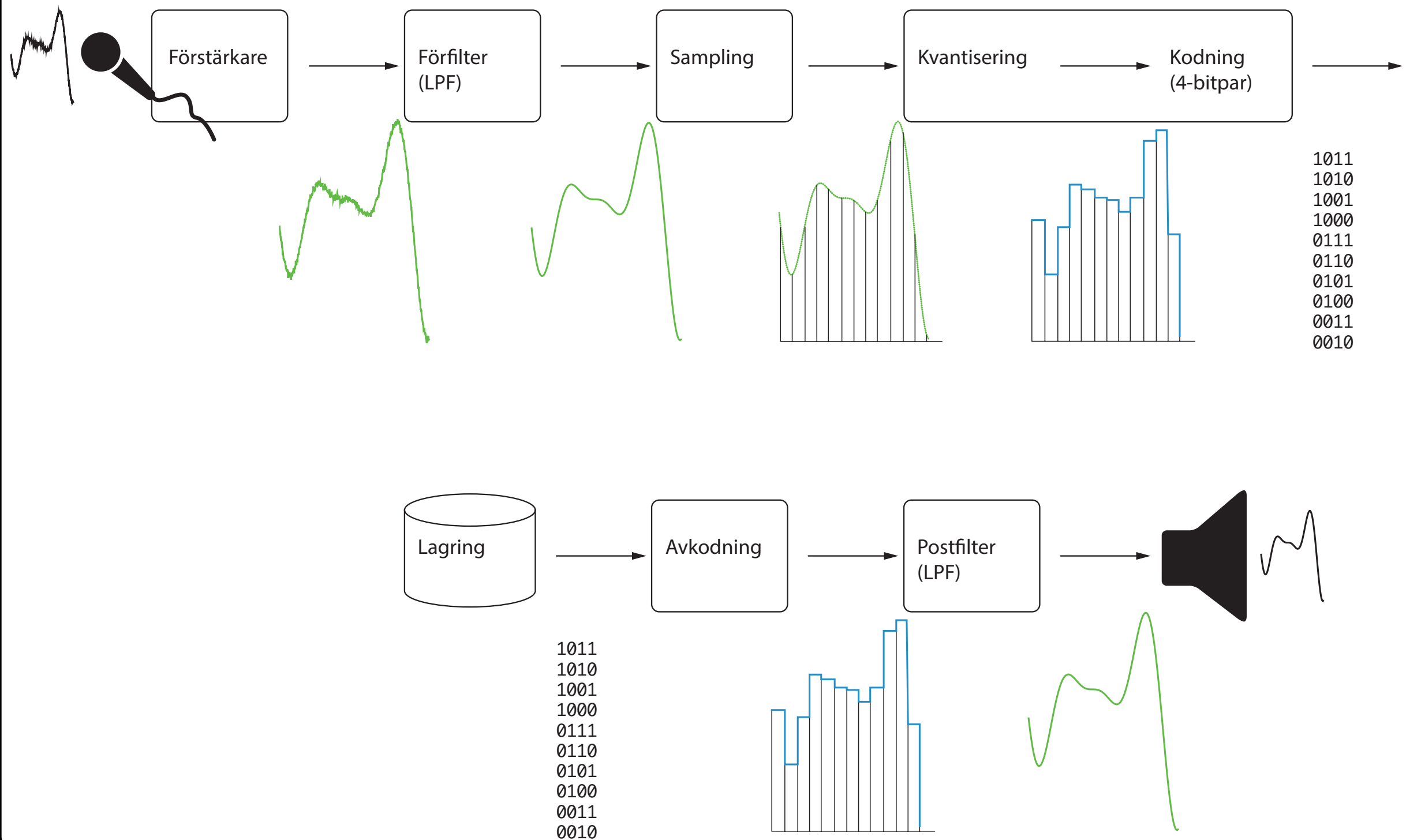
- ✧ Sound is pressure changes in the air.
- ✧ These are continuously changed over time.
- ✧ Thereby, sound is analogue.
- ✧ Analog \leftrightarrow digital.



AD/DA-converter

- ✦ A band limited signal might be sampled and reconstructed without any loss.
- ✦ The numeric values associated with the sampled value can be stored digitally.
- ✦ The sample will be held at the same level during a clock cycle during reconstruction.
- ✦ A low pass filter interpolates to recreate the original sound wave.

AD/DA-converter



Sampling

✧ The Nyquist theorem

- A sound (band limit time continuous signal), with bandwidth of f_0 Hz, that are sampled with the sample frequency of f_s can be reproduced from the sampled signal if $f_s > 2f_0$.

CD = 44.1kHz

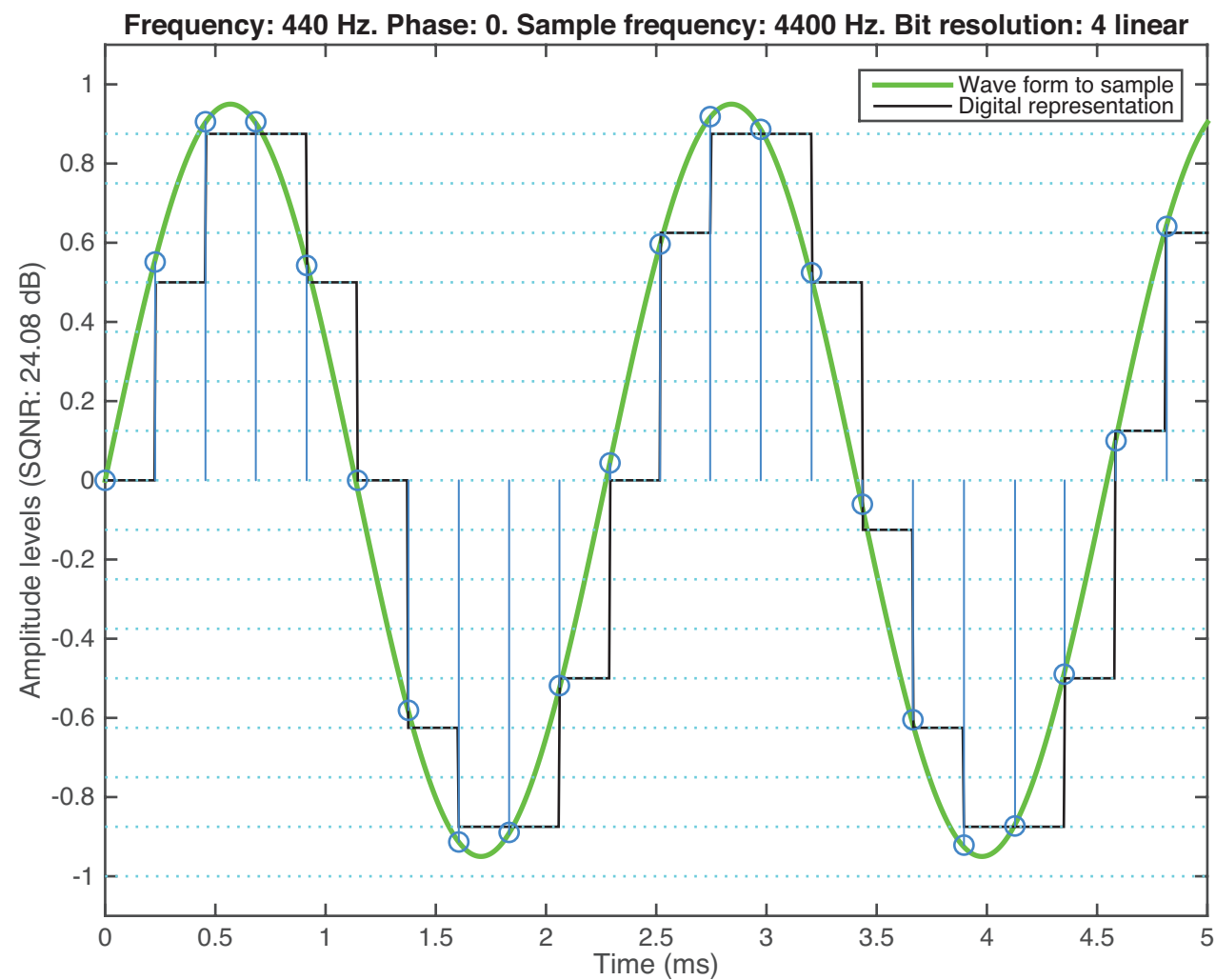
96kHz

DAT/DV = 48kHz

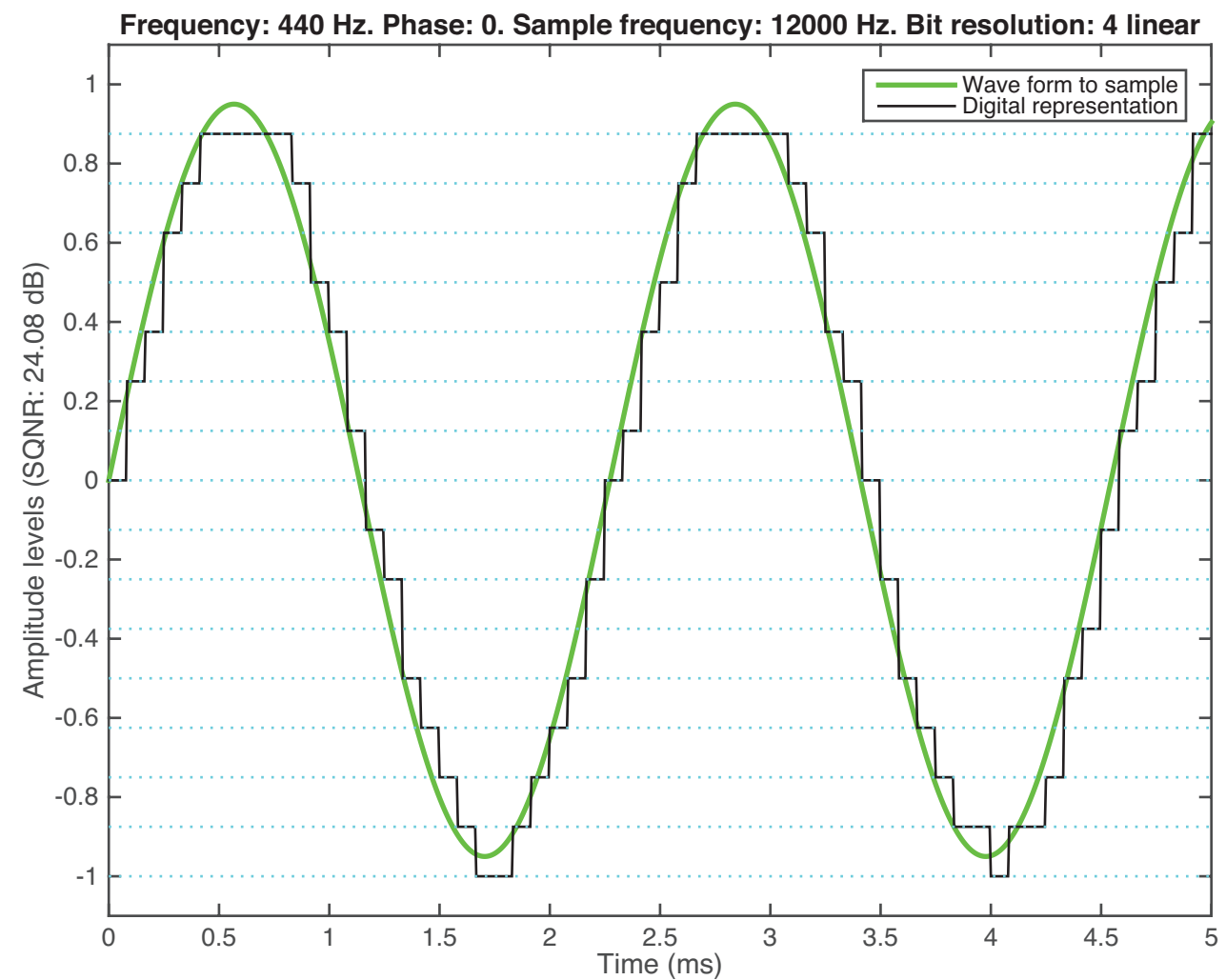
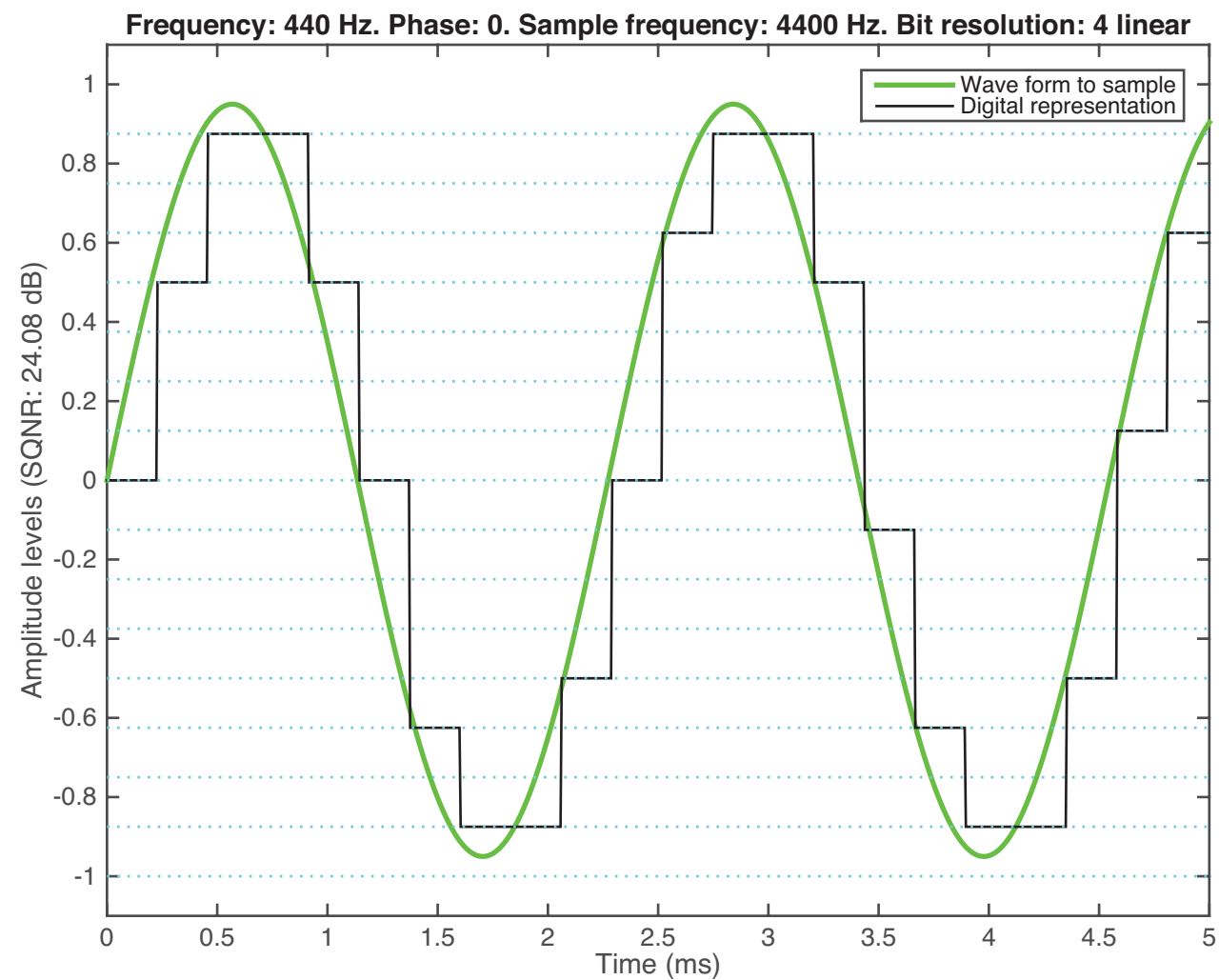
192kHz

DVDA & SACD

Sample frequency



Sample frequency



Quantising

- ✦ The sampled analogue value is converted in an AD converter to a digital value.
- ✦ The value is represented digitally with a restricted number of levels, these levels are restricted due to the number of bits used.
 - The number of intervals is 2^n , where n is the number of bits.

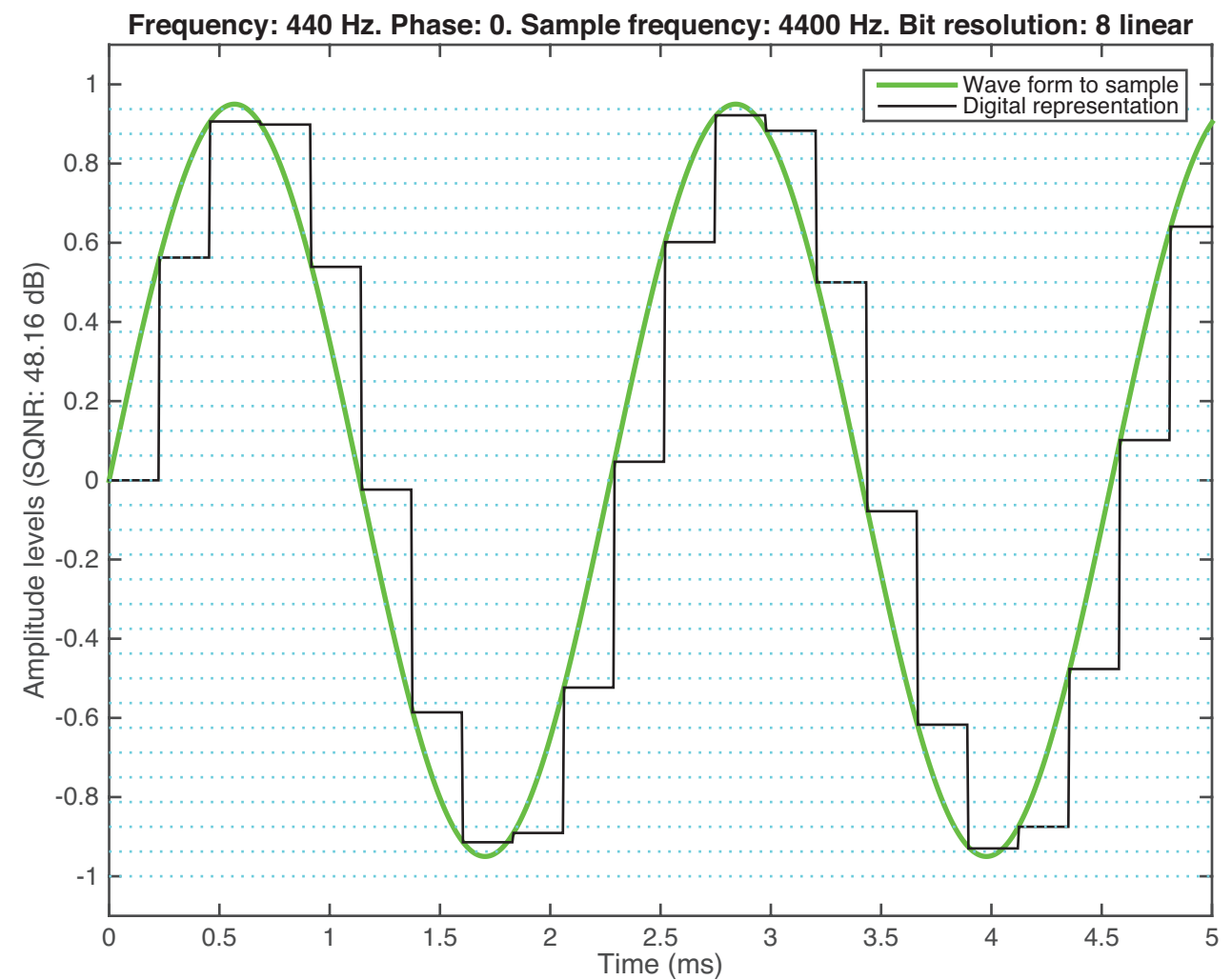
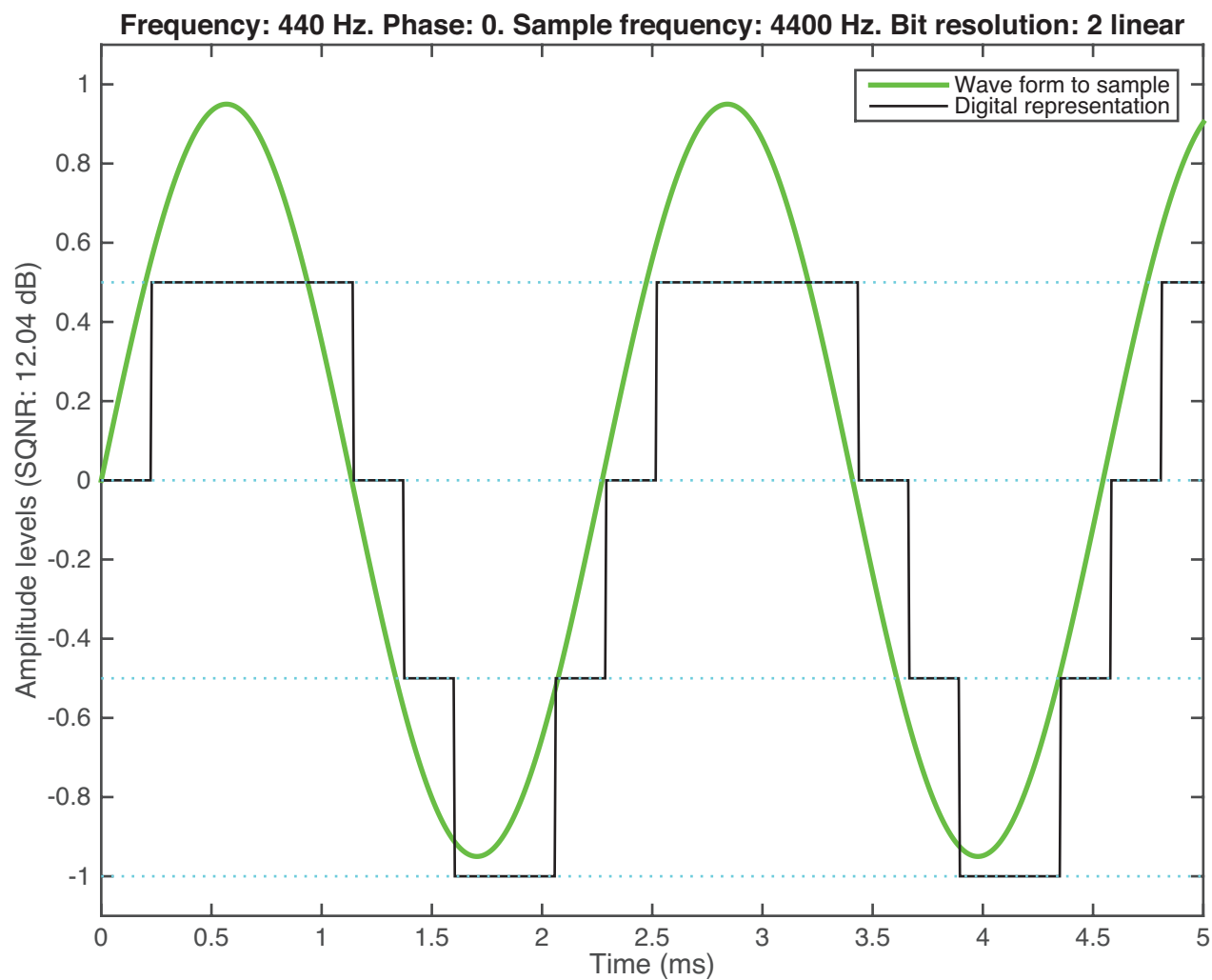
$$2^4 = 16$$

$$2^{12} = 4096$$

$$2^{16} = 65536$$

$$2^{24} = 16777216$$

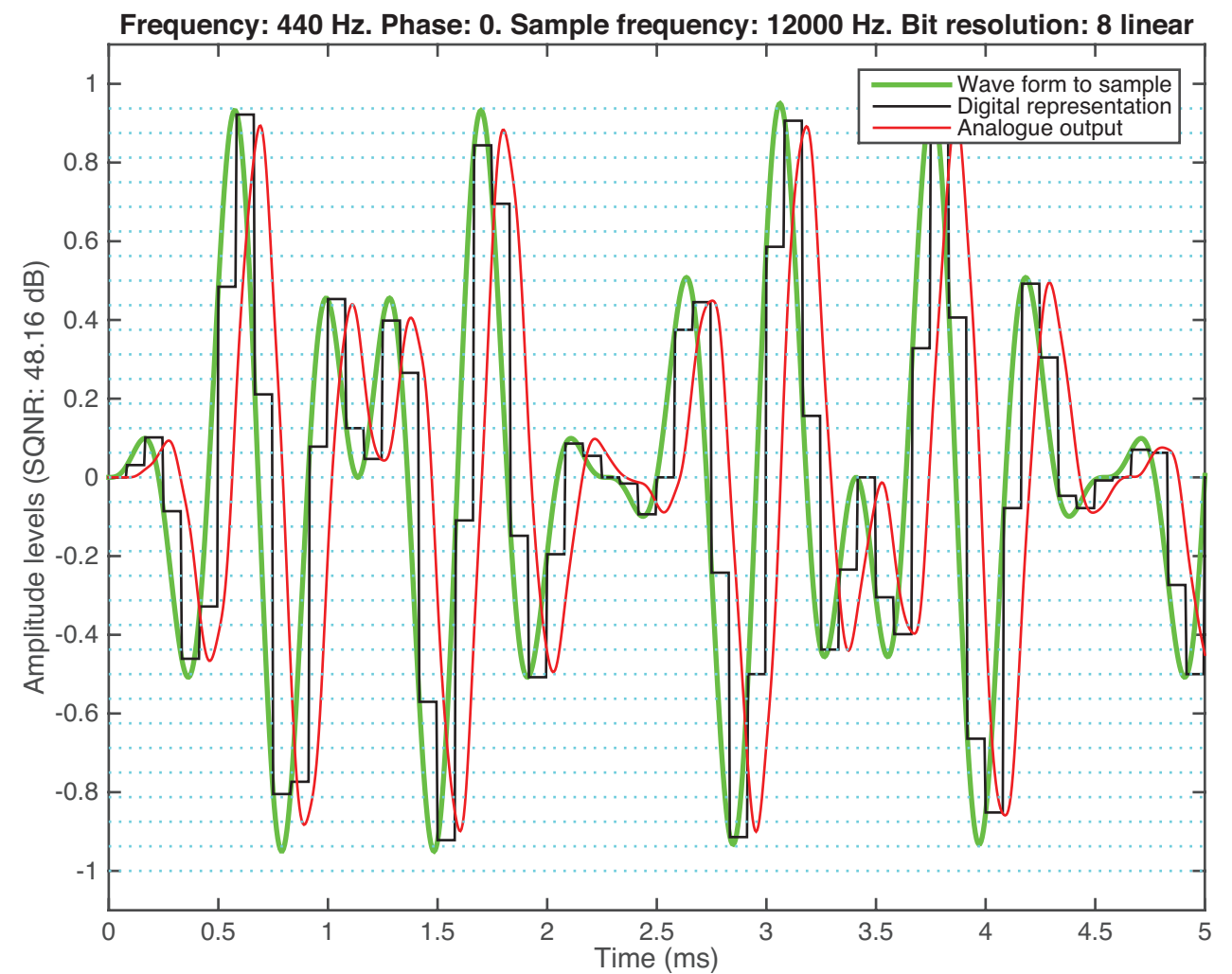
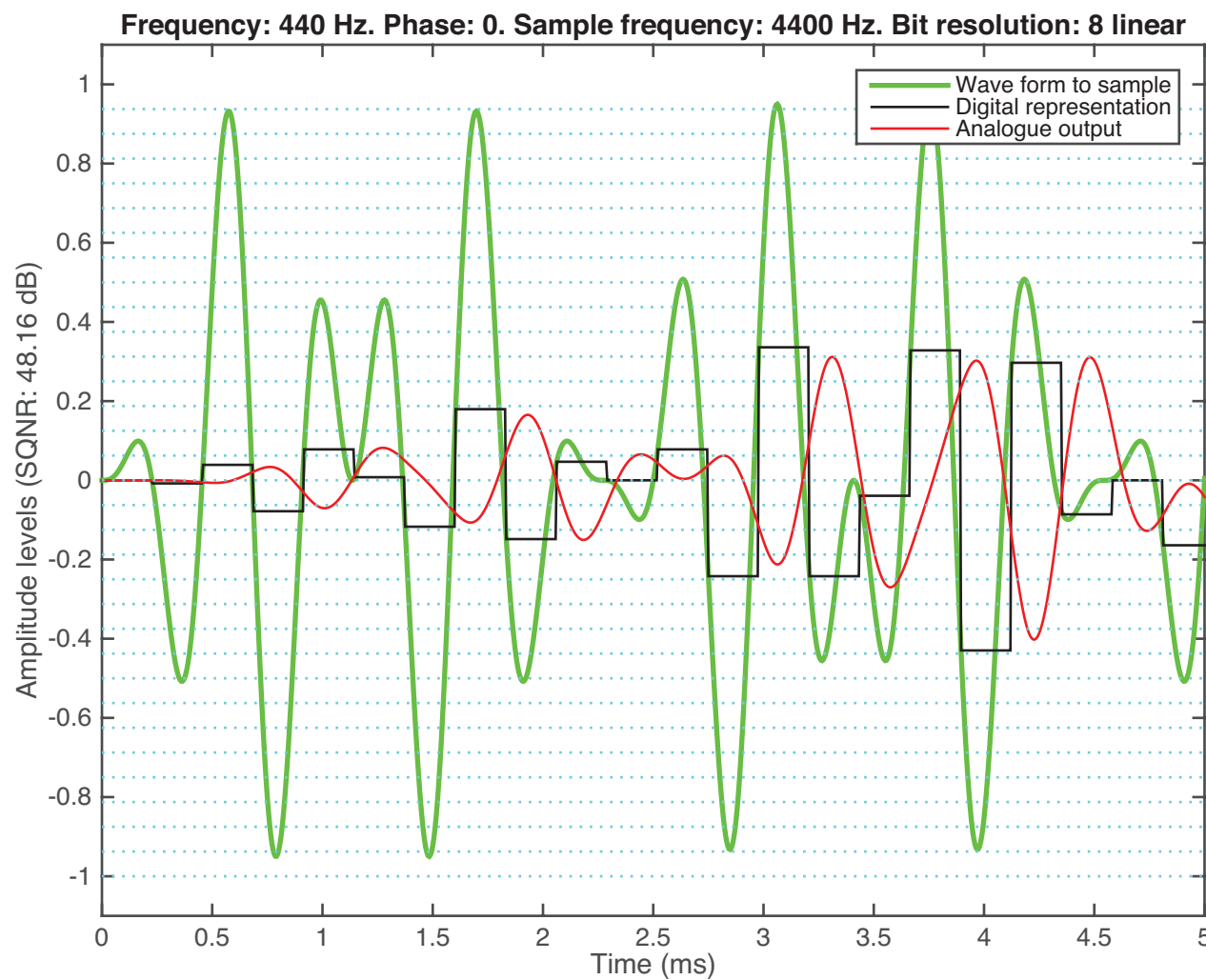
Quantising (cont.)



Dynamic range

- ✦ When recording a higher number of bits is best with regards to dynamic range and head-room.
- ✦ Dynamic range is the difference in amplitude between the loudest sample and quiet, or
- ✦ Signal-to-error ratio
- ✦ 8-bit = 49.8dB
- ✦ 16-bit = 97.8dB
- ✦ 24-bit = 145.8dB

Digital sound representation



The microphone

TNGD10 - Moving media

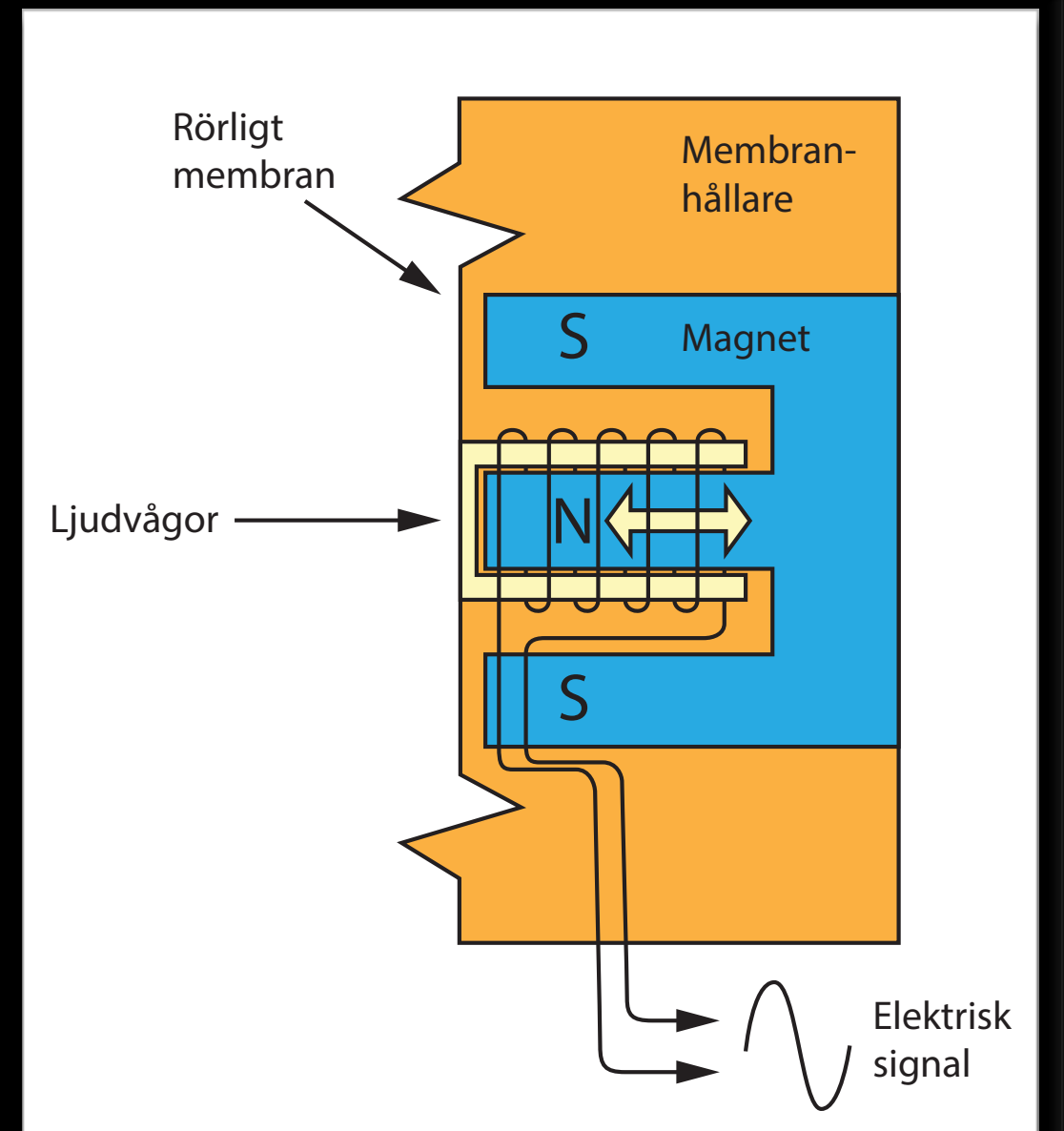
The microphone

- ✦ The funnel...
- ✦ The microphone was invented 1925
- ✦ Chrystal microphone
- ✦ Band microphone
- ✦ Dynamic microphone
- ✦ Condenser microphone (electret microphone)



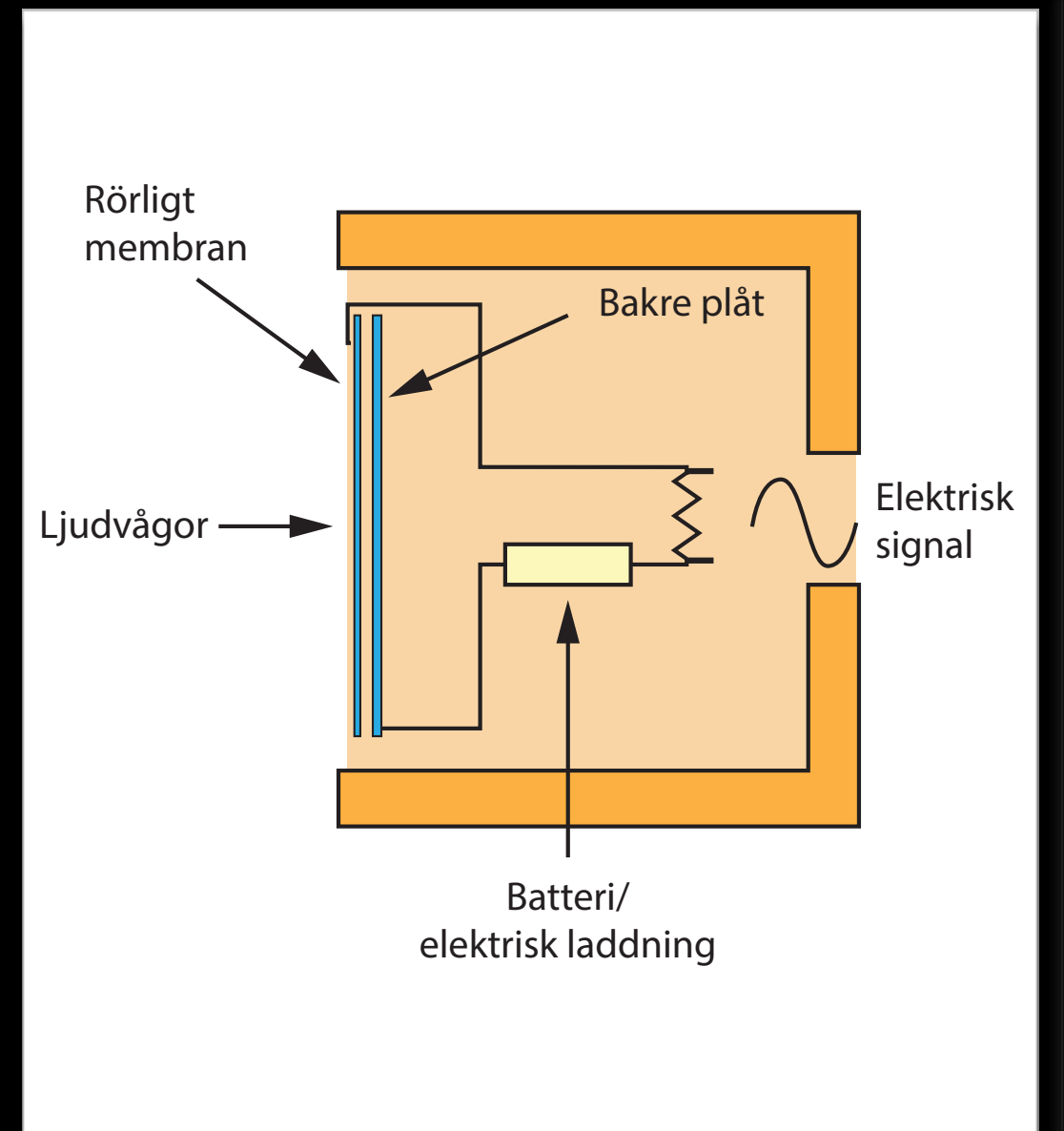
Dynamic microphone

- ✦ Withstand high sound pressures
- ✦ “Inherit” slowness
- ✦ Drums, amplifier, vocals
- ✦ Membrane
- ✦ Coil
- ✦ Magnet



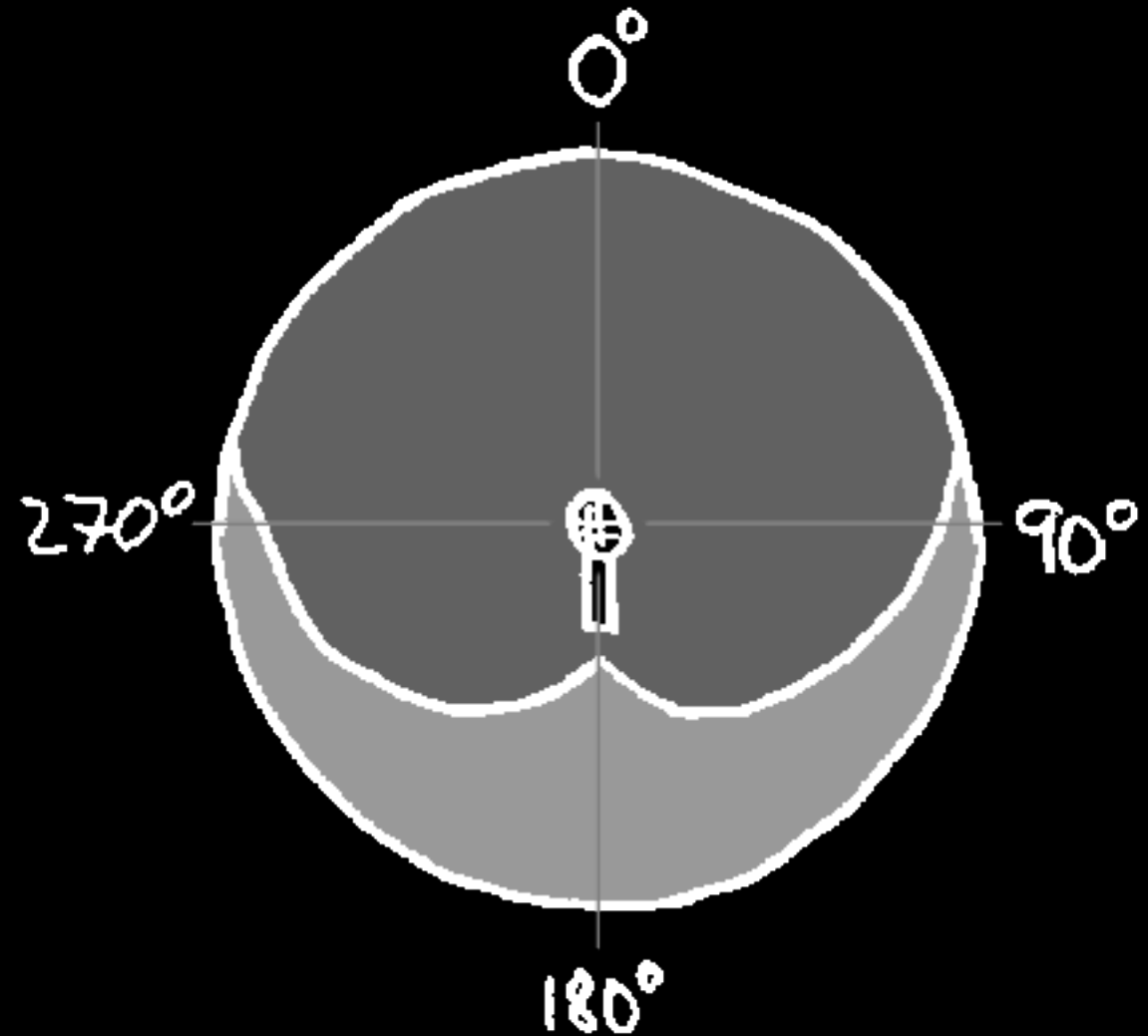
Condenser microphone

- ✦ Higher sensitivity
- ✦ Requires phantom power
- ✦ Vocals, acoustical instruments
- ✦ Membrane
- ✦ Fixed electrode



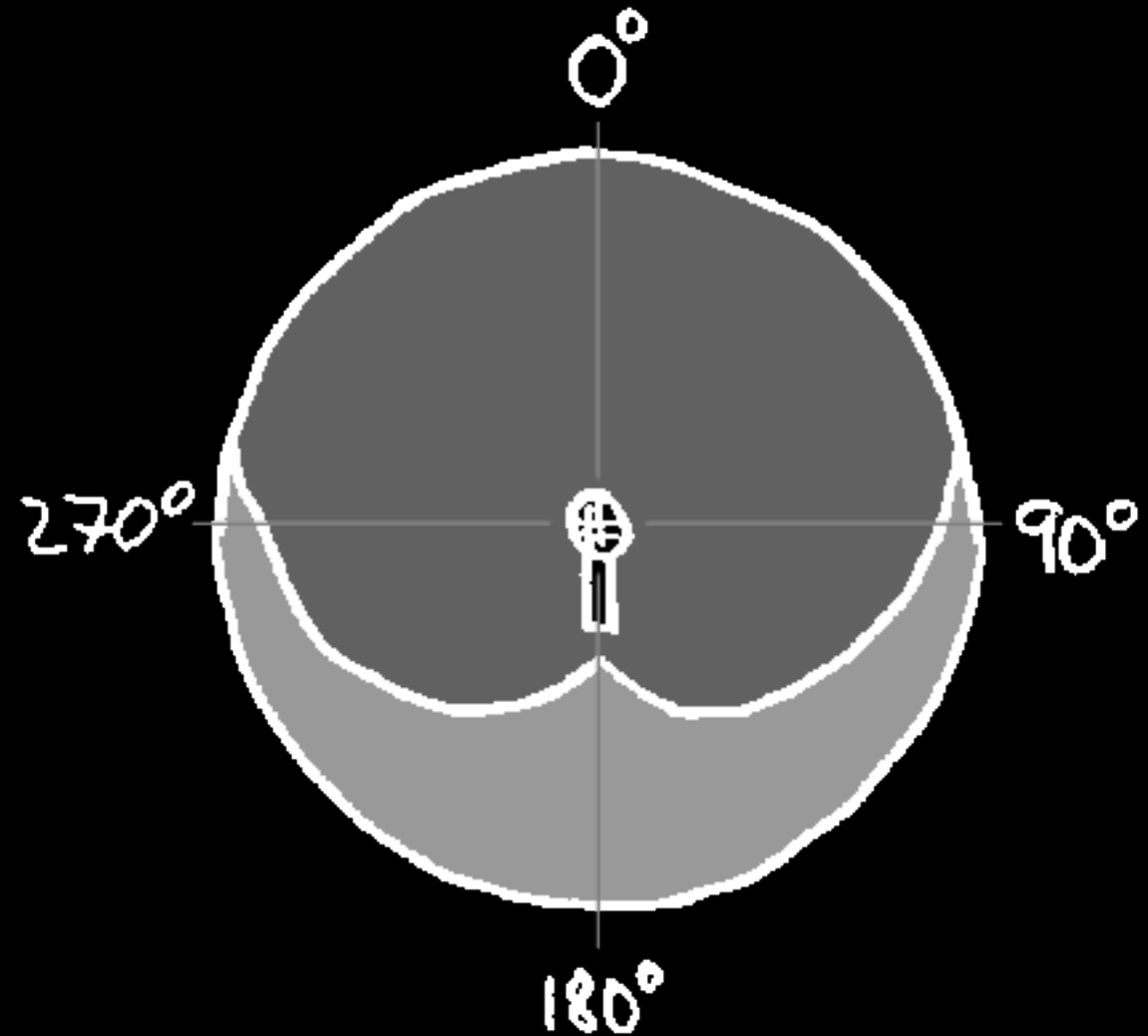
Directional microphone

- ✧ Velocity microphone/
pressure gradient
microphone
- ✧ Proximity effect
- ✧ Cardioid
- ✧ Super cardioid
- ✧ Hyper cardioid



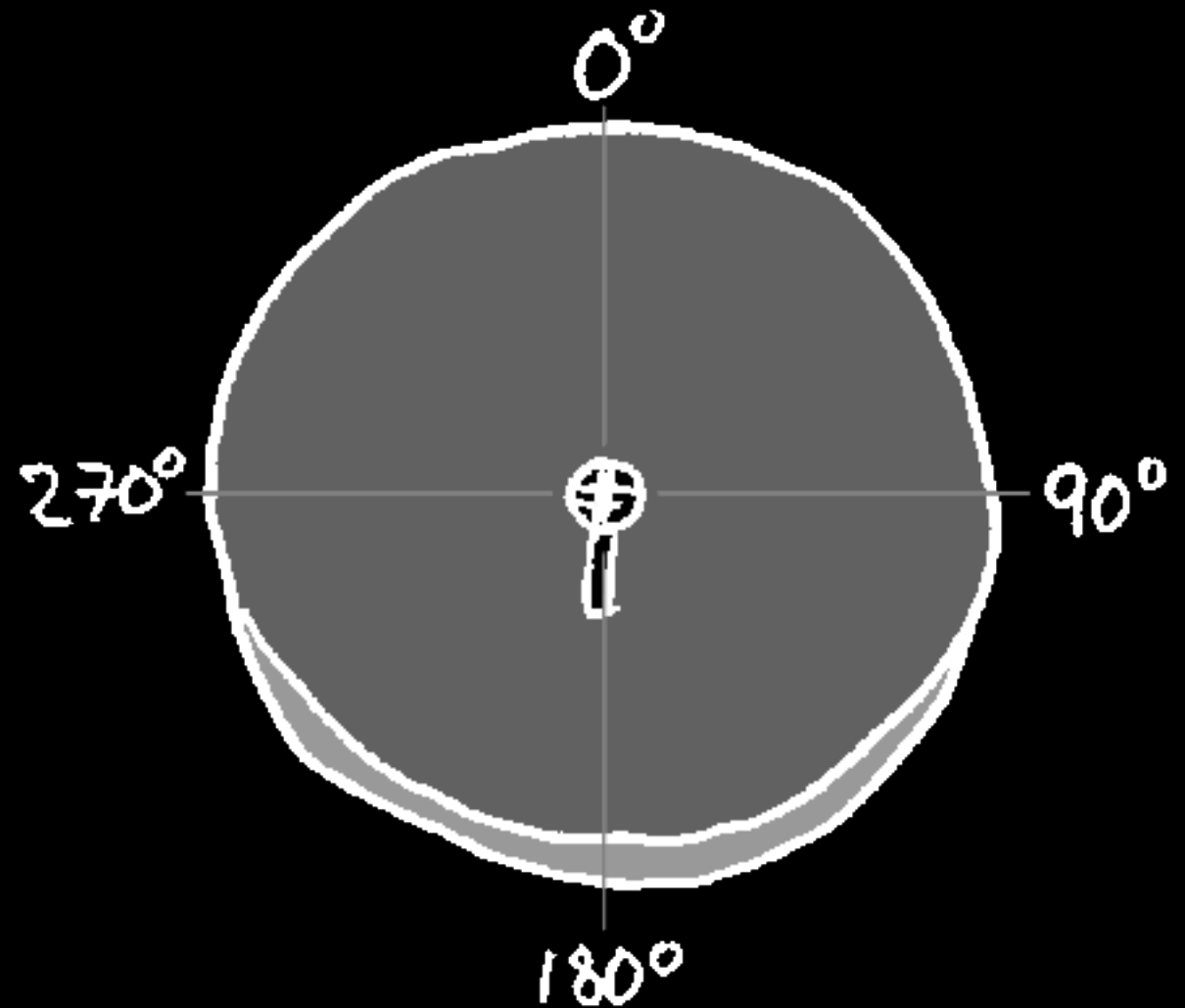
Directional microphone (cont.)

- ✦ Vocal and instrument microphone
- ✦ AB-stereo, ORTF
- ✦ XY-stereo
- ✦ (MS-stereo)



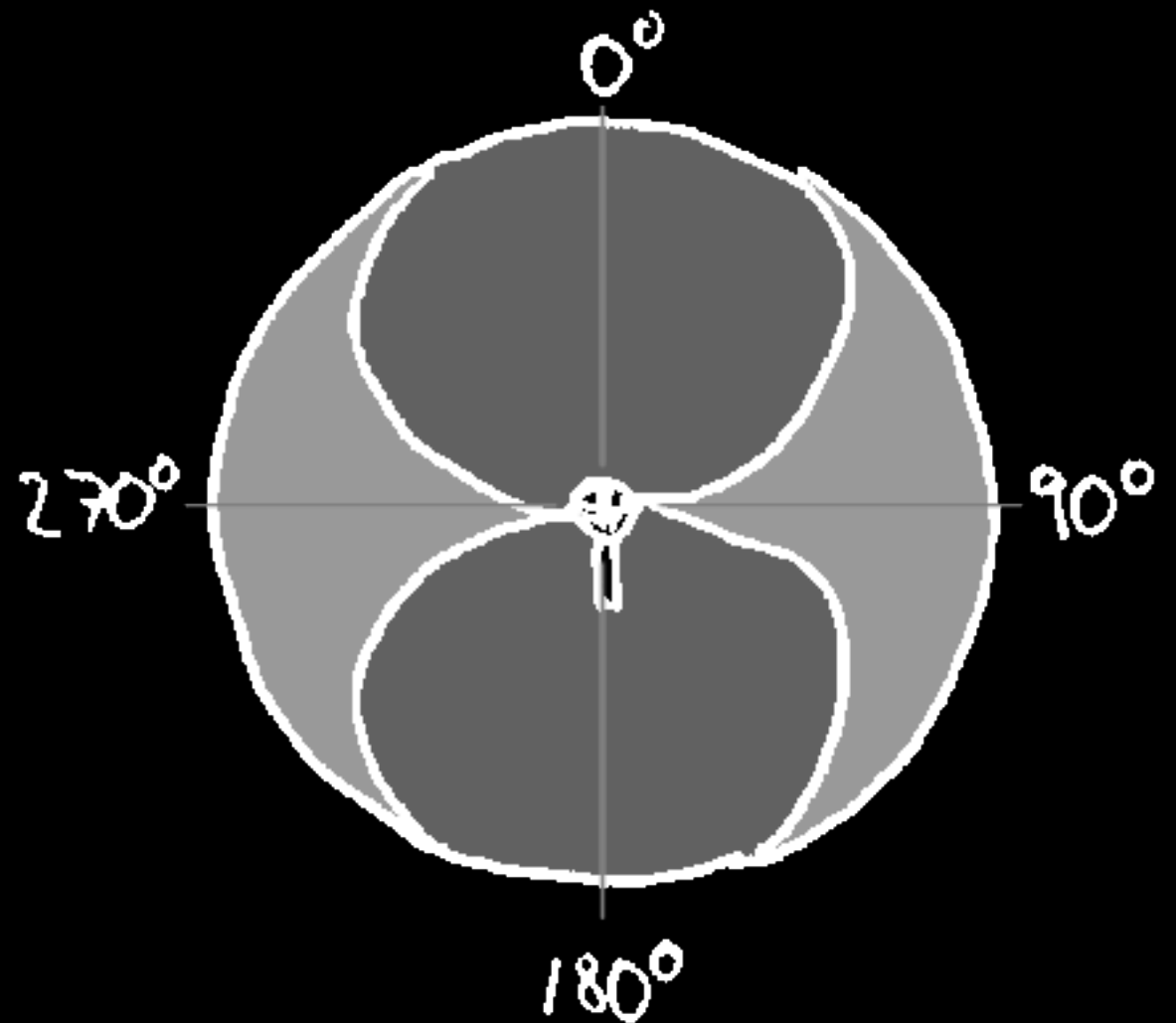
Omnidirectional microphone

- ✦ Pressure microphone
- ✦ No proximity effect
- ✦ Linear low frequency
- ✦ Ambience microphone
- ✦ MS-stereo



Bidirectional microphone

- ✧ Velocity microphone
- ✧ Proximity effect
- ✧ Reversed polarity
- ✧ Blumlein stereo
- ✧ MS-stereo



Microphones (cont.)

- ✧ - High pass filter
 - Attenuation, pad
 - Selectable patterns
- ✧ Balanced XLR
- ✧ Puff protection
 - Socks...
 - Windshield
 - Pens
- ✧ Phantom power, +48V



Sensitive signal

- ✦ Use balanced XLR.
- ✦ Be aware of parallel 230V cables.
- ✦ Avoid old computer & TV screens, fluorescent lamps, mobile phones...
- ✦ Ground hum (50Hz), use different outlets.



The preamplifier

- ✧ Amplifies the signal to line level.
- ✧ High pass filter
- ✧ Phantom power
- ✧ Change polarity



Microphone placement

TNGD10 - Moving media

Stereo

- ✦ Intensity stereo/coincident stereo
- ✦ The intensity/sound level varies with the direction of the sound.
- ✦ The sound is in phase, no comb filter effect when summed to mono.
- ✦ X/Y, Blumlein, MS



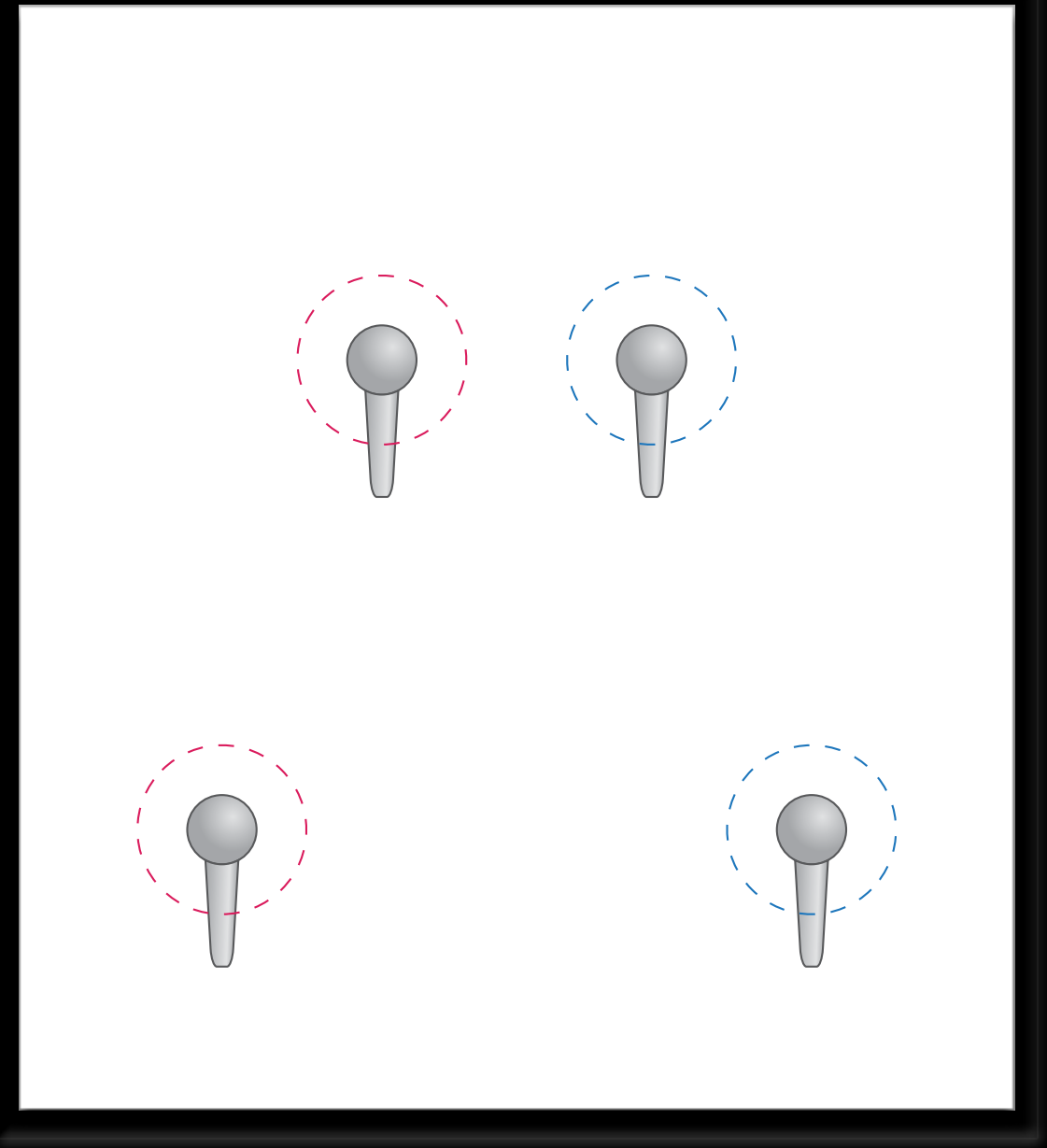
Stereo (cont.)

- ✧ Near coincident stereo/
spaced pair/ear spaced
stereo
- ✧ Phase difference
- ✧ A slightly more spacey
stereo
- ✧ Not as mono compatible
- ✧ ORTF, AB



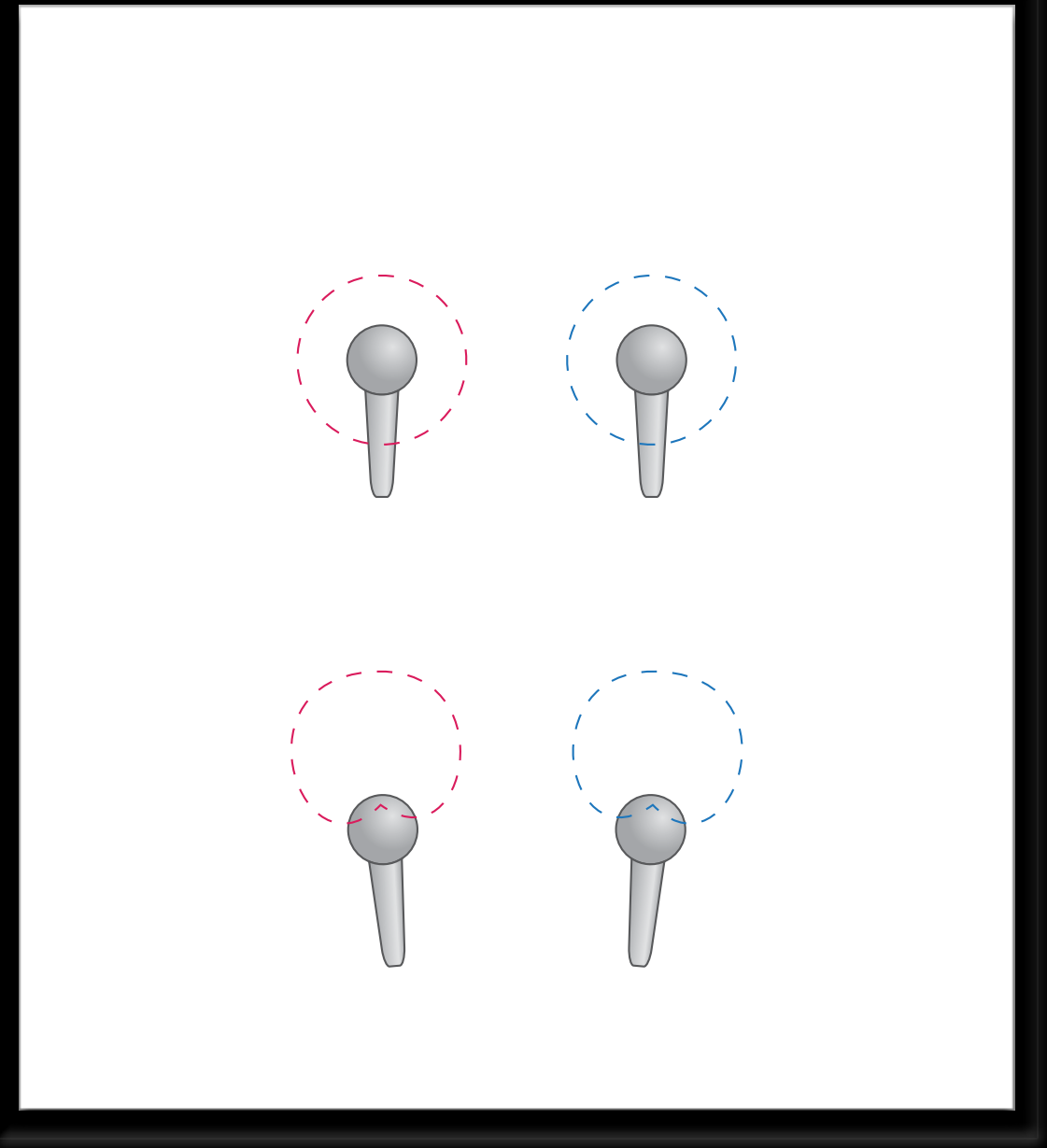
Stereo (cont.)

- ✧ The quality on the stereo recording is due to three factors:
- ✧ The distance



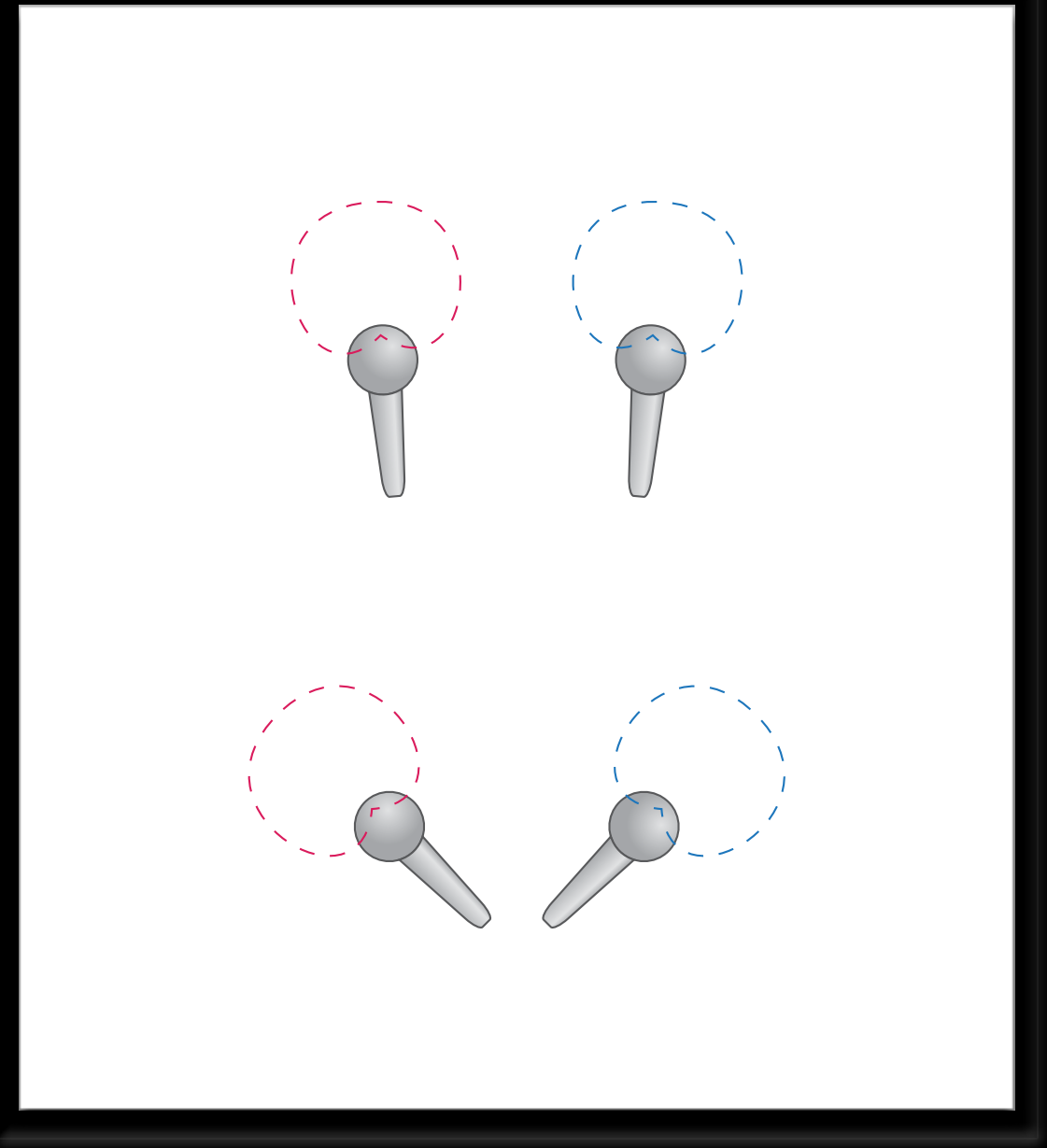
Stereo (cont.)

- ✧ The quality on the stereo recording is due to three factors:
- ✧ The distance
- ✧ The pattern



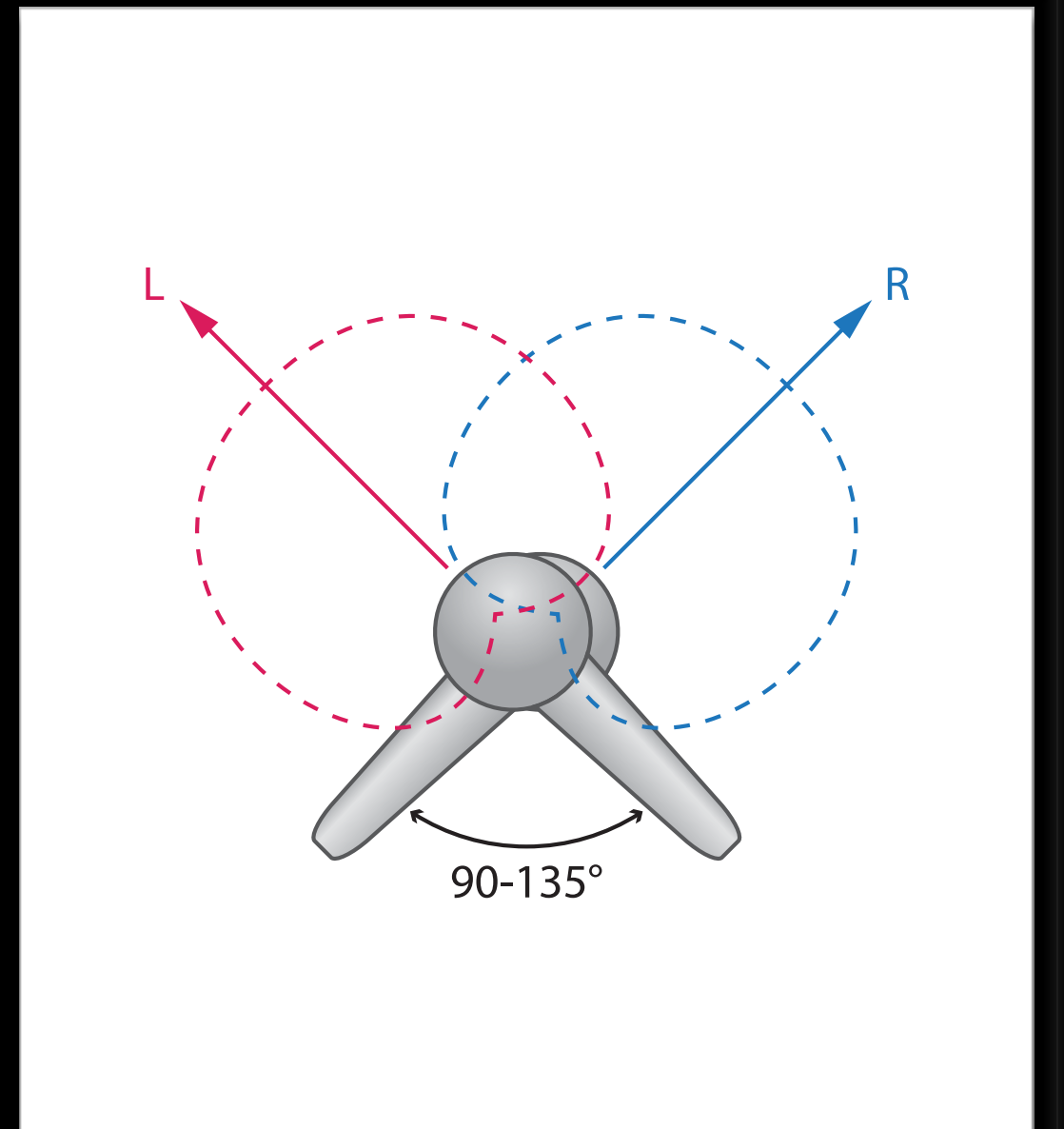
Stereo (cont.)

- ✧ The quality on the stereo recording is due to three factors:
- ✧ The distance
- ✧ The pattern
- ✧ The spacing angle



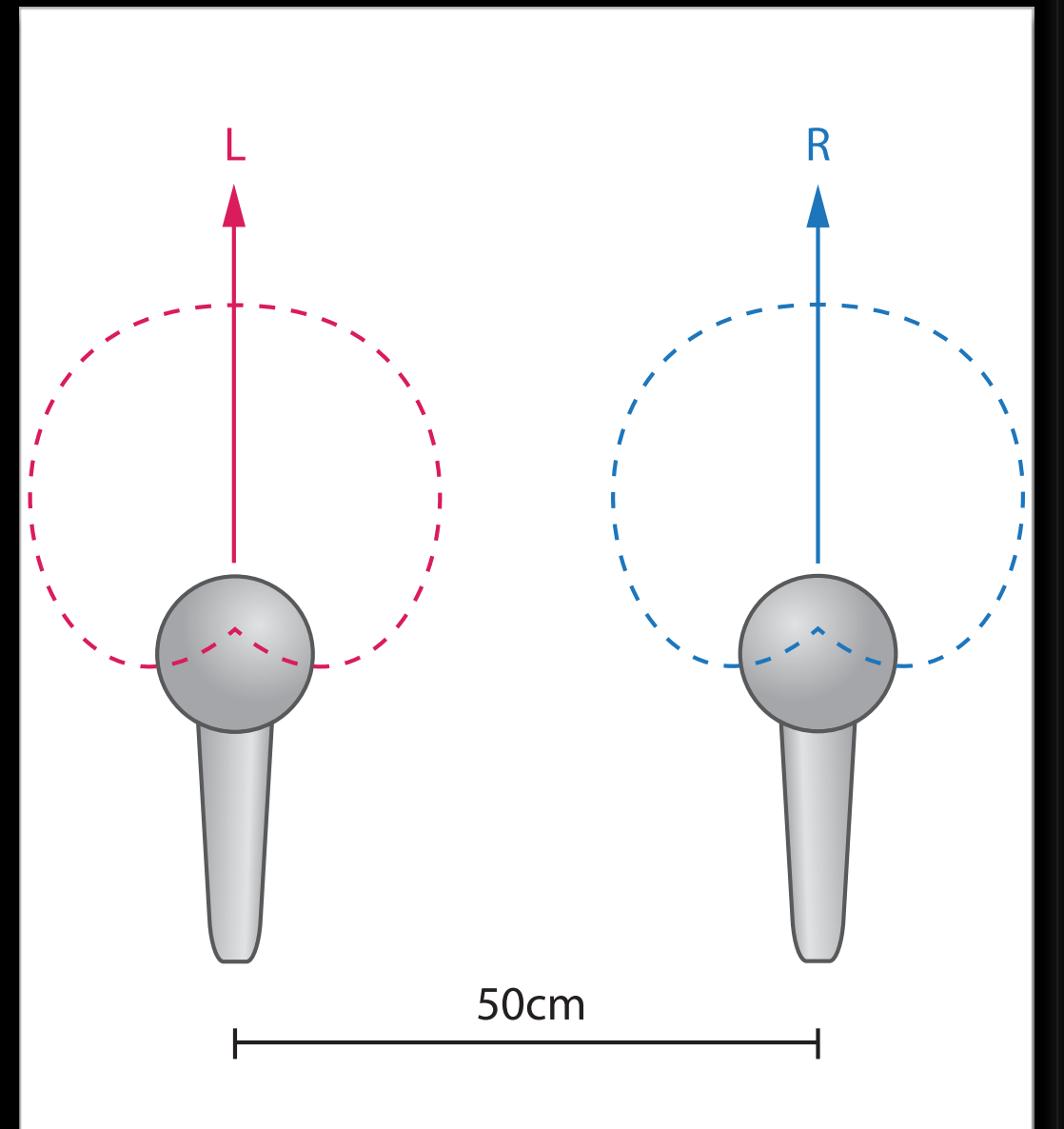
XY-stereo

- ✧ Two directional microphones (cardioid, super, or hyper)
- ✧ 90° - 135°
- ✧ A too tight angle creates a rather poor stereo image.
- ✧ Good mono compatibility.



AB-stereo

- ✦ Two omni or directional microphones.
- ✦ Omni directional records more of the room acoustics.
- ✦ The distance between the microphone 40-100cm.
- ✦ A too wide separation creates a "hole" in the middle.
- ✦ Less good mono compatibility.



Recording for video

TNGD10 - Moving media

Recording

- ✧ Always keep as short distance between the sound source and the microphone
- ✧ Make sure that the stage/room is quiet.
- ✧ Use stands or boom
- ✧ Use a windshield



Recording (cont.)

- ✧ Take care of the cable
- ✧ Use gaffa
- ✧ Avoid microphonics
- ✧ Use a clapper board
- ✧ Write down recording list



Zoom H5

- ✦ Hand held audio recording device
- ✦ Two channels microphone input
- ✦ Built in XY stereo
- ✦ XLR + phantom power for external microphones
- ✦ USB connection



Recording sound

- ✦ 48kHz 16-bit
- ✦ Remove the “built in” microphones
- ✦ Choose the right input
- ✦ Check the input sound level
- ✦ Write down audio file name/time



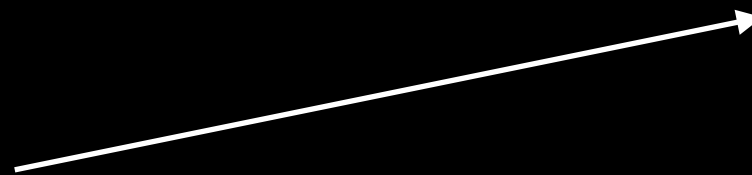
Before using H5

- ✦ Buy batteries
- ✦ Make sure to have the XLR cable and microphones
- ✦ Microphone stands
- ✦ USB-cable and headphones
- ✦ And the memory card!



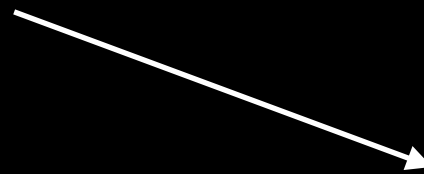
Using H5

- ✦ Remove built in microphones



Using H5

- ✦ Remove built in microphones
- ✦ Connect headphones



Using H5

- ✦ Remove built in microphones
- ✦ Connect headphones
- ✦ Connect XLR cable (use left channel if mono)



Using H5

- ✦ Remove built in microphones
- ✦ Connect headphones
- ✦ Connect XLR cable (use left channel if mono)
- ✦ Start the H5



Using H5

- ✦ Remove built in microphones
- ✦ Connect headphones
- ✦ Connect XLR cable (use left channel if mono)
- ✦ Start the H5
- ✦ Set Stereo recording mode with Selector and Menu



Using H5

- ✦ Select the input channels



Using H5

- ✦ Select the input channels
- ✦ Set the input level



Using H5

- ✦ Select the input channels
- ✦ Set the input level
- ✦ Check the input level



Using H5

- ✦ Select the input channels
- ✦ Set the input level
- ✦ Check the input level
- ✦ Start recording



Using H5

- ✦ Select the input channels
- ✦ Set the input level
- ✦ Check the input level
- ✦ Start recording
- ✦ Pause/resume recording



Using H5

- ✦ Select the input channels
- ✦ Set the input level
- ✦ Check the input level
- ✦ Start recording
- ✦ Pause/resume recording
- ✦ Stop recording



Using H5

- ✦ Select the input channels
- ✦ Set the input level
- ✦ Check the input level
- ✦ Start recording
- ✦ Pause/resume recording
- ✦ Stop recording
- ✦ Headphone sound level



Using H5

- ✦ Select REC in the menu
- ✦ Select REC FORMAT
- ✦ Set format, 48kHz/16bit
- ✦ In IN/OUT
 - ✦ Set LOW CUT 80 Hz
 - ✦ Set COMP/LIMITER to LIMITER(GENERAL)
 - ✦ Set In1/2PHANTOM to +48V
- ✦ Set for INPUT 1/2



Using H5

- ✦ Play back



Using H5

- ✦ Play back
- ✦ Selecting recordings



Using H5

- ✦ Play back
- ✦ Selecting recordings
- ✦ Connect to computer via USB



Using H5

https://www.zoom-na.com/sites/default/files/products/downloads/pdfs/E_H5_0.pdf

Sound in Adobe

TNGD10 - Moving media

Sound in Adobe

- ✦ Audition - sound editing
- ✦ Dynamic processing, filters & EQ, room effects
- ✦ Dynamic link workflow between Premiere and Audition
 - In Premiere
 - Edit -> Edit in Adobe Audition -> Sequence
 - In Audition
 - Export to Adobe Premiere Pro
- ✦ Premiere
 - keyframes and automation
 - sound level
 - effects

Dynamics

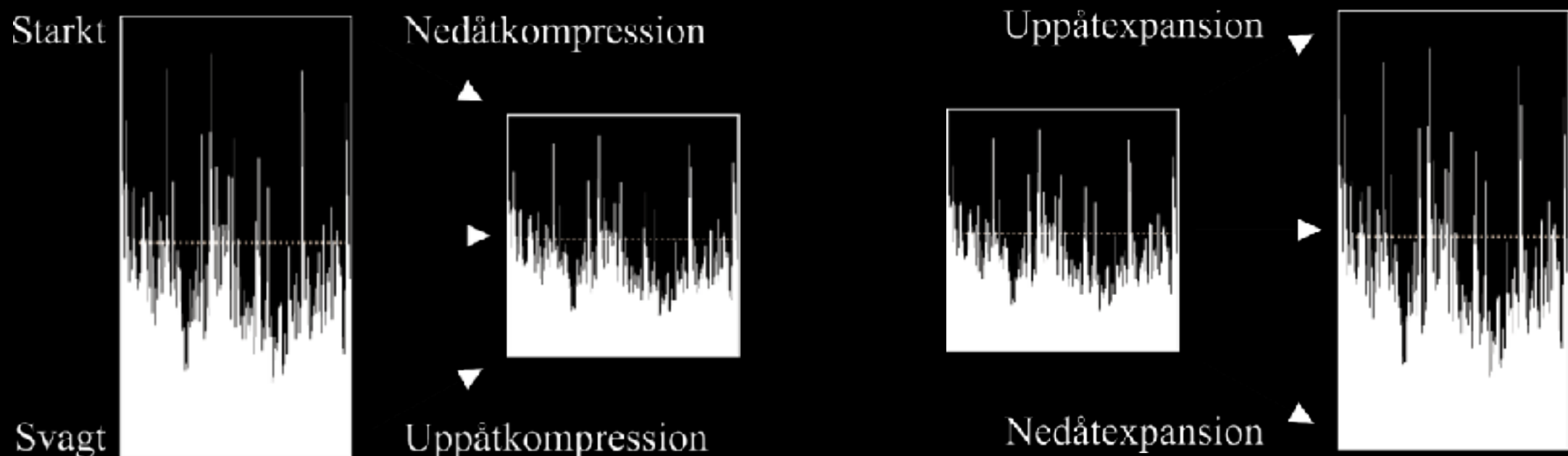
TNGD10 - Moving media

Dynamics

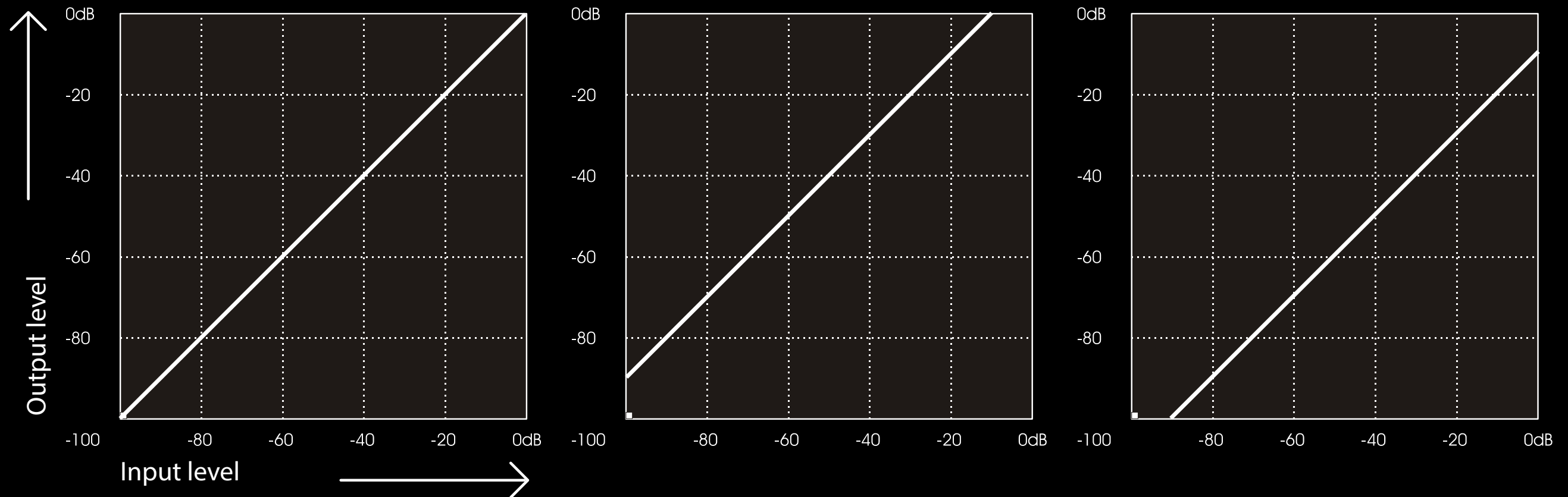
- ✧ Dynamics/dynamic range is equal to difference.
- ✧ Modern pop music has a dynamic range of about 9dB.
- ✧ The CD format, 44.1kHz and 16 bits can theoretically give 96dB in dynamic range.
- ✧ Macro and micro dynamics

Upwards or downwards...

- ✧ Downward compression - the normal
- ✧ Upward compression - for noise reduction
- ✧ Downward expansion - the normal
- ✧ Upward expansion - for noise reduction

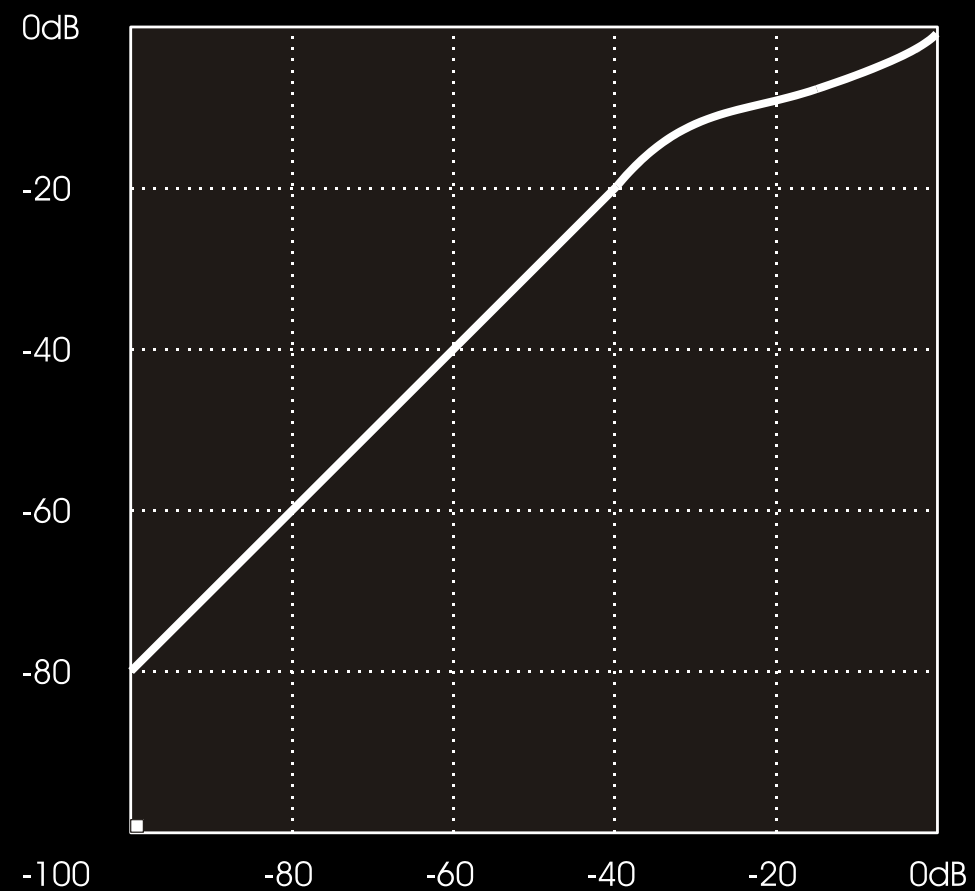
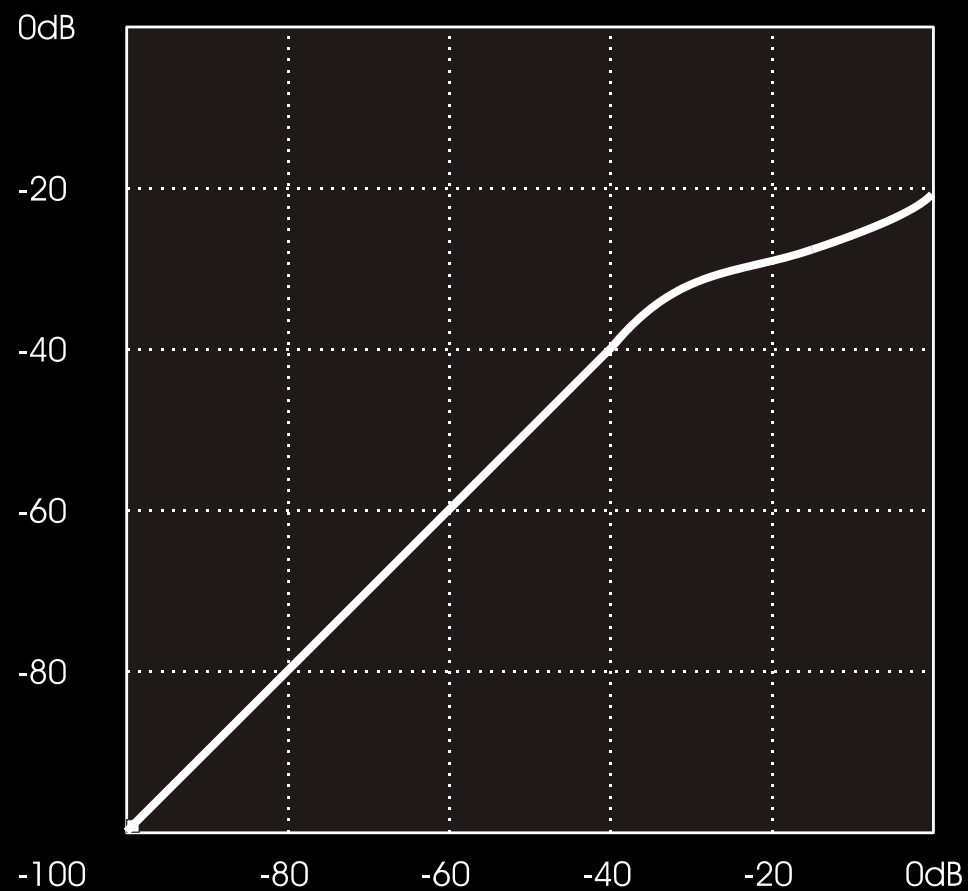


Transfer curves



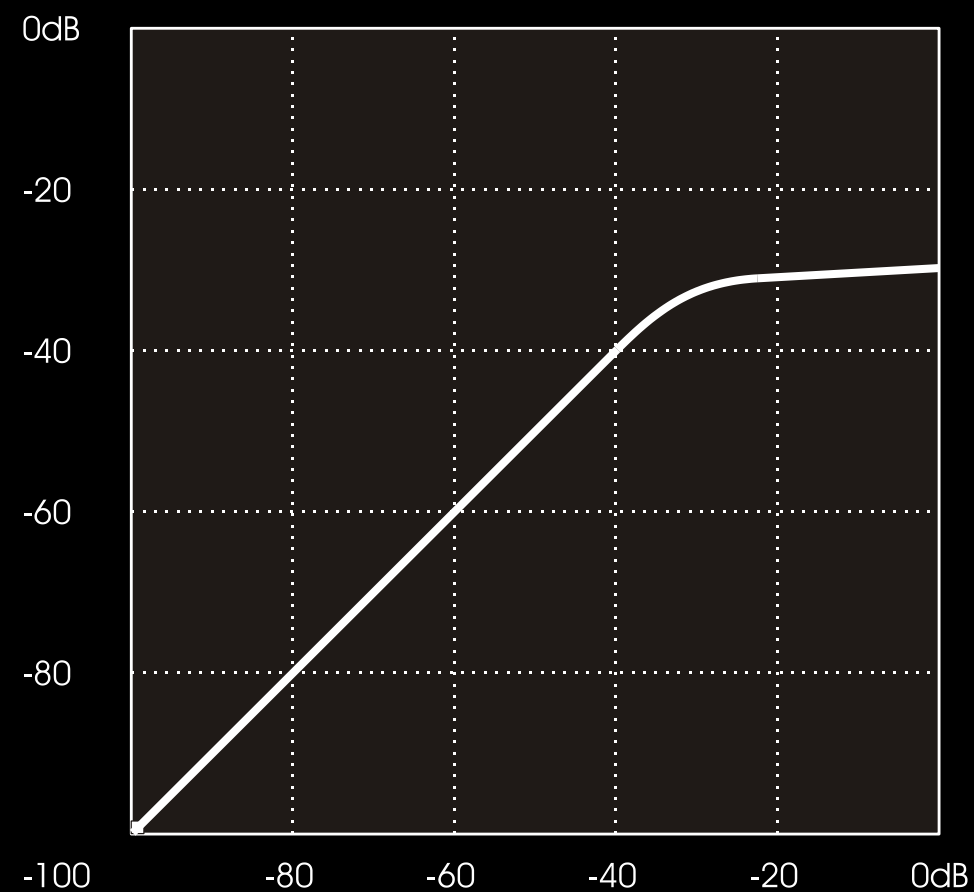
Compression

- ✧ Threshold - when the compression starts
- ✧ Compression ratio - the amount of compression



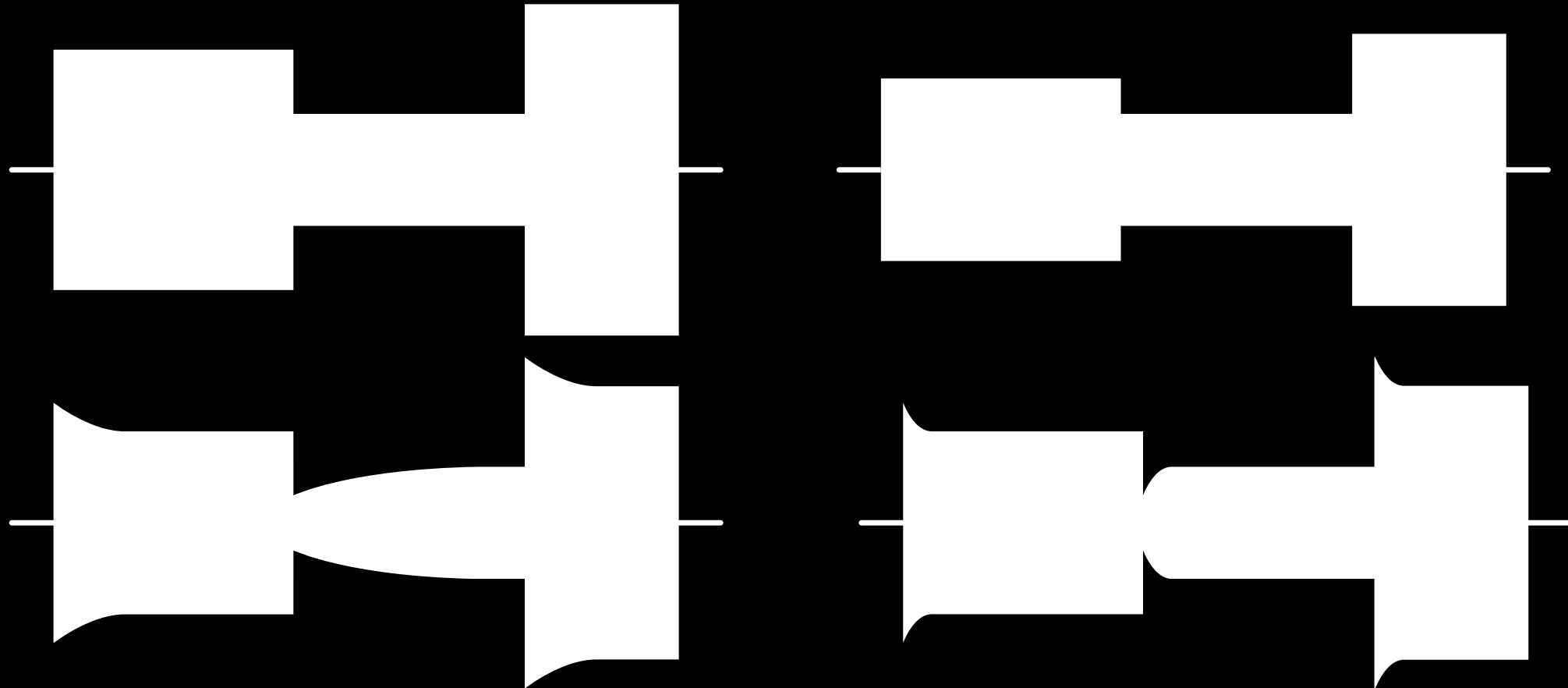
Compression (cont.)

- ✧ Limiting around ratios of 10:1 or more (∞ :1)



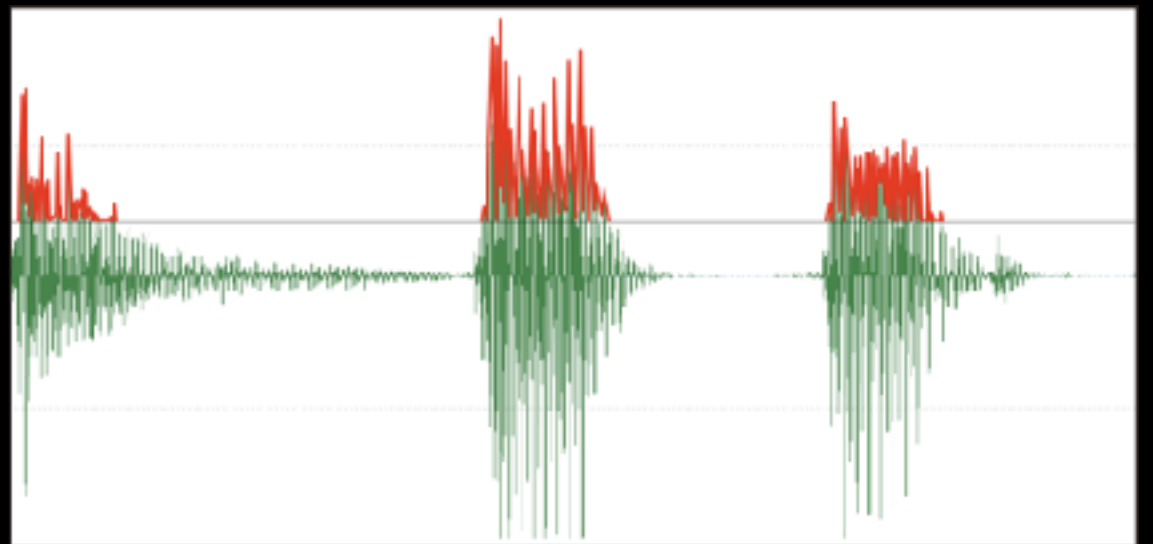
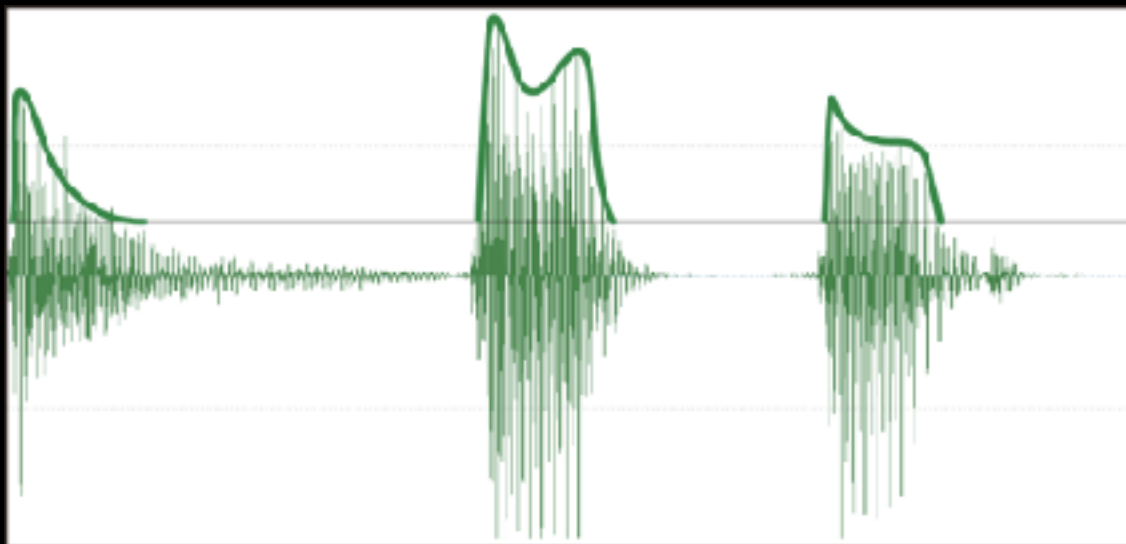
Time settings

- ✧ Attack - how fast attenuation will start
- ✧ Release - how fast attenuation will stop



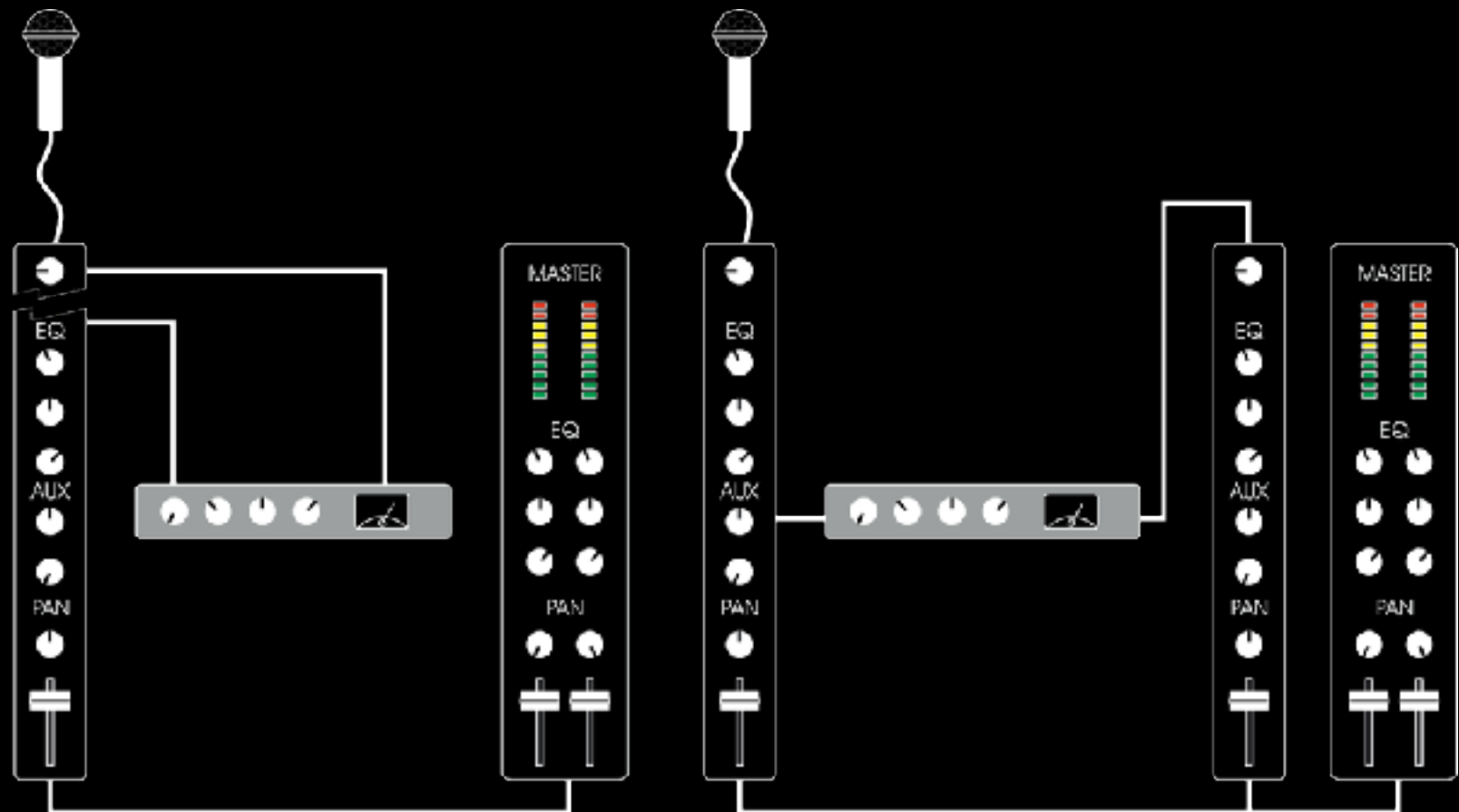
Time and compression

- ✱ Too short release time setting might lead to distortion



What is compressed?

- ✦ Everything. Dialogue, sound Fx, movie score...
- ✦ Side-chaining to open up the dynamics to let through the dialogue.
- ✦ Serial and/or parallel compression

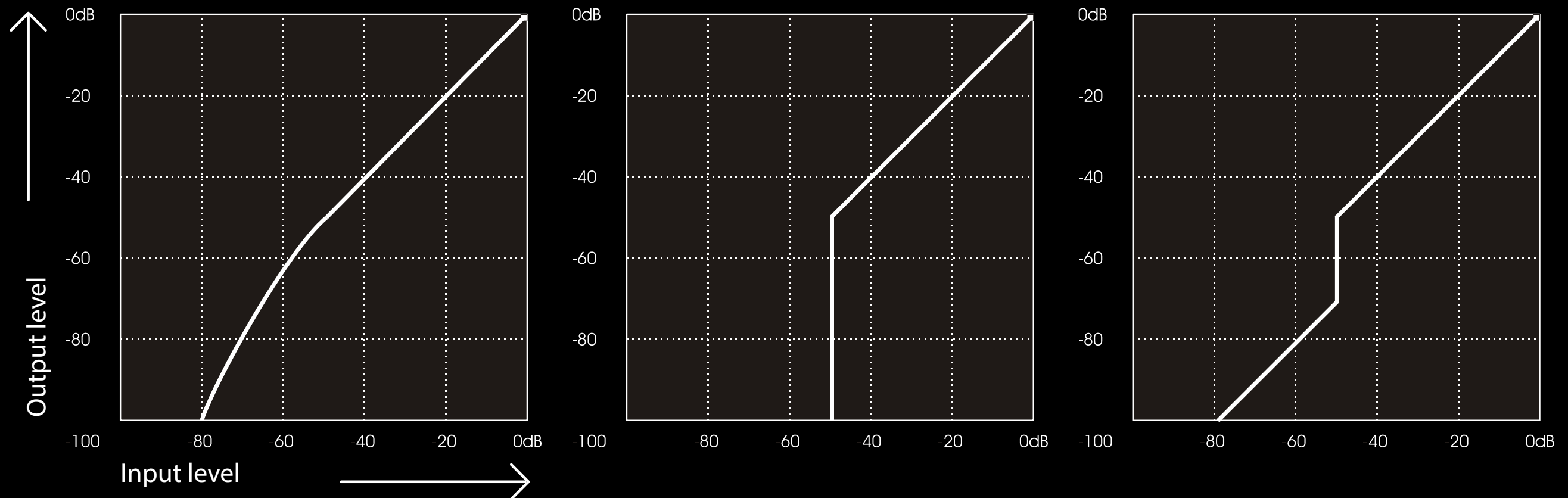


Effects on the sound

- ✦ Listen for changes on the transients (the attack) in the sound.
- ✦ Listen after frequency changes in the sound...
- ✦ Keep an eye on the master level on the output...

Expansion

- ✧ Expander, gate, noise-gate
- ✧ Expansion ratio - the amount of expansion



Expansion / gating

- ✧ Frequency settings in a gate
- ✧ Attack, release and hold

What is expanded/gated?

- ✧ Noisy recordings
- ✧ Reverb (for 80ies snare drums)
- ✧ Side-chaining to create effects and give room for other sounds in the dynamics.
- ✧ Listen for changes in the transient in the sound.