

## EE 102: Signal Processing and Linear Systems

**Homework #10: Filter Design**

Name: \_\_\_\_\_

Submission Date: \_\_\_\_\_

**Problem 1** An underdamped series RLC circuit is excited by an input at time  $t_0$ . The circuit outputs a voltage that is a damped oscillation at its natural frequency. This output signal is modeled as

$$x(t) = A e^{-\alpha(t-t_0)} \cos(\omega_0(t-t_0)) u(t-t_0), \quad \alpha > 0, \omega_0 > 0,$$

where  $u(\cdot)$  is the unit step.

(a) **(6 points)** Write two time-domain properties of  $x(t)$  that you can observe directly from the signal equation or by graphing it.

(b) **(2 points)** Write a signal  $\bar{x}(t)$  that when shifted by  $t_0$  gives  $x(t)$ .

(c) **(10 points)** Compute the Fourier transform  $X(\omega)$  of  $x(t)$ .

Hint: Use Euler's formula to express the cosine as a sum of complex exponentials — don't solve integration by parts!

(d) **(5 points)** Compute the Fourier transform  $\bar{X}(\omega)$  of  $\bar{x}(t)$ . Using the time-shift property, prove that  $X(\omega) = e^{-j\omega t_0} \bar{X}(\omega)$  to verify your result.

(e) **(5 points)** Sketch  $|X(\omega)|$ . Indicate the spectral peaks (the frequency peaks) at  $\pm\omega_0$ .

(f) **(6 points)** Define an ideal low-pass filter as follows:

$$H_{LP}(\omega) = \begin{cases} 1, & |\omega| \leq \Omega_c, \\ 0, & |\omega| > \Omega_c. \end{cases}$$

Sketch  $H_{LP}(\omega)$  for positive and negative  $\omega$ . Label  $\pm\Omega_c$  on your graph.

(g) **(8 points)** By using the inverse Fourier transform, compute the impulse response of the low-pass filter

**Hint:** use even symmetry of  $H_{LP}(\omega)$ . Solved example 4.5 in Oppenheim & Willsky 2nd Edition has the answer.

**(h) (10 points)** We know that if we ‘apply’ the filter system to the input. That is, if there is a system with impulse response  $h_{\text{LP}}(t)$ , and we apply an input  $x(t)$  to this system to obtain an output  $y(t)$ . Using the convolution theorem, we know that in the frequency domain, we have that  $Y(\omega) = X(\omega) H_{\text{LP}}(\omega)$ . Compute  $Y(\omega)$  and discuss the properties of  $Y(\omega)$ .

**(i) (10 points)** Now, let’s do the same in time-domain. Write the convolution integral

$$y(t) = (x * h_{\text{LP}})(t) = \int_{-\infty}^{\infty} x(\tau) h_{\text{LP}}(t - \tau) d\tau.$$

Solve the convolution integral to compute  $y(t)$  in time-domain. Compute the inverse Fourier transform of  $Y(\omega)$  to obtain  $y(t)$  and verify your result.

**Important:** You will not be able to solve the convolution integral in closed-form. Instead, express your answer as an integral expression that can be evaluated numerically. Things can be quite difficult in time-domain!

**(j) (10 points)** Using Parseval’s theorem, we can define a new type of energy in the frequency domain that is band-limited to  $|\omega| \leq \Omega_c$ :

$$E_{\text{LP}} = \frac{1}{2\pi} \int_{-\Omega_c}^{\Omega_c} |X(\omega)|^2 d\omega,$$

this is the energy contained in the signal  $x(t)$  that lies within the passband of the low-pass filter.

Compute  $E_{\text{LP}}$  in terms of  $\alpha, \omega_0, \Omega_c$ .

**Hint:** use  $|a + j(b - \omega)|^{-2} = \frac{1}{a^2 + (b - \omega)^2}$  and  $\int \frac{d\omega}{a^2 + (\omega - b)^2} = \frac{1}{a} \arctan \frac{\omega - b}{a}$ . You may find it useful to ignore cross terms if  $a \ll \omega_0$ . How do you justify this approximation physically?

**(k) (10 points)** Explore other possible filters in frequency domain: low-pass, band-pass, and high-pass. Comment on which one might better isolate the ring-down mode in the RLC circuit. Refer to sketches,  $h(t)$ , and band-energy calculations. You don’t need to solve it out completely but make convincing arguments using Fourier analysis properties.

**Problem 2** Validate Problem 1 using computer programs. You may start on the coding problem before finishing Problem 1. It is **highly recommended that you start early with the homework so that you can get help during TA's Friday office hour.**

With a sampling rate  $F_s$  Hz, we have a time interval of  $\Delta t = 1/F_s$  and we can write the continuous-time signal above by substituting  $t = n\Delta t$  as a discrete-time signal:

$$x[n] = A e^{-\alpha(n-n_0)\Delta t} \cos(\omega_0(n - n_0)\Delta t) u[n - n_0].$$

- (a) **(6 points)** Choose parameters  $(A, \alpha, \omega_0, t_0)$  and  $F_s$ . Implement the equation above in a function. Then, plot  $x[n]$  versus  $n\Delta t$ . Label your graph.
- (b) **(8 points)** Re-use the previous homework assignment code for DFT implementation to compute the DFT with length  $N$  for the signal above. Plot  $|X[k]|$  versus discrete frequency. Indicate indices corresponding to  $\pm\omega_0$ . You may need to center the spectrum for better visualization.
- (c) **(8 points)** Implement the ideal low-pass filter by masking  $|\omega| \leq \Omega_c$  in the DFT domain. Compute  $y[n]$  by inverse DFT. Plot the filtered signal  $y[n]$  and comment on the filter's operation.
- (d) **(10 points)** Implement at least one other filter type (high-pass or band-pass, or something else of your choice). Provide the DFT mask, impulse response, and plot the filtered signal. Comment on how well the filter isolates the ring-down mode in the RLC circuit.
- (e) **(10 points)** Estimate band-limited input energy by summing  $|X[k]|^2$  over passband indices to verify your answer in Problem 1. Find out how well your filter isolates the ringing mode for your chosen parameters. Experiment with different parameters.