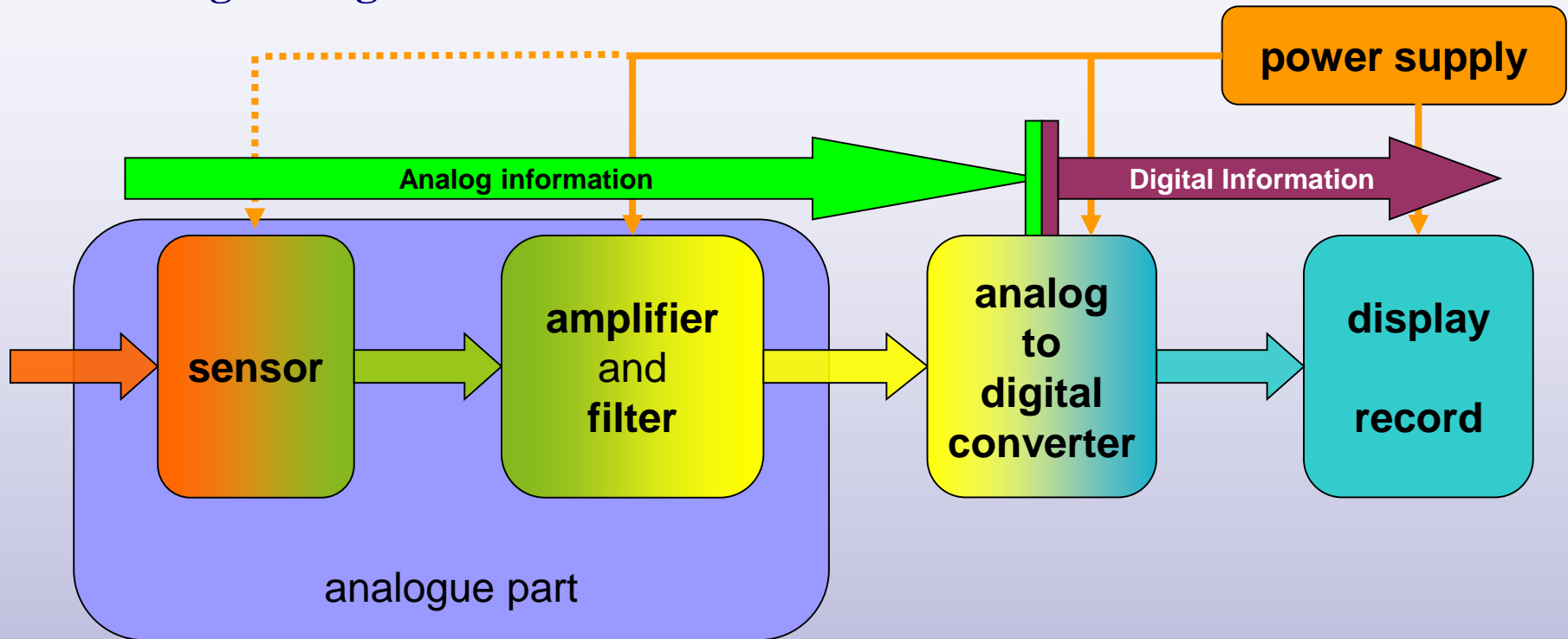


Measurement device component properties

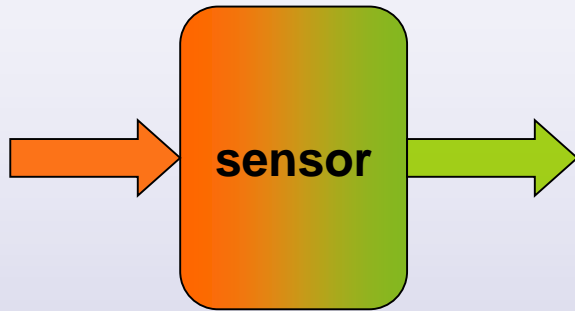
Analogue part



Analogue - digital measurement device



1. Sensor



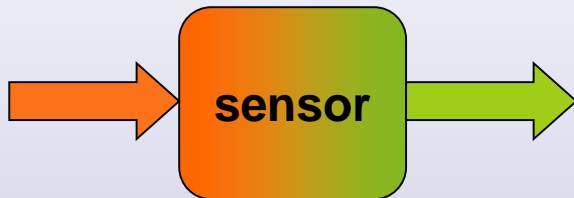
sensor converts the monitored physical quantity into electrical physical quantity (voltage, current,)

- **1.1. sensor static properties**
- **1.2. static transmission characteristic errors**
- **1.3. sensor dynamic properties**
- **1.4. sensor properties = summary**

1.1. Sensor static properties

- **input range**

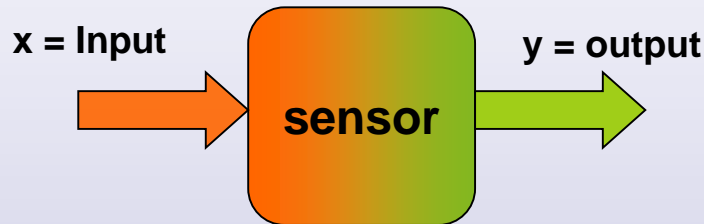
- the minimum and maximum values of the input quantity that the sensor is able to process



- if the input quantity is inside the sensor range the output signal is proportional to the input quantity
- if the input quantity is outside the sensor range
 - the output signal does not match the input quantity
 - the sensor may be destroyed

1.1. Sensor static properties

- **sensor static transmission characteristic (sensitivity)**
 - the relationship between the output and input signal



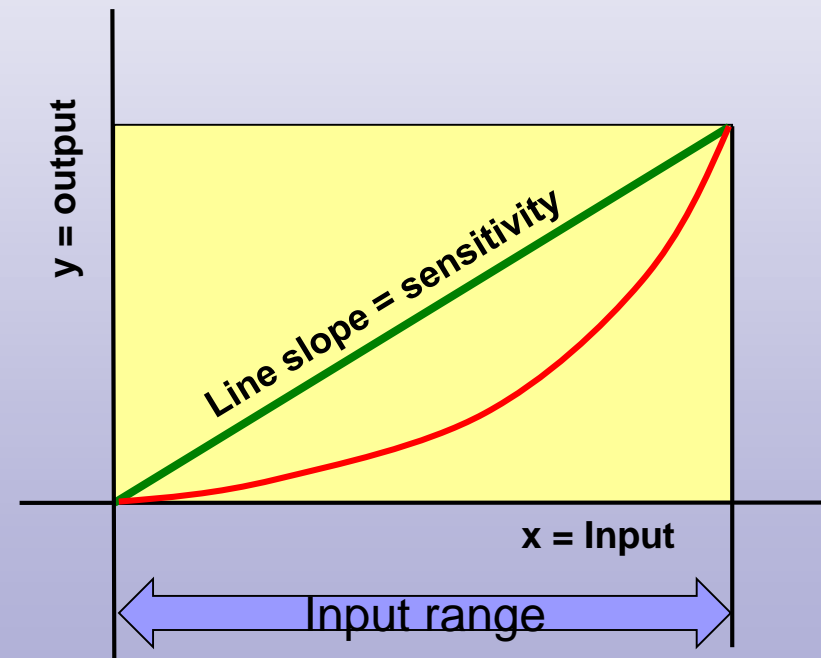
- generally it can be a polynomial function

$$y = a_0 + a_1x + a_2x^2 + \dots + a_nx^n$$
- usually the function is linear

$$y = a_1x$$
- the symbol S is used instead of the coefficient a_1

$$y = S * x$$

where S is the sensor sensitivity



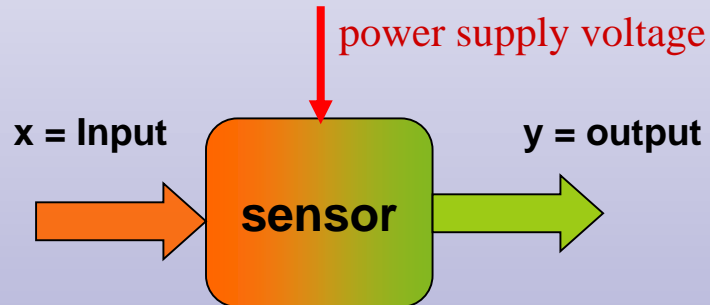
- **static conversion characteristic is valid only if the input quantity is within the range !!!!**

1.1. Sensor static properties

- **examples of the sensor description**

- input range + sensitivity input 0 - 100°C, sensitivity 5mV/°C
- input range + output range input 0 - 100°C, output 0 – 500mV

- **the sensitivity can be depended on power supply voltage for some types of sensors** (eg strain gauge force sensors)



$$y = S * V * x$$

where S is the sensor sensitivity
 V is the power

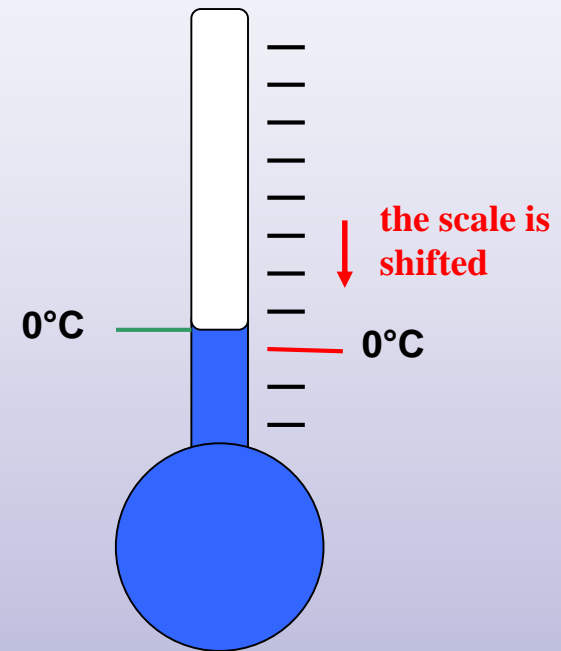
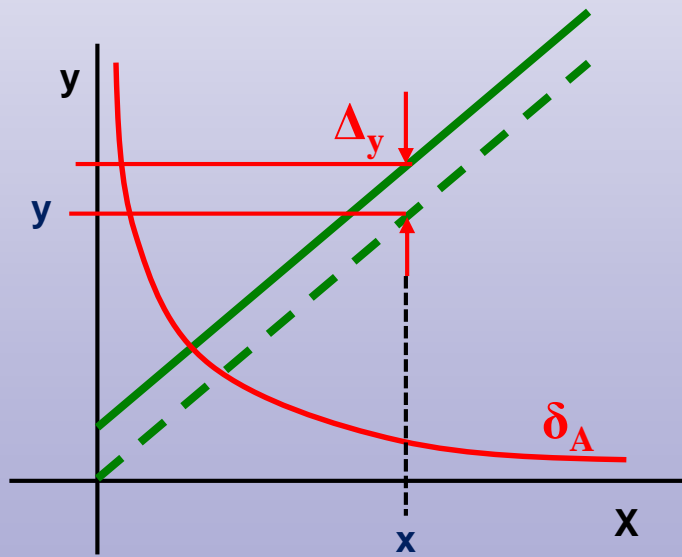
voltage

- **example of this sensor description**

- input range **10kN**
- output **2mV/V** this information means that for the maximum force 10 kN the output will be 2mV per each one volt of power supply
- for example, if the power supply is 10V, the sensor output will be 20mV at full load by force of 10kN

1.2. static transmission characteristic errors

- additive error (zero shift) δ_A
 - the zero input does not correspond to the zero output
 $y = S \cdot x + q$
 - deviation from the correct value $\Delta y = q$
 - the deviation is constant over the entire range
 - error calculation: $\delta_A = \frac{\Delta y}{y} (* 100 \text{ [%]})$



Example:

thermometer range: 0 - 100°C

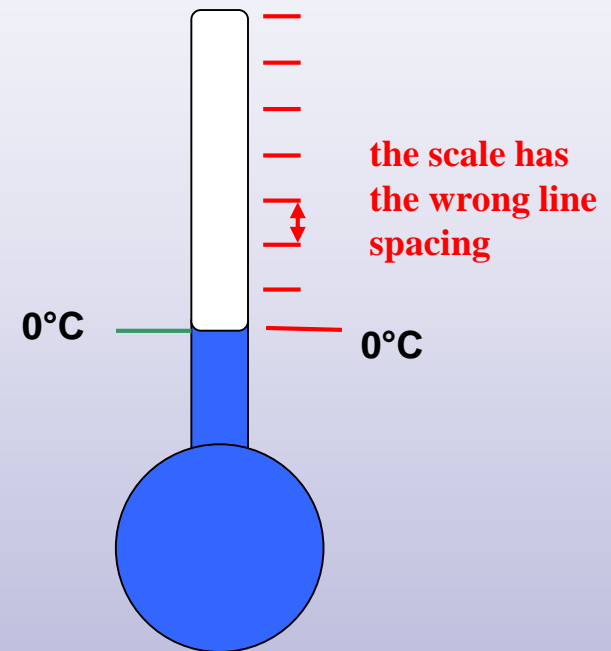
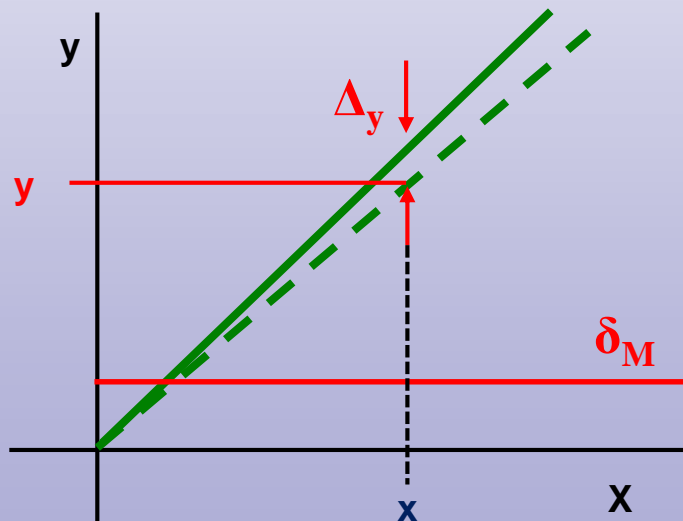
zero shift $\Delta y = 1^\circ\text{C}$

reality	thermometr value	error δ_A
1°C	2°C	100%
100°C	101°C	1%

- the additive error hyperbolically distorts the output, that is, most at the beginning of the range

1.2. static transmission characteristic errors

- **multiplicative error (sensitivity error) δ_M**
 - the sensitivity is not S , but $S + \Delta S$
 - $y = (S + \Delta S) * x$
 - deviation from the correct value $\Delta y = \Delta K * x$
 - the deviation increases with value
 - error calculation: $\delta_M = \frac{\Delta y}{y} = \frac{\Delta K * x}{K * x} (* 100 [\%])$



Example:

thermometer range: 0 - 100°C

scale division $S + \Delta S = 1.01^\circ\text{C}$

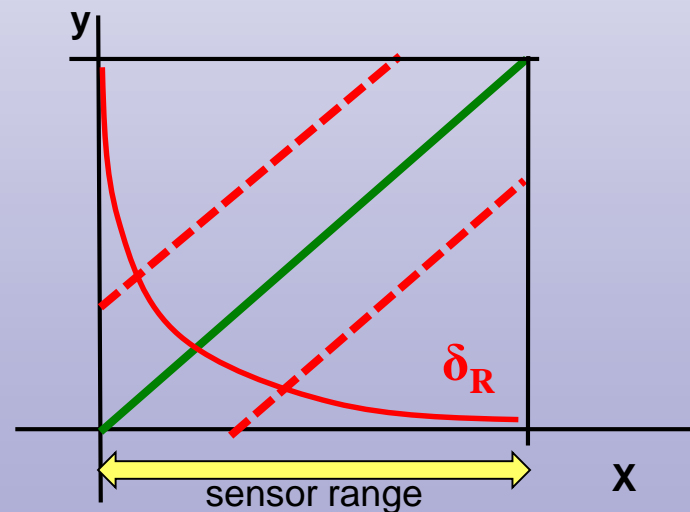
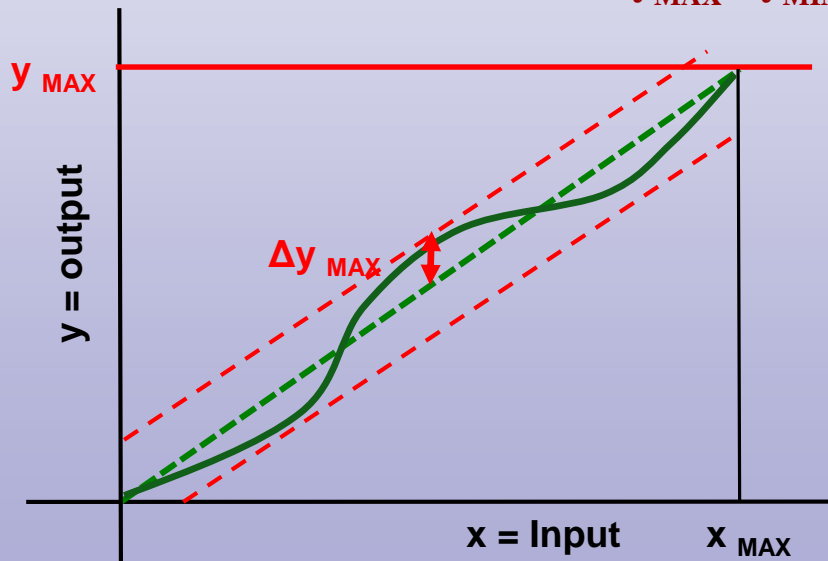
reality	thermometer value	error δ_M
1°C	1.01°C	1%
100°C	101°C	1%

- the multiplicative error is constant over the entire range

1.2. static transmission characteristic errors

- the sensor relative error (the sensor accuracy) δ_r
 - multiplicative and additive error can occur simultaneously
 - static characteristic can have other errors - nonlinearity, hysteresis, etc.
 - **determining the resulting error would be very difficult**
 - therefore the sensor producer publishes the maximum output deviation Δy_{MAX} and guarantees that deviations in the whole range are always smaller than this maximum

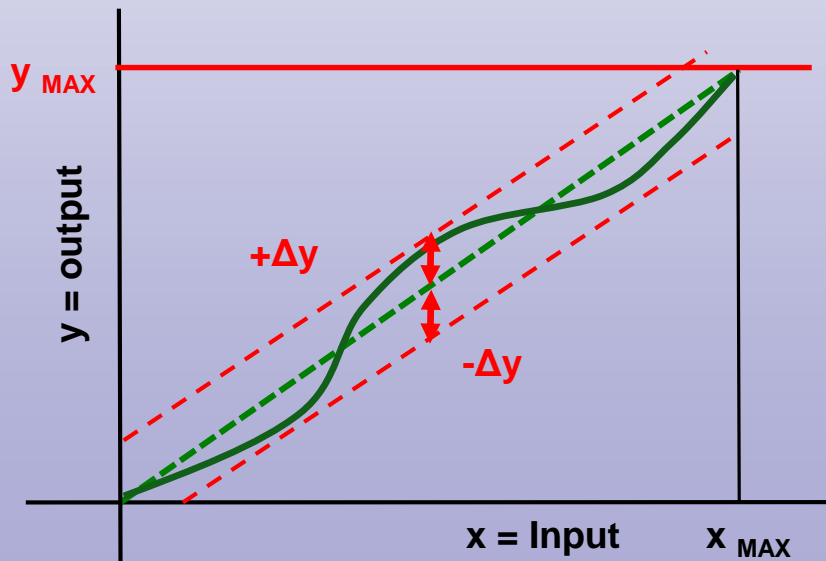
• sensor relative error $\delta_r = \frac{\Delta y_{MAX}}{y_{MAX} - y_{MIN}} = \frac{\Delta y_{MAX}}{\text{sensor range}}$



- the relative error hyperbolically distorts the output

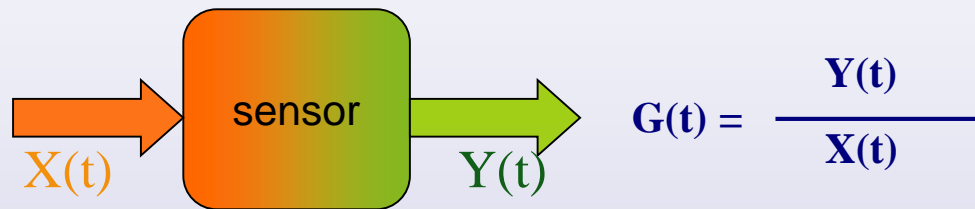
1.2. static transmission characteristic errors

- the sensor resolution
- the deviation Δy caused by the relative error can be **positive or negative**
- the sensor resolution is the smallest change in a sensed quantity that can be unambiguously determined
- the resolution $r = \frac{1}{\frac{y_{MAX}-y_{MIN}}{2*\Delta y_{MAX}} + 1} \cong \frac{2*\Delta y_{MAX}}{y_{MAX}-y_{MIN}} = 2 * \delta_R$
- the sensor resolution is thus essentially twice the relative error (accuracy) of the sensor

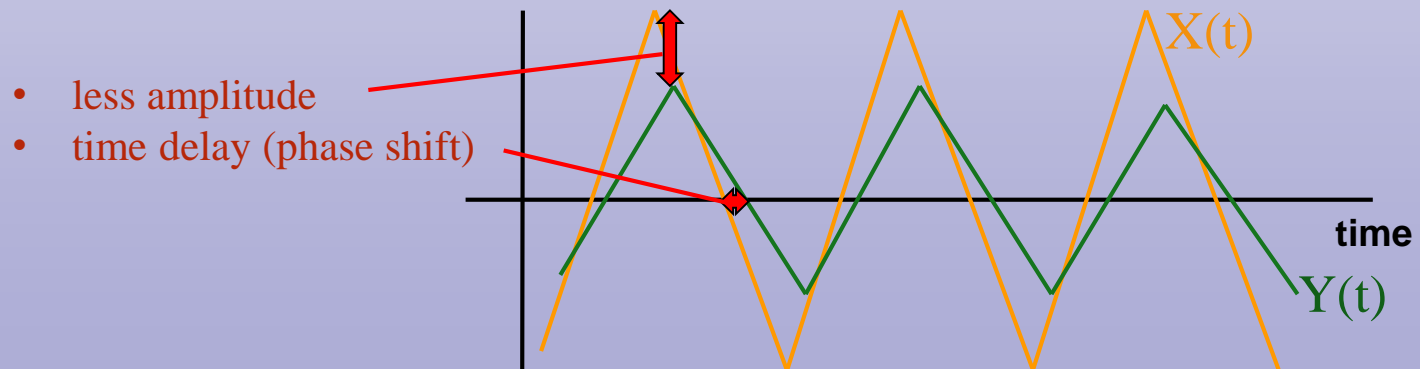


1.3. sensor dynamic properties

the sensor transfer characteristic $G(t)$ is defined as ratio of the output function to the input function

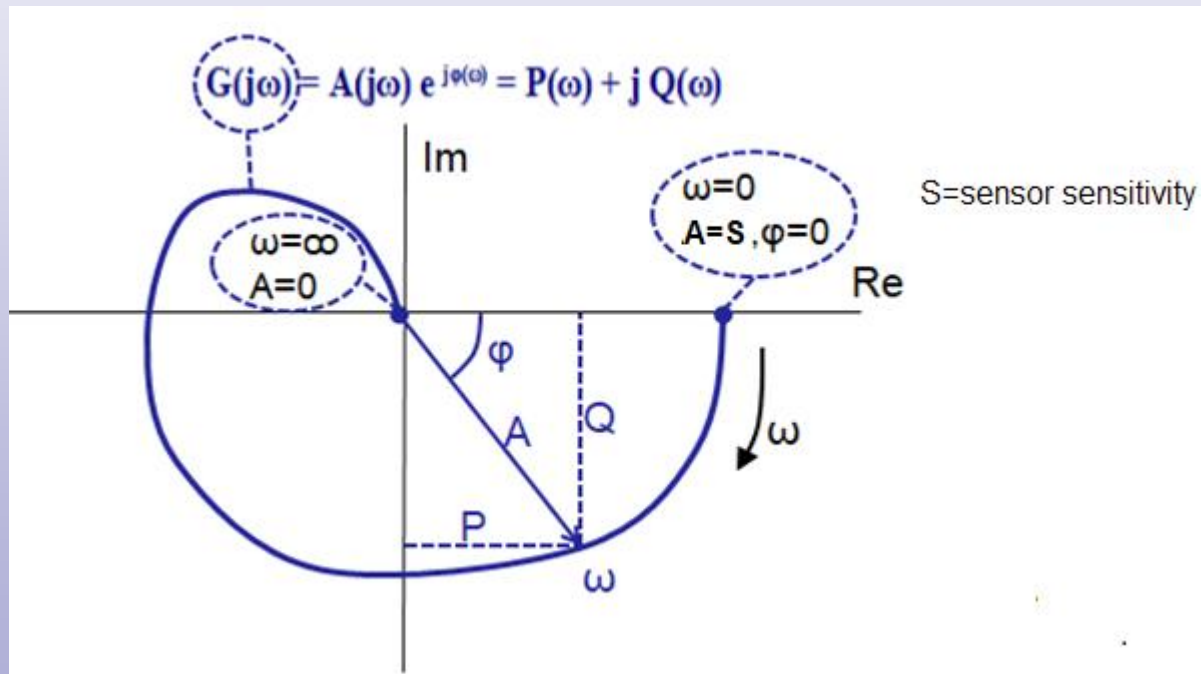


- the ideal state is when the output function exactly follows the input function, thus $G(t)$ is constant and it is $G(t) = S$, where S is the sensor sensitivity
- the input function is variable over time
- the internal circuits of the sensor work at some maximum speed
- **if the rate of change of the input function exceeds the capabilities of the sensor, $G(t)$ ceases to be constant and the output signal is distorted**



1.3. sensor dynamic properties

- **the sensor frequency characteristic**
- this time function $G(t) = \frac{Y(t)}{X(t)}$ is converted by the Fourier Transform into the frequency domain
 - $G(j\omega) = \frac{Y(j\omega)}{X(j\omega)} = A(j\omega) * e^{j\varphi(\omega)} = P(\omega) + j * Q(\omega) \quad \text{where } \omega = 2\pi f$
 - the frequency characteristic can be drawn in a complex plane:



1.3. sensor dynamic properties

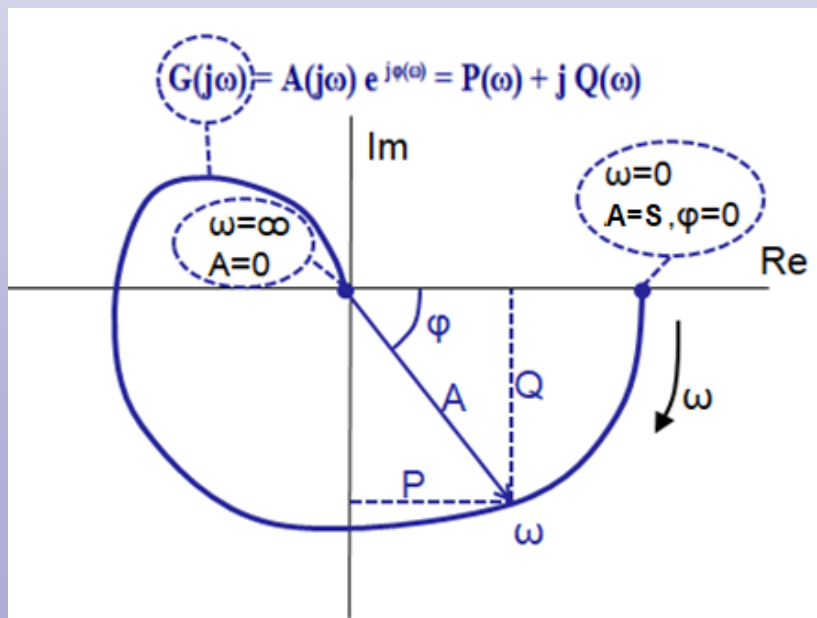
- the sensor amplitude and phase characteristic
- the representation of the frequency response in the complex plane may be incomprehensible therefore, the amplitude and phase characteristics are displayed more often as a dependence of amplitude and phase on frequency

$$G(j\omega) = P(\omega) + j Q(\omega)$$



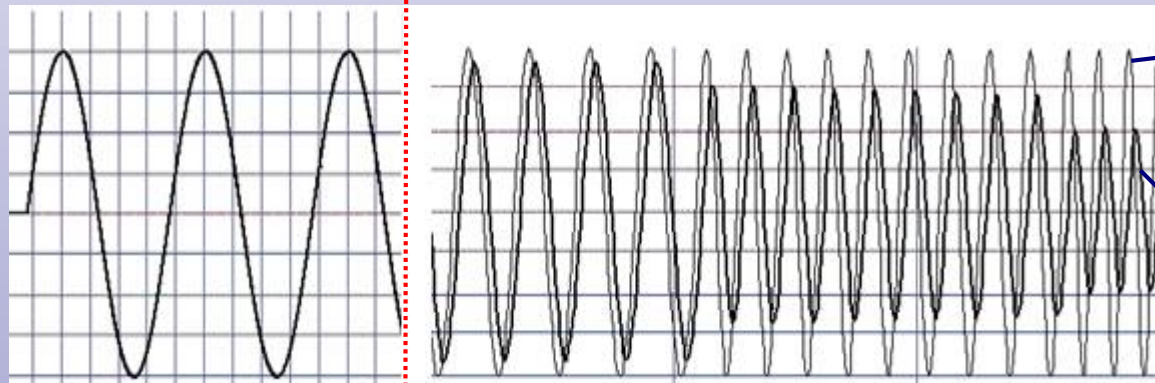
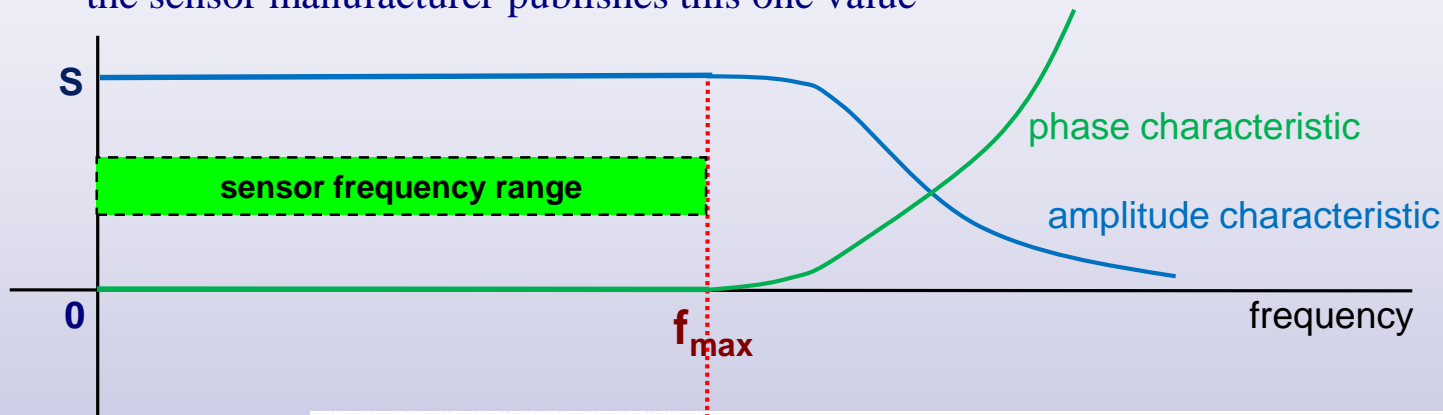
$$A(\omega) = \sqrt{P^2(\omega) + Q^2(\omega)}$$

$$\varphi(\omega) = \arctan \frac{Q(\omega)}{P(\omega)}$$



1.3. sensor dynamic properties

- the sensor frequency range
 - it is the maximum frequency of the input signal where no output distortion
 - the sensor manufacturer publishes this one value



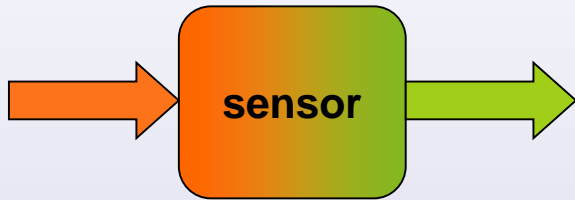
input signal

output signal

the input and output signals are the same

the output signal is distorted

1.4. sensor properties = summary



the manufacturer writes the following basic sensor properties in it data sheet:

- **sensor input range**
 - the minimum and maximum values of the input quantity that the sensor is able to process
- **sensor sensitivity**
 - the relationship between the output and input signal
- **sensor accuracy and resolution**
 - the maximum output deviation in the whole range are always smaller than sensor accuracy
 - the sensor resolution as twice the sensor accuracy
- **sensor frequency range**
 - the maximum frequency of the input signal where no output signal distortion

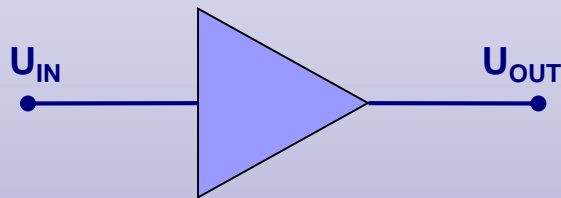
2. Amplifier

- amplifier regulates (increases) the size of the output sensor signal to the level required for its processing in the A/D converter



amplifier gain G

- one value gain for single-purpose input
- adjustable gain for universal input (set by the user program today)

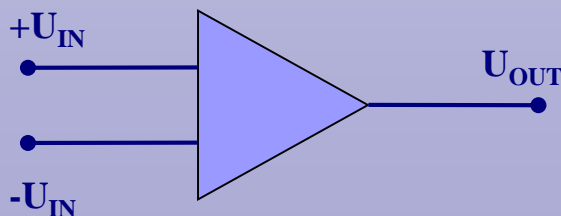


$$U_{OUT} = G * U_{IN}$$

where G is gain

differential amplifier

- used for sensors with differential output (eg strain gauges force sensors)
- it has two inputs + IN and -IN and amplifies the signal difference between these inputs



$$U_{OUT} = G * (+U_{IN} - -U_{IN})$$

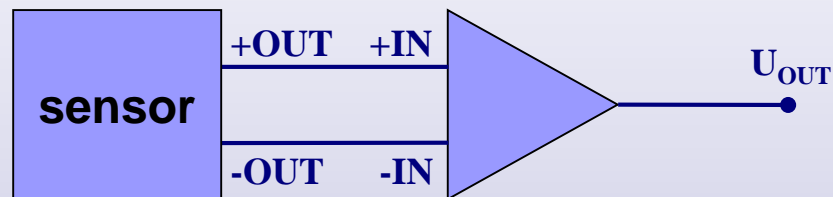
2. Amplifier

- **the amplifier properties are practically identical to the properties of the sensor so the amplifier also has 4 basic parameters**
 - input range
 - gain (usually can be set to several values)
 - relative error (accuracy)
 - frequency range (maximum rate of input change)
- **since the parameter definitions are identical to the sensor, they will not be explained again here** (only instead of sensitivity is here gain)
- amplifier
 - is an electronic circuit
 - relative error and frequency range are usually much better than the sensor properties
- **measurement accuracy depends predominantly on the sensor**

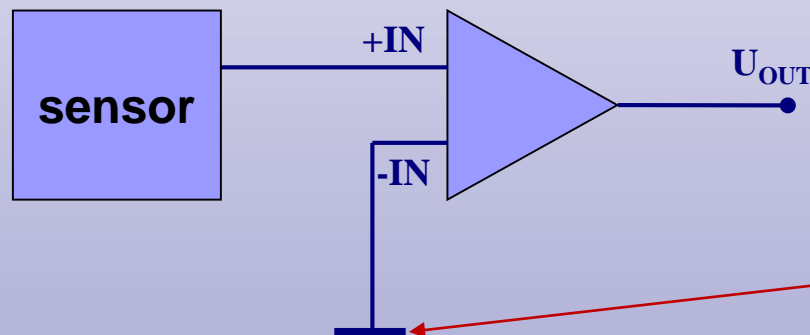
2. Amplifier

differential amplifier connection

- if a sensor with a differential output is used, the sensor outputs + OUT and -OUT are connected to the + IN and -IN amplifier inputs:



- if a sensor with a unipolar output is used, its output is connected to the amplifier + IN input:



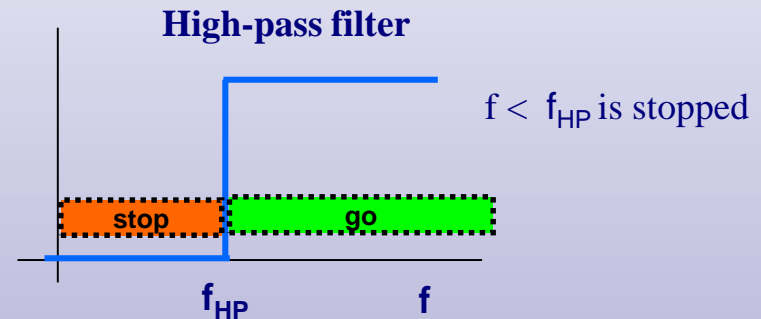
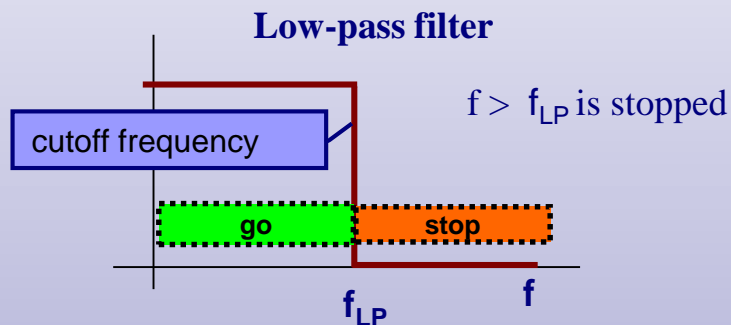
$$U_{OUT} = G * (+U_{IN} - -U_{IN})$$

$$U_{OUT} = G * (+U_{IN} - 0) = G * +U_{IN}$$

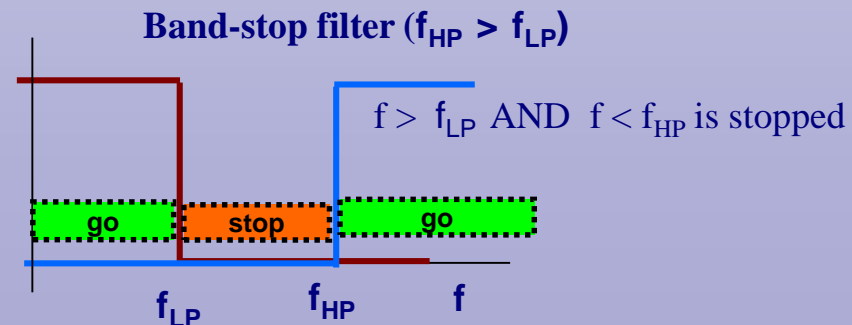
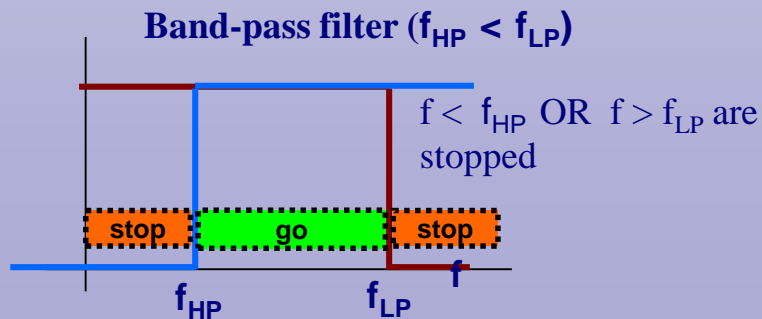
- the amplifier input - IN must be connected to ground so that it has a zero signal
- not connected amplifier input - IN is an antenna for noise reception

3. Filter

- **removes unwanted components (noise) from the signal**
 - it is part of the amplifier
 - the filter principle is not to transfer certain frequencies in the signal, so it can prevent the transfer of noise
 - there exist 4 types of filters according to the band of non transferred frequencies
- **filters classification according to the non transferred band:**
 - two basic filters

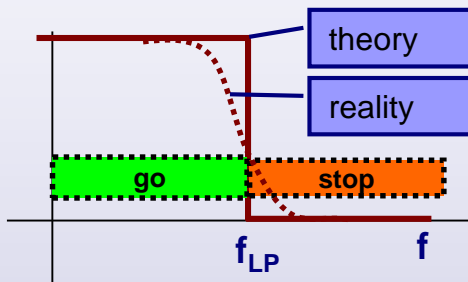


- two compound filters - consist of a pair of basic filters



3. Filter

- it is not possible to realize a rectangular filter frequency characteristic in practice



- the course of the characteristic (slope) is given by:

- **filter type**

type of electrical circuit

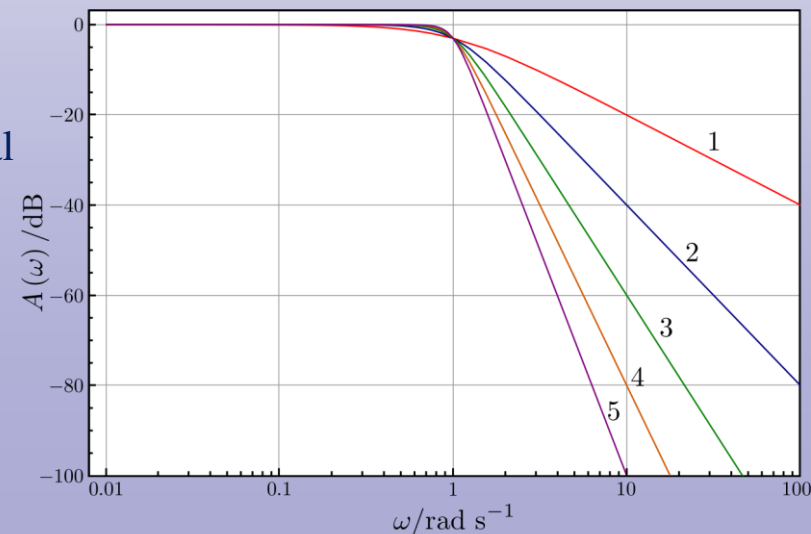
the filter type is named after the its author – Bessel, Butterworth

- **filter order**

can be simply explained as the number of filters included in the series - the first order filter is one filter, the second order filter is two identical filters in a serie, etc.

higher filter order = steeper characteristic slope

- the slope of the filter amplitude characteristic is given in dB/decade or dB/octave for sound



3. Filter

▪ decibel (dB) definition

- base unit **bel** (B) is named after A. G. Bell (he used it for attenuation in the telephone line)
- **1B = ratio of 10 : 1** between **power** of sound at the beginning and end of the telephone line
- **1dB** (decibel) is **the tenth** of the base unit **bel**
- ie power ratio 10 : 1 = 1B (10dB), 100 : 1 = 2B (20dB), 1000 : 1 = 3B (30dB)

- $1B = \log\left(\frac{P_2}{P_1}\right) \quad [-, W, W]$

- $1dB = 10 * \log\left(\frac{P_2}{P_1}\right) \quad [-, W, W]$

- if we need to express the ratio of the signal amplitudes instead of signal powers, we can define:

$$P = \frac{U^2}{R} \text{ and } \log(x^2) = 2 * \log(x)$$

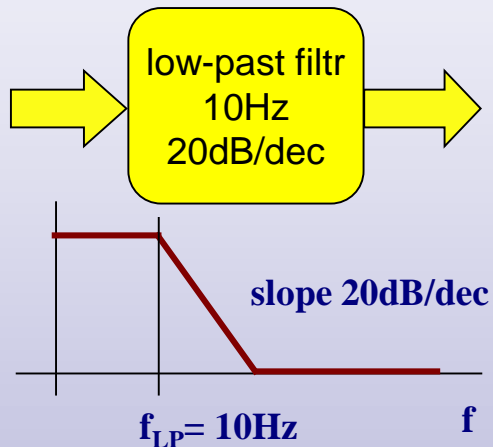
- $1dB = 20 * \log\left(\frac{U_2}{U_1}\right) \quad [-, V, V]$

U [V]		P [W]
reduction	dB	reduction
1	0	1
0,89	1	0,79
0,79	2	0,63
0,707	3	0,5
0,5	6	0,25
0,35	9	0,125
0,316	10	0,1
0,22	13	0,05
0,1	20	0,01
0,01	40	0,0001

3. Filter

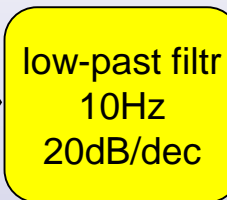
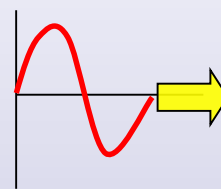
example of the low-pass filter

- cutoff frequency 10Hz
- slope 20dB/dec

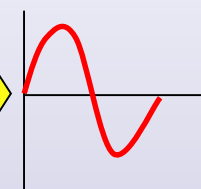


U [V]		P [W]
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0,01	40	0,0001

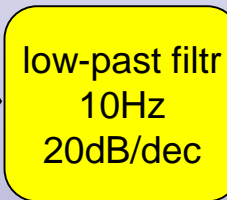
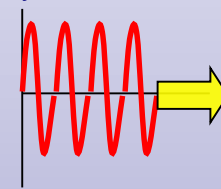
signal amplitude 10V
frequency 10Hz



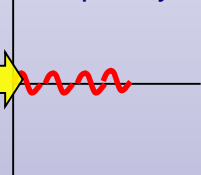
signal amplitude 10V
frequency 10Hz



signal amplitude 10V
frequency 100Hz



signal amplitude 1V
frequency 100Hz



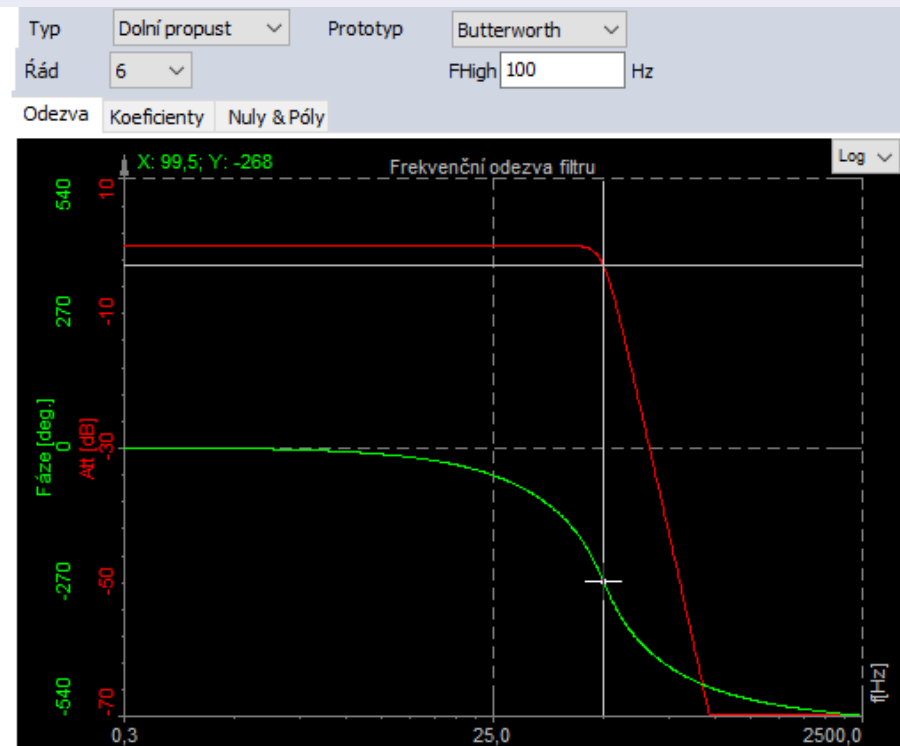
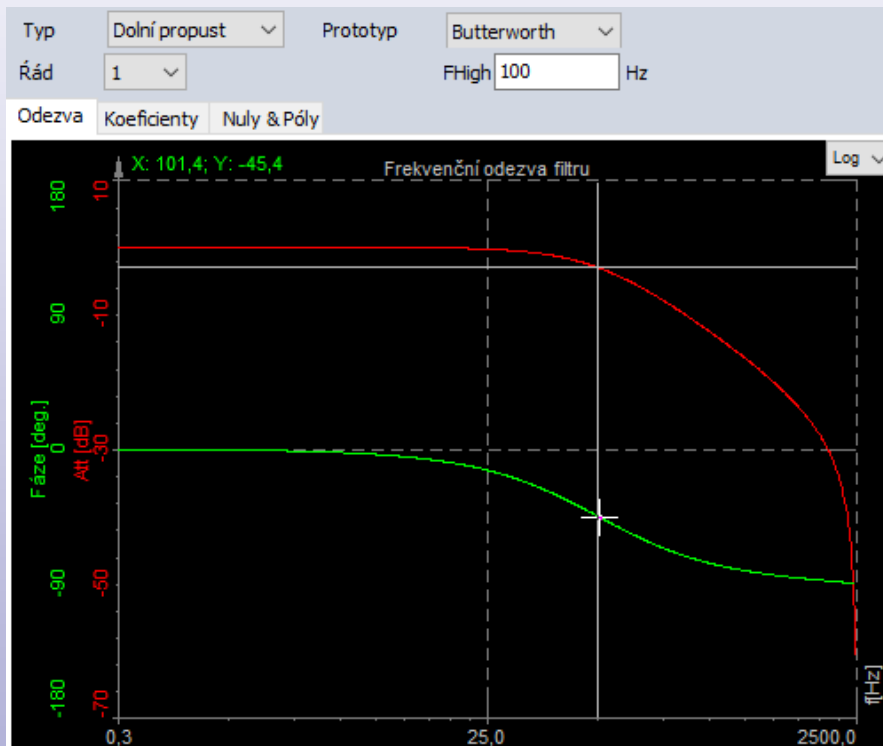
- the amplitude of the signal with a frequency lower than the cutoff frequency did not change
- 20dB/dec means that on the range of one order (for example from 10 to 100Hz) the signal amplitude is reduced 10x
- the amplitude of the signal with a frequency of 100Hz is reduced to one tenth, ie 1V

3. Filter

- **example of the real low-pass filter**

- low-pass filter, type Butterworth, cutoff frequency 100Hz
- left first order, right sixth order

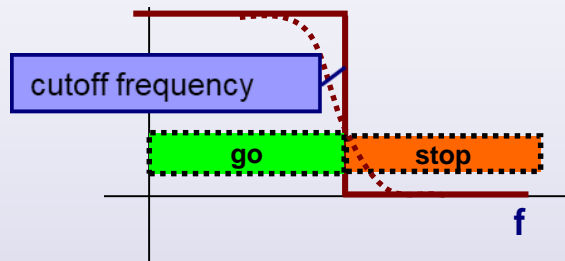
amplitude characteristic is red
phase shift is green



- the decrease in the real filter characteristic may not be linear
- **each filter also causes a phase shift of the signal !!!!!**
- **the higher order has a steeper slope of the amplitude characteristic, but also a larger phase shift**

3. Filter

- Summary of the filter properties



- filter type according to the non transferred band
- cutoff frequency [Hz]
- the slope of the filter amplitude characteristic [dB/dec]
 - type of electrical circuit (Bessel, Butterworth)
 - filter order

3. Filter

- **final controversy whether to use the filter or not**
 - **it will never be in the measured data what the filter stops!!!**
 - an incorrectly set filter can invalidate the measured data
 - the data will not contain some signal component
 - there will be a phase shift on the signals
 - **the filter can also be applied mathematically to the measured data**
- **for short-term non-repetitive measurements**
 - it is better not to use any HW filter, to record all even with noise and then do the filtering mathematically on data files
 - it can be judged (compared) how the filter will affect the signal, but it is added work
- **for long-term repetitive measurements**
 - to measure for the first time without a filter
 - using mathematical filtering on the measured data for "tune" the filter parameters to remove noise and not affect the signals
 - to set the real filter parameters by obtained parameters and use it for further repeated measurements
 - no further mathematical filtering of the data is then required

Exam questions

- sensor static and dynamic properties
 - input range (p. 3)
 - sensor static transmission characteristic (sensitivity) (p. 4)
 - additive error and multiplicative error (p. 7, 8)
 - the sensor relative error (accuracy) and the sensor resolution (p.9, 10)
 - the sensor frequency characteristic, amplitude and phase characteristic (p. 12, 13)
 - the sensor frequency range (p. 14)
 - sensor properties summary (p. 15)
- amplifier
 - unipolar and differential amplifier, principal, gain (p. 16)
 - amplifier properties summary (p. 17)
- filter
 - filters classification according to the non transferred band (p. 19)
 - the slope of the filter amplitude characteristic, unit (p. 20, 21)
 - filter type and filter order and their influence on the characteristic slope (p. 20, 23)
 - filter properties summary (p. 24)