

Homework 2

github repository link - <https://github.com/Volodimirich/DSP/tree/master/HomeWork2>

Task №1

Design a low pass FIR filter with parameters: passband $F_{pass}=5$ MHz; stopband $F_{stop}=6$ MHz; attenuation at least 60dB in the stopband (out-of-band attenuation). Let the sampling frequency be $F_s=50$ MHz. Determine the design with the lowest computational complexity.

Provide code.

Solution:

Nyquist frequency is equal to $F_N = 50 \text{ MHz} / 2 = 25 \text{ MHz}$.

Then normalized passband and stop frequency are equal:

$$F_{pass} = \frac{5 \cdot 10^6}{25 \cdot 10^6} = 0.2, F_{stop} = \frac{6 \cdot 10^6}{25 \cdot 10^6} = 0.24$$

The function that checks that the filter conditions are fulfilled is as follows:

```
# This function check filter correctness
def filter_analyse(b, graph = False):
    result = True
    w, h = signal.freqz(b)
    w_n = w/max(w)
    h_n = 20 * np.log10 (abs(h))

    if graph:
        plot_graph(w_n, h_n)

    max_passband = h_n[0]
    min_passband = h_n[0]

    for x, y in zip(w_n, h_n):
        #If x < Fpass then amplitude must be greater than -3dB
        if x < 0.2:
            if y < -3:
                return False
            min_passband = min_passband if y < min_passband else y
            max_passband = max_passband if y > max_passband else y
        #Ripple not greater than 1db
        if (max_passband - min_passband) > 1:
            return False
```

```

        #If x > Fstop then amplitude must be less than attenuation
        if x > 0.24 and y > -60:
            return False

    return True

```

Then the function performing the brute-force has the following form:

```

def filter_brute_force():
    res_fin = 1000
    data = (0, 0)
    windows = ['boxcar', 'triang', 'blackman', 'hamming', 'hann',
               'bartlett', 'flattop', 'parzen', 'bohman',
               'blackmanharris', 'nuttall', 'barthann', 'cosine',
               'exponential', 'tukey', 'taylor']
    for shift in range(100):
        for window_name in windows:
            res = 100
            while True:
                b = signal.firwin(res, 0.2 + 0.0001 * shift,
                                   pass_zero = 'lowpass', window = window_name)
                if filter_analyse(b):
                    if res < res_fin:
                        data = (res, window_name, 0.2 + 0.0001 * shift)
                        res_fin = res
                        break
                else:
                    res += 1
            if res > 500:
                break

    return data

```

In this function we went through the possible windows from the existing ones, and the shift where -6dB is located. By going through the possible values and selecting the best parameter by the order parameter we get a filter with the following parameters:

Order - 149, Window type - blackman, Shift - 0.2063

In this case our plot has the next form:

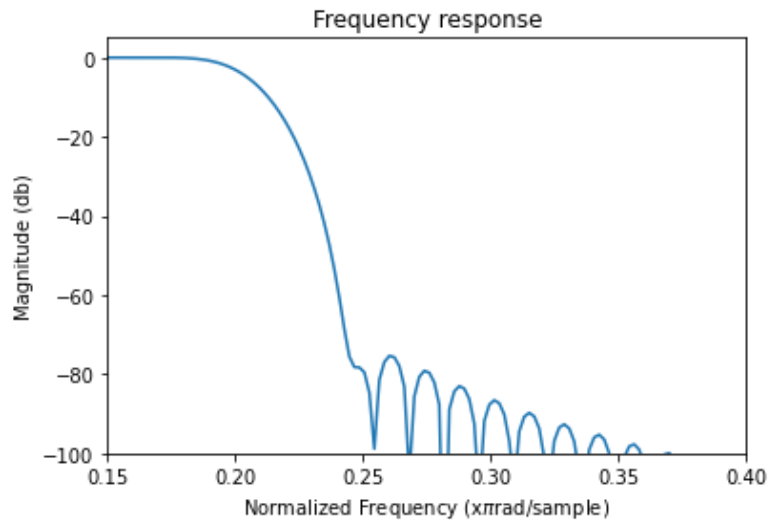


Figure 1: Fir1 filter with best parameters

Also let's analyse firls filter:

```
order_min = 1000
for i in range(1, 200):
    for order in range(101, 201, 2):
        res = filter_analyse(signal.firls(order, [0, 0.001 * i, 0.24, 1],
        [1, 1, 0, 0]))
        if res:
            if order_min > order:
                data = (order, 0.001 * i)
                order_min = order
            break
```

In this case we are trying to find a minimum order value with a changing shift from 0 where magnitude is still 1.

In this case we have:

Order - 123, Shift - 0.173

In this case our plot has the next form:

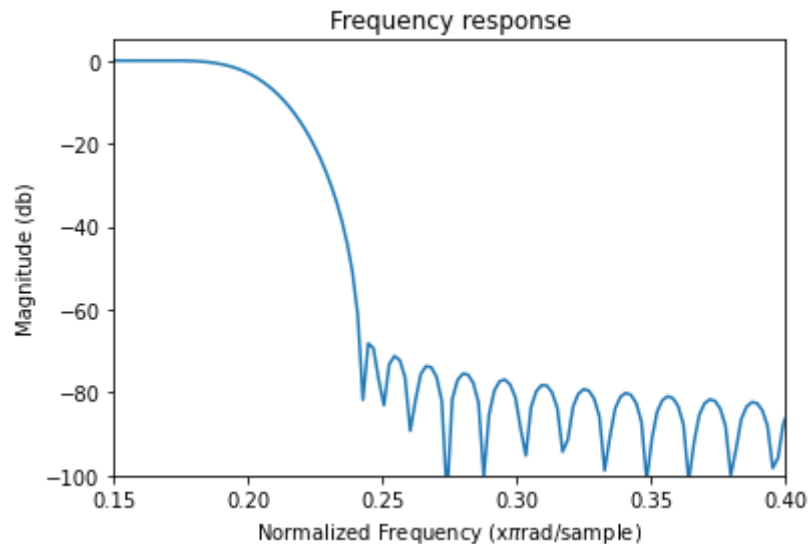


Figure 2: Firls filter with the best parameters

Fir2 gives us the same result as the firls. Then the best order, which we were able to achieve - 123.

Task №2

Using the impulse invariance method for analog to digital filter conversion, calculate the Chebyshev lowpass digital filter with parameters: passband 20MHz; passband ripple 0.2dB; stopband (out-of-band) attenuation 60dB; sampling frequency $F_s = 100\text{MHz}$.

- Plot the impulse response for both analog and digital systems.
- Plot the magnitude response for analog and digital systems in the frequency domain.

Provide code.

Solution:

In this task we don't have stopband frequency. Let's set this value as $\text{stop_freq} = 42\text{MHz}$. Just now we need filter order:

```
sample_freq = 100 * 10^6;
nyquist_freq = sample_freq / 2;
ripple = 0.2;
stopband = 60;
passband = 20 * 10^6;
passband_normalized = passband / nyquist_freq;
stop_freq = 42 * 10^6;
stop_normalized = stop_freq / nyquist_freq;
order = cheb1ord(passband_normalized, stop_normalized, ripple,
stopband);
```

As a result of this code we can understand that filter order = 4. After a little lab code modernization we can plot impulse responses for analog and digital systems.

```
[bz,az] =impinvar(b, a, sample_freq);
[r,p] = residue(b,a);
t = linspace(0, 40/sample_freq, 1000);
h = real(r.*exp(p.*t)/sample_freq);
#' useless comment, little fix because of ' in operation
figure(1);
plot(t, h);
hold;
impz(bz, az, [], sample_freq);
legend('Analog', 'Digital');
hold off;
```

The plot is as follows:

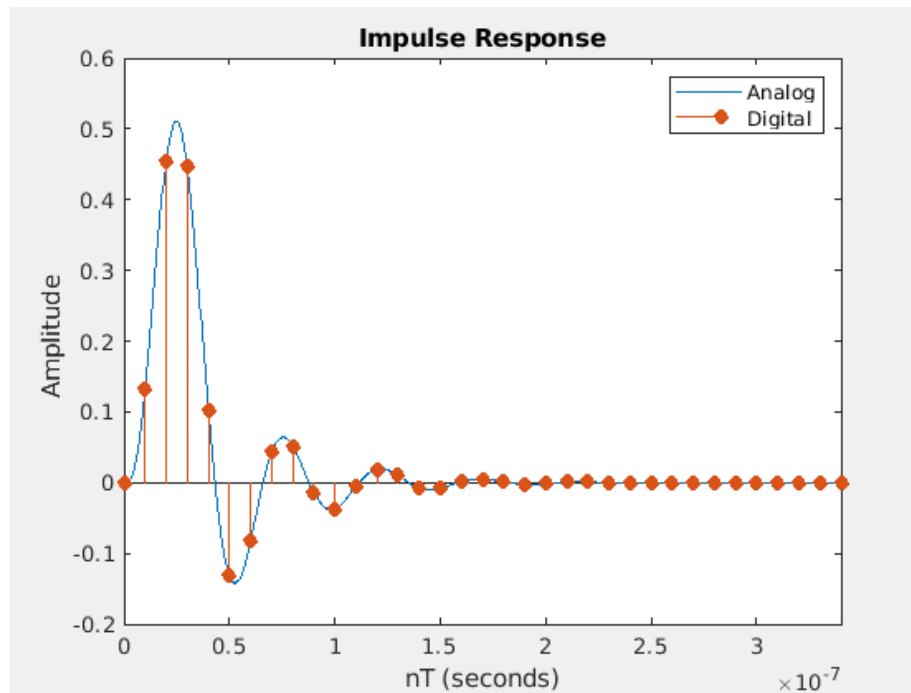


Figure 1: Impulse response

The code for point b was also obtained by slightly modifying the lab work code.

```
figure(2);
[H,W] = freqz(bz, az);
F = linspace(0,sample_freq/2, length(W));
semilogy(F, abs(H))
hold on;
[H] = freqs(b,a,W*sample_freq);
semilogy(F, abs(H));
legend("Digital", "Analog");
```

```
hold off;
```

The plot is as follows:

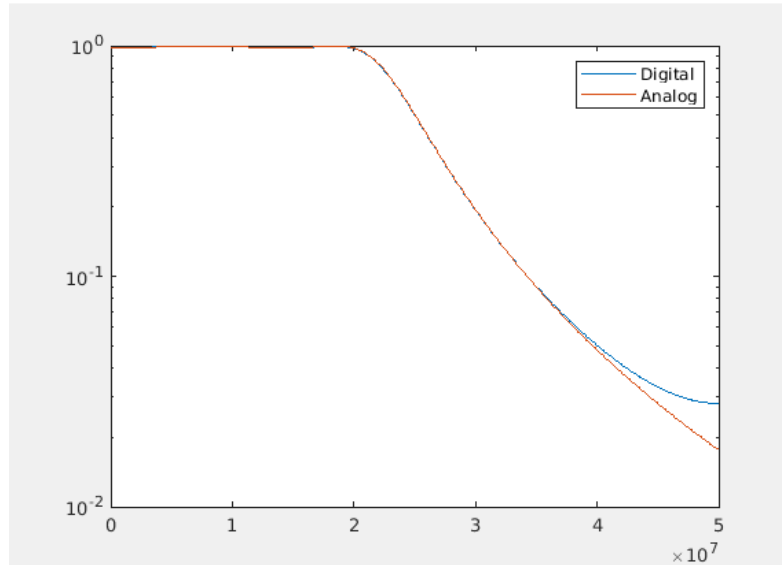


Figure 2: Frequency response

Task №3

Implement a digital prototype of the analog filter with the transfer function

$$H(s) = \frac{s + 2.5}{s^2 + 2.5s + 4}$$

using the Bilinear Transformation. The sample clock frequency is $F_s = 20\text{Hz}$.

a) Determine the Linear Difference Equation of the digital filter.

b) Plot impulse and frequency responses for digital and analog filters.

Provide code.

Solution:

$$\frac{s + 2.5}{s^2 + 2.5s + 4} = \frac{2s + 5}{2s^2 + 5s + 8} = \frac{A}{B} \quad \text{Bilinear transformation: } s = \frac{2(z - 1)}{T(z + 1)}$$

On the first step, let's calculate B part:

$$\begin{aligned} 2s^2 + 5s + 8 &= 2 \frac{4(z - 1)^2}{T^2(z + 1)^2} + 5 \frac{2(z - 1)}{T(z + 1)} + 8 = \frac{8(z - 1)^2 + 10(z - 1)T(z + 1) + 8T^2(z + 1)^2}{T^2(z + 1)^2} = \\ &= \frac{8z^2 - 16z + 8 + 10z^2T - 10T + 8T^2z^2 + 16T^2z + 8T^2}{T^2(z + 1)^2} = \frac{(8 + 10T + 8T^2)z^2 + (16T^2 - 16)z + 8 - 10T + 8T^2}{T^2(z + 1)^2} \end{aligned}$$

After that calculate A part:

$$2s+5 = 2 \frac{2(z-1)}{T(z+1)} + 5 = \frac{4(z-1) + 5T(z+1)}{T(z+1)} = \frac{4(z-1)T(z+1) + 5T^2(z+1)^2}{T^2(z+1)^2} = \frac{4Tz^2 - 4T + 5T^2z^2 + 10T^2z + 5T^2}{T^2(z+1)^2} = \frac{(4T + 5T^2)z^2 + 10T^2z + 5T^2 - 4T}{T^2(z+1)^2}$$

Then we have:

$$\frac{A}{B} = \frac{(4T + 5T^2)z^2 + 10T^2z + 5T^2 - 4T}{(8 + 10T + 8T^2)z^2 + (16T^2 - 16)z + 8 - 10T + 8T^2}$$

According to $T = \frac{1}{F} = \frac{1}{20}$ we have:

$$\frac{A}{B} = \frac{0.2125z^2 + 0.025z - 0.1875}{8.52z^2 - 15.96z + 7.52} = \frac{0.0249 + 0.00293z^{-1} - 0.022z^{-2}}{1 - 1.8732z^{-1} + 0.8827z^{-2}} = \frac{Y(z)}{X(z)}$$

$$0.0249 X(z) + 0.00293 X(z) z^{-1} - 0.022 X(z) z^{-2} = Y(z) - 1.8732 Y(z) z^{-1} + 0.8827 Y(z) z^{-2}$$

Then linear difference equation has the following form

$$y[n] = -1.8732 y[n-1] + 0.8827 y[n-2] - 0.0249 x[n] + 0.00293 x[n-1] - 0.022 x[n-2]$$

```
clock_freq = 20;
nyquist_freq = clock_freq / 2;
b = [0, 1, 2.5];
a = [1, 2.5, 4];
bz = [0.0249, 0.0029, -0.022];
az = [1, -1.8732, 0.8826];

[freq_resp, ang_resp] = freqz(bz,az);
x_vals = linspace(0, nyquist_freq, length(freq_resp));

figure(1);
semilogy(x_vals,abs(freq_resp))
hold on;
[H] = freqs(b,a, ang_resp*clock_freq);
semilogy(x_vals, abs(H))

legend("Digital", "Analog")
hold off;
fvtool(bz, az)
```

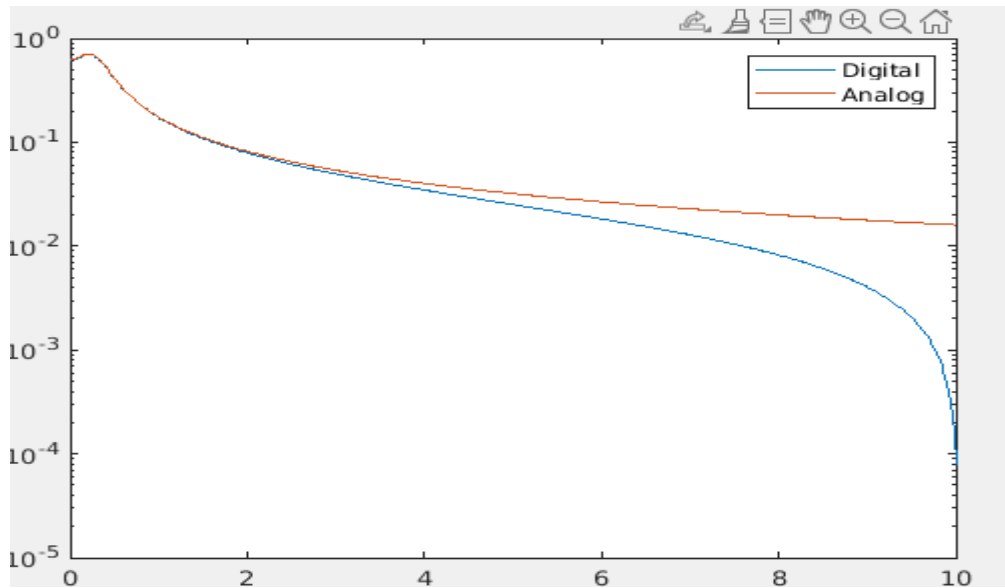


Figure 1: Frequency response

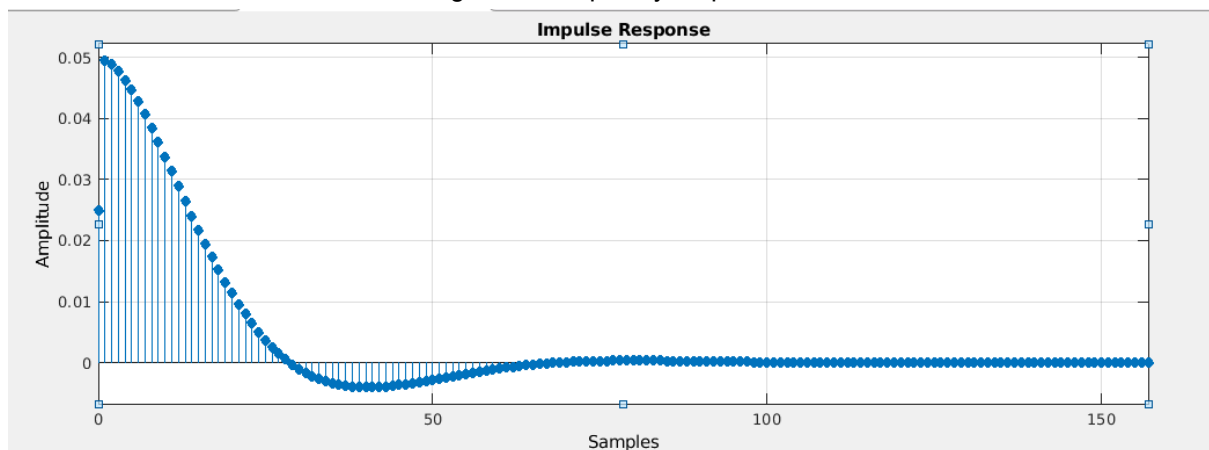


Figure2: Impulse response

Task №4

A filter has the transfer function

$$H(z) = 3 + 4z^{-1} + 6z^{-2} + 8z^{-3}$$

Determine the impulse response of the filter with the modified frequency response

$$F(w) = H(w - \frac{\pi}{4})$$

Solution:

Let's use definition $z = re^{iw}$

$$H(w - \frac{\pi}{4}) = 3 + 4re^{-(w-\frac{\pi}{4})i} + 6re^{-2(w-\frac{\pi}{4})i} + 8re^{-3(w-\frac{\pi}{4})i} = 3 + 4re^{\frac{\pi i}{4}}e^{-wi} + 6re^{\frac{\pi i}{2}}e^{-2wi} + 8re^{\frac{3\pi i}{4}}e^{-3wi}$$

After that we need to calculate IDTFT:

$$x[w] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X[w] e^{wni} dw$$

$$h[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} (3 + 4re^{\frac{\pi i}{4}} e^{-wi} + 6re^{\frac{\pi i}{2}} e^{-2wi} + 8re^{\frac{3\pi i}{4}} e^{-3wi}) e^{wni} dw = \frac{3}{2\pi} \int_{-\pi}^{\pi} e^{nwi} dw +$$

$$\frac{2r}{\pi} \int_{-\pi}^{\pi} e^{(n-1)wi} dw + \frac{3r}{\pi} \int_{-\pi}^{\pi} e^{(n-2)wi} dw + \frac{4r}{\pi} \int_{-\pi}^{\pi} e^{(n-3)wi} dw$$

Let's calculate part of our equation:

$$\int_{-\pi}^{\pi} e^{nwi} dw$$

$$\int_{-\pi}^{\pi} e^{wki} dw = \frac{e^{wki}}{ki} \Big|_{-\pi}^{\pi} = \frac{\cos wk + i \sin wk}{ki} \Big|_{-\pi}^{\pi} = \frac{\cos \pi k + i \sin \pi k}{ki} - \frac{\cos(-\pi)k + i \sin(-\pi)k}{ki} =$$

$$\frac{\cos \pi k + i \sin \pi k}{ki} - \frac{\cos \pi k - i \sin \pi k}{ki} = \frac{2i \sin \pi k}{ki} = \frac{2 \sin \pi k}{k}$$

By replacing k with the corresponding expression with n we obtain:

$$h[n] = \frac{3 \sin \pi n}{\pi n} + \frac{4r \sin \pi(n-1)}{\pi(n-1)} e^{\frac{\pi i}{4}} + \frac{6r \sin \pi(n-2)}{\pi(n-2)} e^{\frac{\pi i}{2}} + \frac{8r \sin \pi(n-3)}{\pi(n-3)} e^{\frac{3\pi i}{4}} =$$

$$3 \cdot \text{sinc}(n) + 4r \cdot \text{sinc}(n-1) e^{\frac{\pi i}{4}} + 6r \cdot \text{sinc}(n-2) e^{\frac{\pi i}{2}} + 8r \cdot \text{sinc}(n-3) e^{\frac{3\pi i}{4}}$$

Then:

$$h[n] = 3\delta[n] + 4r\delta[n-1]e^{\frac{\pi i}{4}} + 6r\delta[n-2]e^{\frac{\pi i}{2}} + 8r\delta[n-3]e^{\frac{3\pi i}{4}}$$

Task №5

For a linear system with the transfer function

$$H(z) = \frac{1z + 2}{3z^3 + 4z^2 + 5z + 6}$$

- Calculate the difference equation relating the input $x[n]$ to the output $y[n]$
- Design block diagram realizations (Direct-Form 1 and Direct-Form 2)
- Plot impulse and frequency responses

Provide code.

Solution:

$$H(z) = \frac{1z + 2}{3z^3 + 4z^2 + 5z + 6} = \frac{1z^{-2} + 2z^{-3}}{3 + 4z^{-1} + 5z^{-2} + 6z^{-3}} = \frac{\frac{1}{3}z^{-2} + \frac{2}{3}z^{-3}}{1 + \frac{4}{3}z^{-1} + \frac{5}{3}z^{-2} + 2z^{-3}} = \frac{Y(z)}{X(z)}$$

$$Y(z) + \frac{4}{3}Y(z)z^{-1} + \frac{5}{3}Y(z)z^{-2} + 2Y(z)z^{-3} = \frac{1}{3}X(z)z^{-2} + \frac{2}{3}X(z)z^{-3}$$

$$Y(z) = -\frac{4}{3}Y(z)z^{-1} - \frac{5}{3}Y(z)z^{-2} - 2Y(z)z^{-3} + \frac{1}{3}X(z)z^{-2} + \frac{2}{3}X(z)z^{-3}$$

$$y[n] = -\frac{4}{3}y[n-1] - \frac{5}{3}y[n-2] - 2y[n-3] + \frac{1}{3}x[n-2] + \frac{2}{3}x[n-3]$$

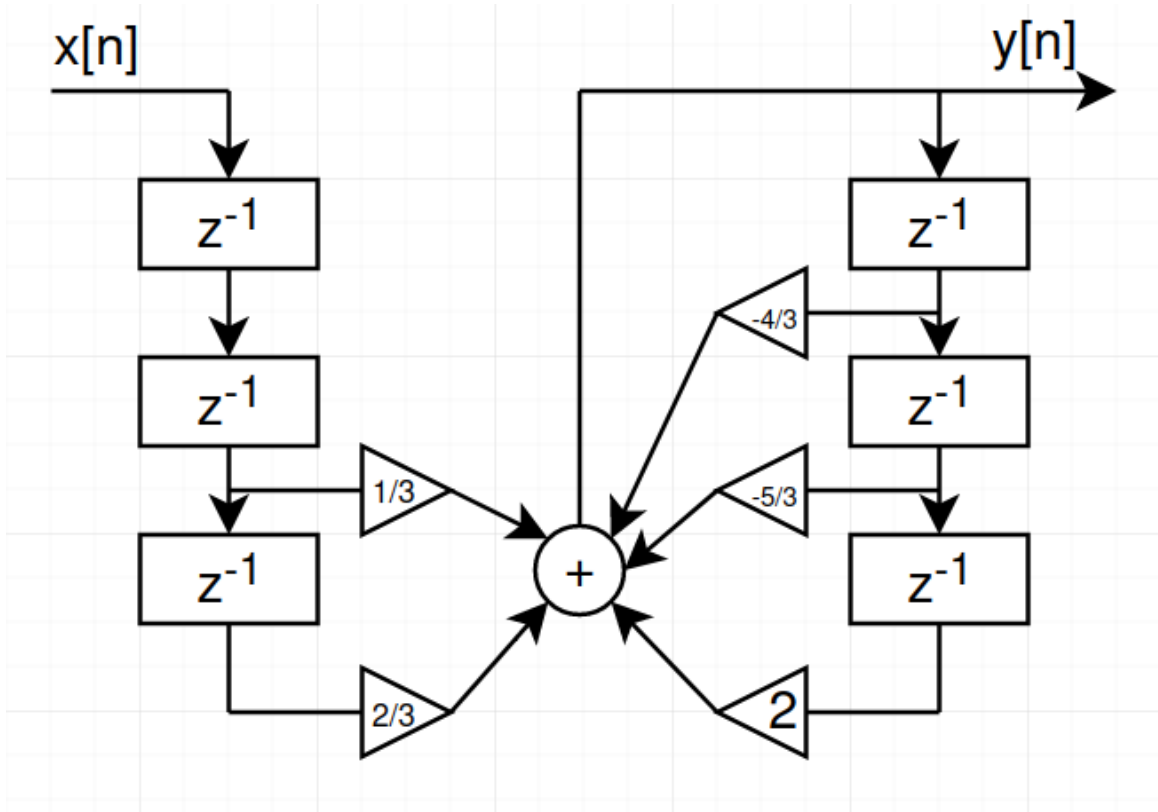


Figure 1: Direct-Form 1

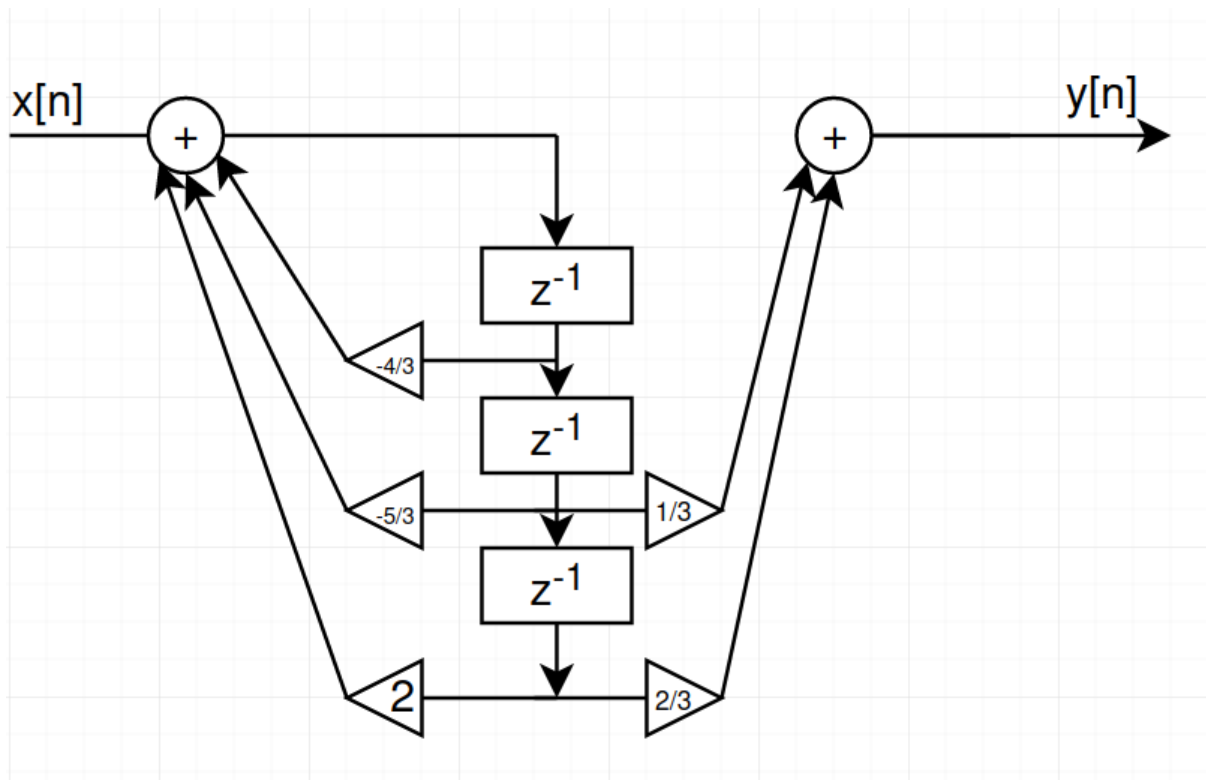


Figure 2: Direct-Form 2

According to $H(z) = \frac{\frac{1}{3}z^{-2} + \frac{2}{3}z^{-3}}{1 + \frac{4}{3}z^{-1} + \frac{5}{3}z^{-2} + 2z^{-3}}$ we can calculate impulse and frequency response with using next code:

```
fvtool([0 0 1/3 2/3],[1 4/3 5/3 2])
```

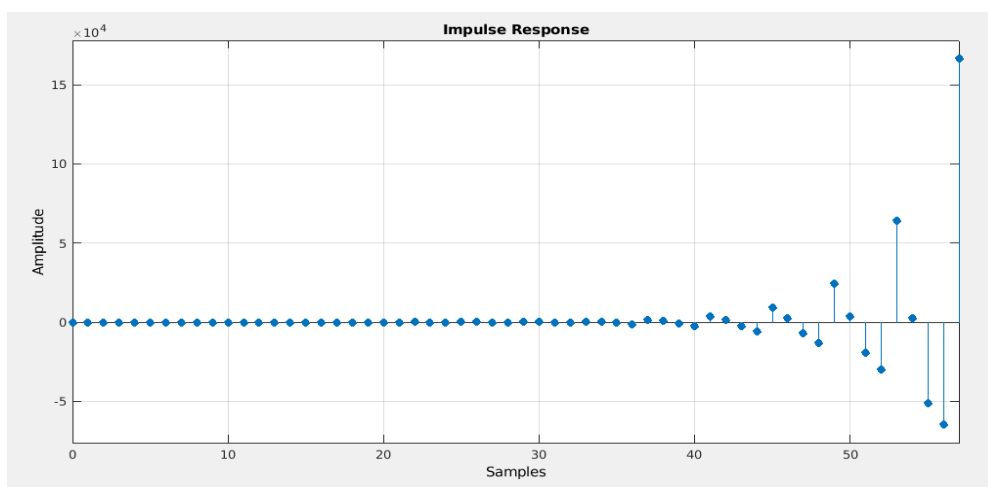


Figure 3: Impulse response

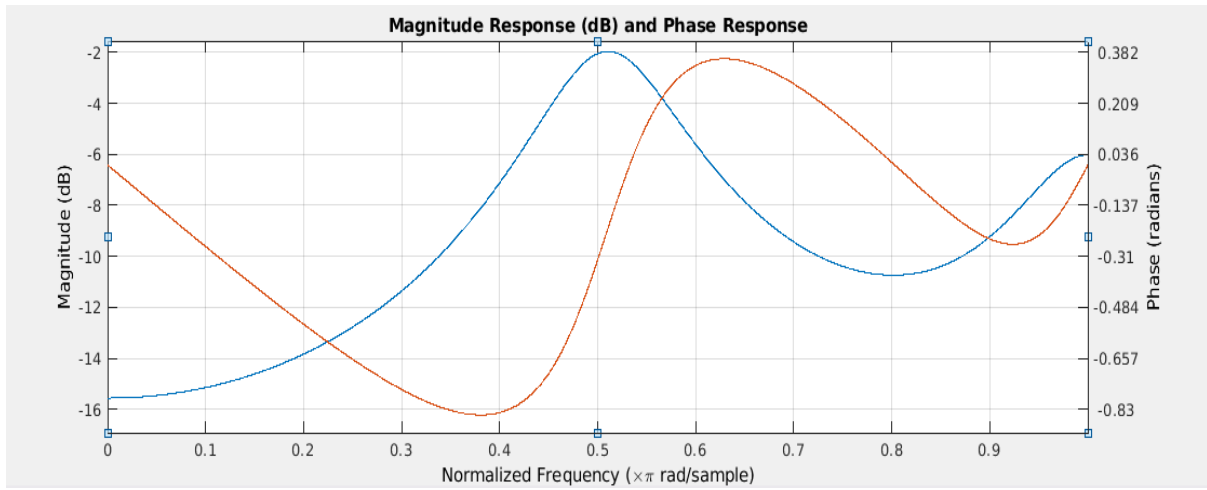


Figure 4: Frequency response