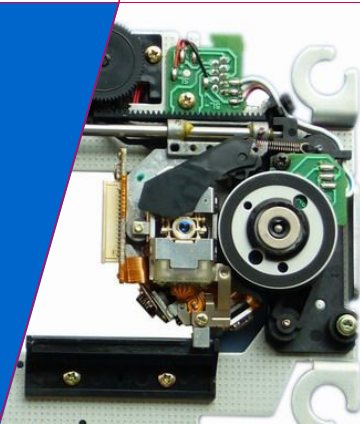


4CM00: Control Engineering *Background and prerequisites*



September 2020

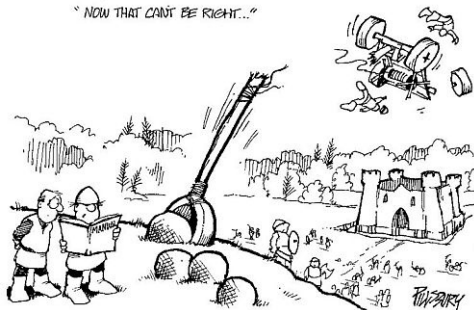
TU / **e**

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Where innovation starts

What is this course all about?

- ▶ Control the output of a dynamic system by manipulating its input
 - via both feedback and feedforward techniques
 - requires knowledge and understanding of the dynamic system
- ▶ We consider only linear systems
- ▶ We consider only single-input-single-output (SISO) systems
- ▶ Because we're mechanical engineers, we apply our knowledge particularly to motion systems



What's so special about motion systems?

- ▶ they behave (fairly) linear
- ▶ they can be of very high order
- ▶ they're often hardly damped
 - they can have very sharp resonances
- ▶ they react fast, and can be sampled at a high rate
 - you can collect lot's of measurement data in a very short time

The theory of this course (4CM00) is

- ▶ in principle *generally* applicable to any linear system
- ▶ *particularly* useful for motion systems

Prerequisites; or things you
should already know...

We consider dynamic systems with input u and output y

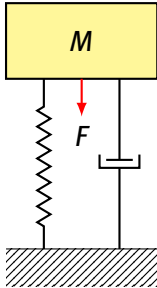
- ▶ The output of a dynamic system is *time dependent* and can be manipulated by its input
 - It can be interpreted as an operation on *signals*
 - It transforms signal $u(t)$ into another signal $y(t)$
- ▶ A dynamic system can be described by a *differential equation*

$$F\left(t, u, \frac{du}{dt}, \frac{d^2u}{dt^2}, \dots, \frac{d^m u}{dt^m}, y, \frac{dy}{dt}, \frac{d^2y}{dt^2}, \dots, \frac{d^n y}{dt^n}\right) = 0$$

$$F\left(t, u, \dot{u}, \ddot{u}, \dots, u^{(m)}, y, \dot{y}, \ddot{y}, \dots, y^{(n)}\right) = 0$$

- ▶ A dynamic system is *linear* if F is linear in its arguments, i.e. if the differential equation is *linear* in the (derivatives of) u and y

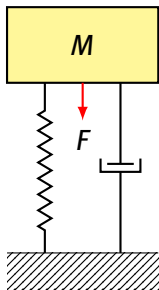
$$a_n y^{(n)} + a_{n-1} y^{(n-1)} + \dots + a_2 \ddot{y} + a_1 \dot{y} + a_0 y = \\ b_m u^{(m)} + b_{m-1} u^{(m-1)} + \dots + b_2 \ddot{u} + b_1 \dot{u} + b_0 u$$



Mass supported by a spring and damper

$$M\ddot{x} + d\dot{x} + kx = F$$

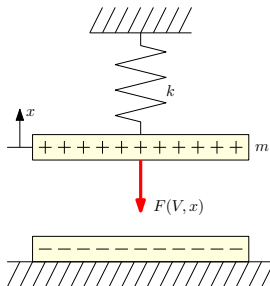
- ▶ Input $u = F$, output $y = x$
- ▶ Linear system



Mass supported by a spring and damper

$$M\ddot{x} + d\dot{x} + kx = F$$

- ▶ Input $u = F$, output $y = x$
- ▶ Linear system

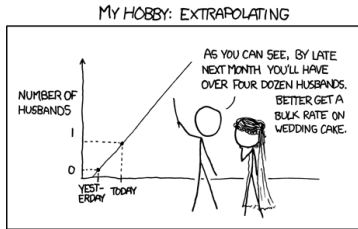


Charged plate suspended on a spring

$$m\ddot{x} + kx = -\alpha \frac{V^2}{(d_0 + x)^2}$$

- ▶ Input $u = V$, output $y = x$
- ▶ Non-linear system (both in u and in y)

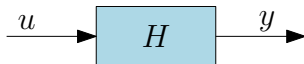
All systems are non-linear...



...but in this course we focus on *linear* systems only. Why?

- ▶ Most systems behave linearly in their relevant operating range
- ▶ Most solutions are linear
 - linear controllers often do the job
- ▶ Linear systems are relatively easy to work with!
 - they have very useful properties

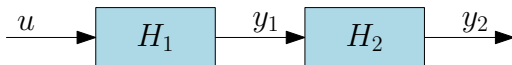
Linear systems have a unique *frequency response*



- ▶ when $u = \cos(\omega t)$, then $y = A \cos(\omega t + \phi)$
 - amplitude A and phase ϕ only depend on frequency ω
 - no other frequencies are induced

Linear systems have a superposition property:

1. When $u_1 \rightarrow y_1$ and $u_2 \rightarrow y_2$,
then $\alpha u_1 + \beta u_2 \rightarrow \alpha y_1 + \beta y_2$ for any (real) α and β
2. Consider the series interconnection



When $u = \cos(\omega t)$, then $y_1 = A_1 \cos(\omega t + \phi_1)$ and
 $y_2 = A_2 A_1 \cos(\omega t + \phi_1 + \phi_2)$

Note that a *system* is actually an operation on one or more *signals*

- ▶ A time-domain signal $x(t)$ can be transformed to s -domain via the Laplace transform:

$$X(s) = \mathcal{L}\{x(t)\} = \int_0^{\infty} e^{-st} x(t) dt \quad (1)$$

- ▶ Consequently, time derivatives of signals can be expressed as:

$$\begin{aligned} \mathcal{L}\{\dot{x}(t)\} &= \int_0^{\infty} e^{-st} \dot{x}(t) dt = [e^{-st} x(t)]_0^{\infty} + s \int_0^{\infty} e^{-st} x(t) dt \\ &= sX(s) - x(0) \end{aligned} \quad (2)$$

$$\mathcal{L}\{\ddot{x}(t)\} = s^2 X(s) - sx(0) - \dot{x}(0) \quad (3)$$

- ▶ Since we're often only interested in input–output behavior, the transients $x(0)$ and $\dot{x}(0)$ are often omitted.

Time domain $x(t)$	Laplace domain $X(s)$
$\delta(t)$	1
$1(t)$	$1/s$
t	$1/s^2$
t^2	$2!/s^3$
t^m	$m!/s^{m+1}$
e^{-at}	$\frac{1}{s+a}$
te^{-at}	$\frac{1}{(s+a)^2}$
$\frac{1}{(m-1)!} t^{m-1} e^{-at}$	$\frac{1}{(s+a)^m}$
$\sin(\omega t)$	$\frac{\omega}{s^2+\omega^2}$
$\cos(\omega t)$	$\frac{s}{s^2+\omega^2}$
$e^{-at} \cos(\omega t)$	$\frac{s+a}{(s+a)^2+\omega^2}$
$e^{-at} \sin(\omega t)$	$\frac{\omega}{(s+a)^2+\omega^2}$

- ▶ Consider the example

$$\ddot{y} + 2\dot{y} + 25y = u$$

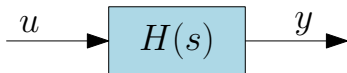
- ▶ Using the Laplace transform, this can be written as

$$s^2 Y(s) + 2sY(s) + 25Y(s) = U(s)$$

- ▶ The ratio between $U(s)$ and $Y(s)$ defines the transfer function $H(s)$

$$H(s) = \frac{Y(s)}{U(s)} = \frac{1}{s^2 + 2s + 25}$$

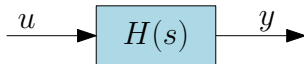
- ▶ For linear systems the transfer function is always *rational*
 - i.e. both numerator and denominator are polynomials in s



The use of the Laplace transform (1)

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How to determine the output $y(t)$ for a given $u(t)$?



- ▶ In time domain you'd need to solve the differential equation
- ▶ Or calculate the convolution

$$y(t) = h(t) \otimes u(t) = \int_{-\infty}^{\infty} h(\tau) u(t - \tau) d\tau$$

where $h(\tau)$ is the *impulse response* of the system $H(s)$.

It's much easier via the Laplace domain (at least for linear systems)

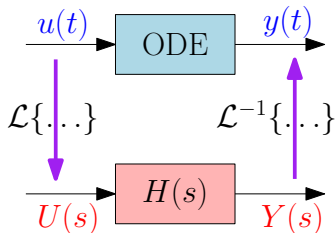
- ▶ Convolution simplifies to

$$Y(s) = \mathcal{L}\{y(t)\} = \mathcal{L}\{h(t) \otimes u(t)\} = H(s) \cdot U(s)$$

The use of the Laplace transform (2)

13/59

Hence, $y(t) = \mathcal{L}^{-1} \{H(s) \cdot \mathcal{L}\{u(t)\}\}$, or schematically

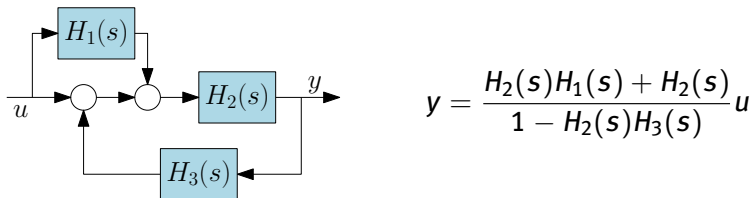
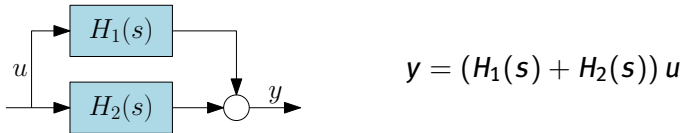
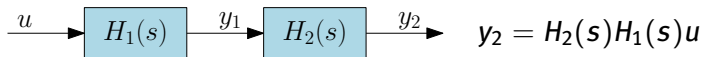


Simple example: $H(s) = \frac{3}{s+3}$ with step input $u(t) = 1(t)$

- ▶ $U(s) = 1/s$
- ▶ $Y(s) = \frac{3}{s(s+3)} = \frac{1}{s} - \frac{1}{s+3}$
- ▶ $y(t) = 1 - e^{-3t}$

Note: Impulse response $h(t) = 3e^{-3t}$

Linear systems + Laplace domain = easy combinations



Note: in this course we only consider SISO systems. For MIMO systems the order of multiplication matters, i.e. $H_2H_1 \neq H_1H_2$!

Transfer functions: time domain

What are poles and zeros?

- ▶ Poles: solutions of “denominator = 0”
- ▶ Zeros: solutions of “numerator = 0”

Example:

$$H(s) = \frac{s + 16}{(s + 2)(s + 4)}$$

- ▶ Two poles: one at $s = -2$, one at $s = -4$
- ▶ One zero: at $s = -16$

But what do these poles and zeros mean?

To understand, consider a step response of the same system.

- ▶ $U(s) = \mathcal{L}\{1(t)\} = \frac{1}{s}$
- ▶ $Y(s) = H(s)U(s) = \frac{s+16}{s(s+2)(s+4)}$

Using partial fraction expansion, we can write this output as

$$Y(s) = \frac{2}{s} - \frac{3.5}{s+2} + \frac{1.5}{s+4}$$

The step response in time domain thus becomes

$$y(t) = \mathcal{L}^{-1}\{Y(s)\} = [2 - 3.5e^{-2t} + 1.5e^{-4t}] \cdot 1(t)$$

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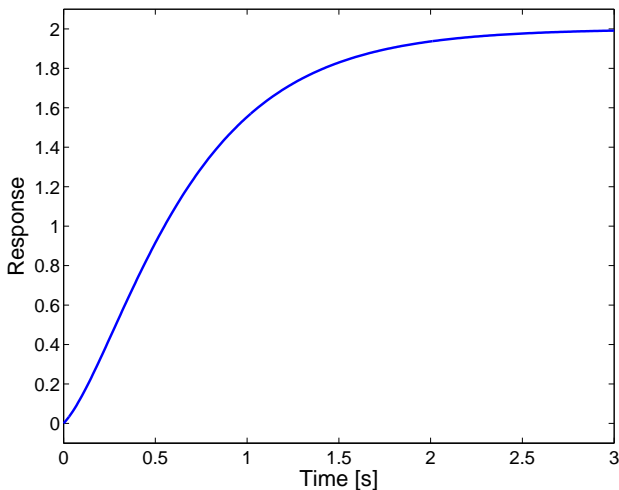
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Note: the exponents are the same as the **poles** at $s = -2$ and $s = -4$.



$$y(t) = [2 - 3.5e^{-2t} + 1.5e^{-4t}] \cdot 1(t)$$

Another example:

$$H(s) = \frac{s + 20}{s^2 + 4s + 40}$$

- ▶ Poles: at $s = -2 - 6j$ and $s = -2 + 6j$.
- ▶ Zero: at $s = -20$

With a step input $U(s) = \frac{1}{s}$ the output becomes

$$\begin{aligned} Y(s) &= \frac{s+20}{s(s^2+4s+40)} \\ &= \frac{0.5}{s} - \frac{0.5s+1}{s^2+4s+40} = 0.5\frac{1}{s} - 0.5\frac{s+2}{(s+2)^2+6^2} \\ y(t) &= \mathcal{L}^{-1}\{Y(s)\} = [0.5 - 0.5e^{-2t}\cos(6t)] \cdot 1(t) \end{aligned}$$

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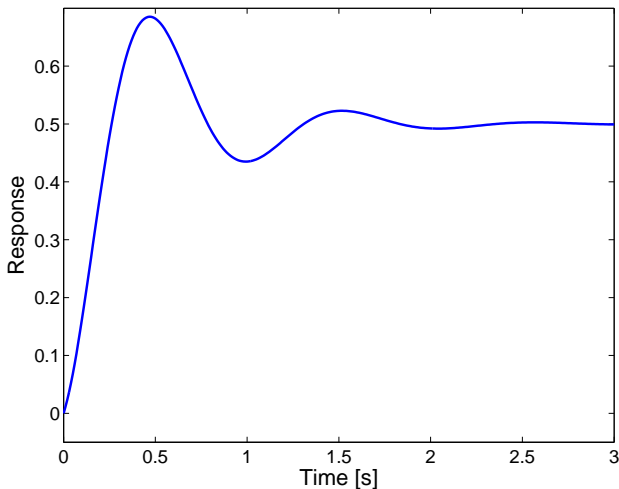
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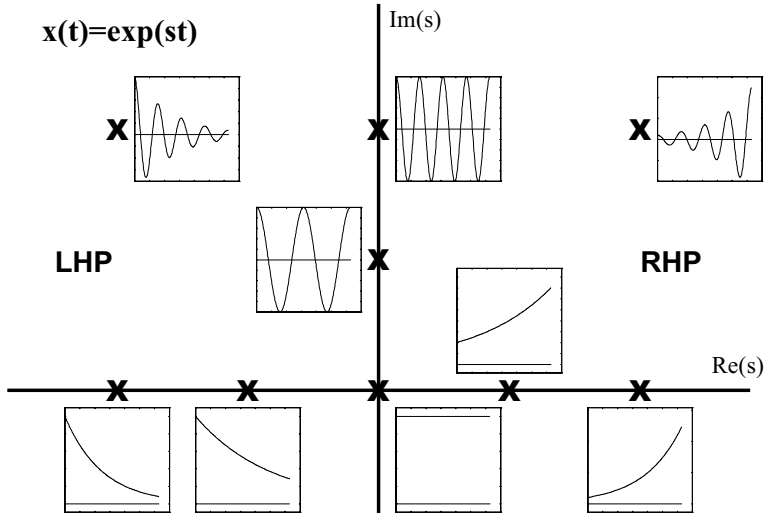
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- ▶ **exponent** is real part of the pole, **frequency** is its imaginary part
- ▶ system is stable iff *all* exponents decay \Leftrightarrow *all* poles are in LHP



$$y(t) = [0.5 - 0.5e^{-2t} \cos(6t)] \cdot 1(t)$$



The effect of the zero is much more subtle

- ▶ determines details of the time response
 - i.e. the gain of each exponent
 - i.e. how much each exponent contributes to the total response
- ▶ most evident when located in the right-half-plane (RHP)

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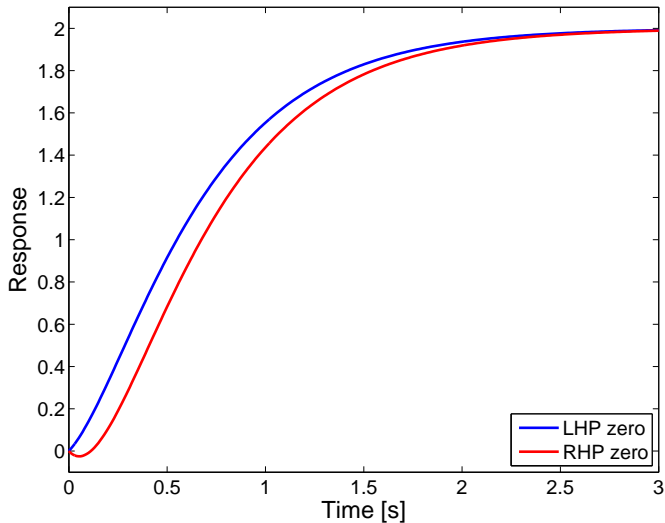
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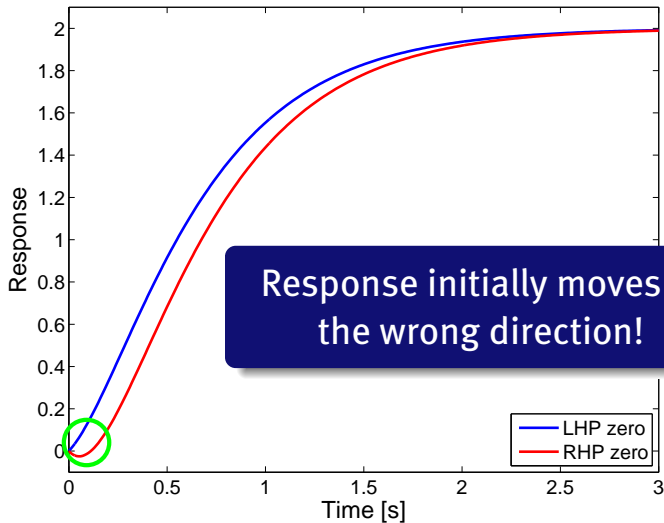
Example:

$$H(s) = \frac{16 - s}{(s + 2)(s + 4)}$$

With a step input $U(s) = \frac{1}{s}$ the output becomes

$$\begin{aligned} Y(s) &= \frac{16-s}{s(s+2)(s+4)} \\ &= \frac{2}{s} - \frac{4.5}{s+2} + \frac{2.5}{s+4} \\ y(t) &= \mathcal{L}^{-1}\{Y(s)\} = [2 - 4.5e^{-2t} + 2.5e^{-4t}] \cdot 1(t) \end{aligned}$$





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With a step input $U(s) = \frac{1}{s}$ the output becomes

$$\begin{aligned} Y(s) &= \frac{s+20}{s(s^2+4s+40)} = \frac{0.5}{s} - \frac{0.5s+3}{s^2+4s+40} \\ &= \frac{1}{2} \cdot \frac{1}{s} - \frac{1}{2} \cdot \frac{s+2}{(s+2)^2 + 6^2} - \frac{1}{3} \cdot \frac{6}{(s+2)^2 + 6^2} \\ y(t) &= \mathcal{L}^{-1}\{Y(s)\} \\ &= \left[\frac{1}{2} - \frac{1}{2}e^{-2t} \cos(6t) - \frac{1}{3}e^{-2t} \sin(6t) \right] \cdot 1(t) \end{aligned}$$

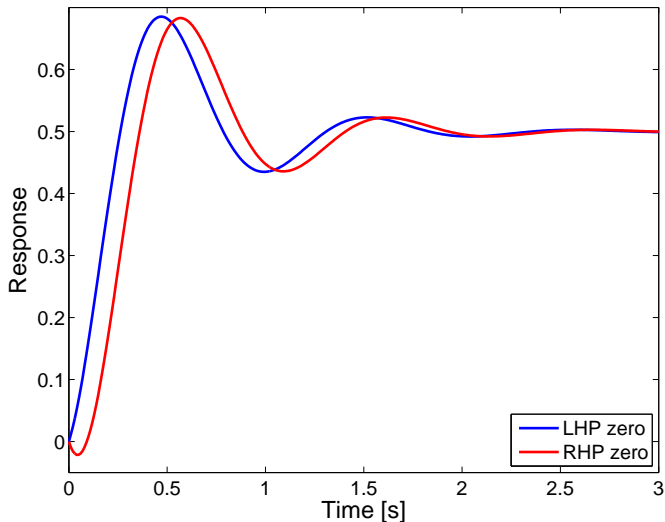
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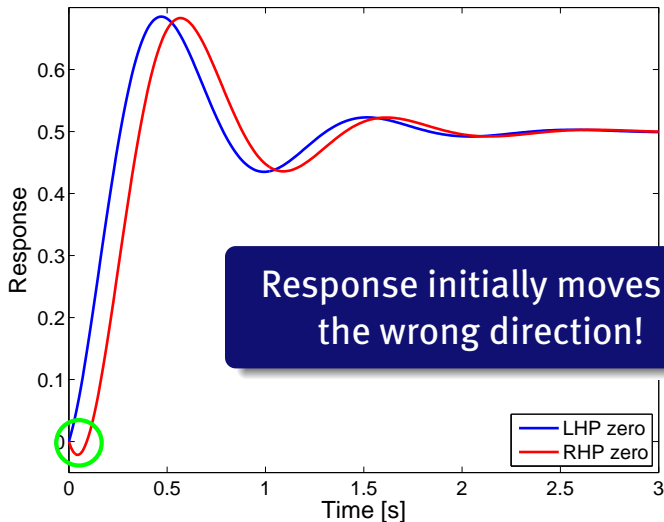
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What happens if poles and zeros are close to each other?

- ▶ Example:

$$H(s) = \frac{s + 1.92}{(s + 2)(s + 4)}$$

- ▶ Zero at $s = -1.92$ is very close to pole at $s = -2$

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With a step input $U(s) = \frac{1}{s}$ the output becomes

$$\begin{aligned} Y(s) &= \frac{s + 1.92}{s(s + 2)(s + 4)} \\ &= \frac{0.24}{s} + \frac{0.02}{s + 2} - \frac{0.26}{s + 4} \\ y(t) &= \mathcal{L}\{Y(s)\} = [0.24 + \textcolor{red}{0.02}e^{-2t} - 0.26e^{-4t}] \cdot 1(s) \end{aligned}$$

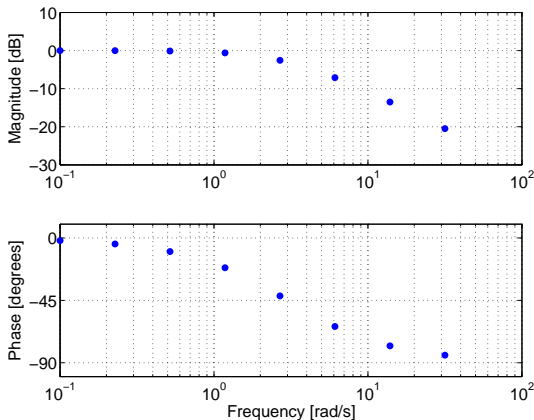
Hence, the pole at $\textcolor{red}{s} = -\textcolor{red}{2}$ hardly contributes to the time response.

Transfer functions: frequency domain

Remember: linear systems have a unique *frequency response*

- ▶ when $u = \cos(\omega t)$, then $y = A \cos(\omega t + \phi)$

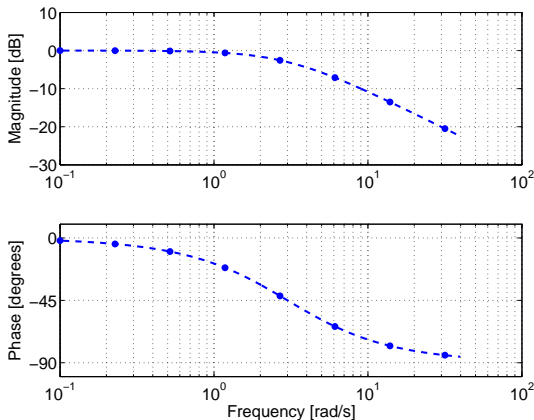
Hence, there is a unique *frequency response function* (FRF) $H(j\omega)$



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Hence, there is a unique *frequency response function* (FRF) $H(j\omega)$



Frequency response $H(j\omega)$ is often represented in a *Bode diagram*

- ▶ considers *steady state* at each frequency
- ▶ gives amplitude amplification and phase shift at ω_1 in steady state
- ▶ superposition: also valid for a combination of frequencies ω_i

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- ▶ superposition: also valid for a combination of frequencies ω_i

FRF follows directly from the transfer function $H(s)$

- ▶ by substitution of $s = j\omega$ into $H(s)$
- ▶ yields a complex valued function $H(j\omega)$
- ▶ then $|H(j\omega)| = A(\omega)$ and $\angle H(j\omega) = \phi(\omega)$

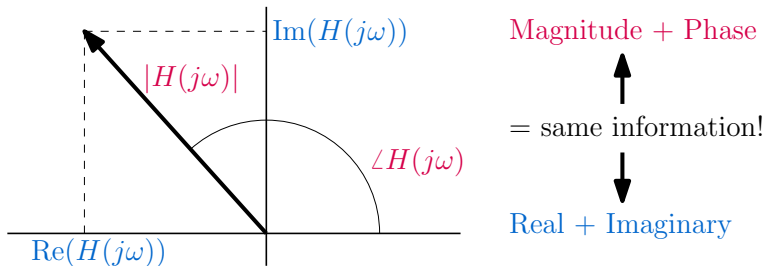
Example: $H(s) = \frac{3}{s+3}$

- ▶ substituting $s = j\omega$ yields

$$H(j\omega) = \frac{3}{j\omega + 3} = \underbrace{\frac{9}{\omega^2 + 9}}_{\text{real part}} - \underbrace{\frac{3j\omega}{\omega^2 + 9}}_{\text{imaginary part}}$$

Hence, a frequency response function (FRF) is a *complex function*!

- ▶ Frequency response can also be drawn in the complex plane!



However, note that

- ▶ all input/output signals are of course still real valued
- ▶ the complex number represents the *operation* on these signals

- ▶ frequency axis is logarithmic
- ▶ magnitude plot (vertical axis) is also logarithmic
 - amplitude A is expressed in decibel (dB):

$$\text{dB} = 20 \cdot \log_{10} A$$

- amplitude axis is thus linear on dB-scale
- ▶ consequence: simple powers ω^x are *straight lines*
 - E.g. mass line:

$$\begin{aligned}\log |H| &= \log \left| -\frac{1}{m\omega^2} \right| \\ &= \log(1/m) - 2 \cdot \log \omega\end{aligned}$$

- ▶ these asymptotes are therefore characterized by their *slopes*

Another example: $H(s) = \frac{1}{s^2 + 2s + 25}$

- ▶ substituting $s = j\omega$ yields

$$H(j\omega) = \frac{1}{25 - \omega^2 + 2j\omega} = \underbrace{\frac{25 - \omega^2}{(25 - \omega^2)^2 + 4\omega^2}}_{\text{real part}} - \underbrace{\frac{2j\omega}{(25 - \omega^2)^2 + 4\omega^2}}_{\text{imaginary part}}$$

But how to easily construct the Bode diagram from this?

The FRF approximates certain asymptotes, e.g.

- ▶ $\lim_{s \rightarrow 0} H(s) = \frac{1}{25}$ hence $\lim_{\omega \rightarrow 0} H(j\omega) = \frac{1}{25} \Rightarrow 0\text{-slope}$
- ▶ $\lim_{s \rightarrow \infty} H(s) = \frac{1}{s^2}$ hence $\lim_{\omega \rightarrow \infty} H(j\omega) = -\frac{1}{\omega^2} \Rightarrow -2\text{-slope}$

And what happens in between?

Complex-valued poles (or zeros) have an (anti-)resonance with *damping*

Normalized second-order denominator:

$$H(s) = \frac{\omega_n^2}{s^2 + 2\xi\omega_n s + \omega_n^2}$$

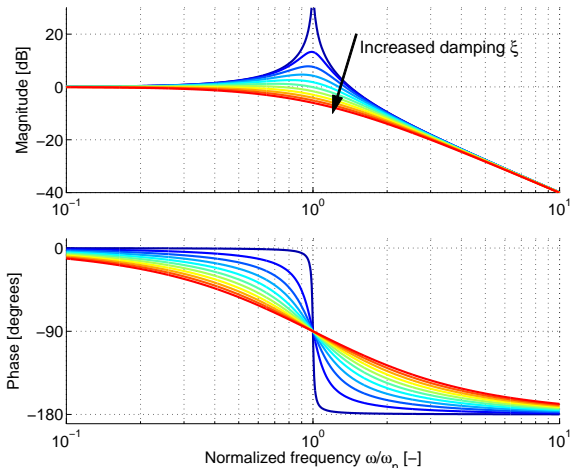
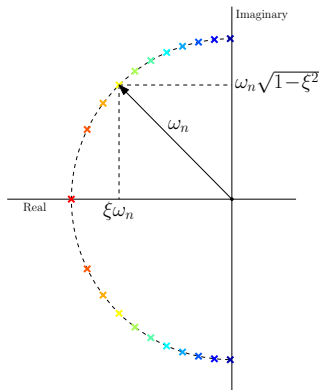
- ▶ ω_n : undamped natural frequency
- ▶ ξ : normalized damping

Poles are located at (using ABC-formula):

$$s = \underbrace{-\xi\omega_n}_{\text{exponential decay}} \pm j \underbrace{\omega_n \sqrt{1 - \xi^2}}_{\text{oscillation frequency } \omega_d}$$

- ▶ for small damping $\omega_d \approx \omega_n$
- ▶ for $\xi \geq 1$ both poles are real

$$H(s) = \frac{\omega_n^2}{s^2 + 2\xi\omega_n s + \omega_n^2}$$

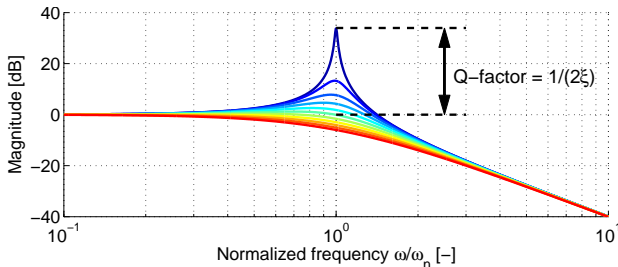


How to estimate the damping from a Bode diagram?

- By evaluating the magnitude at the eigenfrequency ω_n , since

$$|H(j\omega_n)| = \left| \frac{\omega_n^2}{-\omega_n^2 + 2\xi\omega_n j\omega_n + \omega_n^2} \right| = \left| \frac{1}{j \cdot 2\xi} \right| = \frac{1}{2\xi}$$

- This is known as the *Q-factor* of a resonance.
- The damping is then $\xi = \frac{1}{2Q}$.

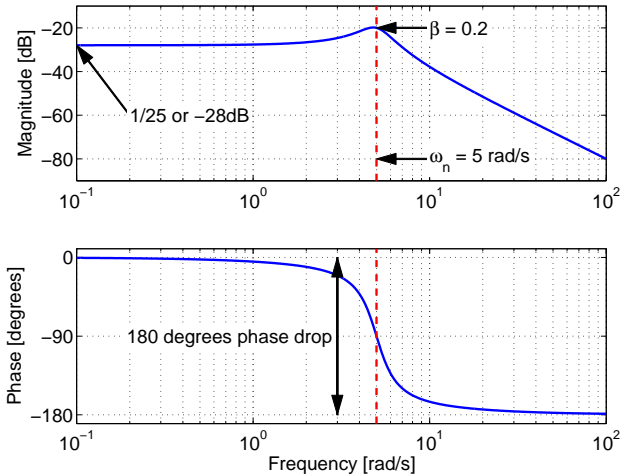


Draw the frequency response function $H(j\omega)$

▶ for $H(s) = \frac{1}{s^2 + 2s + 25}$

Draw the frequency response function $H(j\omega)$

► for $H(s) = \frac{1}{s^2 + 2s + 25}$



A transfer function

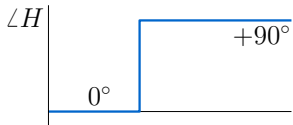
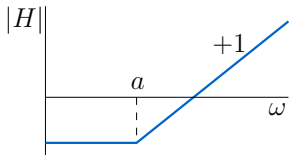
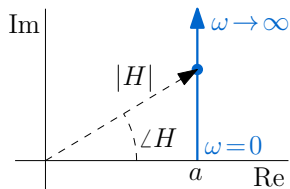
- ▶ consists of only **first** and **second** order terms in both **numerator** and **denominator**
- ▶ thus has **real** and **complex** valued **zeros** and **poles**

The shape of the Bode diagram is determined by the location of the poles and zeros of the transfer function

- ▶ a pole/zero at $s = \lambda$ manifests itself at frequency $f = |\lambda|$ rad/s
- ▶ at every zero the slope increases +1
- ▶ at every pole the slope decreases -1
- ▶ the phase depends on LHP or RHP

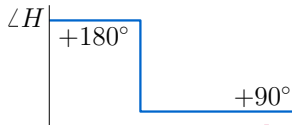
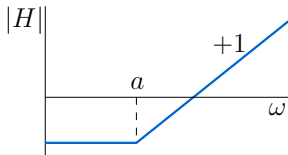
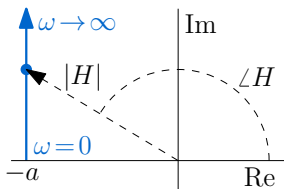
$$H(s) = s + a$$

$$H(j\omega) = j\omega + a$$



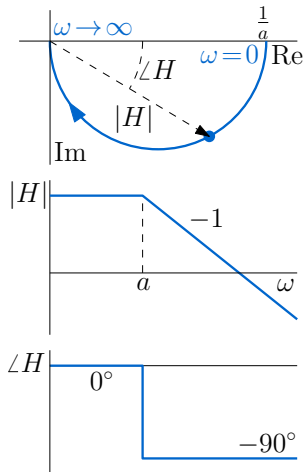
$$H(s) = s - a$$

$$H(j\omega) = j\omega - a$$



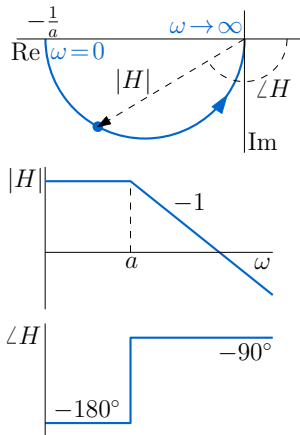
$$H(s) = \frac{1}{s+a}$$

$$H(j\omega) = \frac{1}{j\omega+a} = -\frac{j\omega-a}{\omega^2+a^2}$$



$$H(s) = \frac{1}{s-a}$$

$$H(j\omega) = \frac{1}{j\omega-a} = -\frac{j\omega+a}{\omega^2+a^2}$$



Summarizing:

	Magnitude	Phase
LHP pole	slope decrease: -1	<i>phase lag: -90°</i>
RHP pole	slope decrease: -1	<i>phase lead: $+90^\circ$</i>
LHP zero	slope increase: $+1$	<i>phase lead: $+90^\circ$</i>
RHP zero	slope increase: $+1$	<i>phase lag: -90°</i>

Note: For complex conjugated poles or zeros (i.e. *two poles* or *two zeros*) the effect is doubled.

If all poles and zeros are in LHP we have:

$$\text{angle}(H) = \text{slope}(H) \times 90^\circ$$

- ▶ known as Bode gain-phase relation

Does not hold for:

- ▶ unstable systems (RHP poles)
- ▶ non-minimum phase systems (RHP zeros)

Fortunately, most *motion* systems are stable and minimum phase:

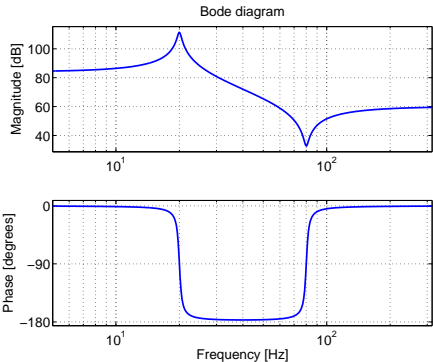
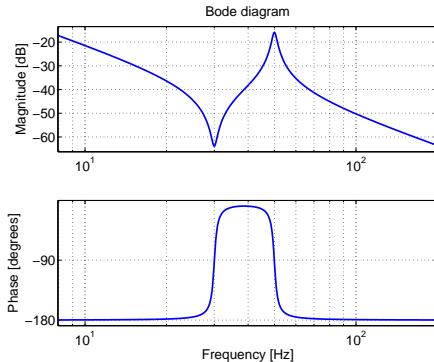
- ▶ simplifies stability analysis
- ▶ simplifies controller design

Note1: Poles or zeros in RHP can easily be identified from the Bode diagram.

Note2: Often, you can even determine stability by close examination of the Bode diagram.

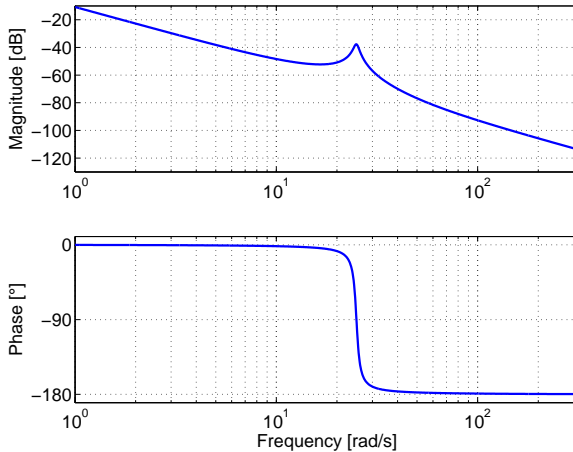
Poles, zeros and resonances can be read from the Bode diagram.

Some examples:



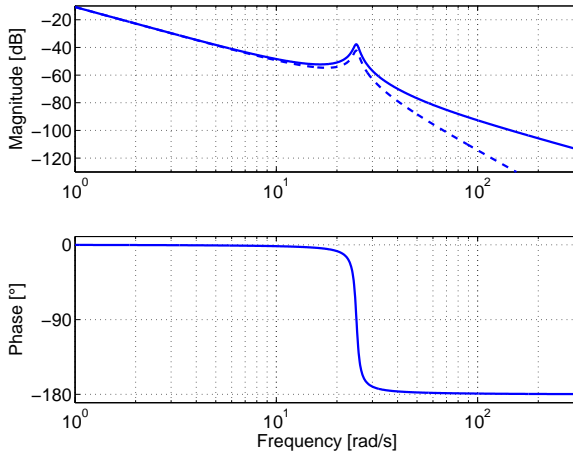
Note: $\angle(H) = \text{slope}(H) \times 90^\circ$

Another example:



How many poles and zeros are there, and are they LHP or RHP?

Another example:



How many poles and zeros are there, and are they LHP or RHP?

Bode contains more information:

- ▶ Amplitude for $\omega \rightarrow 0$ gives the steady state gain in the output for a *constant* input
- ▶ Amplitude for $\omega \rightarrow \infty$ denotes the *direct feedthrough*: it denotes the output gain *at* a step on the input

The first one obviously follows from the *Final Value Theorem*:

$$\lim_{t \rightarrow \infty} y(t) = \lim_{s \rightarrow 0} sY(s)$$

- ▶ for a step input $U(s) = \frac{1}{s}$ this yields

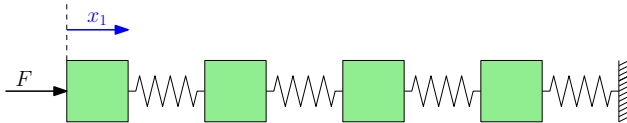
$$\lim_{t \rightarrow \infty} y(t) = \lim_{s \rightarrow 0} s \frac{H(s)}{s} = H(0).$$

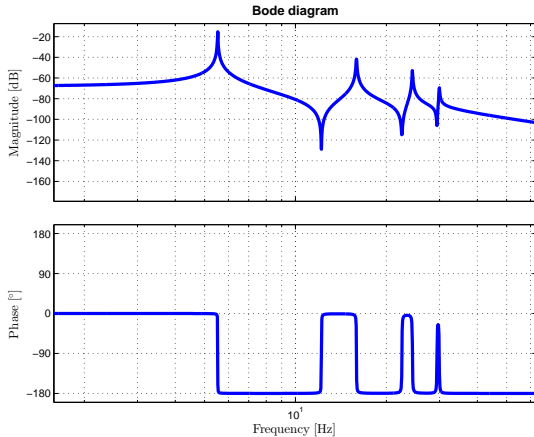
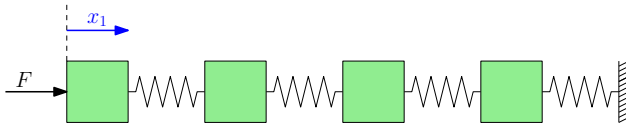
Note: This holds for any transfer function, including closed loops!

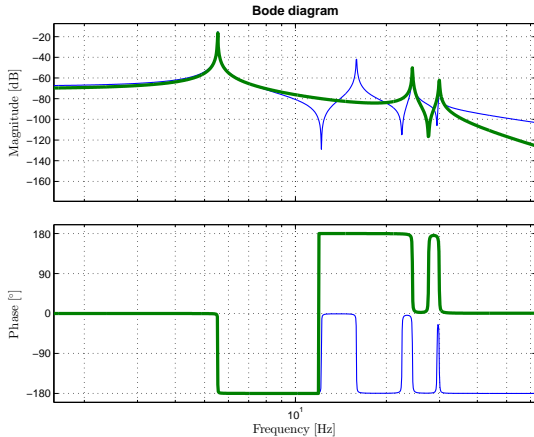
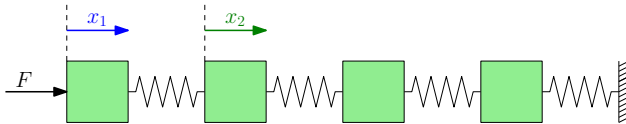
Poles and zeros in physical systems

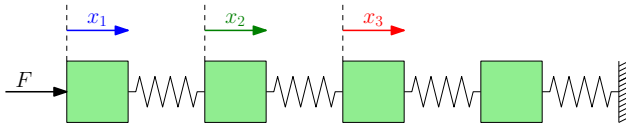
Poles and zeros: 1-D example

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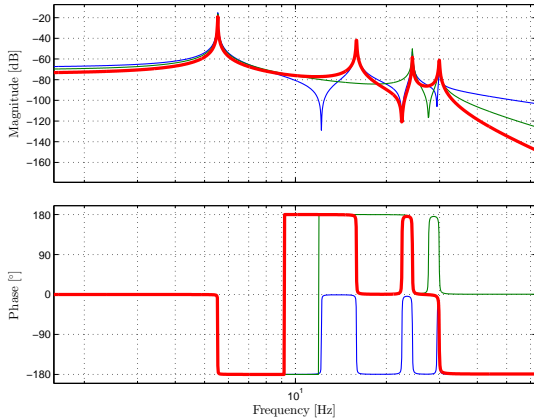


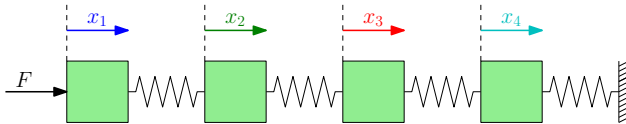




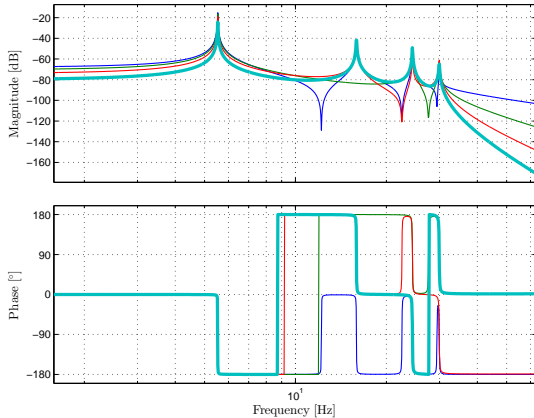


Bode diagram





Bode diagram



Summarizing, the example shows that

- ▶ all plants had exactly the same resonances, regardless of sensor position;
- ▶ each plant had different anti-resonances; even the number of anti-resonances varies.

This is an important fundamental difference between poles and zeros.

- ▶ Poles are a *system property*
- ▶ Zeros are the result of specific actuator and/or sensor placement

Note that zeros can make poles *unobservable* or *uncontrollable*, but they cannot remove or shift poles.

This can be understood by looking at the State-Space (SS) notation of dynamical systems.

Most general form of linear dynamical system is State Space (SS):

$$\dot{x} = Ax + Bu$$

$$y = Cx + Du$$

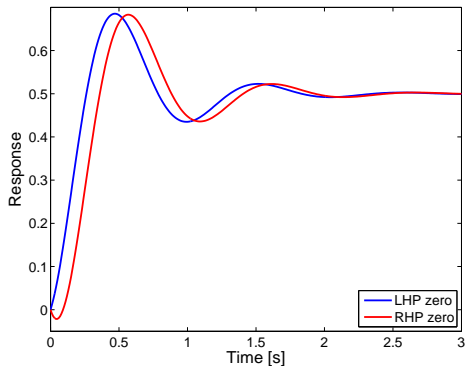
- ▶ The transfer function is then given by

$$Y(s) = \underbrace{\left[C (sI - A)^{-1} B + D \right]}_{H(s)} \cdot U(s)$$

- ▶ The poles of the system are the eigenvalues of A
- ▶ The zeros follow from B , C and D , which are determined by the sensor and actuator locations
- ▶ System matrix A is unaffected by sensor/actuator locations!

Here we only use the Transfer Function (TF) representation $H(s)$.

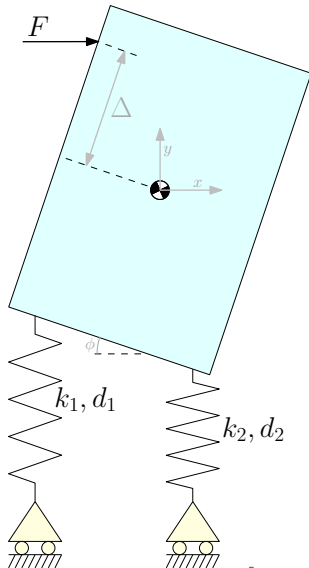
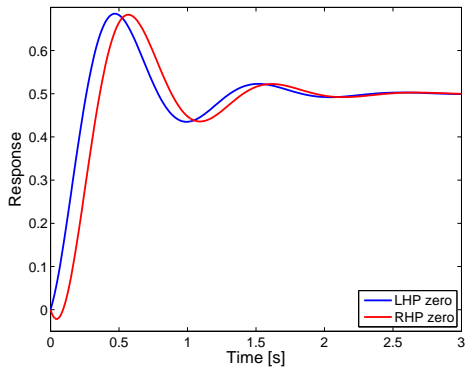
Remember how we recognized RHP zeros in the time domain?



When do we get RHP zeros in practice?

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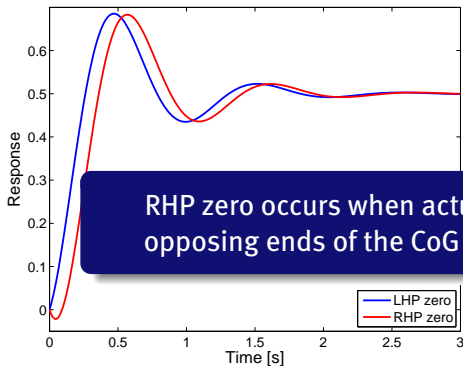
Remember how we recognized RHP zeros in the time domain?



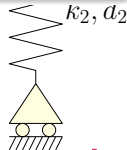
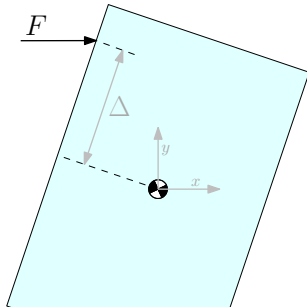
When do we get RHP zeros in practice?

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Remember how we recognized RHP zeros in the time domain?



RHP zero occurs when actuator and sensor are on opposing ends of the CoG with sufficient distance



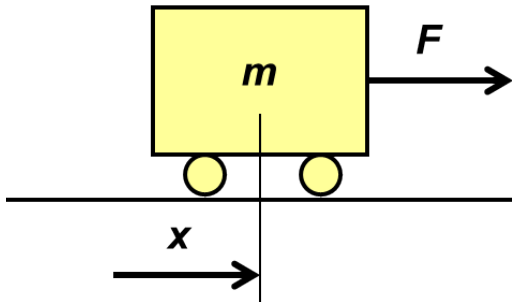
Other classic examples:

- ▶ Throwing coal on a fire: temperature first drops before it rises
- ▶ Parallel parking: car front initially goes to the left when you steer to the right

But what's wrong with RHP zeros?

- ▶ It takes a while before the system moves in the right direction
- ▶ Time domain: the response effectively becomes 'slower' or delayed
- ▶ Frequency domain: the system loses extra phase (non-minimum phase)
- ▶ A controller has to 'wait' for that delayed response
- ▶ Hence it limits the achievable bandwidth
 - will become clear when we look at it in frequency domain (Nyquist)

Basic controller design



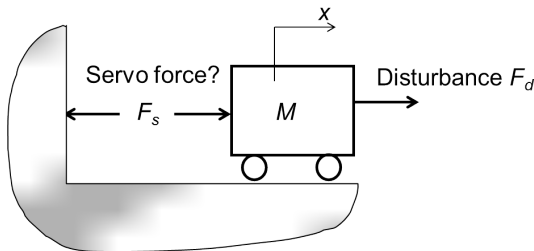
$$F = m \cdot a = m\ddot{x}$$

\Downarrow

$$X(s) = \frac{1}{ms^2} \cdot F(s)$$

Problem: there is a disturbance force F_d on the mass

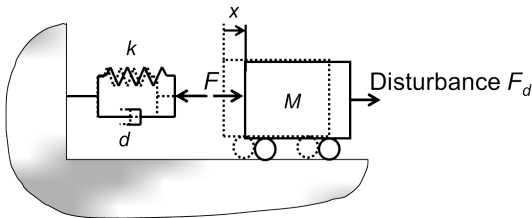
Goal: how to maintain a certain position x in the presence of this disturbance?



In other words, what must F_s look like?

Mechanical solution:

- ▶ make a stiff connection between mass and base

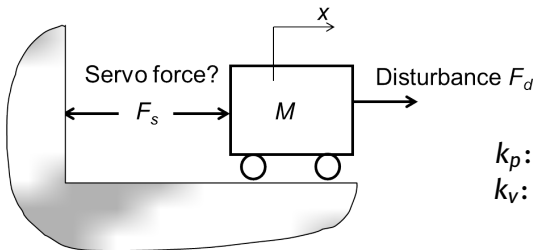


Resulting force on mass: $F = -k \cdot x - d \cdot \dot{x}$

Resulting eigenfrequency: $f = \frac{1}{2\pi} \sqrt{\frac{k}{M}}$

Electrical solution / servo analogy:

- ▶ measure distance x between mass and base
- ▶ compute servo force based on x and its velocity \dot{x}
 - ⇒ PD-controller



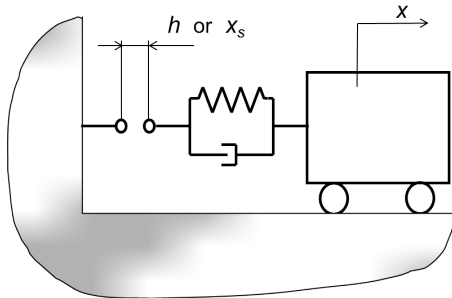
k_p : servo stiffness
 k_v : servo damping

Resulting force on mass: $F_s = -k_p \cdot x - k_v \cdot \dot{x}$

Resulting eigenfrequency: $f = \frac{1}{2\pi} \sqrt{\frac{k_p}{M}}$

To follow a non-zero setpoint:

- ▶ make a 'stiff connection' between x and the setpoint

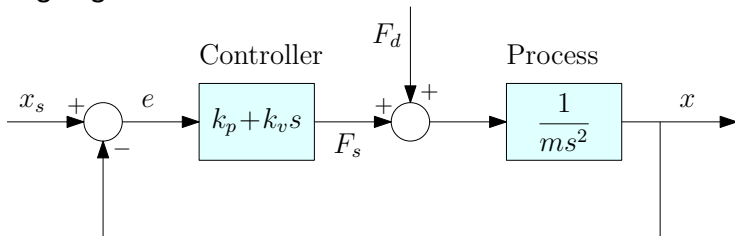


Mechanical: $F = k(h - x) + d(\dot{h} - \dot{x})$

Servo: $F_s = k_p(x_s - x) + k_v(\dot{x}_s - \dot{x}) = k_p e + k_v \dot{e}$

What are we actually doing?

⇒ designing a PD-controller!



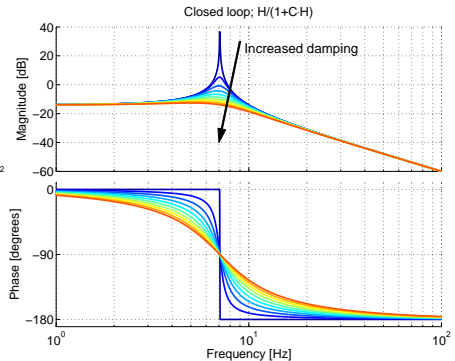
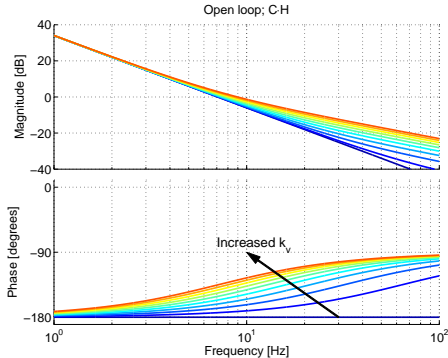
What is the transfer function from F_d to x ?

Open loop:

$$\frac{x}{F_d} = \frac{1}{ms^2}$$

Closed loop:

$$\frac{x}{F_d} = \frac{\frac{1}{ms^2}}{1 + \frac{k_p + k_v s}{ms^2}} = \frac{1}{ms^2 + k_v s + k_p}$$



- ▶ all lecture slides
- ▶ book: “Feedback control of dynamic systems” (Franklin, Powell, Emami-Naeini)
- ▶ document: “Summary_Control_Engineering”
- ▶ document: “Tips and tricks”

