

# DSP Lab Solved problems

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## 1 FIR Filter Design for Signal Restoration

You are given a discrete-time signal  $x(n)$  which is a combination of a desired low-frequency signal and an unwanted high-frequency noise component. The signal is defined as:

$$x(n) = \cos(0.1\pi n) + 0.5\sin(0.8\pi n) \quad \text{for } 0 \leq n \leq 100$$

Question is:

1. Design a digital FIR low-pass filter to remove the high-frequency noise component.
2. Filter the input signal using designed filter

## Program

```
clearvars; close all;

% given signal
n = 0:100;
x = cos(0.1 * pi * n) + 0.5*sin(0.8 * pi * n);

%%% design LPF
wp = 0.2*pi;
ws = 0.5*pi;
TW = (ws - wp);
fc = (ws + wp)/2/pi;

% As = 60dB; so use blackman which has As = 74dB
M = ceil(11 * pi / TW);
m = 0:M-1;
m = m - ceil((M-1)/2);
w = blackman(M);

hd = fc * sinc(fc * m);
h = hd .* w'; % required filter
```

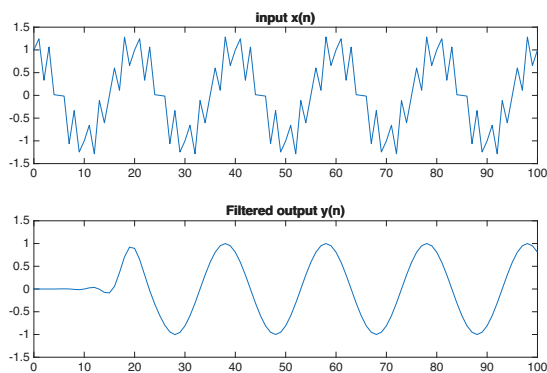
```

%% Filter the input signal
y = filter(h, 1, x);

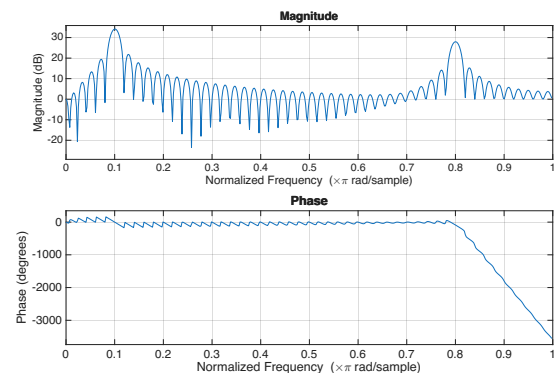
tiledlayout(2, 1);
nexttile; plot(n, x); title("input x(n)");
nexttile; plot(n, y); title("Filtered output y(n)");
% plot frequency responses and spectrums
figure; freqz(h);
figure; freqz(x);
figure; freqz(y);

```

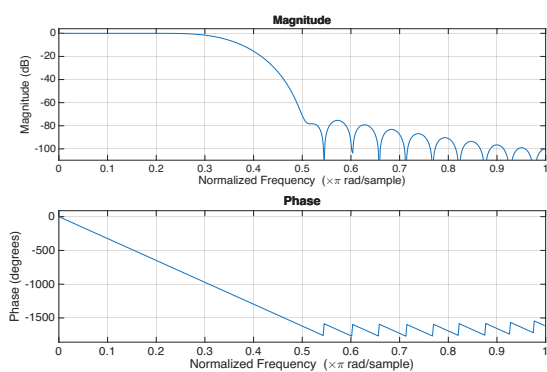
Plotted signal



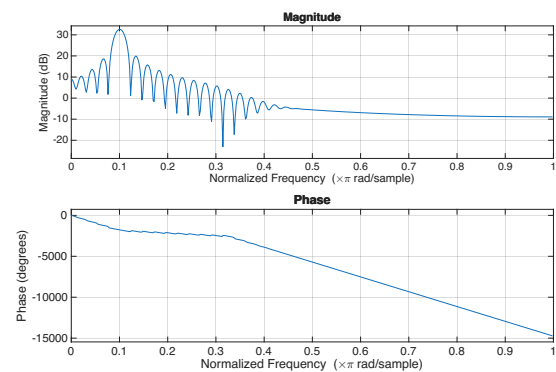
Input message spectrum



Filter's frequency response



Filtered Output spectrum



## 2 LTI System Stability Analysis and Modification

Consider the LTI system described by the following difference equation:

$$y(n) - 1.2y(n-1) + 0.8y(n-2) = x(n)$$

Question:

1. Determine if this system is stable or not
2. Plot poles and zeros of the system

3. Compute and plot the impulse response of the system for  $0 \leq n \leq 50$
4. Modify the system by changing only the coefficient of  $y(n-1)$  (the -1.2 term) so that the poles are located at  $p = 0.5 \pm 0.5j$ . Plot the pole-zero diagram for new system

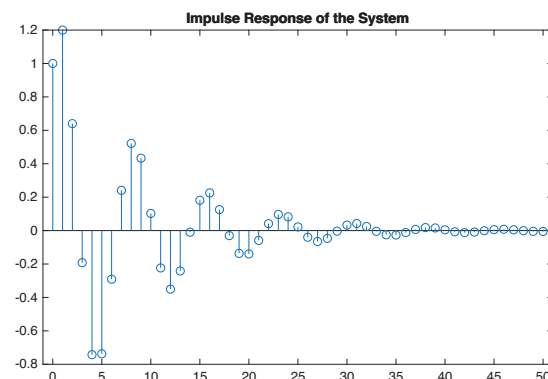
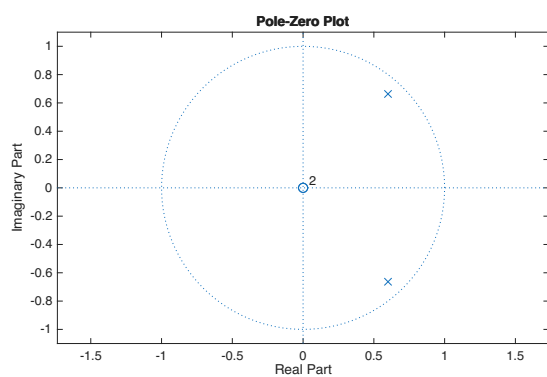
### Program for Question 1-3:

```
close all; clearvars;
% given system:  $y(n) - 1.2y(n-1) + 0.8y(n-2) = x(n)$ 
num = 1; den = [1, -1.2, 0.8];

%% plot zeros and poles
zplane(num, den);

%% check if system is stable or not
% by checking if all poles lie inside unit circle
system = filter(num, den);
if (abs(pole(system)) < 1)
    disp("stable");
else
    disp("unstable");
end

%% compute impulse response
n = 0:50;
impulse = n == 0;
response = filter(num, den, impulse);
figure;
stem(n, response);
title('Impulse Response of the System');
```



### Program For Question 4

Given system has 2 zeros at  $p = 0$  and 2 poles. To change the poles location to  $p = 0.5 \pm 0.5j$ , First write it in system function form, change denominator expressions and work backwards:

(here  $z^2$  implies double zero at  $p = 0$ )

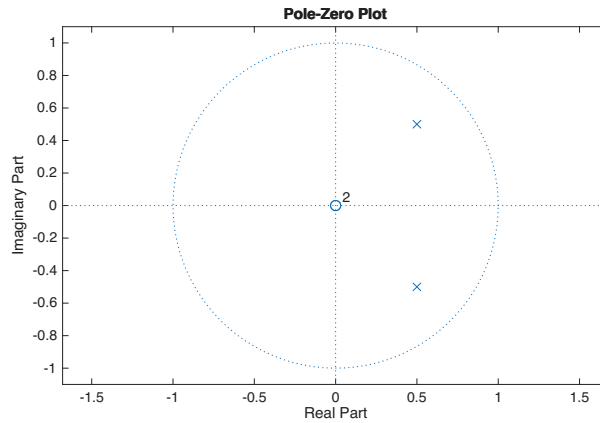
$$\begin{aligned}
 H'(z) &= \frac{z^2}{(z - p_1)(z - p_2)} \\
 H'(z) &= \frac{z^2}{[z - (0.5 + 0.5i)][z - (0.5 - 0.5i)]} \\
 &= \frac{z^2}{z^2 - 1z + 0.5} \\
 \frac{Y(z)}{X(z)} &= \frac{1}{1 - 1z^{-1} + 0.5z^{-2}} \\
 \Rightarrow X(z) &= Y(z) [1 - 1z^{-1} + 0.5z^{-2}]
 \end{aligned}$$

Applying inverse Z-transform gives the required system:

$$x(n) = y(n) - y(n - 1) + 0.5y(n - 2)$$

Hence the implementation:

```
num = 1; den = [1, -1, 0.5];
zplane(num, den);
```



new system function

### 3 Gibbs oscillation

Simulate Gibbs oscillation on a given signal  $s(t)$ .

#### Background

The Fourier series of a complex-valued periodic function  $s(t)$ , integrable over the interval  $[a, b]$  on the real line, is defined as:

$$s(t) = \sum_{n=-\infty}^{\infty} c_n e^{i2\pi \frac{n}{T} t}$$

Where  $T = b - a$  is the period of function. Fourier coefficients  $c_n$  are:

$$c_n = \frac{1}{T} \int_a^b s(t) e^{-i2\pi \frac{n}{T} t} dt$$

## Program

```
clearvars; close all;

a = -1;
b = 1;
t = linspace(a, b, 1000);
T = b - a;

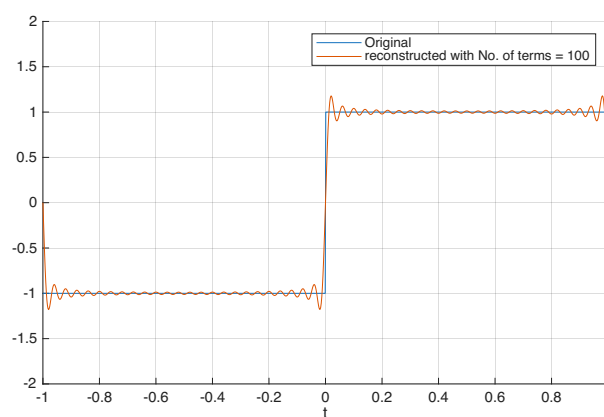
% input signal
s = @(t) sign(t);

% Reconstruct signal with N harmonics on each side
N = 50;
f_series = zeros(size(t));

% coefficients
c = @(k) 1 / T * integral(@(t) s(t) .* exp(-2j * pi * t * k / T), a, b);

% Fourier series summation
for k = -N:N
    f_series = f_series + c(k) * exp(2j * pi * k * t / T);
end

% Plot the actual and reconstructed signal
hold on;
plot(t, s(t));
plot(t, real(f_series));
xlabel('t');
legend(["Original", "reconstructed with No. of terms = " + N*2]);
grid on;
ylim([-2, 2]);
```



Notice the wide oscillation near discontinuities