

Linear Systems (034032)

lecture no. 9

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Previously on Linear Systems . . .

Let G be a stable continuous-time LTI system with the impulse response g . Its frequency response $G(j\omega) = (\mathfrak{F}\{g\})(j\omega)$ or

$$G(j\omega) = G(s)|_{s=j\omega} = (\mathfrak{L}\{g\})(j\omega).$$

In the discrete-time case, the frequency response $G(e^{j\theta}) = (\mathfrak{F}\{g\})(e^{j\theta})$ or

$$G(e^{j\theta}) = G(z)|_{z=e^{j\theta}} = (\mathfrak{Z}\{g\})(e^{j\theta}).$$

The frequency response shapes

- the response to harmonic inputs
- the response to periodic inputs
- the steady-state response to test sine wave input

Outline

Systems as filters I

Frequency response plots

Systems as filters II

Outline

Systems as filters I

Frequency response plots

Systems as filters II

Frequency-domain response of LTI systems

By the convolution property of the Fourier transform,

$$y(t) = (Gu)(t) \iff Y(j\omega) = G(j\omega)U(j\omega)$$

whenever the corresponding Fourier transforms exist. Hence, the frequency response of G just scales every harmonic component of the input, so that

- harmonics with frequencies ω at which $|G(j\omega)| > 1$ are amplified
- harmonics with frequencies ω at which $|G(j\omega)| < 1$ are attenuated
- harmonics with frequencies ω at which $|G(j\omega)| < 1/\sqrt{2}$ do not pass
- this is a convention, facilitating categorical conclusions; take it with a grain of salt

This multiplication property

- facilitates the use of LTI system as filters, whose task is to shape the spectrum of signals of interest (to pass “desired” components and to block “unwanted” ones).

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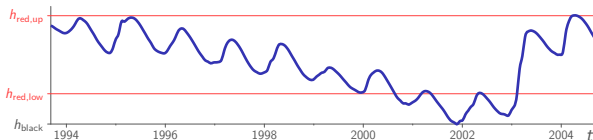
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Example

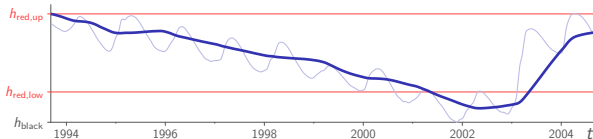
Kinneret water level h from Sep 1993 to Sep 2004



If processed by the finite-memory integrator, so that $h_a = G_{fmi,1}h$ or

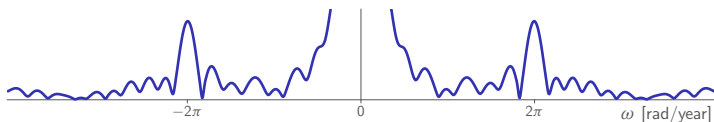
$$h_a(t) = \int_{t-1}^t h(s) ds \iff H_a(s) = \frac{1 - e^{-s}}{s} H(s)$$

(average over the last year), we end up with



Example: frequency-domain insight

The amplitude spectrum of h (with $h_{\text{red,up}}$ taken as zero) is

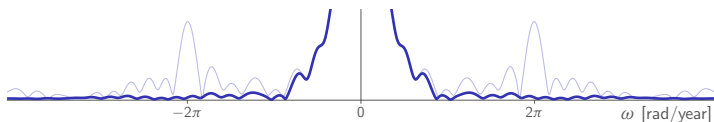


The frequency response magnitude of $G_{\text{fmi},1}$,

$$|G_{\text{fmi},1}(j\omega)| = \left| \frac{1 - e^{-j\omega}}{j\omega} \right| = \left| \text{sinc}\left(\frac{\omega}{2}\right) \right|$$

A plot of the magnitude of the frequency response $|G_{\text{fmi},1}(j\omega)|$. The horizontal axis is labeled ω and has tick marks at 0 , 2π , and 4π . The vertical axis has a tick mark at 1 . The plot shows a sinc function that starts at 1 at $\omega = 0$ and has zeros at $\omega = 2\pi$ and $\omega = 4\pi$.

is zero at $\omega = 2\pi$, so the peaks at $\omega = \pm 2\pi$ [rad/year] are filtered out,



eliminating the effect of annual cycles.

Classification

Depending on their purpose, filters may be categorized as

- **low-pass filters** allow only harmonics with $\omega \leq \omega_b$ to pass
i.e. $|G(j\omega)| \geq \frac{1}{\sqrt{2}} \iff \omega \leq \omega_b$, which is known as the **bandwidth** of G
- **high-pass filters** allow only harmonics with $\omega \geq \omega_c$ to pass
i.e. $|G(j\omega)| \geq \frac{1}{\sqrt{2}} \iff \omega \geq \omega_c$, which is known as the **cutoff frequency** of G
- **band-pass filters** allow only harmonics with $\omega_1 \leq \omega \leq \omega_2$ to pass
i.e. $|G(j\omega)| \geq \frac{1}{\sqrt{2}} \iff \omega \in [\omega_1, \omega_2]$
- **band-stop filters** allow only harmonics with $\omega \leq \omega_1$ and $\omega \geq \omega_2$ to pass
i.e. $|G(j\omega)| \geq \frac{1}{\sqrt{2}} \iff \omega \notin (\omega_1, \omega_2)$, which is known as the **stopband** of G

The question is

- what are systems, whose frequency responses are such filters?

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Ideal filters

- ideal low-pass has

$$F_{\text{ilp}, \omega_b}(j\omega) = \text{rect}_{2\omega_b}(\omega) \iff f_{\text{ilp}, \omega_b}(t) = \frac{\omega_b}{\pi} \text{sinc}(\omega_b t)$$

(see Lect. 3, Slide 37) and is non-causal and even BIBO *unstable*¹, for $f_{\text{ilp}, \omega_b} \notin L_1$. Hence, **not quite practical**.

— ideal high-pass $F_{\text{ihp}, \omega_c} = 1 - F_{\text{ilp}, \omega_c}$

— ideal band-pass has

$$F_{\text{ibp}, [\omega_1, \omega_2]}(j\omega) = \text{rect}_{2\omega_2}(\omega) - \text{rect}_{2\omega_1}(\omega)$$

$$\Downarrow$$

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¹But, strangely enough, it is L_2 -stable, try to prove it with the material of Lects. 3 & 6.

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Decibels

Decibel (dB) is a unit of measurement expressing the ratio of two values of a root-power quantity on a logarithmic scale. Applying to $|G(j\omega)|$, it is

$$|G(j\omega)|_{(\text{dB})} := 20 \log_{10} |G(j\omega)|$$

Useful properties:

- $|G_1(j\omega)G_2(j\omega)|_{(\text{dB})} = |G_1(j\omega)|_{(\text{dB})} + |G_2(j\omega)|_{(\text{dB})}$
- $|[G(j\omega)]^n|_{(\text{dB})} = n|G(j\omega)|_{(\text{dB})}$ for all $n \in \mathbb{R}$
- $\left| \frac{1}{G(j\omega)} \right|_{(\text{dB})} = -|G(j\omega)|_{(\text{dB})}$
- $\left| \frac{G_1(j\omega)}{G_2(j\omega)} \right|_{(\text{dB})} = |G_1(j\omega)|_{(\text{dB})} - |G_2(j\omega)|_{(\text{dB})}$

Some common values² (memorize those in blue):

gain	1	$\sqrt{2}$	2	4	5	10	25	50	100	1000
dB	0	≈ 3	≈ 6	≈ 12	≈ 14	20	≈ 28	≈ 34	40	60

²The `mag2db` and `db2mag` commands of Matlab are handy.

Bode plot

Consists of

- Bode magnitude plot of $|G(j\omega)|$ (in dB) vs. ω (in logarithmic scale)
- Bode phase plot of $\arg(G(j\omega))$ (in deg) vs. ω (in logarithmic scale)

In the logarithmic scale the distance between ω_0 and $N\omega_0$ does *not* depend on ω_0 for a given $N \in \mathbb{R}_+$ ($N = 2$ is an octave, $N = 10$ is a **decade**).

Example: For $G_{\text{therm}}(s) = 1/(\tau s + 1)$,

$$G_{\text{therm}}(j\omega) = \frac{1}{\sqrt{1 + \tau^2\omega^2}} e^{-j\arctan(\tau\omega)}$$

with the magnitude and phase as

respectively. The same on the Bode plot:

Bode plot

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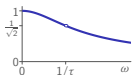
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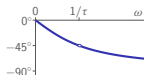
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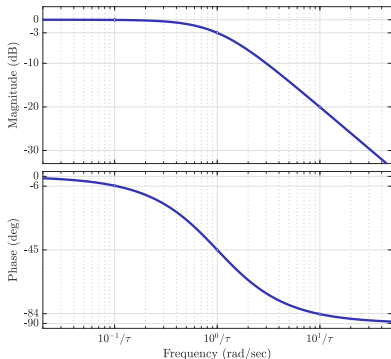
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and



respectively. The same on the Bode plot:



Bode plot: advantages

- factors add up on both magnitude and phase plots, meaning the Bode plots of systems with real-rational transfer functions, like

$$G(s) = \frac{b_m \prod_i (s - z_i) \prod_j (s^2 + 2\zeta_{z,i} \omega_{z,i} s + \omega_{z,i}^2)}{\prod_i (s - p_i) \prod_i (s^2 + 2\zeta_{p,i} \omega_{p,i} s + \omega_{p,i}^2)}$$

can be built by superposing frequency responses of 3 basic blocks³

0. static gain, like k for $k \in \mathbb{R} \setminus \{0\}$
 1. first-order factor, like $s + a$, for $a \in \mathbb{R}$
 2. second-order factor, like $s^2 + 2\zeta \omega_n s + \omega_n^2$, for $\zeta \in (-1, 1)$ & $\omega_n > 0$
- very large magnitudes are not that large in dB
 - logarithmic frequency scale facilitates viewing wider frequency ranges

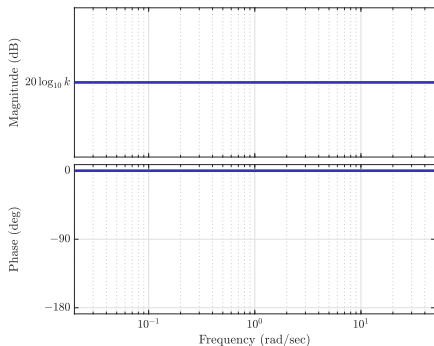
³If they are in denominators, then both their magnitude (dB) and phase change sign.

Basic blocks: static gain

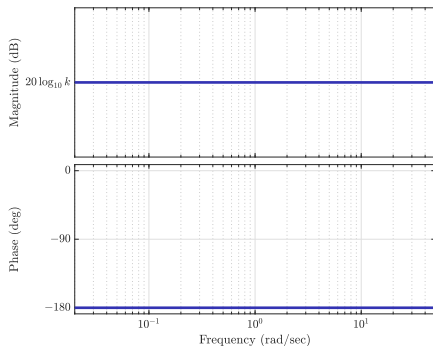
If $G(s) = k$, then

$$G(j\omega) = k = |k| \begin{cases} e^{j0} & \text{if } k > 0 \\ e^{-j\pi} & \text{if } k < 0 \end{cases}$$

and both magnitude and phase plots are horizontal lines:



if $k > 0$



if $k < 0$

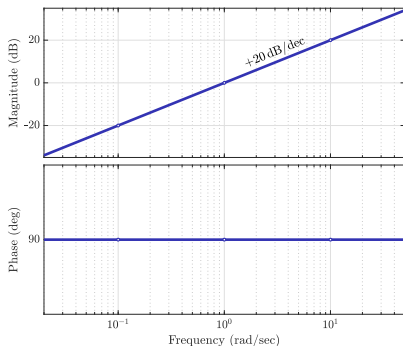
Basic blocks: 1-order factor, $a = 0$

If $G(s) = s$, then

$$G(j\omega) = j\omega = \omega e^{j\pi/2}$$

and

- the Bode magnitude plot is a straight line with the $+20$ dB/dec slope, passing through the 0 dB level at $\omega = 1$ rad/sec
- the Bode phase plot is a horizontal line



Basic blocks: 1-order factor, $a \neq 0$

If $a \neq 0$, then it is convenient to normalize the static gain of $s + a$. Hence, the basic block is $G(s) = \tau s + 1$ for $\tau = 1/a \neq 0$, for which

$$G(j\omega) = 1 + j\tau\omega = \sqrt{1 + \tau^2\omega^2} e^{j\arctan(\tau\omega)}$$

The magnitude can be approximated as

$$|G(j\omega)| = \sqrt{1 + \tau^2\omega^2} \approx \begin{cases} 1 & \text{if } |\tau|\omega < 1 \\ |\tau|\omega & \text{if } |\tau|\omega > 1 \end{cases}$$

which corresponds to straight lines (the frequency $\omega = 1/\tau$ is known as the corner frequency). The phase can be approximated as

$$\arg(G(j\omega)) = \arctan(\tau\omega) \approx \begin{cases} 0 & \text{if } |\tau|\omega < 1 \\ \pm 90^\circ & \text{if } |\tau|\omega > 1 \end{cases}$$

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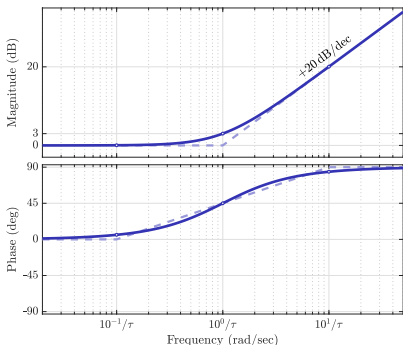
which corresponds to straight lines (the frequency $\omega = 1/\tau$ is known as the **corner frequency**). The phase can be approximated as

$$\arg(G(j\omega)) = \arctan(\tau\omega) \approx \text{sign } \tau \begin{cases} 0^\circ & \text{if } |\tau|\omega < \frac{1}{10} \\ 45^\circ(1 + \log_{10} \omega) & \text{if } \frac{1}{10} \leq |\tau|\omega \leq 10 \\ 90^\circ & \text{if } |\tau|\omega > 10 \end{cases}$$

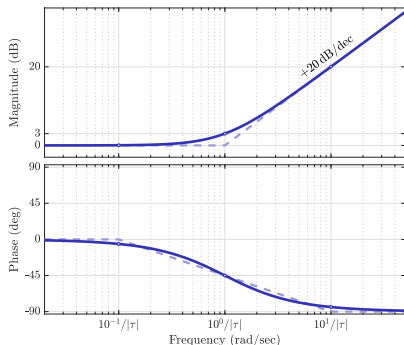
(error within $\pm 5.711^\circ$), which corresponds to straight lines too for $\log \omega$.

Basic blocks: 1-order factor, $a \neq 0$ (contd)

Thus, precise (solid) and approximate (dashed) Bode plots are



if $\tau > 0$



if $\tau < 0$

- drawing precise Bode is easy nowadays (e.g. `bode` in Matlab)
- approximate Bode is still useful for a quick mental grasping

Basic blocks: 2-order factor, $\zeta \neq 0$

If $G(s) = (s/\omega_n)^2 + 2\zeta(s/\omega_n) + 1$ (with the normalized static gain), then

$$G(j\omega) = 1 - \frac{\omega^2}{\omega_n^2} + j2\zeta\frac{\omega}{\omega_n} = \sqrt{\left(1 - \frac{\omega^2}{\omega_n^2}\right)^2 + 4\zeta^2\frac{\omega^2}{\omega_n^2}} e^{j\arg(G(j\omega))},$$

where, assuming $\arctan \phi \in [-\pi/2, \pi/2]$,

$$\arg(G(j\omega)) = \arctan \frac{2\zeta\omega/\omega_n}{1 - \omega^2/\omega_n^2} + \begin{cases} 0 & \text{if } \omega \leq \omega_n \\ \pi & \text{if } \omega \geq \omega_n \wedge \zeta > 0 \\ -\pi & \text{if } \omega \geq \omega_n \wedge \zeta < 0 \end{cases}$$

which is a continuous and monotonic function of ω (increasing if $\zeta > 0$ and decreasing if $\zeta < 0$). Both the magnitude and phase may be approximated by piecewise linear functions, but this is accurate only around $|\zeta| = 1/\sqrt{2}$.

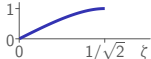
Basic blocks: 2-order factor, $\zeta \neq 0$ (contd)

The derivative of the magnitude

$$\frac{|G(j\omega)|}{d\omega} = \underbrace{\frac{2}{|G(j\omega)|} \frac{\omega}{\omega_n}}_{>0, \forall \omega > 0} \left(\frac{\omega^2}{\omega_n^2} + 2\zeta^2 - 1 \right).$$

Hence,

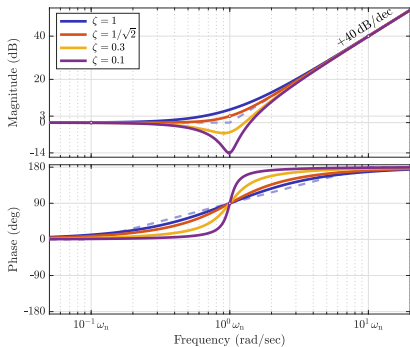
- if $|\zeta| \geq \frac{1}{\sqrt{2}}$, then $|G(j\omega)|$ is monotonically increasing
- if $|\zeta| < \frac{1}{\sqrt{2}}$, then $|G(j\omega)|$
 - monotonically decreases for $\omega < \sqrt{1 - 2\zeta^2} \omega_n$
 - monotonically increases for $\omega > \sqrt{1 - 2\zeta^2} \omega_n$
 - has

$$\min_{\omega} |G(j\omega)| = 2\zeta \sqrt{1 - \zeta^2} = 0$$


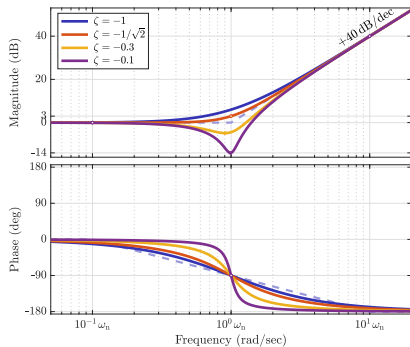
attainable at $\omega = \sqrt{1 - 2\zeta^2} \omega_n < \omega_n$

Basic blocks: 2-order factor, $\zeta \neq 0$ (contd)

Thus, precise (solid) and approximate (dashed) Bode plots are



if $\zeta > 0$



if $\zeta < 0$

If ζ is “far” from $1/\sqrt{2}$, the

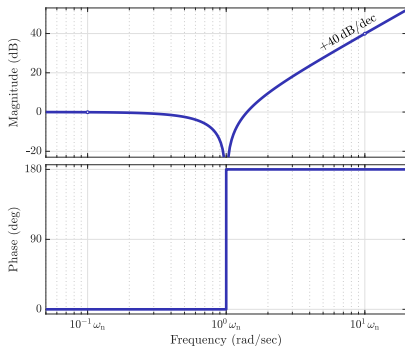
- approximate Bode plots are not quite accurate around $\omega = \omega_n$

Basic blocks: 2-order factor, $\zeta = 0$

If $G(s) = (s/\omega_n)^2 + 1$ (the static gain is again normalized), then

$$G(j\omega) = 1 - \frac{\omega^2}{\omega_n^2} = \left| 1 - \frac{\omega^2}{\omega_n^2} \right| \begin{cases} e^{j0} & \text{if } \omega < \omega_n \\ e^{j\pi} & \text{if } \omega > \omega_n \end{cases}$$

resulting in

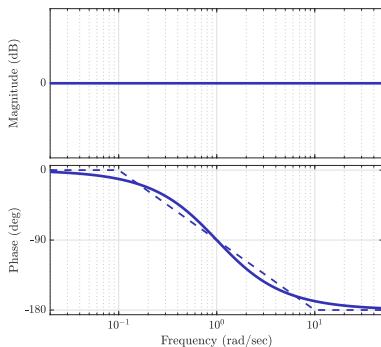
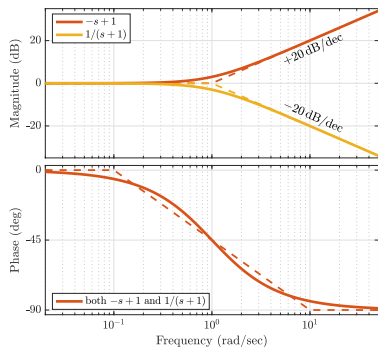


Example 1

If

$$G_1(s) = \frac{-s + 1}{s + 1}$$

then



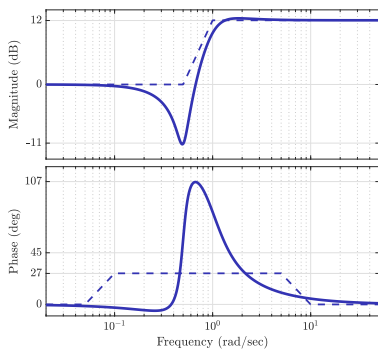
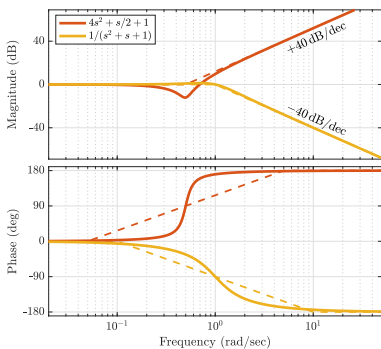
and the high-frequency gain is $G_1(\infty) = 1 = 0$ dB.

Example 2

If

$$G_2(s) = \frac{4s^2 + s/2 + 1}{s^2 + s + 1}$$

then



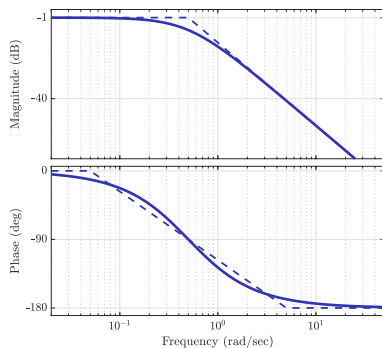
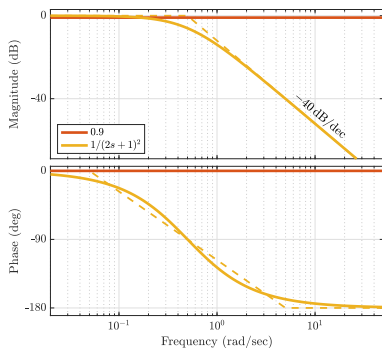
and the high-frequency gain is $G_2(\infty) = 4 \approx 12$ dB.

Example 3

If

$$G_3(s) = \frac{0.9}{(2s + 1)^2}$$

then



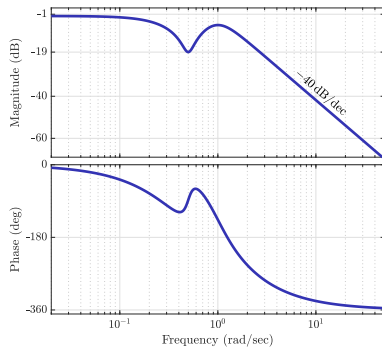
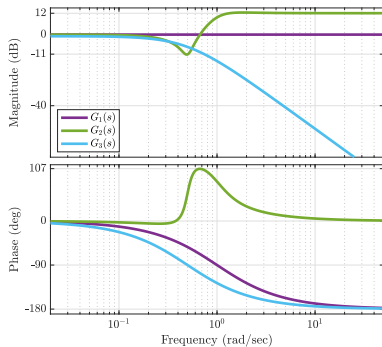
and the magnitude decays at high frequencies with a slope of -40 dB/dec.

Example 4

If

$$G(s) = G_1(s)G_2(s)G_3(s) = \frac{0.9(-s + 1)(4s^2 + s/2 + 1)}{(s + 1)(2s + 1)^2(s^2 + s + 1)}$$

then



Asymptotic properties of Bode plots: magnitude

At low frequencies:

- every zero at the origin contributes a slope of $+20$ dB/dec
- every integrator (pole at the origin) contributes a slope of -20 dB/dec
- if no poles/zeros at the origin, starts as a horizontal line at $|G(0)|_{(\text{dB})}$

At high frequencies:

- every zero adds a slope of $+20$ dB/dec
- every pole adds a slope of -20 dB/dec
- if $G(s)$ is bi-proper, ends as a horizontal line at $|G(\infty)|_{(\text{dB})}$

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Asymptotic properties of Bode plots: phase

At low frequencies:

- every zero at the origin contributes a phase lead of $+90^\circ$
- every integrator (pole at the origin) contributes a phase lag of -90°

At high frequencies:

- every zero in $\mathbb{C} \setminus \mathbb{C}_0 = \{s \in \mathbb{C} \mid \operatorname{Re} s \leq 0\}$ adds a phase lead of 90°
- every pole in $\mathbb{C} \setminus \mathbb{C}_0 = \{s \in \mathbb{C} \mid \operatorname{Re} s \leq 0\}$ adds a phase lag of -90°
- every zero in $\mathbb{C}_0 = \{s \in \mathbb{C} \mid \operatorname{Re} s > 0\}$ adds a phase lag of -90°
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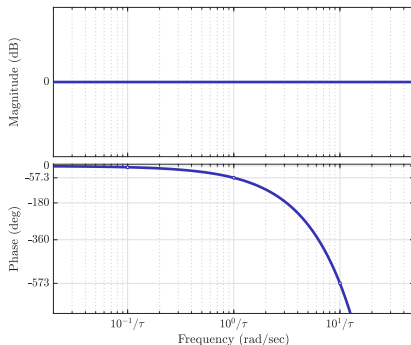
- every zero in $\mathbb{C} \setminus \mathbb{C}_0 = \{s \in \mathbb{C} \mid \operatorname{Re} s \leq 0\}$ adds a phase lead of 90°
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- every pole in $\mathbb{C}_0 = \{s \in \mathbb{C} \mid \operatorname{Re} s > 0\}$ adds a phase lead of 90°

Bode for non-rational transfer functions

If

$$G(s) = \bar{D}_\tau(s) = e^{-\tau s}, \quad \tau > 0$$

then $G(j\omega) = e^{-j\tau\omega}$, so that $|G(j\omega)| = 1$ and $\arg(G(j\omega)) = -\tau\omega$ [rad]. We therefore have



Bode for non-rational transfer functions (contd)

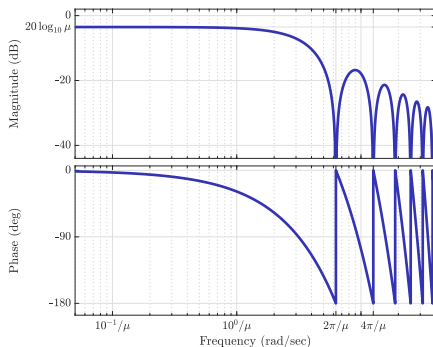
If

$$G(s) = G_{fmi,\mu}(s) = \frac{1 - e^{-\mu s}}{s}, \quad \mu > 0$$

then

$$G(j\omega) = \frac{1 - e^{-j\mu\omega}}{j\omega} = \mu \frac{e^{j\mu\omega/2} - e^{-j\mu\omega/2}}{j2\mu\omega/2} e^{-j\mu\omega/2} = \mu \operatorname{sinc}\left(\frac{\mu\omega}{2}\right) e^{-j\mu\omega/2}$$

and we have



Polar plot

Shows $\text{Im } G(j\omega)$ vs. $\text{Re } G(j\omega)$ as the frequency ω grows from 0 to ∞ , with an arrow indicating the growth direction of ω .

Example 4:

$$G(s) = \frac{0.9(-s+1)(4s^2+s/2+1)}{(s+1)(2s+1)^2(s^2+s+1)} \implies$$

Polar plots are

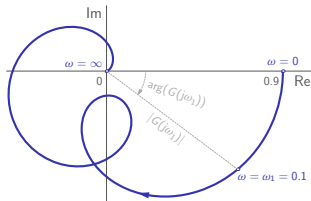
- less informative than Bode
- harder to draw manually than Bode
- produced by the `nyquist` command of Matlab
- Draws the plot for $-\infty < \omega < \infty$ (aka the Nyquist diagram). To produce the plain polar plot, use `nyquistz` (also `nyquistplot`).
- very important in feedback control applications (the Nyquist criterion)

Polar plot

Shows $\text{Im } G(j\omega)$ vs. $\text{Re } G(j\omega)$ as the frequency ω grows from 0 to ∞ , with an arrow indicating the growth direction of ω .

Example 4:

$$G(s) = \frac{0.9(-s + 1)(4s^2 + s/2 + 1)}{(s + 1)(2s + 1)^2(s^2 + s + 1)}$$



Polar plots are

- less informative than Bode
- harder to draw manually than Bode
- can be easily and accurately produced by the `nyquist` command of Matlab
- Draws the plot for $-\infty < \omega < \infty$ (aka the Nyquist diagram). To produce the plain polar plot, use `nyquist(G, w)` or `nyquist(G, w, 's')`
- very important in feedback control applications

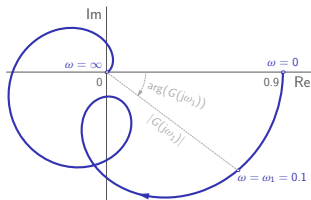
Polar plot

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Example 4:

$$G(s) = \frac{0.9(-s + 1)(4s^2 + s/2 + 1)}{(s + 1)(2s + 1)^2(s^2 + s + 1)}$$

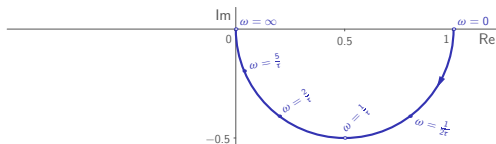
\Rightarrow



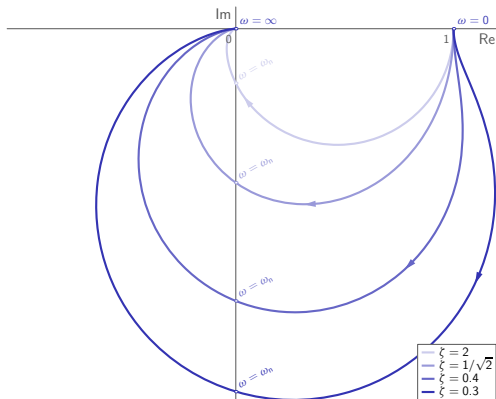
Polar plots are

- less informative than Bode the frequency is hidden
 - harder to draw manually than Bode no superposition rules hold
 - produced by the `nyquist` command of Matlab
- Draws the plot for $-\infty < \omega < \infty$ (aka the Nyquist diagram). To produce the plain polar plot, use `setoptions(nyquistplot(G), 'ShowFullContour', 'off')`
- very important in feedback control applications (the Nyquist criterion)

Some polar plots

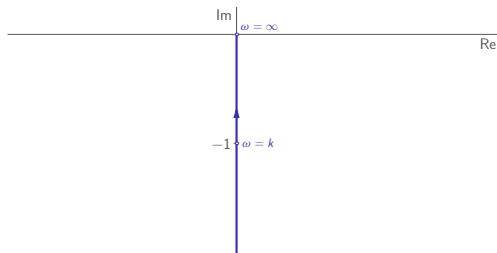


$$G(s) = \frac{1}{\tau s + 1}$$

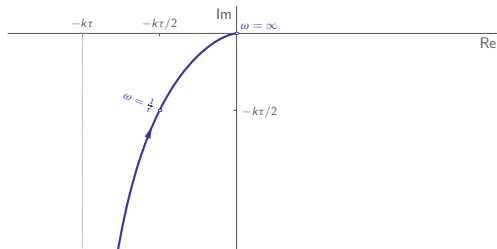


$$G(s) = \frac{1}{s^2/\omega_n^2 + 2\zeta s/\omega_n + 1}$$

Some polar plots (contd)

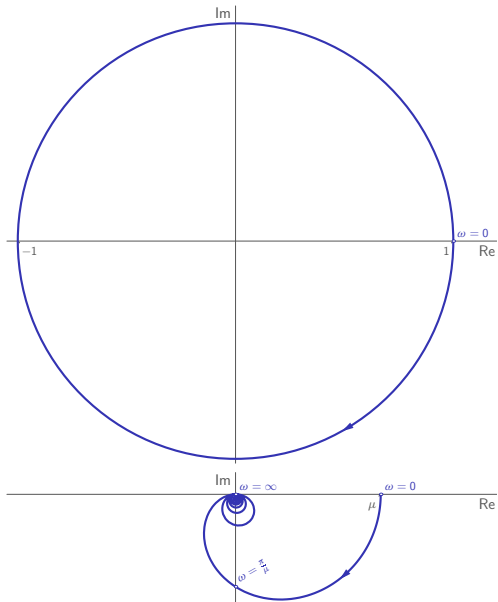


$$G(s) = \frac{k}{s}, \quad k > 0$$



$$G(s) = \frac{k}{s(\tau s + 1)}, \quad k > 0$$

Some polar plots (contd)



$$G(s) = e^{-\tau s}$$

or

$$G(s) = \left(\frac{-\tau s + 1}{\tau s + 1} \right)^2$$

$$G(s) = \frac{1 - e^{-\mu s}}{s}, \mu > 0$$

Outline

Systems as filters I

Frequency response plots

Systems as filters II

Butterworth polynomials

The Butterworth polynomial of degree n , $B_n(s)$, is the Hurwitz polynomials such that

$$|B_n(j\omega)|^2 = 1 + \omega^{2n}$$

Its general form is (depending on whether n is even or odd)

$$B_n(s) = \prod_{i=1}^{n/2} (s^2 + 2\xi_i s + 1) \quad \text{or} \quad B_n(s) = (s + 1) \prod_{i=1}^{(n-1)/2} (s^2 + 2\xi_i s + 1)$$

where

$$\xi_i := \sin\left(\frac{2i-1}{2n}\pi\right) \in (0, 1), \quad i \in \mathbb{Z}_{1..\lfloor n/2 \rfloor}$$

Roots of $B_n(s)$ are at equally-spaced points in $\{s \in \mathbb{C} \mid \operatorname{Re} s < 0 \wedge |s| = 1\}$.

Particular cases:

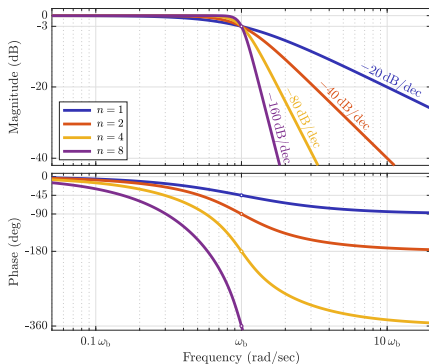
$$B_1(s) = s + 1, \quad B_2(s) = s^2 + \sqrt{2}s + 1, \quad B_3(s) = (s + 1)(s^2 + s + 1).$$

Low-pass Butterworth filter

The n -order low-pass filter

$$F(s) = \frac{1}{B_n(s/\omega_b)} \quad \Rightarrow \quad |F(j\omega)| = \frac{1}{\sqrt{1 + (\omega/\omega_b)^{2n}}}$$

is monotonically decreasing with $|F(j\omega_b)| = \frac{1}{\sqrt{2}}$ (i.e. ω_b is its bandwidth):

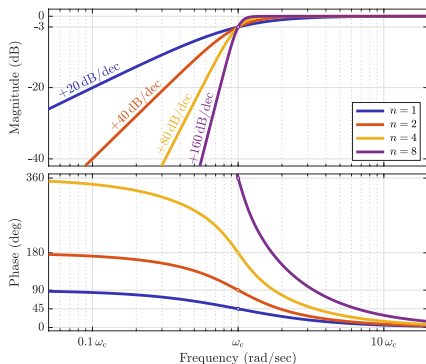


High-pass Butterworth filter

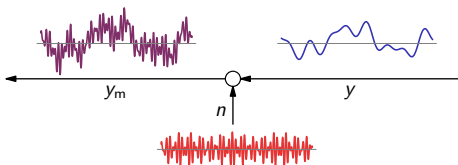
The n -order high-pass filter

$$F(s) = \frac{(s/\omega_c)^n}{B_n(s/\omega_c)} \quad \Rightarrow \quad |F(j\omega)| = \frac{(\omega/\omega_c)^n}{\sqrt{1 + (\omega/\omega_c)^{2n}}}$$

is monotonically increasing with $|F(j\omega_c)| = \frac{1}{\sqrt{2}}$ (i.e. ω_c is its cut-off freq.):



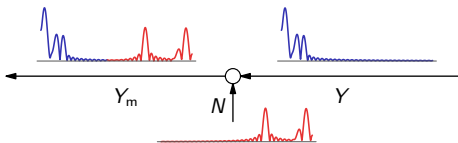
Example from Lect. 3



The question was

- how signal (y) can be recovered from its corrupt measurements (y_m)?

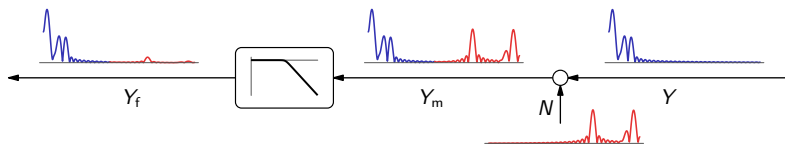
This might appear a tough task, for there is no way we can separate y from n in the time domain... But in the frequency domain these signals are well separated,



suggesting that the frequency-domain viewpoint is valuable.

Example from Lect. 3 (contd)

If we process the measurement by a low-pass filter (4-order Butterworth in this case), then the result,

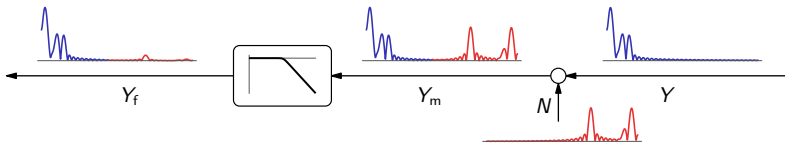


separates the slow signal y from fast noise n . *In the time domain,*

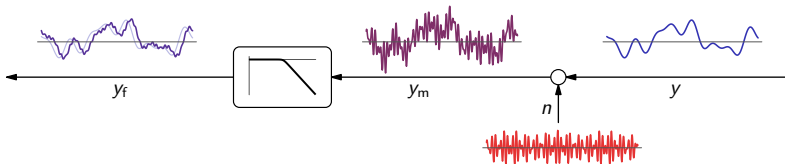
although with certain phase lag

Example from Lect. 3 (contd)

If we process the measurement by a low-pass filter (4-order Butterworth in this case), then the result,



separates the slow signal y from fast noise n . In the time domain,



although with certain phase lag...

Notch filter

Is a narrow stopband band-stop filter of the form

$$F(s) = \frac{s^2 + \omega_0^2}{s^2 + 2\zeta\omega_0 s + \omega_0^2} \quad \Longrightarrow \quad |F(j\omega)| = \sqrt{\frac{(\omega^2 - \omega_0^2)^2}{(\omega^2 - \omega_0^2)^2 + 4\zeta^2\omega_0^2\omega^2}}$$

is

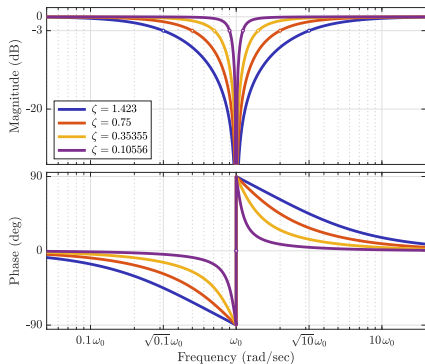
- monotonically decreasing in $\omega < \omega_0$
- monotonically increasing in $\omega > \omega_0$

with

$$|F(j\omega)| = \frac{1}{\sqrt{2}} \iff \omega = \begin{cases} \omega_1 := (\sqrt{1 + \zeta^2} - \zeta)\omega_0 < \omega_0 \\ \omega_2 := (\sqrt{1 + \zeta^2} + \zeta)\omega_0 > \omega_0 = 1/\omega_1 \end{cases}$$

i.e. (ω_1, ω_2) is its stopband (a decade if $\zeta = 0.45\sqrt{10} \approx 1.423$).

Notch filter



with stopbands $(\omega_0/\sqrt{10}, \sqrt{10}\omega_0)$ (decade), $(0.5\omega_0, 2\omega_0)$, $(\omega_0/\sqrt{2}, \sqrt{2}\omega_0)$ (octave), and $(0.9\omega_0, \omega_0/0.9)$.