

## Multimedia Technology – labs in audio technology

- 1) Measurement of acoustic power radiated by a loudspeaker
- 2) Measurement of input electric impedance of a loudspeaker in a closed and bass-reflex enclosures
- 3) Measurement of sound absorption coefficient
- 4) Audio signals and their assesment

## Measurement of acoustic power radiated by a loudspeaker

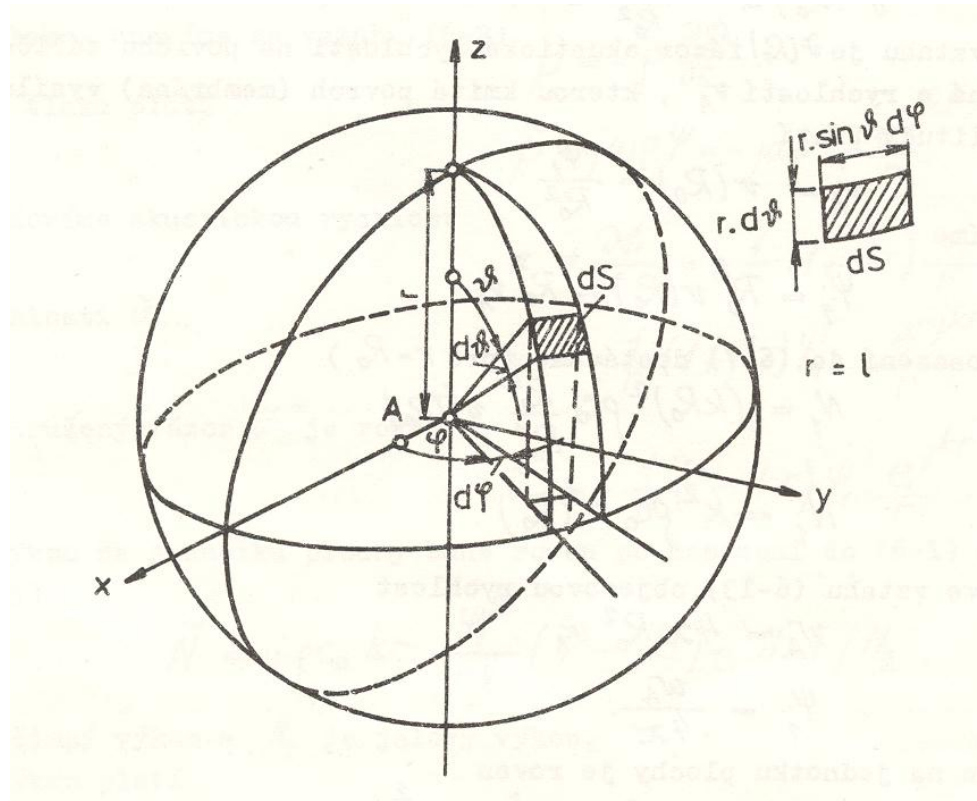
$$P = \vec{F} \cdot \vec{v} = p \cdot \vec{S} \cdot \vec{v} = p \cdot v \cdot S \cdot \cos(\psi)$$

P is the power, F force, v velocity, S surface,  $\psi$  phase angle

On condition of in-phase quantities (plane wave):

$$dP = I dS = p_{\text{ef}} v_{\text{ef}} dS = \frac{p_{\text{ef}}^2}{\rho c_0} dS$$

$$P = \frac{1}{\rho c_0} \iint_S p_{\text{ef}}^2 dS$$



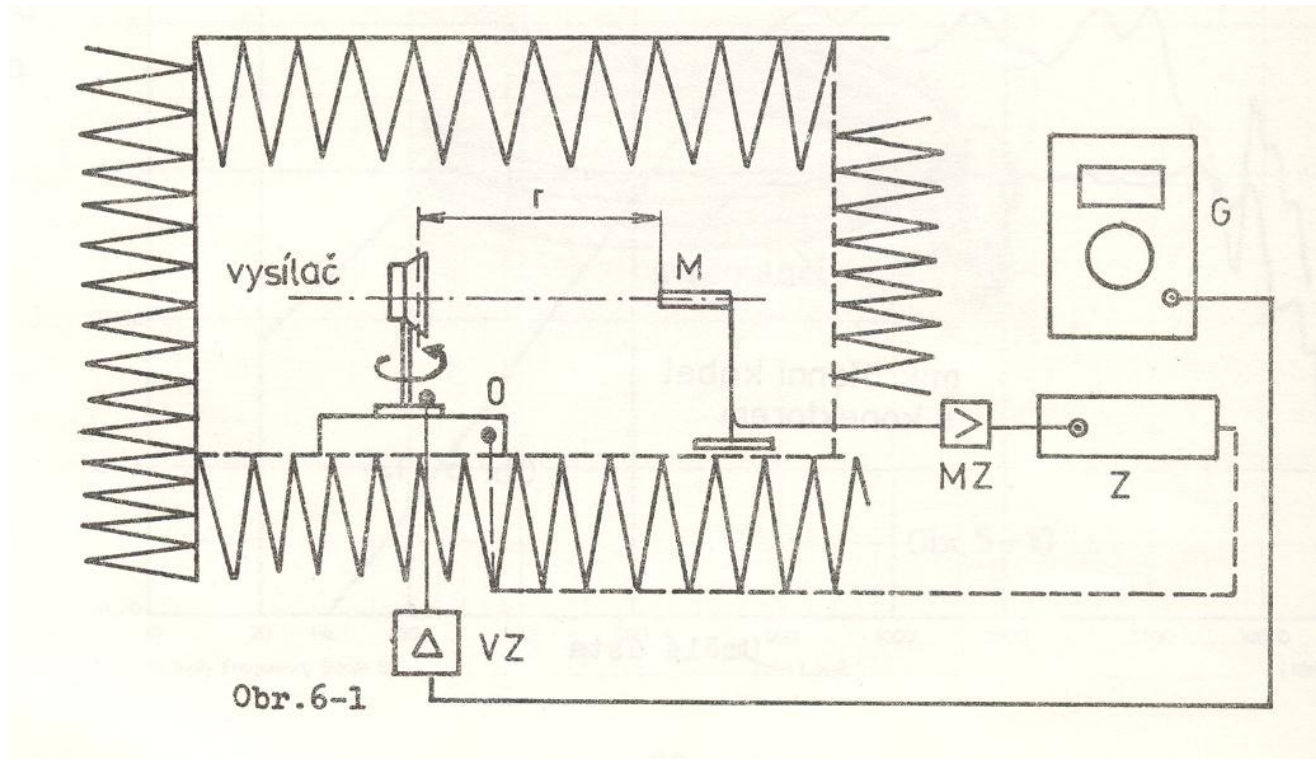
In spherical coordinates:  $dS = l^2 \sin(\vartheta) d\varphi d\vartheta, \varphi \in < 0; 2\pi; \vartheta \in < 0; \pi >$

Total radiated power: 
$$P = \frac{l^2}{\rho c_0} 2\pi \int_0^\pi p_\vartheta^2 \sin(\vartheta) d\vartheta$$

Pressure of the omnidirectional source (0th order transmitter),  
having the same power as the measured transmitter

$$P = 4\pi l^2 \frac{p_s^2}{\rho c_0} \Rightarrow p_s^2 = \frac{P \rho c_0}{4\pi l^2}$$

$$\int_0^\pi p_s^2 \sin \vartheta \, d\vartheta = 2p_s^2$$



## Numerical characteristics of transmitter directivity

Directivity factor

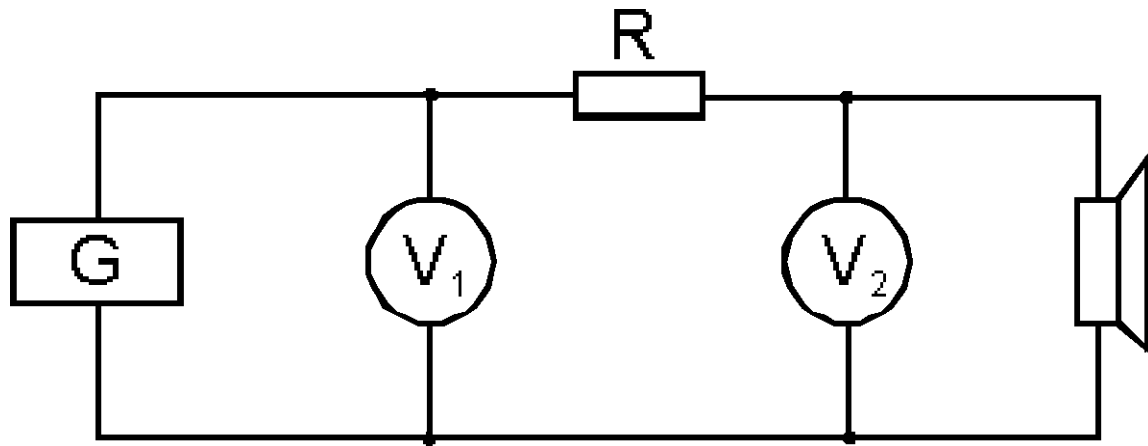
$$\sigma = \frac{p_{\vartheta}^2(\vartheta = 0, l = \textit{const.})}{p_s^2(l = \textit{const.})}$$

Directivity index

$$G = 10.\log \sigma$$

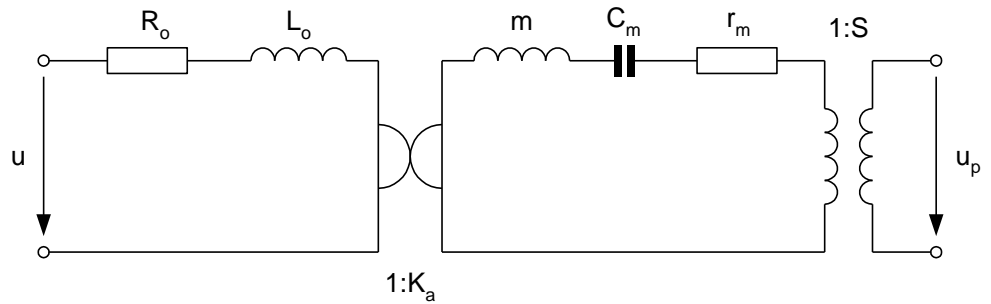
Measurement output: directional characteristics, total power,  $\sigma$  and  $G$ .

Measurement of input electric impedance of an electrodynamic loudspeaker in a closed and bass-reflex enclosures

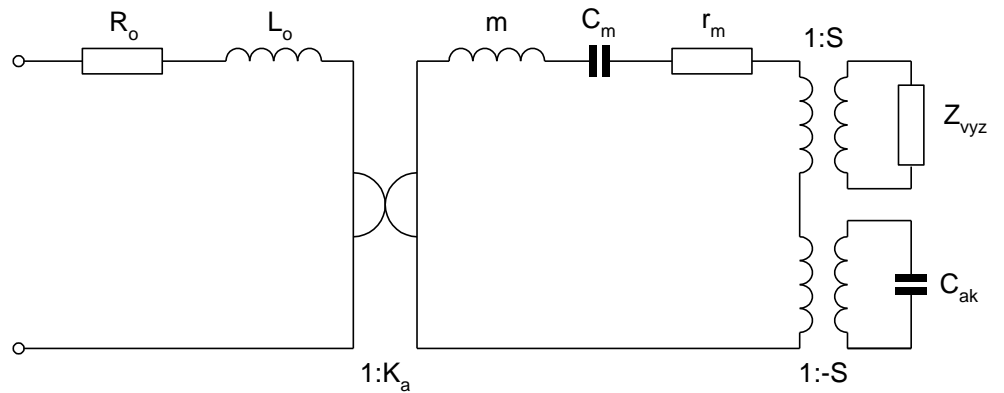


$$Z = \frac{U_2 R}{U_1 - U_2}$$

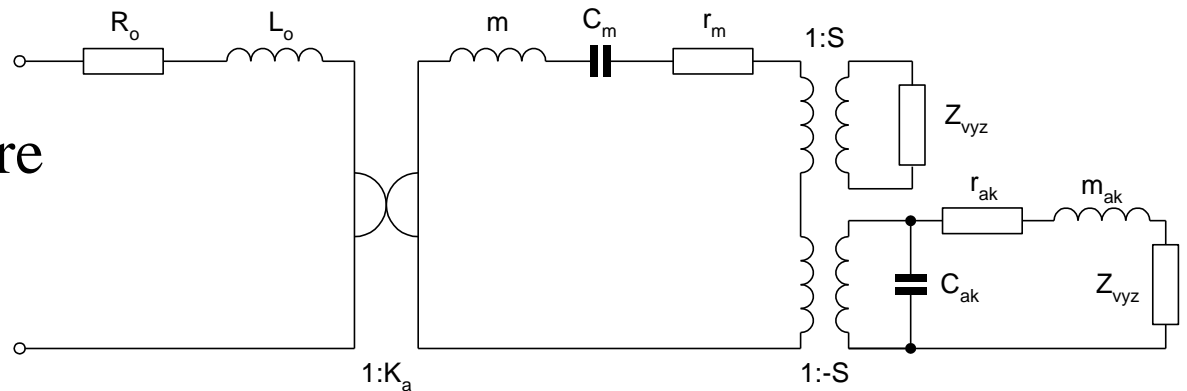
Without an enclosure



In a closed enclosure

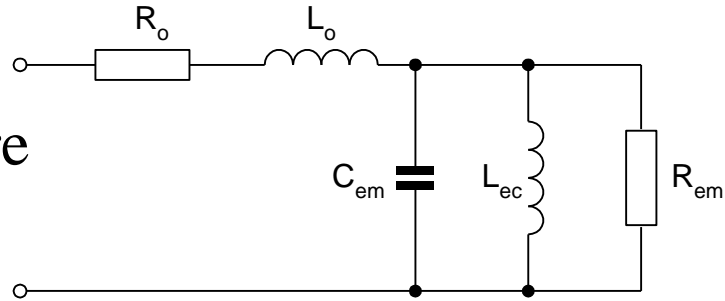


In a bass-reflex enclosure

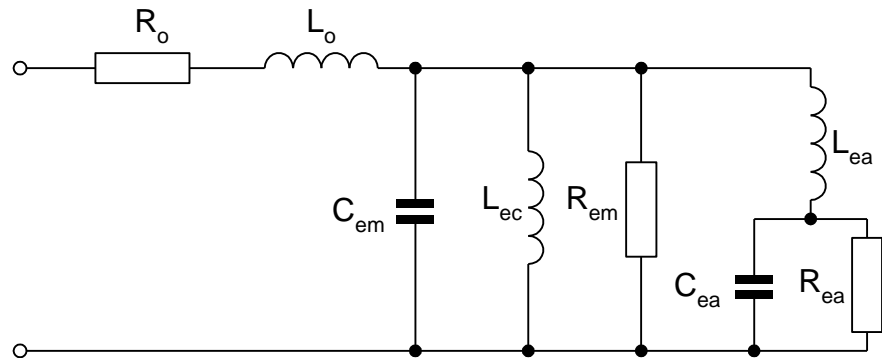


## Transformation to electric sides

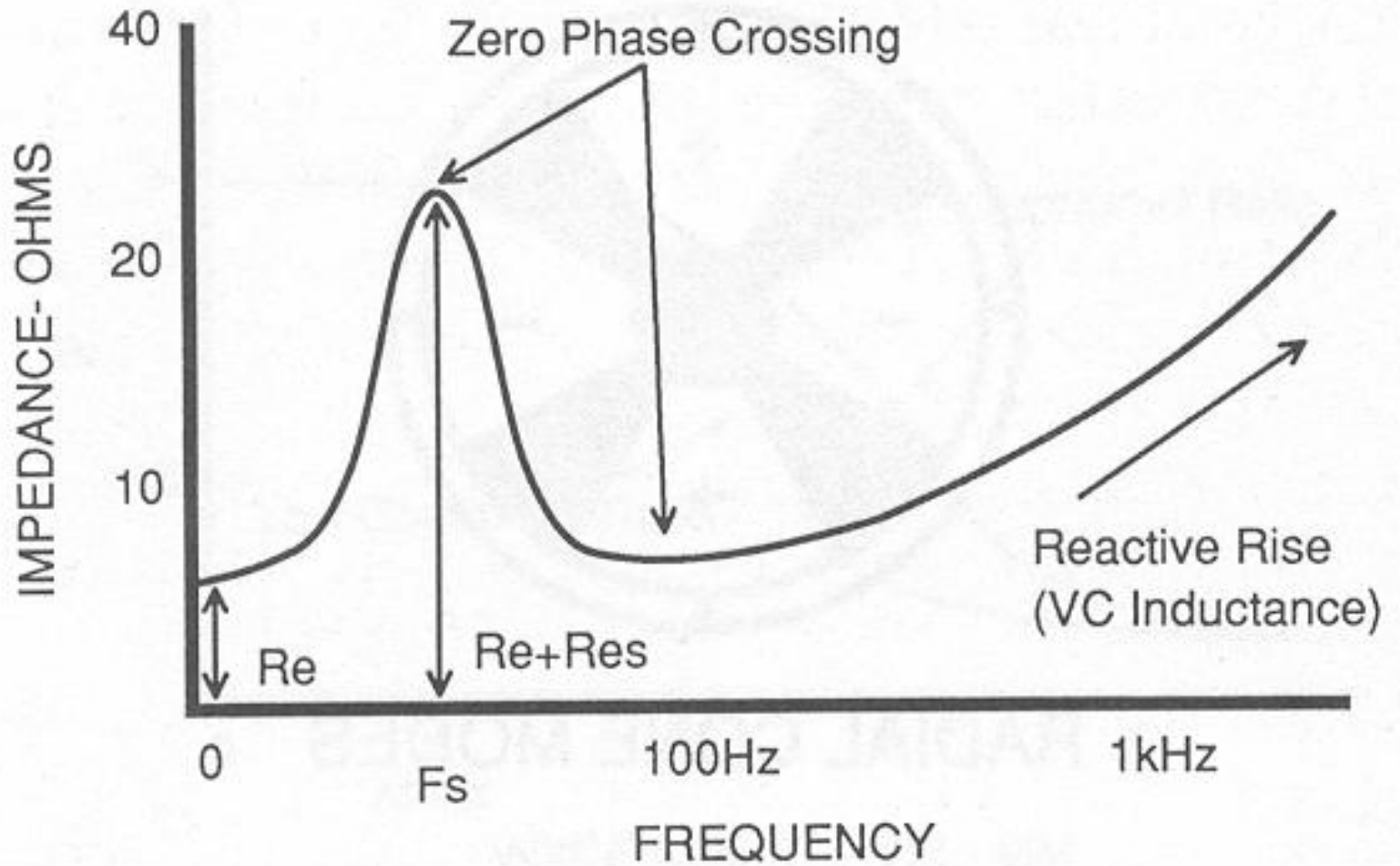
Loudspeaker without an enclosure  
and in a closed enclosure



Loudspeaker  
in the bassreflex enclosure

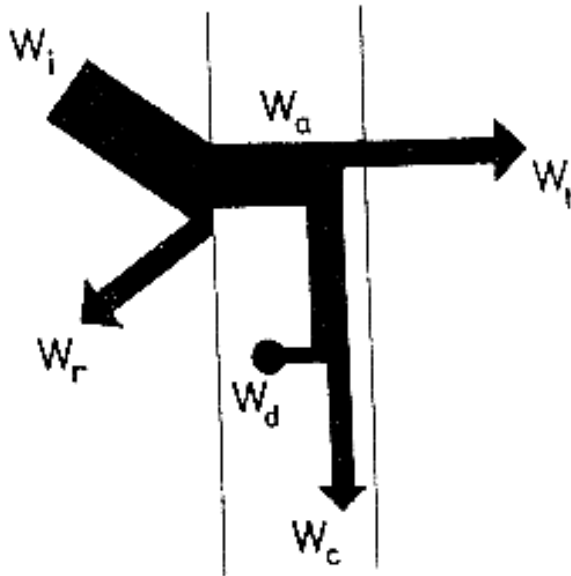






Output: Input electric impedances traced in one graph

## Measurement of the sound absorption coefficient in the Kundt tube



$$\alpha = \frac{W_a}{W_i}$$

Characterizes basic acoustic quality of a surface of a wall

Absorption of a wall  $A = \alpha \cdot S$

Sabine's reverberation time  $T = 0,164 \frac{V}{A}$

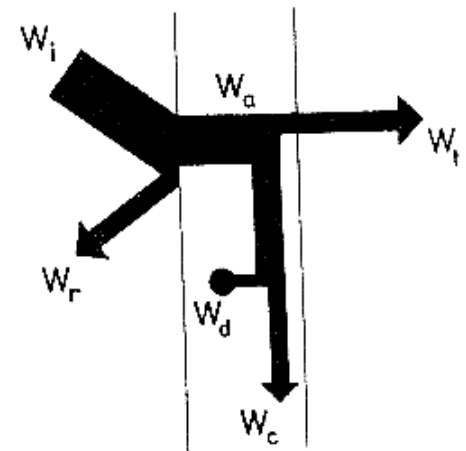
Measurement principle: termination of a waveguide (its acoustic parameters) affects spatial distribution of a standing wave and its character

$$W_i = W_r + W_a \qquad W_a = W_d + W_c + W_t$$

$$\alpha = \frac{W_a}{W_i} = 1 - \frac{W_r}{W_i} = 1 - |\beta|^2 \qquad \tau = \frac{W_t}{W_i}$$

$\beta$ ...reflexion coefficient

$\tau$ ..transmission coefficient



Measurement of pressure with a probe at maxima and minima

$$\mu_0 = \frac{|p_{\min}|}{|p_{\max}|} = \frac{(1 - |\beta|)}{(1 + |\beta|)}$$

$\mu_0$  ... standing wave ratio

$$\alpha_0 = \frac{4}{2 + \mu + \frac{1}{\mu}} = \frac{1}{\frac{1}{2} + \frac{1}{4} \left( \mu + \frac{1}{\mu} \right)}$$

Sound absorption coefficient for perpendicular incidence

Sound absorption coefficient is frequency dependent

Usually measured at following frequencies

64 Hz

125 Hz

250 Hz

500 Hz

1kHz

2kHz

4kHz

8kHz

Because of the method the measurement range is 350-750 Hz

Output:  $\alpha$  versus frequency curves

# Sound signals and their assessment

- Characteristics of human hearing
- Review of basics from Signals and Systems
  - Time and frequency domain
  - Digital signal (sampling & quantization)
- Lossy compression of audio signals

Adobe Audition is used

# Basics

- Frequency spectrum – analysis of unknown (periodic) signal – frequencies and levels
- Influence of sampling and quantization on frequency spectrum
- Analysis of Dirac pulse, impulse response of LP and HP creation

# Hearing characteristics

- Frequency range of your hearing (swept signal)
- Dynamic range of your hearing (1 kHz), bit depth



# Lossy Audio Compression

- Comparizon of 3 codecs (mp3, ogg vorbis, wma) artifacts on low bit rates – audible artifacts, time and frequency domain analysis
- The lowest bit rate for superior quality (for the three codecs)

# Outputs

- Table – unknown signal spectrum
- Impulse response of LP and HP
- Your hearing frequency range for 3 levels
- Your hearing dynamic range at 1 kHz
- Compression artifacts in time, frequency and audible domain for 3 codecs
- Minimal bitrate for 3 codecs