



Continuous-Time Signals and Systems

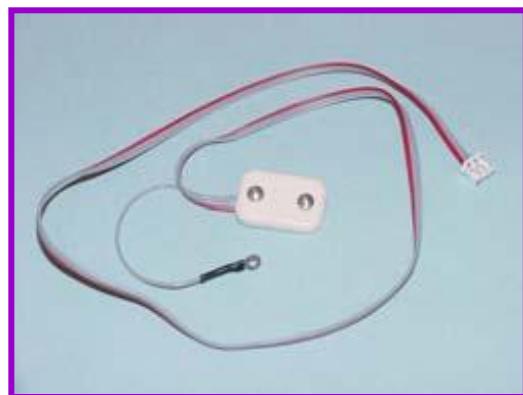
Second-Order Sallen-Key Circuit Analysis

[Edward \(Ted\) A. Clancy](#)

Department of Electrical and Computer Engineering
Department of Biomedical Engineering
Worcester Polytechnic Institute
Worcester, MA U.S.A.

EMG Amplitude (EMGamp) Estimation

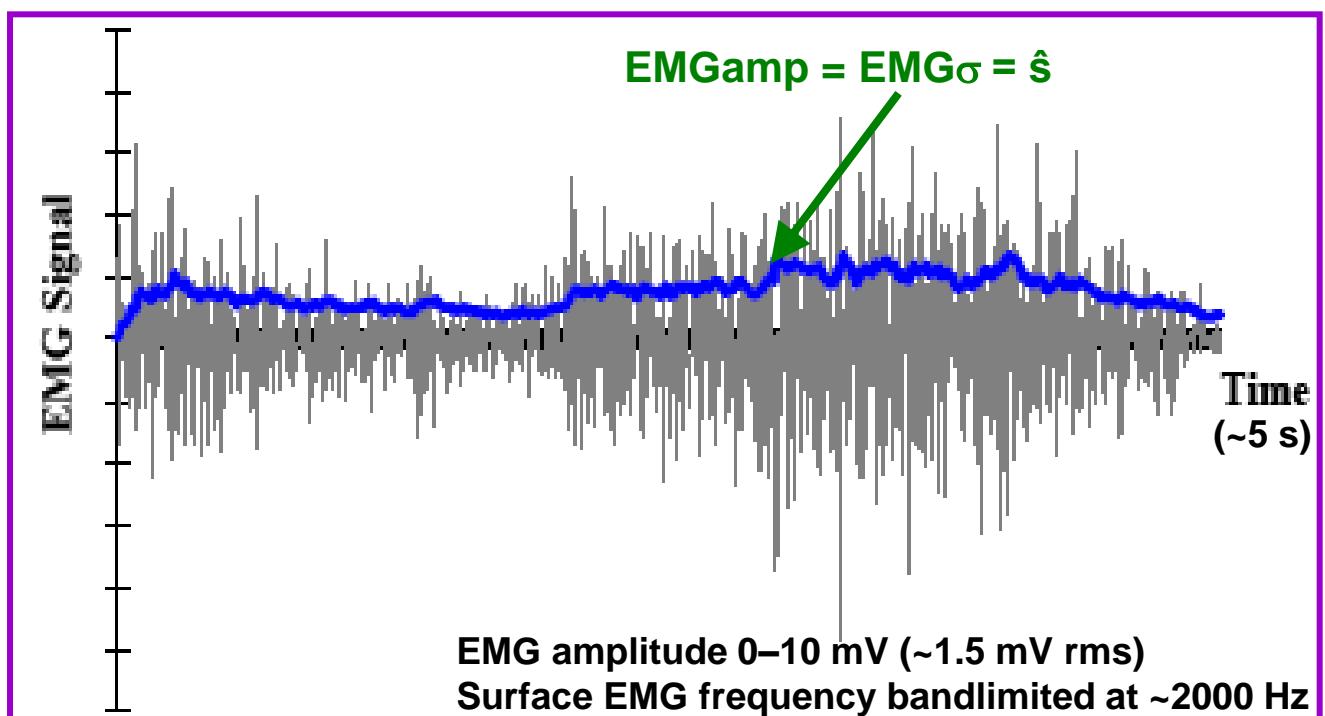
- EMG Amplitude (EMGamp or $\text{EMG}\sigma$): “Intensity” of recorded EMG
 - Time-varying standard deviation of EMG signal
 - EMG amp varies with muscle tension, localized muscle fatigue
- Original estimator: Inman *et al.* [EEG Clin Neurophysiol 4: 187–194, 1952]
 - Analog full-wave rectify and RC low pass filter



Electrode-Amplifier

[Liberating Technologies]

EMG signal well modeled as amplitude-modulated, random process. EMGamp is the modulation.



Introduction

Surface EMG Recordings (2)

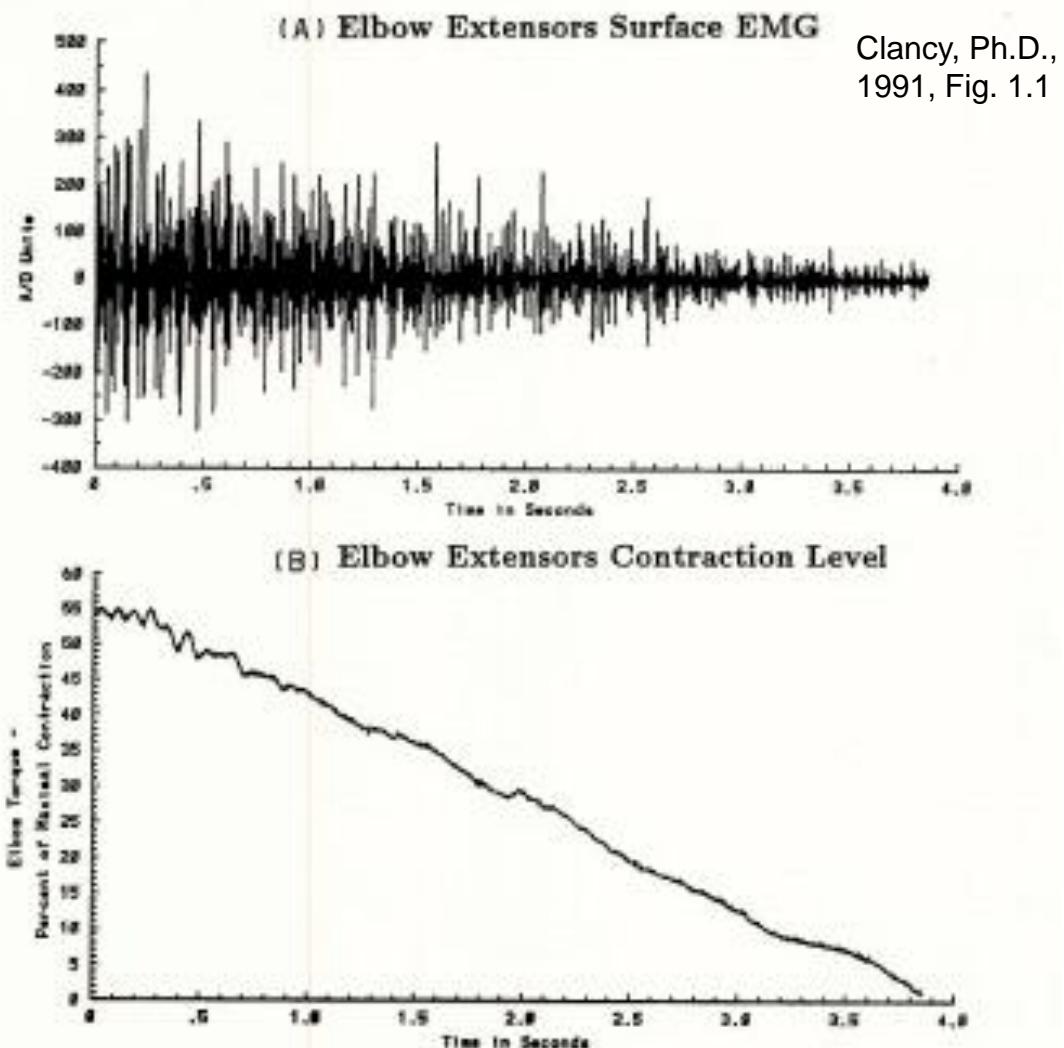


Figure 1.1: Surface EMG Waveform and Corresponding Joint Torque

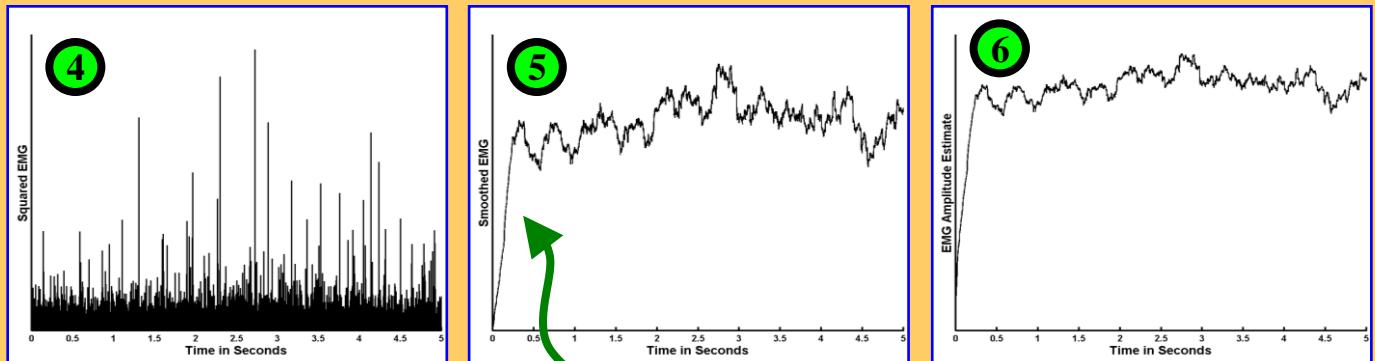
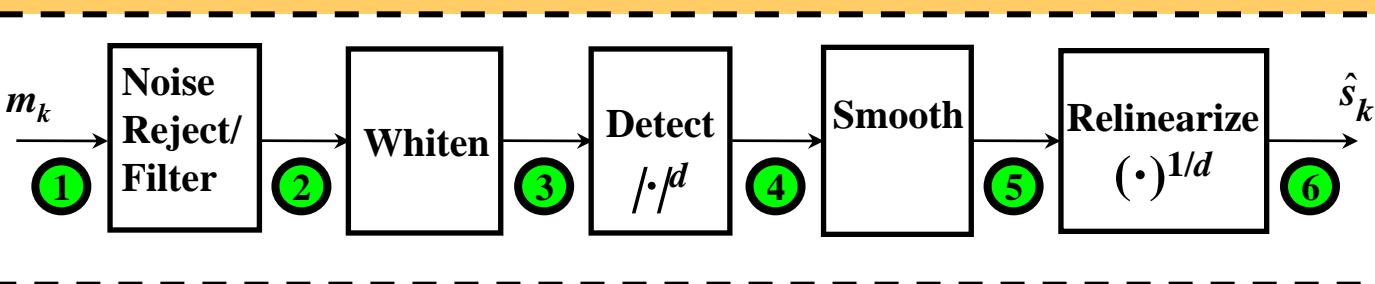
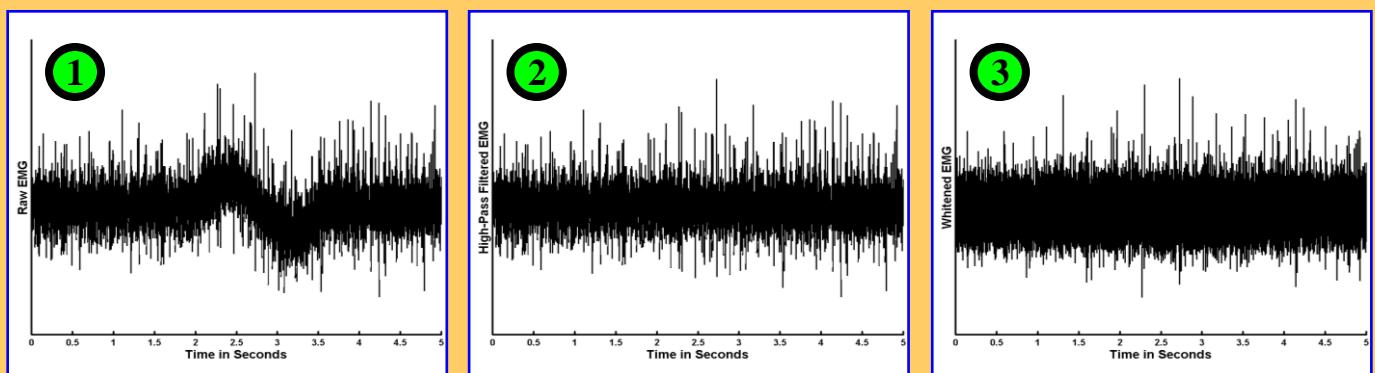
- A) Surface EMG waveform recorded from the triceps muscle with a bipolar electrode during an isometric non-fatiguing contraction. EMG is normalized to its maximum value in this trial.
- B) Torque generated about the elbow during elbow extension for the same trial shown in (A) above. Torque is normalized to its maximum value in this trial.

Effort level \leftrightarrow Time-varying standard deviation

Skeletal Muscle Electrical Activity

Optimal Single-Channel EMG Amplitude Estimation

- More mature/detailed algorithms exist for EMG amplitude estimation



Discard “start-up” transient

Clancy, Morin, Merletti, 2004.

Can also combine information from multiple electrodes placed on same muscle, ...

General Modeling Information

Normal Rate, Sinus Tachycardia, Sinus Bradycardia

- Normal rate: **60–100 bpm**



Rate ≈ 75 bpm

<http://www.technion.ac.il/~eilamp/arrhythmiasintro.html>

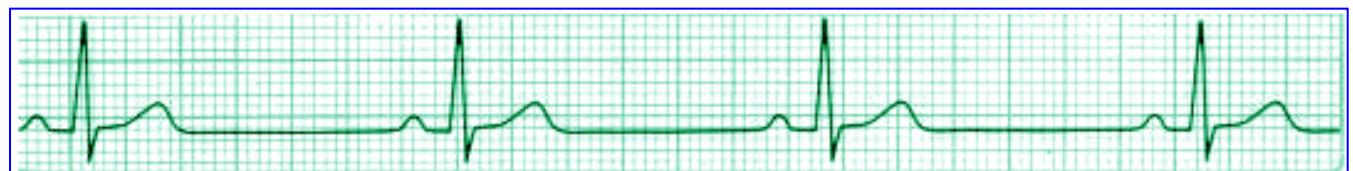
- Sinus Tachycardia: >100 bpm



Rate ≈ 136 bpm

<http://www.technion.ac.il/~eilamp/sinustachycardia.html>

- Sinus Bradycardia: <60 bpm



Rate ≈ 45 bpm

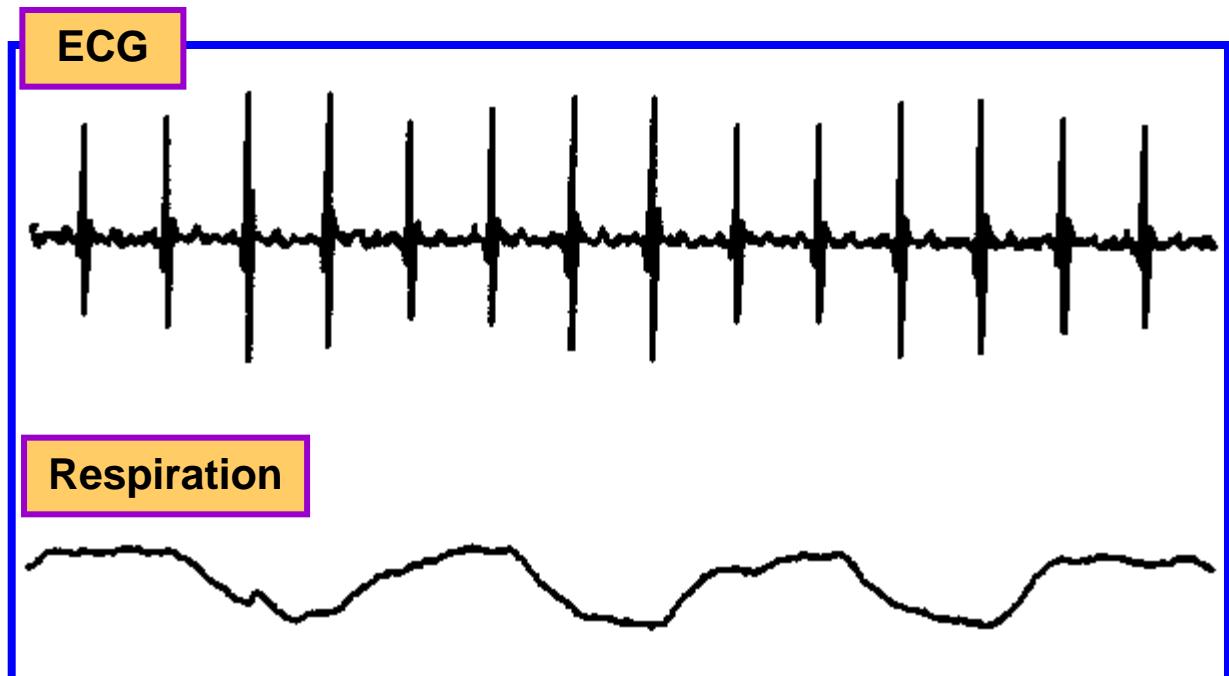
<http://www.technion.ac.il/~eilamp/sinusbrady.html>

Many, many other defined rhythms.

Electrical Aspects of the Cardiac Cycle

Respiratory Modulation

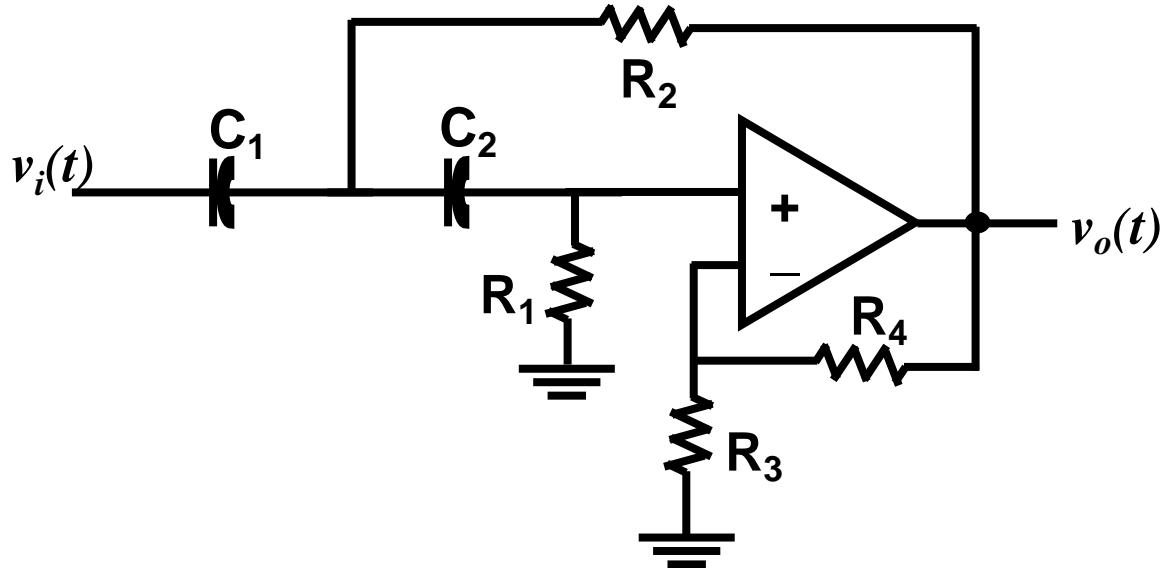
- Respiration modulates ECG rate, amplitude
- Rate increases during inspiration, decreases during expiration



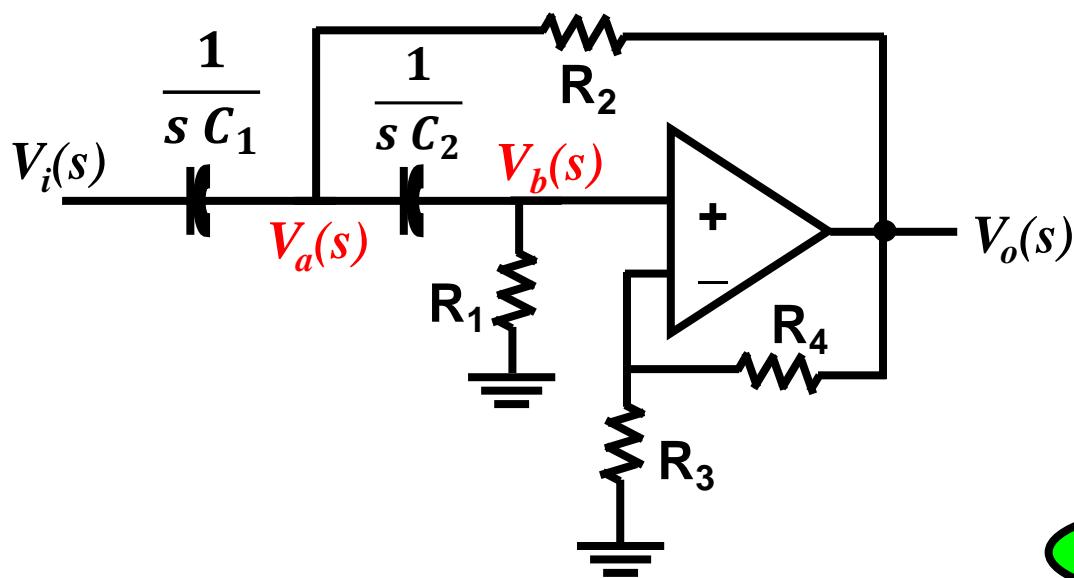
Moody et al., *Computers in Cardiology* 12:113–116, 1985.

Frequency Response of Sallen-Key Highpass Filter (1)

- Find the frequency response of the Sallen-Key highpass filter, shown below:



- In doing so: (1) Define $A_\infty = 1 + \frac{R_4}{R_3} = \frac{R_3+R_4}{R_3}$
(2) Use "s" as the frequency variable



Continued

Frequency Response of Sallen-Key Highpass Filter (2)

(1) By voltage division: $V_b(s) = V_o(s) \cdot \frac{R_3}{R_3+R_4} = \frac{V_o(s)}{A_\infty}$

(2) KCL at node V_a , then substituting (1) for $V_b(s)$:

$$[V_a(s) - V_i(s)] \cdot s C_1 + \frac{V_a(s) - V_o(s)}{R_2} + [V_a(s) - V_b(s)] \cdot s C_2 = 0$$

$$\begin{aligned} V_a(s) \cdot s R_2 C_1 - V_i(s) \cdot s R_2 C_1 + V_a(s) - V_o(s) + V_a(s) \cdot s R_2 C_2 \\ - V_o(s) \cdot \frac{s R_2 C_2}{A_\infty} = 0 \end{aligned}$$

$$\begin{aligned} V_a(s) \cdot A_\infty [s R_2 (C_1 + C_2) + 1] \\ = V_i(s) \cdot A_\infty [s R_2 C_1] + V_o(s) \cdot A_\infty \left[1 + \frac{s R_2 C_2}{A_\infty} \right] \end{aligned}$$

(3) KCL at node V_b , then substituting (1) for $V_b(s)$:

$$[V_b(s) - V_a(s)] \cdot s C_2 + \frac{V_b(s)}{R_1} = 0$$

$$V_o(s) \cdot [s R_1 C_2] - V_a(s) \cdot A_\infty [s R_1 C_2] + V_o(s) = 0$$

$$V_a(s) = \frac{1}{A_\infty} \cdot V_o(s) \left[\frac{1 + s R_1 C_2}{s R_1 C_2} \right]$$

Continued

Frequency Response of Sallen-Key Highpass Filter (3)

- Substituting (3) into (2) :

$$V_o(s) \cdot \frac{[1 + sR_1C_2][sR_2(C_1 + C_2) + 1]}{sR_1C_2} = V_i(s) \cdot A_\infty [sR_2C_1] + V_o(s) [A_\infty + sR_2C_2]$$

Thus,

$$\begin{aligned} & V_o(s) \\ & \cdot \left\{ \frac{[1 + sR_1C_2][sR_2C_1 + sR_2C_2 + 1] - [sR_1C_2][A_\infty + sR_2C_2]}{sR_1C_2} \right\} \\ & = V_i(s) \cdot A_\infty \cdot sR_2C_1 \end{aligned}$$

Giving,

$$\begin{aligned} H(s) &= \frac{V_o(s)}{V_i(s)} \\ &= \frac{A_\infty \cdot s^2 R_1 R_2 C_1 C_2}{s^2[R_1 R_2 C_1 C_2] + s[R_2(C_1 + C_2) + R_1 C_2(1 - A_\infty)] + 1} \end{aligned}$$

or

$$\begin{aligned} H(\omega) & \\ &= \frac{A_\infty \cdot (j\omega)^2 R_1 R_2 C_1 C_2}{(j\omega)^2[R_1 R_2 C_1 C_2] + (j\omega)[R_2(C_1 + C_2) + R_1 C_2(1 - A_\infty)] + 1} \end{aligned}$$

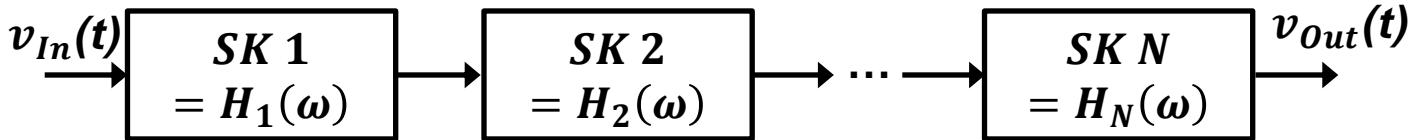
At $\omega = 0$, $H(\omega) = 0$.

At $\omega = \infty$, $H(\omega) = A_\infty$.

Note: Second-order filter

Higher-Order Filters (1)

- Common approach: Cascade filters

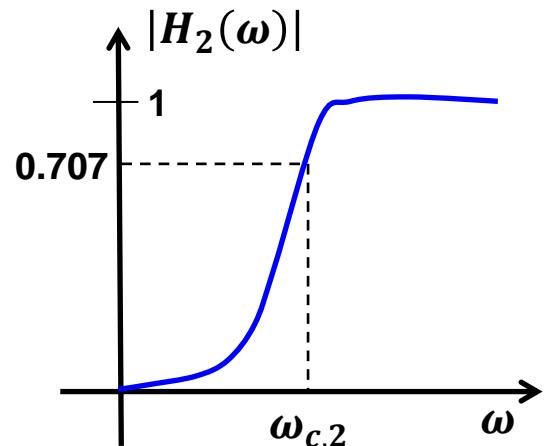
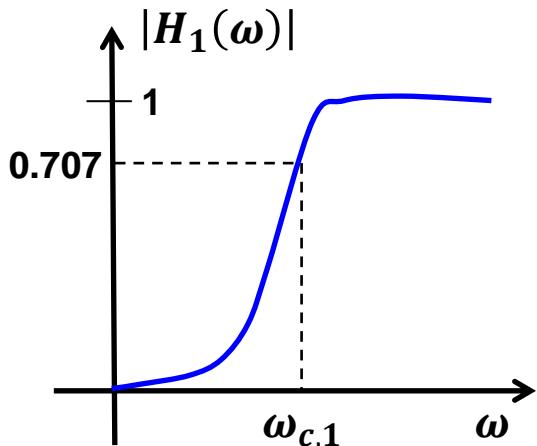


– Overall: $H_T(\omega) = \mathbf{H}_1(\omega) \cdot \mathbf{H}_2(\omega) \cdots \mathbf{H}_N(\omega)$

- Issue: Consider cascade of two filters:

$$|H_T(\omega)| = |H_1(\omega)| \cdot |H_2(\omega)|$$

– For simplicity, let $H_1(\omega) = H_2(\omega) \Rightarrow \omega_{c,1} = \omega_{c,2}$



Question: What is ω_c for $H_T(\omega)$?

1. $\omega_{c,T} = (\omega_{c,1} = \omega_{c,2})$
2. $\omega_{c,T} > (\omega_{c,1} = \omega_{c,2})$
3. $\omega_{c,T} < (\omega_{c,1} = \omega_{c,2})$



Continued

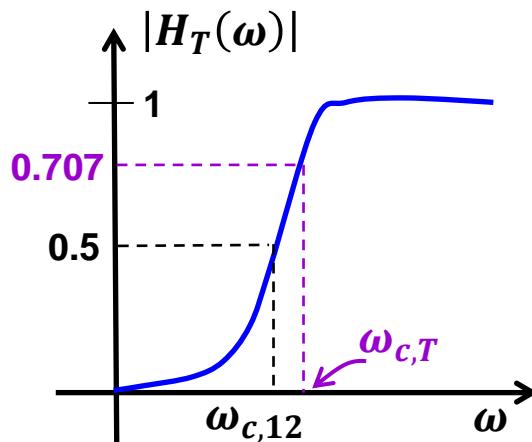
Higher-Order Filters (2)

- Denote: $\omega_{c,12} \equiv \omega_{c,1} = \omega_{c,2}$

– Know: $|H_{1 \text{ or } 2}(\omega_{c,12})| = \frac{\sqrt{2}}{2} = 0.707$

– Thus,

$$|H_T(\omega_{c,12})| = |H_1(\omega_{c,1})| \cdot |H_2(\omega_{c,2})| = \frac{\sqrt{2}}{2} \cdot \frac{\sqrt{2}}{2} = \frac{1}{2}$$



- Cascaded $\omega_{c,T}$ is greater than $\omega_{c,12}$ (for highpass cascade)
- Cascade design must account for $\omega_{c,T}$ shift
 - Complicated analysis
or
– Normalized design tables !!!

Highpass Design: Selecting R's, C's (1)

- Re-write $H(\omega)$ [Divide numerator, denom. by $(j\omega)^2 R_1 R_2 C_1 C_2$]:

$$H(\omega) = \frac{A_\infty}{1 + \frac{1}{j\omega} \cdot \left[\frac{R_2(C_1 + C_2) + R_1 C_2(1 - A_\infty)}{R_1 R_2 C_1 C_2} \right] + \frac{1}{(j\omega)^2} \cdot \left[\frac{1}{R_1 R_2 C_1 C_2} \right]}$$

- To normalize cut-off freq., replace: $\omega = \frac{\omega}{\omega_c} \cdot \omega_c \equiv \omega_n \cdot \omega_c \equiv \omega \cdot \omega_c$:

$$H_{Norm}(\omega) = \frac{A_\infty}{1 + \frac{1}{j\omega} \cdot \left[\frac{R_2(C_1 + C_2) + R_1 C_2(1 - A_\infty)}{\omega_c \cdot R_1 R_2 C_1 C_2} \right] + \frac{1}{(j\omega)^2} \cdot \left[\frac{1}{\omega_c^2 \cdot R_1 R_2 C_1 C_2} \right]}$$

- So,

$$H_{Norm}(\omega) = \frac{A_\infty}{1 + \frac{a}{j\omega} + \frac{b}{(j\omega)^2}},$$

$\omega = 1$
 \Rightarrow cut-off

where

$$a = \frac{R_2(C_1 + C_2) + R_1 C_2(1 - A_\infty)}{\omega_c \cdot R_1 R_2 C_1 C_2},$$

$$b = \frac{1}{\omega_c^2 \cdot R_1 R_2 C_1 C_2}$$

and

$$\omega_c = 2 \pi f_c \Rightarrow \text{Desired cut-off frequency}$$

Continued

Highpass Design: Selecting R's, C's (2)

- Tables of normalized a_i, b_i values exist (“ i ” refers to stage)
 - Various filter orders
 - Various filter shapes (e.g., Butterworth, Chebyshev)
 - E.g.: Fourth-order Butterworth highpass filter:
 $a_1 = 1.8478, b_1 = 1$ (Stage 1)
 $a_2 = 0.7654, b_2 = 1$ (Stage 2)

- Any R_1, R_2, C_1, C_2 satisfying a_i, b_i is OK
 - Underdetermined problem. Solution not unique.
 - Common solution (each stage) is:
 1. Select $C_1 = C_2 = \text{Some Easy Value}$
 \rightarrow Pick C 's first, because fewer manufactured values
 2. R_1 :
 - If $A_i = 1$, then $R_1 = \frac{C_1 + C_2}{a_i \omega_c C_1 C_2}$
 - If $A_i > 1$, then

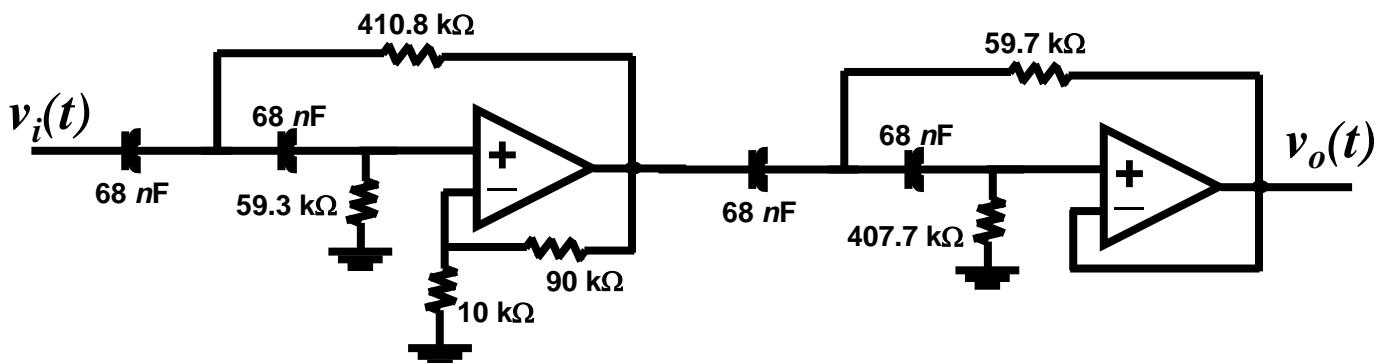
$$R_1 = \frac{a_i \omega_c C_1 C_2 - \sqrt{a_i^2 \omega_c^2 C_1^2 C_2^2 - 4 [\omega_c^2 b_i C_1 C_2 (1-A_i)] (C_1 + C_2)}}{2 \omega_c^2 b_i C_1 C_2 (1-A_i)}$$
 3. $R_2 = \frac{1}{\omega_c^2 \cdot b_i \cdot R_1 \cdot C_1 \cdot C_2}$
 4. Replace ideal R_1, R_2 with manufactured values
 5. Repeat for each stage: $A_T = A_1 \cdot A_2 \cdot A_3 \cdots A_N$

Example Highpass Design (1)

- Desire 4th-order, Butterworth, $f_c = 15 \text{ Hz}$, $A_T = 10$
- From Kugelstadt (Chapter 16 of “Op Amps for Everyone,” literature # SLOD006A, www.ti.com)
 - $a_1 = 1.8478, b_1 = 1$ (Stage 1)
 - $a_2 = 0.7654, b_2 = 1$ (Stage 2)
- Stage 1: Choose $C_1 = C_2 = 68 \text{ nF}$
 - $\rightarrow R_1 = 59.3 \text{ k}\Omega, R_2 = 410.8 \text{ k}\Omega$
 - Choose $R_4 = 90 \text{ k}\Omega, R_3 = 10 \text{ k}\Omega \Rightarrow A_1 = 10$
- Stage 2: Choose $C_1 = C_2 = 68 \text{ nF}$
 - $\rightarrow R_1 = 407.7 \text{ k}\Omega, R_2 = 59.7 \text{ k}\Omega$
 - Short R_4 , omit $R_3 \Rightarrow A_2 = 1$

✓ Reasonable range of R, C values.

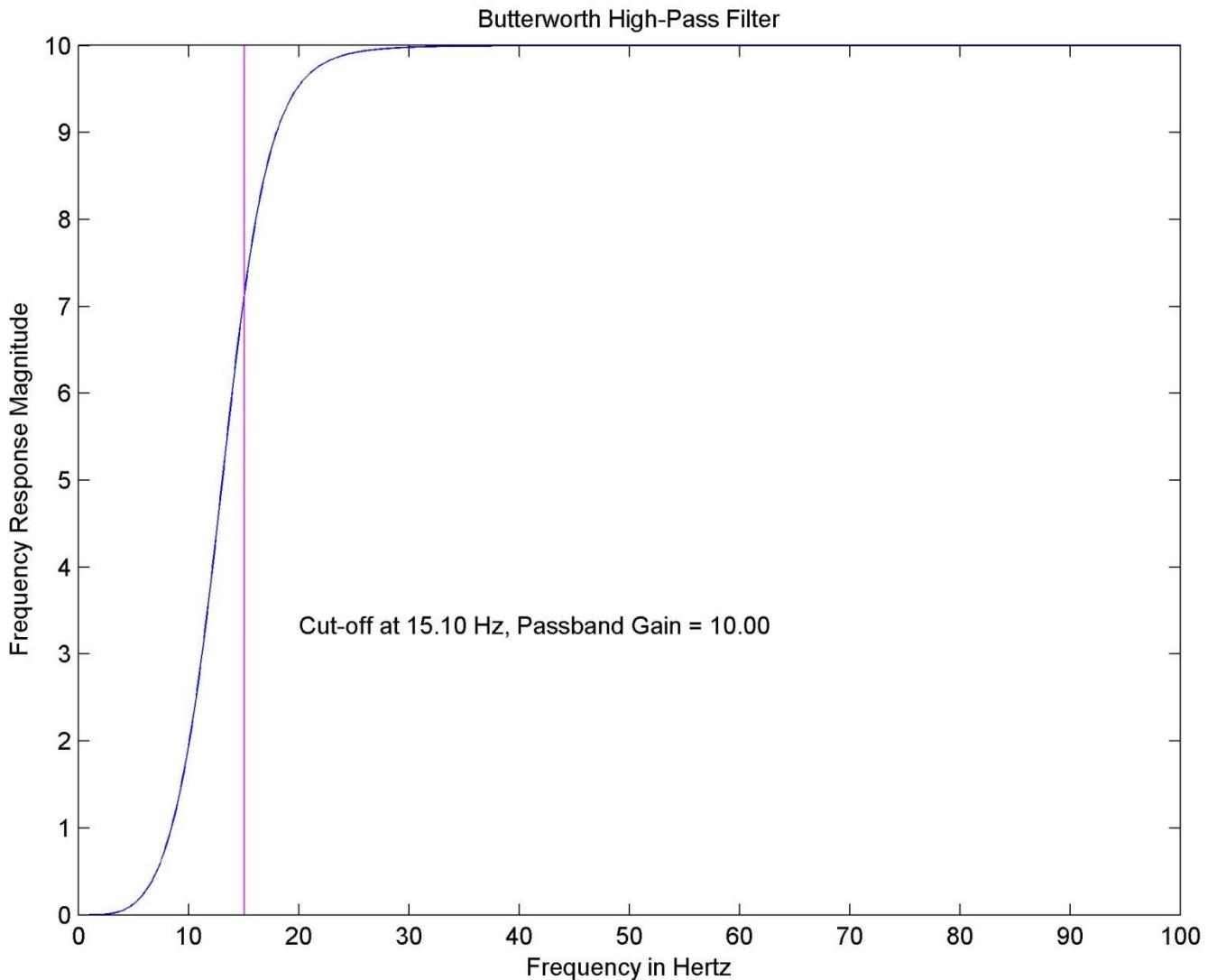
⇒ Circuit:



Continued

Example Highpass Design (2)

**Magnitude response of 4th-order,
Butterworth, $f_c = 15$ Hz, $A_T = 10$**



MATLAB Design Software (1)

9/18/16 8:52 PM C:\Users\Ted\Desktop...\butter_hi_design.m 1 of 3

```

function H = butter_hi_design(f, Fc, A, Stage1, Stage2, Stage3, Stage4, Stage5)
%
% H = b_hi_des(f, Fc, A, Stage1[, Stage2[, Stage3[, Stage4[, Stage5]]]])
%
% Helps design an electronic circuit to build even-order, high-pass
% Butterworth filters of orders 2, 4, 6, 8, 10. Plots the magnitude of
% the resulting frequency response.
% See [Thomas Kugelstadt, "Chapter 16: Active
% Filter Design Techniques," Literature Number SLOA088, Texas Instruments
% Incorporated, Post Office Box 655303, Dallas, Texas 75265, 2001.
% Excerpted from "Op Amps for Everyone," Literature Number SLOD006A,
% Texas Instruments. Available on the Internet at: http://www.ti.com.
%
% f: Frequency axis (Hertz) for all calculations and plotting (vector).
% A: Overall circuit gain (>=1). Applied in 1st stage.
% Fc: Desired cutoff frequency in Hertz (scalar).
% StageX: Up to 5 StageX arguments (one per stage) are permitted. For each stage,
% StageX is a vector of 1-4 elements, corresponding to C1, C2,
% R1 and R2, respectively. C1 must be supplied (in Farads) and
% is the first vector element. If a second argument is supplied,
% then it is C2 (in Farads). If C2 is not supplied, it is set
% equal to C1. If
% a third argument is supplied, it is R1 (Ohms). If not supplied,
% it is set as required to form a Butterworth filter. If a fourth
% argument is supplied, it is R2 (Ohms). If not supplied, it is
% set as required to form a Butterworth filter.
%
% H: Resulting (complex-valued) frequency response corresponding to the
% frequency axis f.
%
% USAGE RECOMENDATIONS: To build a filter, initially call script with only
% C1 and C2 specified for each stage. Use the recommended
% R1 and R2 values to find R1's and R2's that are manufactured. Call the
% script a second time with all values to see the resultant nominal frequency
% response.

% Table of ai values. For even-order Butterworth, all bi values equal 1.
aiTable = [1.4142    NaN    NaN    NaN    NaN; % For 1-stage filter.
            1.8478 0.7654    NaN    NaN    NaN; % For 2-stage filter.
            1.9319 1.4142 0.5176    NaN    NaN; % For 3-stage filter.
            1.9616 1.6629 1.1111 0.3902    NaN; % For 4-stage filter.
            1.9754 1.7820 1.4142 0.9080 0.3129]; % For 5-stage filter.

%
% Determine the number of stages.
Nstage = nargin - 3;
if Nstage<1 | Nstage>5, error('Must have 1-5 stages.');" end
if A<1, error('"A" must be >= 1.');" end

%
% Extract/develop parameters for each stage and build the frequency response.

H = ones( 1, length(f) ); % Initialize frequency response to unity.

```



Continued

MATLAB Design Software (2)

9/18/16 8:52 PM C:\Users\Ted\Desktop...\butter_hi_design.m 2 of 3

```
w = 2*pi*f; % Convert to radian frequency.
MagBig = A; % Cascade passband gain.

for S = 1:Nstage

    if S>1, A = 1; end % Set gain to one after first stage.

    % Prepare to parse parameter vector.
    eval(['Param = Stage' int2str(S) ';']); % Copy parameters to a scratch vector.
    if length(Param)<1, error(['Stage ' int2str(S) ' parameter list < 1.']); end;
    if length(Param)>4, error(['Stage ' int2str(S) ' parameter list > 4.']); end;
    ai = aiTable(Nstage,S);

    % Parse parameter vector.
    % C1.
    C1 = Param(1);
    % C2.
    if length(Param)>1, C2 = Param(2); else, C2 = C1; end
    % R1.
    if length(Param)>2, R1 = Param(3);
    elseif A==1
        R1 = ( C1+C2 ) / ( 2*pi*Fc*ai*C1*C2 );
    else
        a = ((2*pi*Fc)^2) * C1*C2*C2*(1-A);
        b = -ai * 2*pi*Fc * C1*C2;
        c = C1 + C2;
        R1 = ( -b - sqrt(b*b - 4*a*c) ) / (2*a);
    end
    % R2.
    if length(Param)>3, R2 = Param(4);
    else, R2 = 1 / ( ((2*pi*Fc)^2)*R1*C1*C2 ); end

    % Print component values.
    fprintf('Stage %d: C1=%e F, C2=%e F, R1=%e Ohms, R2=%e Ohms\n', S, C1, C2, R1, R2);

    % Update frequency response.
    a = ( R2*(C1+C2) + R1*C2*(1-A) ) ./ (R1*R2*C1*C2);
    b = 1 ./ (R1*R2*C1*C2);
    Hstage = A ./ ( 1 + ( a./(j*w) ) + ( b./(-w.*w) ) );
    H = H .* Hstage;
end

% Now, plot the frequency response magnitude.
plot( f, abs(H) )
xlabel('Frequency in Hertz')
ylabel('Frequency Response Magnitude')

L1 = find( abs(H) >= (MagBig*sqrt(2)/2) );
hold on, plot([f(L1(1)) f(L1(1))], [0 MagBig], 'm'), hold off
Thing = sprintf('Cut-off at %0.2f Hz, Passband Gain = %0.2f', f(L1(1)), MagBig);
text( max(f)/5, max(abs(H))/3, Thing);
```

Continued

MATLAB Design Software (3)

9/18/16 8:52 PM C:\Users\Ted\Desktop...\butter_hi_design.m 3 of 3

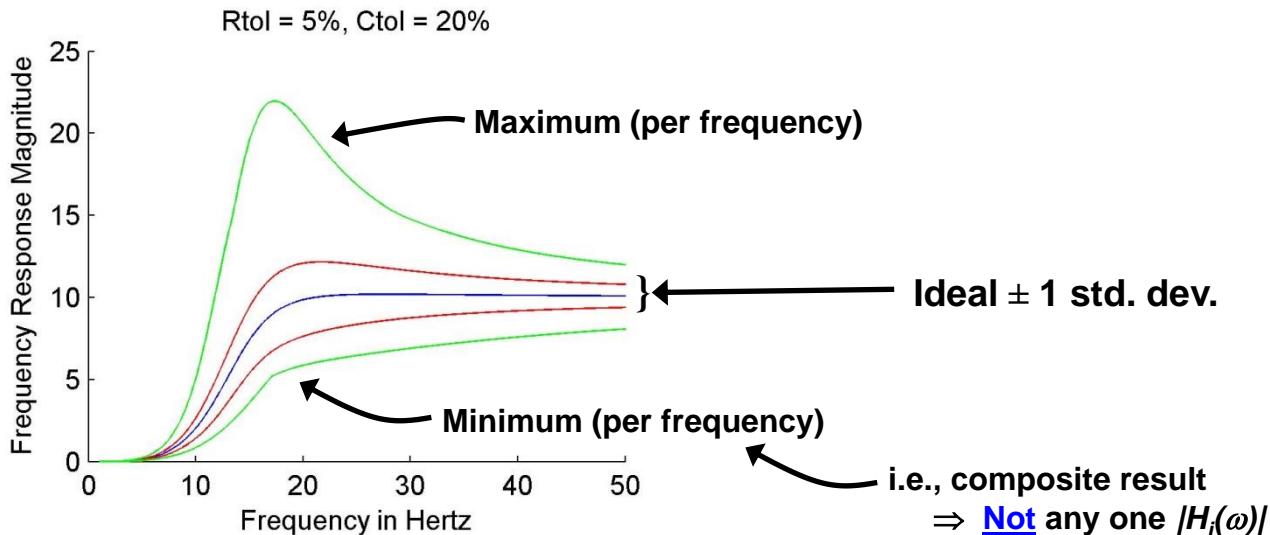
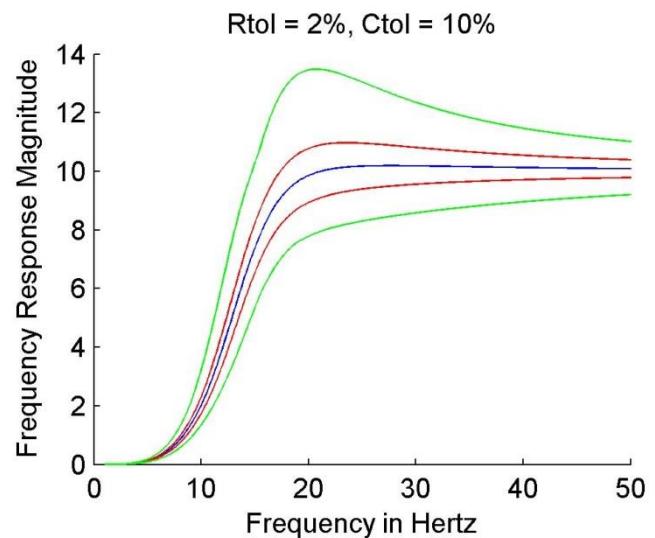
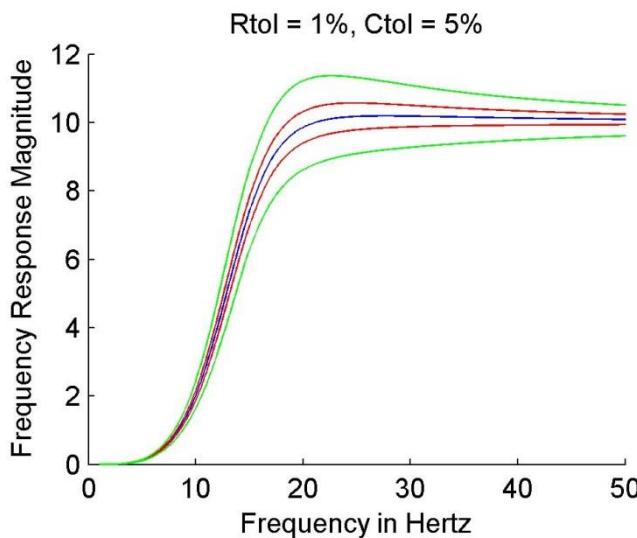
```
title('Butterworth High-Pass Filter');

figure(gcf);

return
```

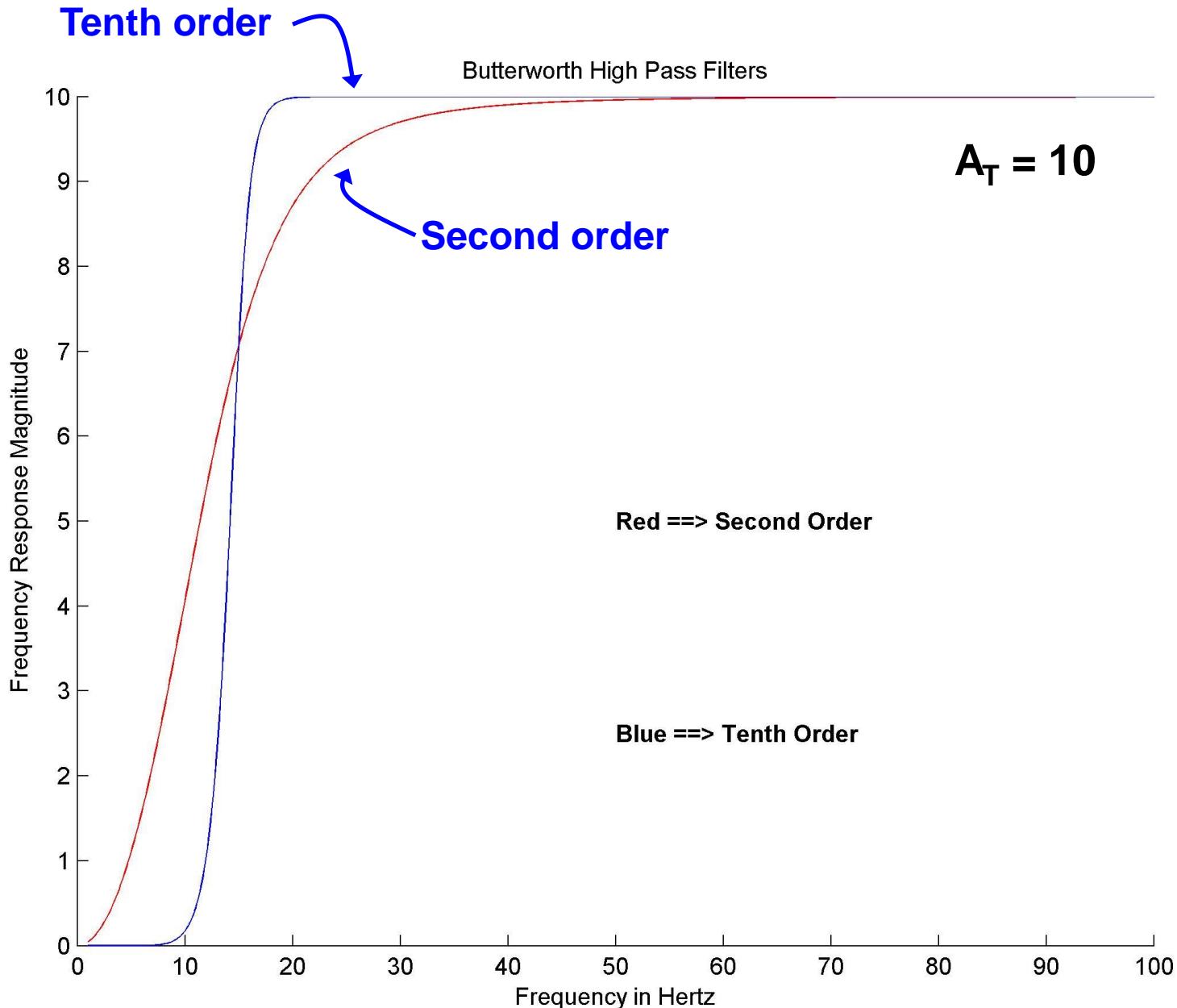
Influence of Component Tolerances

- Fourth-order, Butterworth, $f_c = 15 \text{ Hz}$, $A_T = 10$
 - All gain in first stage
 - Random R 's, C 's using tolerance
 - $N = 1000$ iterations per condition



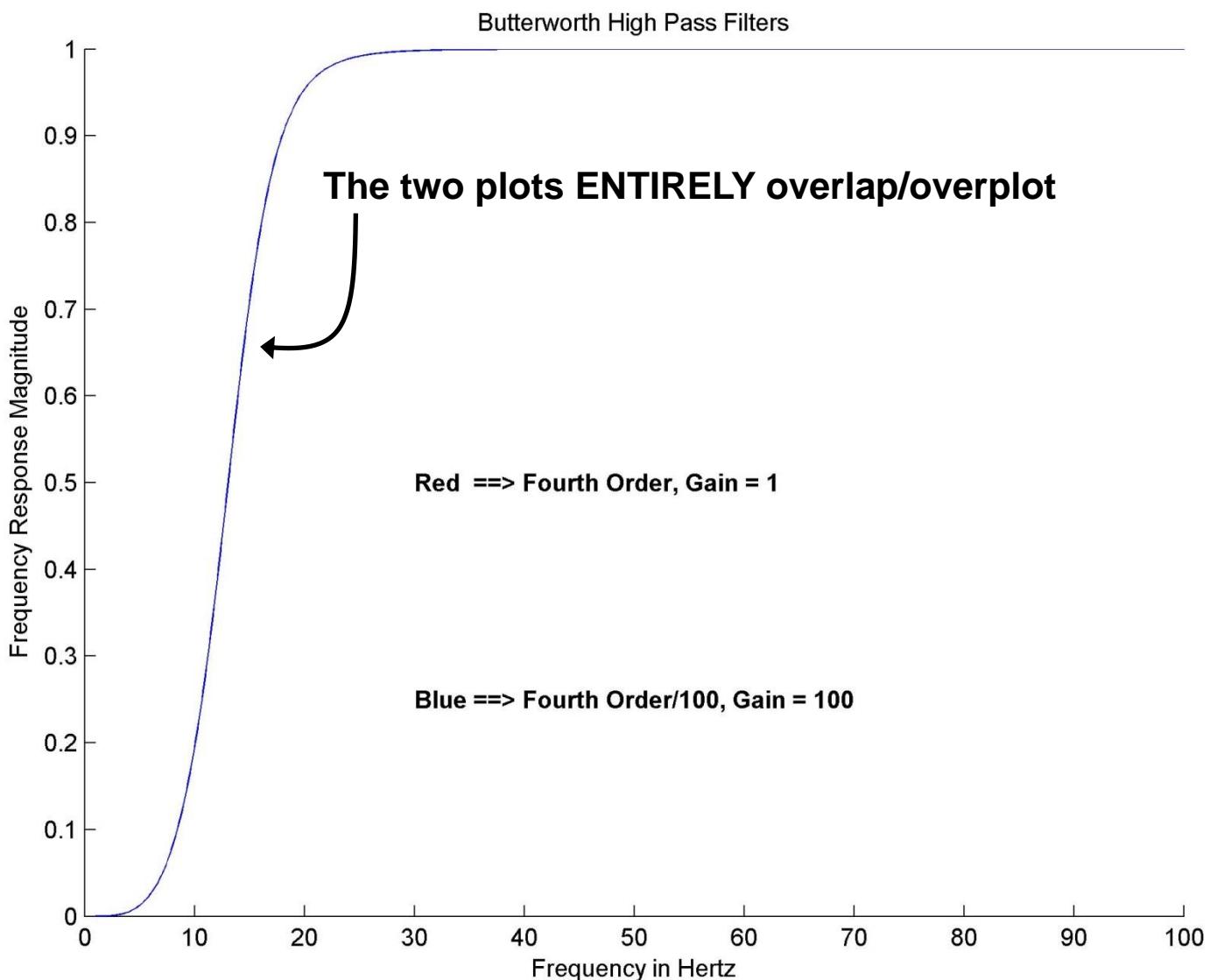
Comparison of Filter Orders

Magnitude Response of Butterworth Filter



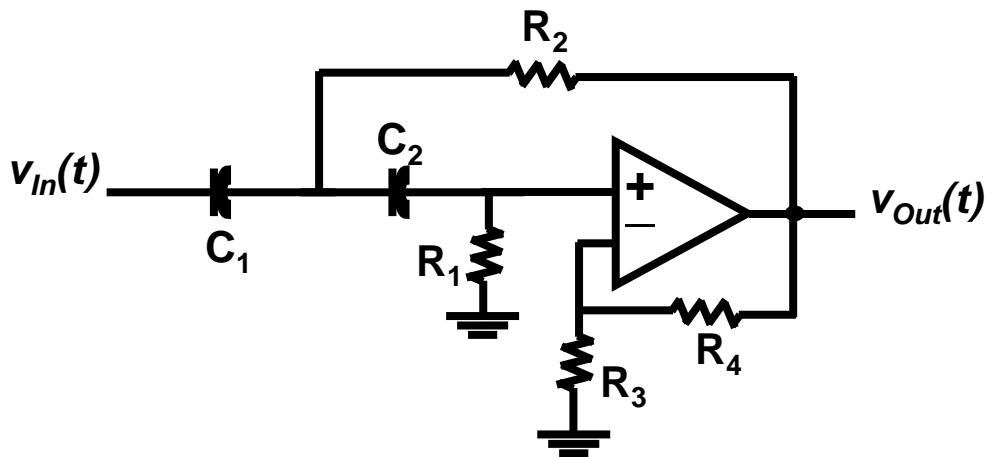
Comparison of Shape vs. Gain

- Red (covered): Fourth-order, Butterworth, $A_T = 1$, $f_c = 15$ Hz
- Blue: Fourth-order, Butterworth, $f_c = 15$ Hz
 - $A_T = 100$, then (to compare shape) divide $\frac{|H_{Blue}(\omega)|}{100}$

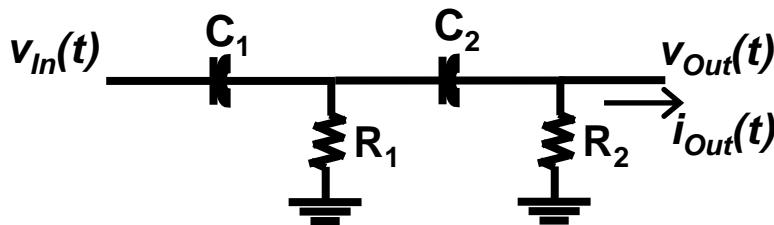


Active vs. Passive Filters

Active:



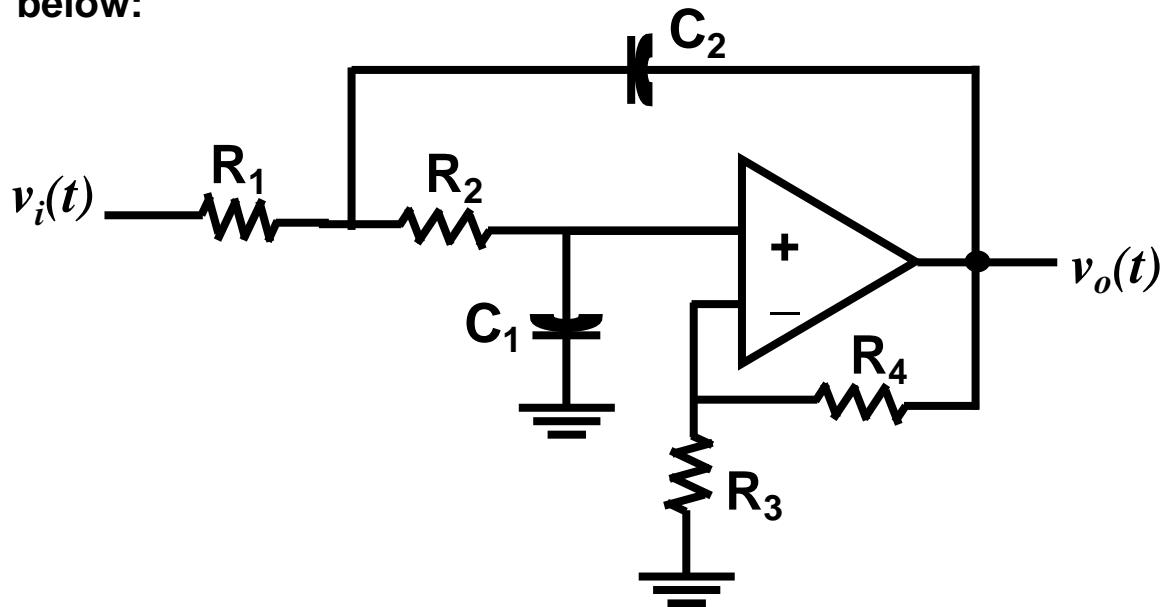
Passive:



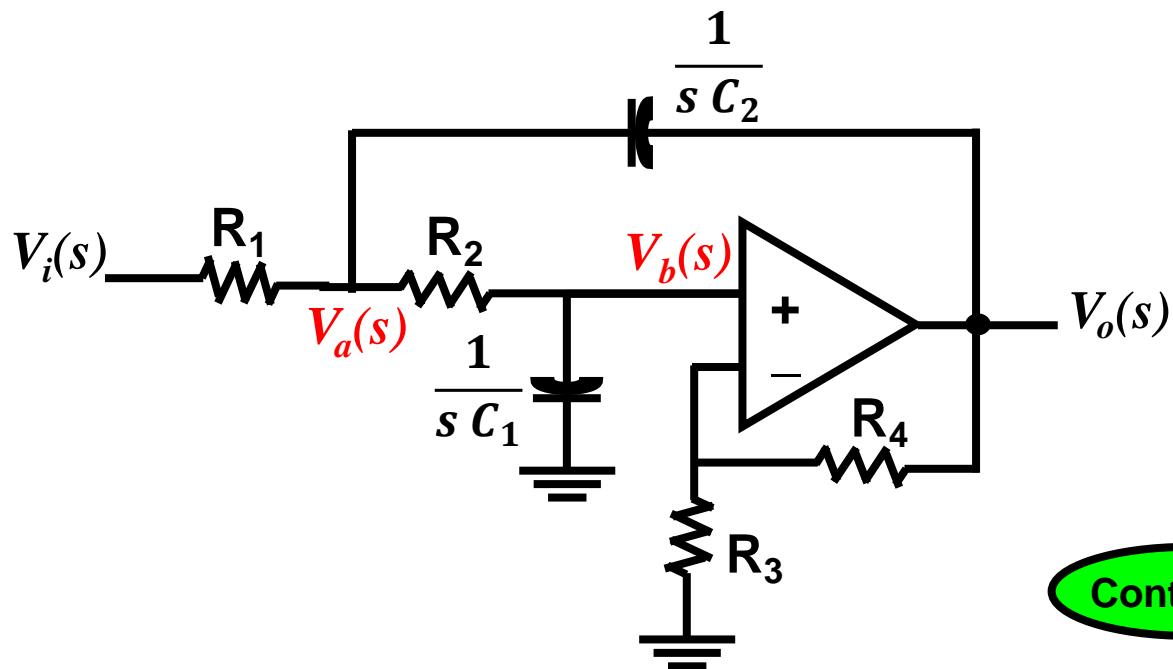
	PROs	CONs
Active	<ul style="list-style-type: none"> • Low output impedance ⇒ easy to cascade • Can apply gain 	<ul style="list-style-type: none"> • More components • Feedback can be unstable • Need a power supply
Passive	<ul style="list-style-type: none"> • Fewer components • Always stable (No feedback) 	<ul style="list-style-type: none"> • No gain • High output impedance ⇒ i_{Out} influences $H(\omega)$ • Complicated to cascade

Frequency Response of Sallen-Key Lowpass Filter (1)

- Find the frequency response of the Sallen-Key lowpass filter, shown below:



- In doing so: (1) Define $A_0 = 1 + \frac{R_4}{R_3} = \frac{R_3 + R_4}{R_3}$
(2) Use "s" as the frequency variable



Continued

Frequency Response of Sallen-Key Lowpass Filter (2)

(1) By voltage division: $V_b(s) = V_o(s) \cdot \frac{R_3}{R_3+R_4} = \frac{V_o(s)}{A_0}$

(2) KCL at node V_a , then multiply by $R_1 R_2$ and substitute (1) for $V_b(s)$:

$$\frac{V_a(s) - V_i(s)}{R_1} + [V_a(s) - V_o(s)] \cdot s C_2 + \frac{V_a(s) - V_b(s)}{R_2} = 0$$

$$V_a(s) \cdot R_2 - V_i(s) \cdot R_2 + V_a(s) \cdot s R_1 R_2 C_2 - V_o(s) \cdot s R_1 R_2 C_2 \\ + V_a(s) \cdot R_1 - V_o(s) \cdot \frac{R_1}{A_0} = 0$$

$$V_a(s) \cdot A_0 [R_1 + R_2 + s R_1 R_2 C_2] \\ = V_i(s) [A_0 R_2] + V_o(s) \left[s R_1 R_2 C_2 + \frac{R_1}{A_0} \right] \cdot A_0$$

(3) KCL at node V_b , then substituting (1) for $V_b(s)$:

$$\frac{V_b(s) - V_a(s)}{R_2} + V_b(s) [s C_1] = 0$$

$$V_o(s) - V_a(s) [A_0] + V_o(s) [s R_2 C_1] = 0$$

$$V_a(s) = \frac{1}{A_0} \cdot V_o(s) [1 + s R_2 C_1]$$

Continued

Frequency Response of Sallen-Key Lowpass Filter (3)

- Substituting (3) into (2) :

$$V_o(s)[1 + sR_2C_1][R_1 + R_2 + sR_1R_2C_2] \\ = V_i(s)[A_0R_2] + V_o(s)[sA_0R_1R_2C_2 + R_1]$$

$$\Rightarrow V_o(s)[s^2R_1R_2^2C_1C_2 + s(R_1R_2C_1 + R_2^2C_1 + R_1R_2C_2) + (R_1 + R_2)] \\ - V_o(s)[sA_0R_1R_2C_2 + R_1] = V_i(s)[A_0R_2]$$

$$\Rightarrow H(s) = \frac{V_o(s)}{V_i(s)} \\ = \frac{A_0R_2}{s^2(R_1R_2^2C_1C_2) + s(R_1R_2C_1 + R_2^2C_1 + R_1R_2C_2 - A_0R_1R_2C_2) + R_2}$$

$$H(s) = \frac{A_0}{s^2(R_1R_2C_1C_2) + s[(R_1 + R_2)C_1 + (1 - A_0)R_1C_2] + 1}$$

or

$$H(\omega) \\ = \frac{A_0}{(j\omega)^2[R_1R_2C_1C_2] + (j\omega)[(R_1 + R_2)C_1 + (1 - A_0)R_1C_2] + 1}$$

At $\omega = 0$, $H(\omega) = A_0$.

At $\omega = \infty$, $H(\omega) = 0$.

Lowpass Design: Selecting R 's, C 's

$$H_{Low, Norm}(\omega) = \frac{A_0}{1 + a(j\omega) + b(j\omega)^2},$$

where, $a = \omega_c \cdot C_1(R_1 + R_2) + \omega_c \cdot (1 - A_0)R_1C_2$

$$b = \omega_c^2 \cdot R_1 R_2 C_1 C_2$$

$\omega_c \Rightarrow$ Desired cut-off frequency

- Again: R_1, R_2, C_1, C_2 underdetermined (each stage)

1. Select $C_1 = C_2 =$ Some Easy Value

→ Pick C 's first, because fewer manufactured values

2. R_1 :

– If $A_i \geq 1$, then

$$R_1 = \frac{a_i \omega_c^2 C_2 - \sqrt{a_i^2 \omega_c^4 C_2^2 - 4 \omega_c^4 b_i C_2 [C_1 + (1 - A_0) C_2]}}{2 \omega_c^3 C_2 [C_1 + (1 - A_0) C_2]}$$

$$3. \quad R_2 = \frac{b_i}{\omega_c^2 \cdot R_1 C_1 C_2}$$

4. Replace ideal R_1, R_2 with manufactured values

5. Repeat for each stage: $A_T = A_1 \cdot A_2 \cdot A_3 \cdots A_N$

Example Lowpass Design

- Design: Fourth-order, Butterworth, $f_c = 1800$ Hz, $A_T = 1$

– Choose all C 's = 10 nF, R_3 's = $\infty \Omega$, R_4 's = 0 Ω



Unity Gain

– Stage 1: $R_1 = R_2 = 8.2$ k Ω

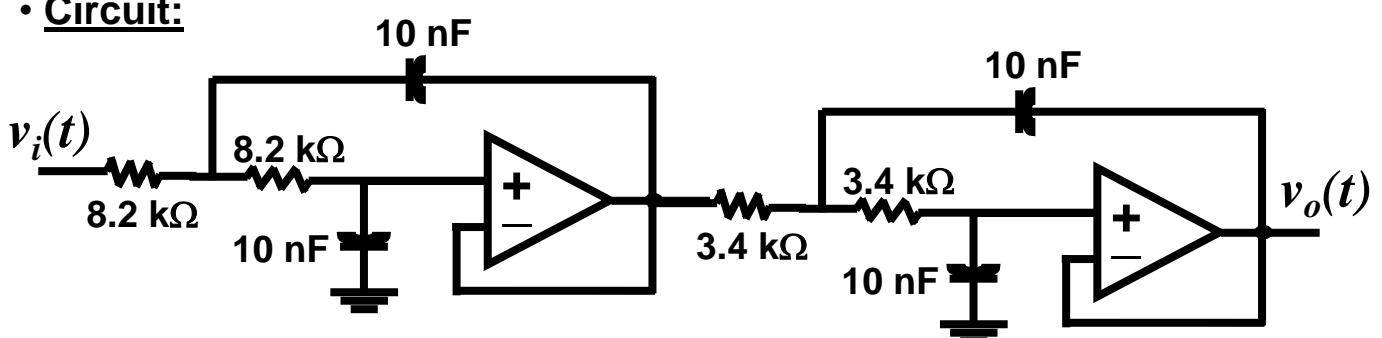
– Stage 2: $R_1 = R_2 = 3.4$ k Ω

From Kugelstadt:

$a_1 = 1.8478$, $b_1 = 1$ (Stage 1)

$a_2 = 0.7654$, $b_1 = 1$ (Stage 2)

- Circuit:



- Magnitude Response:

Butterworth Low-Pass Filter

