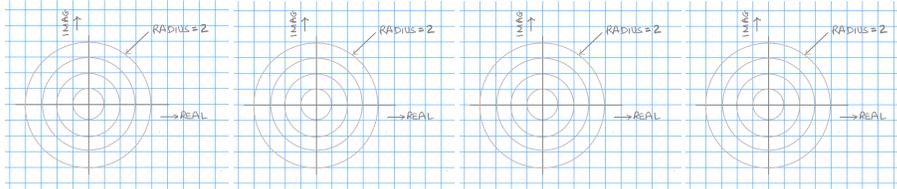


This homework is due on Wednesday, February 26, 2020, at 11:59PM.

Self-grades are due on Monday, March 2, 2020, at 11:59PM.

## 1 Phasors

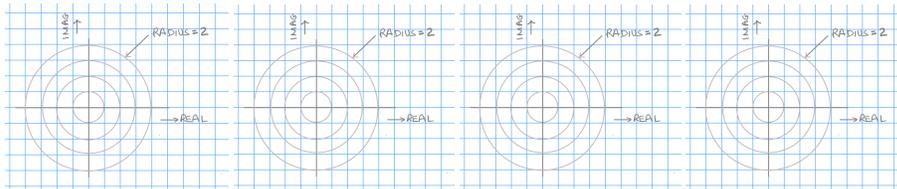
- a) Consider a resistor ( $R = 1.5\Omega$ ), a capacitor ( $C = 1F$ ), and an inductor ( $L = 1H$ ) connected in series. Give expressions for the impedances of  $Z_R, Z_C, Z_L$  for each of these elements as a function of the angular frequency  $\omega$ .
- b) Draw the individual impedances as “vectors” on the same complex plane for the case  $\omega = \frac{1}{2}$  rad/sec. Also draw the combined impedance  $Z_{total}$  of their series combination. Give the magnitude and phase of  $Z_{total}$ . A logically sound graphical argument is sufficient justification.



(a)  $Z_R(@\omega = 0.5)$     (b)  $Z_C(@\omega = 0.5)$     (c)  $Z_L(@\omega = 0.5)$     (d)  $Z_{total}(@\omega = 0.5)$

Figure 1: Impedances at  $\omega = 0.5$ .

- c) Draw the individual impedances as “vectors” on the same complex plane for the case  $\omega = 1$  rad/sec. Also draw the combined impedance  $Z_{total}$  of their series combination. Give the magnitude and phase of  $Z_{total}$ . A logically sound graphical argument is sufficient justification.



(a)  $Z_R(@\omega = 1)$     (b)  $Z_C(@\omega = 1)$     (c)  $Z_L(@\omega = 1)$     (d)  $Z_{total}(@\omega = 1)$

Figure 2: Impedances at  $\omega = 1$ .

- d) Draw the individual impedances as “vectors” on the same complex plane for the case  $\omega = 2$  rad/sec. Also draw the combined impedance

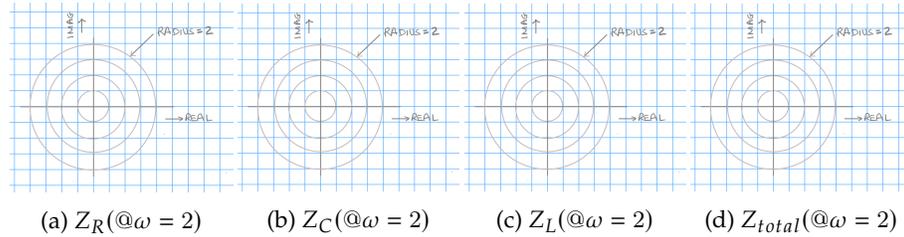


Figure 3: Impedances at  $\omega = 2$ .

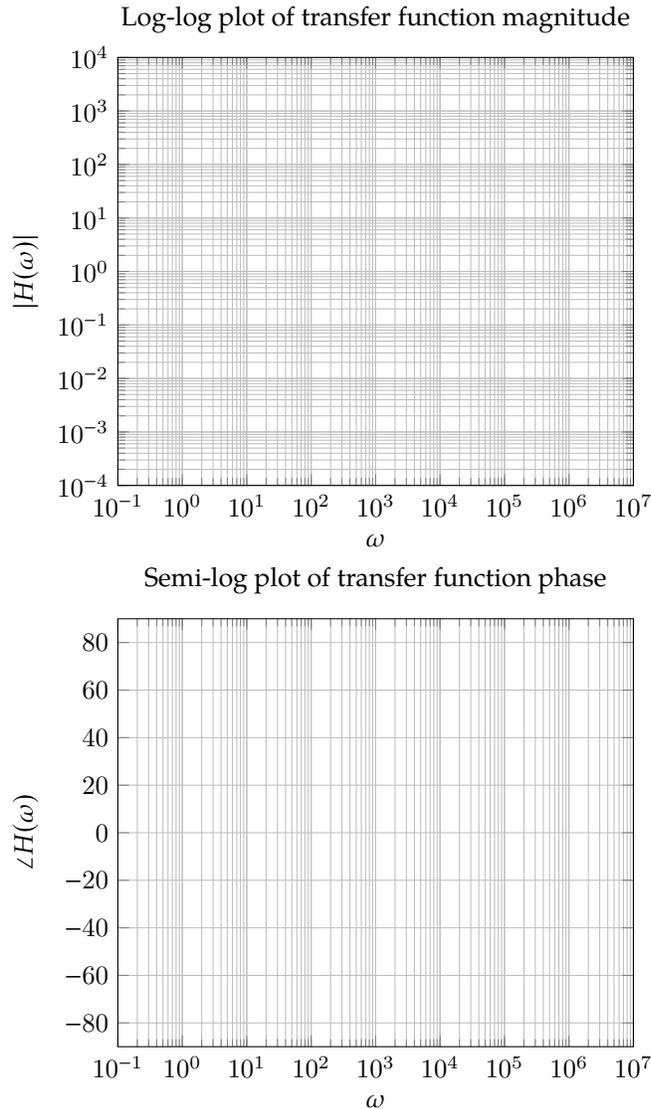
$Z_{total}$  of their series combination. Give the magnitude and phase of  $Z_{total}$ . A logically sound graphical argument is sufficient justification.

- e) For the previous series combination of RLC elements, what is the “natural frequency”  $\omega_n$  where the series impedance is purely real?

## 2 Low-pass Filter

You have a  $1\text{ k}\Omega$  resistor and a  $1\text{ }\mu\text{F}$  capacitor wired up as a low-pass filter.

- Draw the filter circuit, labeling the input node, output node, and ground.
- Write down the transfer function of the filter,  $H(j\omega)$  that relates the output voltage phasor to the input voltage phasor. Be sure to use the given values for the components.
- Write an exact expression for the *magnitude* of  $H(j\omega = j10^6)$ , and give an approximate numerical answer.
- Write an exact expression for the *phase* of  $H(j\omega = j1)$ , and give an approximate numerical answer.
- Write down an expression for the time-domain output waveform  $V_{out}(t)$  of this filter if the input voltage is  $V(t) = 1 \sin(1000t)\text{ V}$ . You can assume that any transients have died out — we are interested in the steady-state waveform.
- Use a computer or calculator to help you sketch the Bode plot (both magnitude and phase) of the filter on the graph paper below.



### 3 Color Organ Filter Design

In the fourth lab, we will design low-pass, band-pass, and high-pass filters for a color organ. There are red, green, and blue LEDs. Each color will correspond to a specified frequency range of the input audio signal. The intensity of the light emitted will correspond to the amplitude of the audio signal.

- a) First, you realize that you can build simple filters using a resistor and a capacitor. Design the first-order **passive** low and high pass filters with

following frequency ranges for each filter using 1  $\mu\text{F}$  capacitors. (“Passive” means that the filter does not require any power supply.)

- Low pass filter – 3-dB frequency at 2400 Hz =  $2\pi \cdot 2400 \frac{\text{rad}}{\text{sec}}$
- High pass filter – 3-dB frequency at 100 Hz =  $2\pi \cdot 100 \frac{\text{rad}}{\text{sec}}$

Draw the schematic-level representation of your designs and show your work finding the resistor values. Also, please mark  $V_{\text{in}}$ ,  $V_{\text{out}}$ , and ground nodes in your schematic. Round your results to two significant figures.

- b) You decide to build a bandpass filter by simply cascading the first-order low-pass and high-pass filters you designed in part (a). Connect the  $V_{\text{out}}$  node of your low-pass filter directly to the  $V_{\text{in}}$  node of your high pass filter. The  $V_{\text{in}}$  of your new band-pass filter is the  $V_{\text{in}}$  of your old low-pass filter, and the  $V_{\text{out}}$  of the new filter is the  $V_{\text{out}}$  of your old high-pass filter. What is  $H_{\text{BPF}}$ , the transfer function of your new band-pass filter? Use  $R_L$ ,  $C_L$ ,  $R_H$ , and  $C_H$  for low-pass filter and high-pass filter components, respectively. Show your work.
- c) Plug the component values you found in (a) into the transfer function  $H_{\text{BPF}}$ . Using MATLAB or iPython, draw a Bode plot from 0.1 Hz to 1 GHz. If you use iPython, you may find the function `scipy.signal.bode` useful. What are the frequencies at which the numerator and denominator of the transfer function become zero? What is the maximum magnitude of  $H_{\text{BPF}}$  in dB? Is that something that you want? If not, explain why not and suggest a simple way (either adding passive or active components) to fix it.
- d) Now that you know how to make filters and amplifiers, we can finally build a system for the color organ circuit below. Before going into the actual schematic design, you must first set specifications for each block. The goal of the circuit is to divide the input signal into three frequency bands and turn the LEDs on based on the input signal’s frequency.

In this problem, assume that the mic board’s transfer function is of the following form:

$$V_{\text{MIC}} = K_{\text{MIC}} \frac{j\omega \left(1 + \frac{j\omega}{\omega_{z1}}\right)}{\left(1 + \frac{j\omega}{\omega_{p1}}\right) \left(1 + \frac{j\omega}{\omega_{p2}}\right) \left(1 + \frac{j\omega}{\omega_{p3}}\right)}$$

where  $K_{\text{MIC}}$  is a constant gain,  $\omega_{z1} = 2\pi \cdot 200\text{Hz}$ ,  $\omega_{p1} = 2\pi \cdot 10\text{Hz}$ ,  $\omega_{p2} = 2\pi \cdot 100\text{Hz}$ , and  $\omega_{p3} = 2\pi \cdot 10\text{KHz}$ . The magnitude of the voltage at the mic board output is 1 V peak-to-peak at 40 Hz. (*Hint*: You can use this information to calculate  $K_{\text{MIC}}$ .)

Suppose that the three filters have transfer functions as below.

- Low pass filter

$$H_{LPF} = \frac{2}{1 + \frac{j\omega}{200\pi}}$$

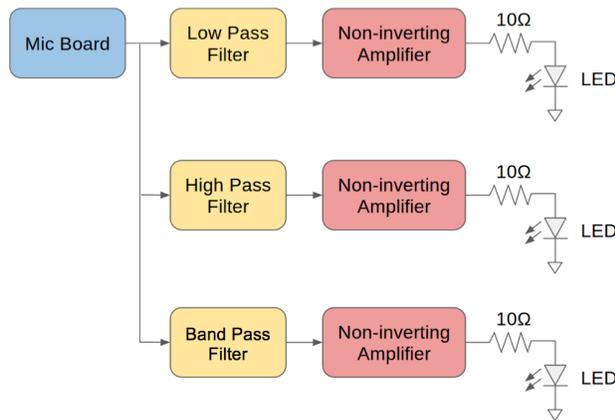
- Band pass filter

$$H_{BPF} = \frac{4.54 \cdot 10^{-4} j\omega}{\left(1 + \frac{j\omega}{400\pi}\right) \left(1 + \frac{j\omega}{4000\pi}\right)}$$

- High pass filter

$$H_{HPF} = \frac{\frac{j\omega}{8000\pi}}{1 + \frac{j\omega}{8000\pi}}$$

What are the phasor voltages at the output of each filter as a function of  $\omega$ ? To clarify,  $\frac{3(1+j\omega(1.5 \cdot 10^3))}{1+j\omega(2 \cdot 100)}$  would be a valid phasor voltage at the output of some filter. Assume that there are ideal voltage buffers before and after each filter.



- e) For 50 Hz, 1000 Hz, and 8000 Hz, what is the voltage gain required of each non-inverting amplifier such that the output peak to peak voltage measured right before the  $10\ \Omega$  resistor is  $5\ V_{pp}$ ?

## 4 Mystery Microphone

You are working for Mysterious Miniature Microphone Multinational when your manager asks you to test a batch of the company's new microphones. You grab one of the new microphones off the shelf, use a tone generator <sup>1</sup> to play pure tones of uniform amplitude at various frequencies, and measure the resultant peak-to-peak voltages using an oscilloscope. You collect data, and then plot it (on a logarithmic scale). The plot is shown below:

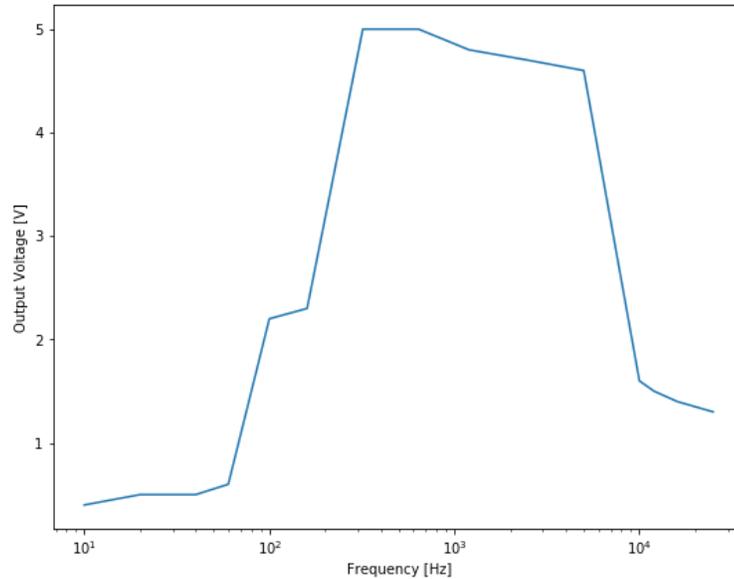


Figure 4: Frequency Response

- a) To which frequencies is the microphone most sensitive, and to which frequencies is the microphone least sensitive?

You report these findings to your manager, who thanks you for the preliminary data and proceeds to co-ordinate some human listener tests. In the meantime, your manager asks you to predict the effects of the microphone recordings on human listeners, and encourages you to start thinking more deeply about the relationships.

- b) For testing purposes, you have a song with sub-bass (150 Hz or less), mid-range ( $\approx 1$  kHz), and some high frequency electronic parts ( $> 12$  kHz).

<sup>1</sup>Note that soundwaves are simply sinusoids at various frequencies with some amplitude and phase. The microphone's diaphragm oscillates with the sound (pressure) waves, moving the attached wire coil back and forth over an internal magnet, which induces a current in the wire. In this way, a microphone can be modeled as a signal-dependent current source. The output current can be converted to a voltage by simply adding a known resistor to the circuit and measuring the voltage across that resistor.

Which frequency ranges of the song would you be able to hear easily, and which parts would you have trouble hearing? Why?

- c) After a few weeks, your manager reports back to you on the findings. Apparently, this microphone causes some people's voices to sound really weird, resulting in users threatening to switch to products from a competing microphone company.

It turns out that we can design some filters to "fix" the frequency response so that the different frequencies can be recorded more equally, thus avoiding distortion. Imagine that you have a few (say up to 4 or so) blocks. Each of these blocks detects a set range of frequencies, and if the signal is within this range, it will switch on a op-amp circuit of your choice. For example, it can be configured to switch on an op-amp filter to double the voltage for signals between 100 Hz and 200 Hz.

What ranges of signals would require such a block, and what gain would you apply to each block such that the resulting peak-to-peak voltage is about 5 V for all frequencies?

## 5 (OPTIONAL) Write Your Own Question And Provide a Thorough Solution.

Writing your own problems is a very effective way to really learn material. Having some practice at trying to create problems helps you study for exams much better than simply solving existing practice problems. This is because thinking about how to create an interesting problem forces you to really consolidate your understanding of the course material.

## 6 Homework Process and Study Group

Citing sources and collaborators are an important part of life, including being a student! We also want to understand what resources you find helpful and how much time homework is taking, so we can change things in the future if possible.

- What sources (if any) did you use as you worked through the homework?**
- If you worked with someone on this homework, who did you work with?** List names and student ID's. (In case of homework party, you can also just describe the group.)
- How did you work on this homework?** (For example, *I first worked by myself for 2 hours, but got stuck on problem 3, so I went to office hours. Then I went to homework party for a few hours, where I finished the homework.*)
- Roughly how many total hours did you work on this homework?**