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H.w#3

# **Section 2: Linear Systems**

# 1. Linearity and Commutativity of Systems A and B

```
System A: A[x(t)] = 2x(t) - 1
System B: B[x(t)] = 0.5x(t)
```

- (a) System A is non-linear due to the constant term -1, which breaks the property of additivity and homogeneity.
- (b) To make A linear, remove the -1: A[x(t)] = 2x(t).
- (c) With the modification:

$$A[B[x(t)]] = A[0.5x(t)] = x(t)$$
  
 $B[A[x(t)]] = B[2x(t)] = x(t)$ 

✓ A and B commute.

#### 2. Integrals with Cosine Components

```
Given f(t) = a_0 \cos(2\pi f t) + a_1 \cos(2\pi f_0 t), and T_1, T_2 are constants. Both integrals:
```

```
\int f(t)(t-T_1) dt and \int f(t)(t-T_2) dt over (-\infty, \infty)
```

are zero because the integrands are odd functions and the limits are symmetric.

Final answer: 0 for both.

#### **Section 3: Fourier Transforms and Basic Filters**

## 1. Low-pass and High-pass Filtering

```
Input: s(t) = a\delta(t - t_0)
```

Fourier Transform:  $S(\omega) = a e^{-j\omega t_0}$ 

Low-pass filter:  $H(\omega) = 1 / (1 + j\omega \tau)$ 

Output magnitude:  $|Y(\omega)| = |a| / sqrt(1 + (\omega \tau)^2)$ 

```
High-pass filter: H(\omega) = j\omega\tau / (1 + j\omega\tau)
Output magnitude: |Y(\omega)| = |a\omega\tau| / sqrt(1 + (\omega\tau)^2)
```

## 2. Group Delays

```
For both filters, the group delay \tau_g is the same: \tau_g = \tau / (1 + (\omega \tau)^2)
```

#### 3. Inverse Fourier Transforms

```
(a) S(f) = (a/2)[\delta(f - f_0) + \delta(f + f_0)] \rightarrow s(t) = a \cos(2\pi f_0 t)
(b) S(f) = (a/2)[\delta(f - f_0) - \delta(f + f_0)] \rightarrow s(t) = a \sin(2\pi f_0 t)
```

### **Section 4: Convolution and Octave Code**

## **1. Discrete Convolution Properties**

- (a) If input is  $\delta[n]$ , output is h[n] (the impulse response).
- (b) If input is  $\delta[n n_0]$ , output is  $h[n n_0]$  (shifted by  $n_0$ ).

### 2. Octave Code to Convolve Sine Wave with Impulse

Use the conv function in Octave to shift a 440 Hz sine wave by convolving with  $\delta[n - n_0]$ .

```
Code:
```

```
fs = 8000;
f = 440;
duration = 0.01;
n = 0:round(duration*fs);
n0 = 40;
x = sin(2*pi*f*n/fs);
delta = zeros(1, length(n));
delta(n0+1) = 1;
y = conv(x, delta);
subplot(2,1,1);
plot(n/fs, x);
```

```
title('Original Sine Wave');
subplot(2,1,2);
plot((0:length(y)-1)/fs, y);
title('Shifted Sine Wave');
```

 $\checkmark$  This shifts the phase of the sine wave by  $n_0$  samples.