



# Digital Communications modulation project (project #3)

# **Submitted by:**

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# **Constellation of different modulation techniques:**

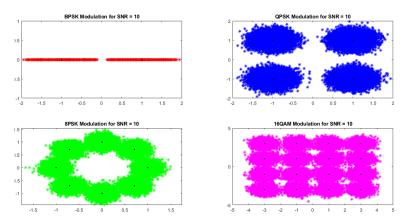


Figure 1: Constellation of the BPSK, QPSK, 8PSK, 16QAM,

# Bit error rate (BER) of different modulation techniques:

## I. Binary phase shift keying (BPSK)

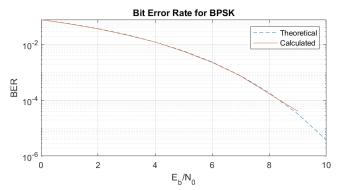


Figure 2: Bit error rate (BER) of BPSK

The theoretical BER of the BPSK is computed given by:

$$BER_{theoritical} = 0.5 \; erfc\left(\sqrt{\frac{E_b}{N_0}}\right)$$
 , where  $E_b = 1$ .

## II. Quadrature phase shift keying (QPSK)

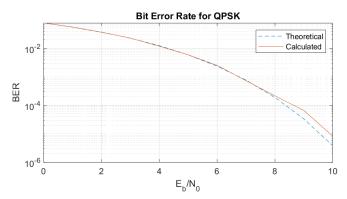


Figure 3: Bit error rate (BER) of QPSK

The theoretical BER of the QPSK is computed given by:

$$BER_{theoritical} = 0.5 \ erfc \left( \sqrt{\frac{E_b}{N_0}} \right)$$
, where  $E_b = 1$ .

## III. Eight-ary phase shift keying (8PSK)

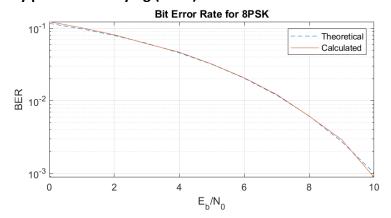


Figure 4: Bit error rate (BER) of 8PSK

The theoretical BER of the QPSK is computed given by:

$$BER_{theoritical} = \frac{1}{3}erfc\left(\sqrt{\frac{E}{N_0}}\sin\left(\frac{\pi}{M}\right)\right)$$
, where  $E = \log_2(M)E_b$  and  $E_b = \frac{1}{3}$ 

So, the law is reduced to:

$$BER_{theoritical} = \frac{1}{3} erfc \left( \sqrt{\frac{1}{N_0}} * \sin(\frac{\pi}{8}) \right)$$

#### IV. 16-Quadrature amplitude modulation (16QAM)

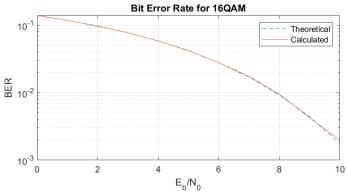


Figure 4: Bit error rate (BER) of 16QAM

The theoretical BER of the 16QAM is computed given by:

$$BER_{theoritical} = \frac{1.5}{4} erfc\left(\sqrt{\frac{E_b}{N_0}}\right)$$
, where  $E_b = 2.5$ 

#### V. BPSK, QPSK, 8PSK and 16QAM modulations on the same graph.

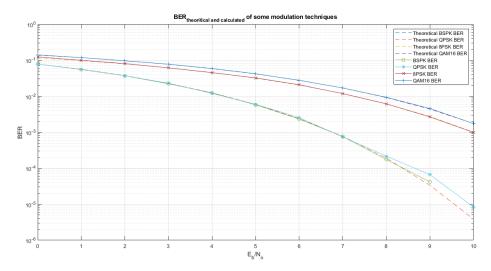


Figure 5: (BER) of BPSK, QPSK, 8PSK, and 16QAM

#### Comment:

- As it is noticed from the previous graph that the 16QAM modulation technique obtains the worst BER of all other simulated modulation techniques. Therefore, as the number of bits per symbol increases, it leads to deterioration in the BER. It is concluded as the 4 bits per symbol (16QAM) is the worst after it the 3 bits per symbol (8PSK).
- The BER in the BPSK and QPSK is quite similar in response.
- By increasing the signal to noise ratio (SNR) which is computed by:

$$SNR = \frac{E_b}{N_0}$$

It leads to a better performance. In other words, less BER.

#### VI. BER of grey encoded QPSK and another encoding method of QPSK

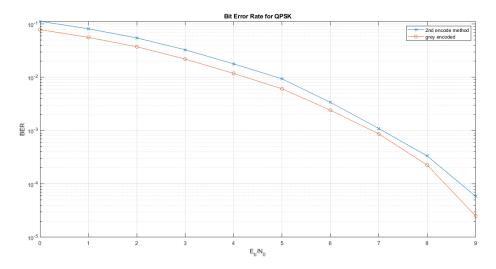


Figure 6: (BER) of grey QPSK and other encoding method

#### Comment:

It was concluded from the graph that the grey encoding method has obtained a better BER than the other encoding method as when the error happens it may change one bit yet the other way of encoding leads to a higher probability of error.

### **Binary Frequency Shift Keying (BFSK)**

#### **System Basis function**

$$\phi_i(t) = \sqrt{\frac{2}{T_b}} \cos{(2\pi f_i t)}$$
 where  $f_i = \frac{n_c + i}{T_b}$ ,  $i = 1,2$ 

Where  $T_{\rm b}$  represents the bit duration or the time taken to transmit a single bit. It is a constant value that defines the duration of each bit in the BFSK signal. And  $n_c$  represents the center frequency offset. It is a constant value that determines the separation between the two carrier frequencies used in BFSK modulation

#### Base band equivalent signal

$$s(t) = \sqrt{\frac{2E_b}{T_b}}\cos(2\pi f_1 t) \text{ or } \sqrt{\frac{2E_b}{T_b}}\cos(2\pi f_2 t)$$

You can suppose that the carrier frequency is any frequency that is close to the frequencies of interest. You can take it as f1, f2, or (f1+f2)/2. Assuming fc = f1, In this case

$$s_1(t) = \sqrt{\frac{2E_b}{T_b}}\cos\left(2\pi f_c t\right)$$

$$s_2(t) = \sqrt{\frac{2E_b}{T_b}} \cos(2\pi (f_c + \Delta f)t)$$

$$= \sqrt{\frac{2E_b}{T_b}} \cos(2\pi f_c t) \cos(2\pi \Delta f t) - \sqrt{\frac{2E_b}{T_b}} \sin(2\pi f_c t) \sin(2\pi \Delta f t)$$

So base band equivalent signal will be

$$s_{1BB}(t) = \sqrt{\frac{2E_b}{T_b}}$$

$$s_{2BB}(t) = \sqrt{\frac{2E_b}{T_b}}\cos(2\pi\Delta f t) + j\sqrt{\frac{2E_b}{T_b}}\sin(2\pi\Delta f t) = \sqrt{\frac{2E_b}{T_b}}e^{(j2\pi\Delta f t)}$$

# **Constellation of BFSK Modulation:**

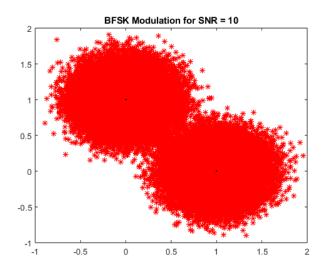


Figure 7 Constellation of BFSK Modulation

# **Bit error rate (BER) of BFSK Modulation:**

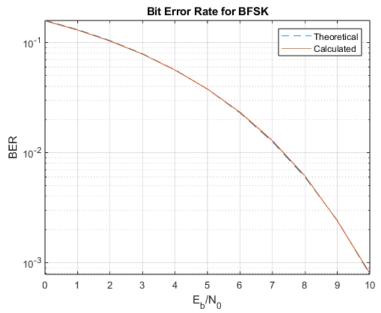


Figure 8 Bit error rate (BER) of BFSK Modulation

The theoretical BER of the BPSK is computed given by:

$$BER_{theoritical} = 0.5 \ erfc\left(\sqrt{\frac{E_b}{2N_0}}\right)$$
, where  $E_b = 1$ 

# Power spectral density (PSD) of BFSK Modulation:

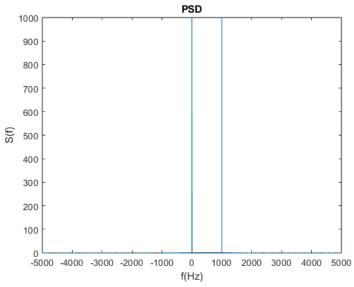


Figure 9 Power spectral density (PSD) of BFSK Modulation

two deltas are observed at the DC frequency (representing the zero bit, denoted as or  $s_{1BB}(t)$ )) and at 1000Hz (representing the one bit, denoted as or  $s_{2BB}(t)$ ) These delta functions satisfy the following properties:  $s_{1BB}(t) = \sqrt{\frac{2E_b}{T_b}}$  is constant (DC), while  $s_{2BB}(t) = \sqrt{\frac{2E_b}{T_b}} e^{(j2\pi\Delta ft)}$ , which is similar to  $s_{1BB}$  but shifted in frequency by  $\Delta f$ . In this case,  $\Delta f$  is equal to  $1/T_b$ , where Tb is 1e-3 seconds, resulting in  $\Delta f = 1000~Hz$ .

#### **MATLAB** code

Please be advised that the version of MATLAB used in this project is R2021a (9.10.0.1602886). Furthermore, the code is accessible on GitHub via the following <u>link</u> (GitHub repository will be made public by the project deadline day).

Please note that the first code snippet provided does not include the implementation of the BFSK modulation technique. The BFSK modulation code is present in a separate code snippet, which is the second one.

```
* FILE DESCRIPTION
* File: source code.m
^{\star} Digital communication project-assignment
* Description: Simulation of different modulation techniques (BPSK, QPSK, 8PSK, 16QAM)
* Author:Omar Muhammad Mustafa, Omar Muhammad Tolba
* Date: 6th May 2023
%% clear all WorkSpace Variables and Command Window
clc;
clear ;
close all;
%% initialization
Bits Number = 1.2e5;
                                                       % number of bits to be generated
SNR_{max} value = 10;
%% Generating the data bits
Bit_Stream = randi([0 1],1,Bits_Number);
%% BPSK Modulation
%initializations needed for Simulation of BPSK
BPSK Eb = ((1 +1)/2)/1;
                                                    %Calculating Eb of BPSK modulation
BPSK BER Theo = zeros(1, SNR max value);
                                                    %Vector to store the theortical BER for
different SNR channel
BPSK BER = zeros(1,SNR_max_value);
                                                    %Vector to store the calculated BER for
different SNR channel
%Mapper
BPSK symbolStream = 2 * Bit Stream - 1;
                                                    %The sent symbol is either 1 or -1
%Channel (AWGN)
BPSK channelNoise = zeros(length(Bit Stream), SNR max value); %Matrix to store the noise in
each channel
BPSK No = zeros(1,SNR_max_value + 1);
%Calculating No of the channel for different SNR
for SNR dB = 1 : (SNR max value + 1)
    %Generating different noise vector for different channels with different SNR
   noise = randn(1,length(Bit_Stream));
   BPSK No(SNR dB) = BPSK Eb/10.^((SNR dB-1)/10);
   BPSK channelNoise(:, SNR dB) = noise * sqrt (BPSK No(SNR dB)/2); %scale the noise.
%Demapper
BPSK demappedSymbol = zeros(1,length(Bit Stream)); %Vector to store the demapped stream
for SNR dB = 1: (SNR max value + 1)
   BPSK DataRx = BPSK symbolStream + BPSK channelNoise(:,SNR dB)'; %Recieved Data
    %Demapping the recieved symbol stream for each channel
    for counter = 1 : Bits Number
        if (BPSK DataRx(1,counter) > 0)
           BPSK demappedSymbol(1,counter) = 1;
           BPSK demappedSymbol(1,counter) = 0;
       end
    [N BER BPSK,BPSK BER(SNR dB)] = symerr(BPSK demappedSymbol,Bit Stream); %Calculated BER
    BPSK BER Theo(SNR dB) = 0.5 * erfc(sqrt(1/BPSK No(SNR dB))); %Theoritical BER
%Plotting constillation of BPSK
figure(1)
subplot(2,2,1,'LineWidth',3)
plot(real(BPSK DataRx),imag(BPSK DataRx),'r*',real(1),imag(0),'k.',real(-1),imag(0),'k.');
title('BPSK Modulation for SNR =
```

```
%Plotting BER
figure(2)
subplot(2,2,1,'LineWidth',3)
EbN0 dB = 1:1:(SNR max value + 1);
%Plotting theoritical BER
semilogy((EbN0 dB - 1), BPSK BER Theo, '--')
%Plotting calculated BER
hold on
semilogy((EbN0 dB - 1),BPSK BER,'-')
grid on
ylabel('BER')
xlabel('E b/N 0')
title('Bit Error Rate for BPSK')
legend('Theoretical','Calculated');
%initializations needed for Simulation of BPSK
QPSK Eb = ((4 * 2) / 4) / 2;
QPSK BER Theo = zeros(1, SNR max value);
                                                   %Vector to store the theortical BER for
different SNR channel
QPSK BER = zeros(1,SNR max value);
                                                   %Vector to store the calculated BER for
different SNR channel
QPSK BER encode2 = zeros(1,SNR max value);
                                                           %Vector to store the calculated
BER for different SNR channel
%Mapper
%Grouping the binary data into groups of 2 bits
QPSK reshaped binary data = reshape(Bit Stream, 2, []) ';
%Mapping the input data to QPSK symbols grey encoded
% 0 0 -> -1-1i
  0 1 -> -1+1i
% 1 0 -> 1-1i
  1 1 -> 1+1i
QPSK map = [-1-1i, -1+1i, 1-1i, 1+1i];
% 0 0 -> -1-1i
  0 1 -> -1+1i
% 1 0 -> 1+1i
  1 1 -> 1-1i
QPSK_map_encode2 = [-1-1i, -1+1i,1+1i, 1-1i];
%The bi2de function is used to convert the binary data to decimal values,
%which are then used as indices to look up the corresponding QPSK symbol in the mapping
QPSK data = QPSK map(bi2de(QPSK reshaped binary data, 'left-msb')+1);
QPSK data encode2 = QPSK map encode2(bi2de(QPSK reshaped binary data, 'left-msb')+1);
%Channel (AWGN)
QPSK channelNoise real = zeros(length(Bit Stream)/2,SNR max value); %Matrix to store the
real noise in each channel
QPSK_channelNoise_complex = zeros(length(Bit_Stream)/2,SNR_max_value); %Matrix to store the
complex noise in each channel
QPSK No = zeros(1, SNR max value + 1);
%Calculating No of the channel for different SNR
for SNR dB = 1: (SNR max value + 1)
    *Generating different noise vector for different channels with different SNR
    QPSK No(SNR dB) = QPSK Eb/10.^((SNR dB-1)/10);
   noise_I = randn(1,length(Bit_Stream)/2); %[todo] get it out of the for loop
   QPSK channelNoise real(:,SNR dB) = noise I * sqrt (QPSK No(SNR dB)/2); %scale the
noise.
    noise_Q = randn(1,length(Bit_Stream)/2); %[todo] get it out of the for loop
   QPSK ChannelNoise complex(:,SNR dB) = noise Q * sqrt (QPSK No(SNR dB)/2); %scale the
noise.
end
%Demapper
QPSK demappedBits = zeros(1,length(Bit Stream)); %Vector to store the demapped stream
QPSK_demappedBits_encode2 = zeros(1,length(Bit_Stream)); %Vector to store the demapped
stream
```

```
QPSK recieved Bits = zeros((Bits Number/2),2);
QPSK recieved Bits encode2 = zeros((Bits Number/2),2);
for SNR dB = 1: (SNR max value + 1)
    %Recieved Data
    QPSK DataRx = (real(QPSK data) + QPSK channelNoise real(:,SNR dB)') ... %[todo] get this
out of the loop
    + 1i *(imag(QPSK data) + QPSK channelNoise complex(:,SNR dB)');
    QPSK DataRx encode2 = (real(QPSK data encode2)+ QPSK channelNoise real(:, SNR dB)')
...%[todo] get this out of the loop
    + 1i *(imag(QPSK data encode2) + QPSK_channelNoise_complex(:,SNR_dB)');
    %Demapping the recieved symbol stream for each channel
    %Grey encoded
    for counter = 1 : Bits Number/2
        if(real(QPSK DataRx(counter)) > 0)
            QPSK recieved Bits(counter,1) = 1;
            QPSK recieved Bits(counter,1) = 0;
        end
        if(imag(QPSK DataRx(counter)) > 0)
            QPSK recieved Bits(counter, 2) = 1;
            QPSK_recieved_Bits(counter,2) = 0;
        end
    QPSK demappedBits = reshape(QPSK recieved Bits',1,[]);
    %Demapping the recieved symbol stream for each channel
    %second encoding method
    for counter = 1 : Bits Number/2
        if (imag(QPSK DataRx encode2(counter)) > 0)
            if(real(QPSK_DataRx_encode2(counter)) > 0)
                QPSK recieved Bits encode2(counter,1) = 1;
                QPSK recieved Bits encode2(counter,2) = 0;
            else
                QPSK_recieved_Bits_encode2(counter,1) = 0;
                QPSK recieved Bits encode2(counter,2) = 1;
            end
        else
            if(real(QPSK_DataRx_encode2(counter)) > 0)
                QPSK_recieved_Bits_encode2(counter,1) = 1;
                QPSK recieved Bits encode2(counter,2) = 1;
                QPSK_recieved_Bits_encode2(counter,1) = 0;
QPSK_recieved_Bits_encode2(counter,2) = 0;
            end
        end
    QPSK demappedBits encode2 = reshape(QPSK recieved Bits encode2',1,[]);
    [N BER QPSK, QPSK BER(SNR dB)] = symerr(QPSK demappedBits, Bit Stream); %Calculated BER
    [N BER QPSK 2,QPSK BER encode2(SNR dB)] = symerr(QPSK demappedBits encode2,Bit Stream);
%Calculated BER
    QPSK BER Theo(SNR dB) = 0.5 * erfc(sqrt(1/QPSK No(SNR dB))); %Theoritical BER
%Plotting constillation of OPSK
figure(1)
subplot(2,2,2,'LineWidth',3)
plot(real(QPSK DataRx),imag(QPSK DataRx),'b*',real(1),imag(1i),'k.',real(-1),imag(-1i),'k.'
    ,real(1),imag(-1i),'k.',real(-1),imag(1i),'k.');
title('QPSK Modulation for SNR = 10')
%Plotting BER
figure (2)
subplot(2,2,2,'LineWidth',3)
EbN0 dB = 1:1:(SNR max value + 1);
%Plotting theoritical BER
semilogy((EbN0_dB - 1),QPSK_BER_Theo,'--')
%Plotting calculated BER
```

```
hold on
semilogy((EbN0 dB - 1),QPSK BER,'-')
grid on
ylabel('BER')
xlabel('E b/N 0')
title('Bit Error Rate for QPSK')
legend('Theoretical','Calculated');
figure (4)
EbN0_dB = 1:1:(SNR_max_value + 1);
%Plotting theoritical BER
semilogy((EbN0 dB - 1),QPSK BER encode2,'-x')
%Plotting calculated BER
hold on
semilogy((EbN0 dB - 1),QPSK BER,'-o')
grid on
ylabel('BER')
xlabel('E b/N 0')
title('Bit Error Rate for OPSK')
legend('2nd encode method','grey encoded');
%% 8PSK
%initializations needed for Simulation of BPSK
M = 8;
                        %Numbers of symboles
log2M = log2(M);
                       %Numbers of bit that represnts 8PSK
M8PSK Eb = 1/3;
M8PSK BER Theo = zeros(1, SNR max value);
                                                     %Vector to store the theortical BER
for different SNR channel
M8PSK BER = zeros(1, SNR max value);
                                                     %Vector to store the calculated BER
for different SNR channel
%Grouping the binary data into groups of 3 bits
M8PSK reshaped binary data = reshape(Bit Stream, log2M, [])';
%Mapping the input data to 8PSK symbols grey encoded
% 1. 0 0 0-> theta = 0
% 2. 0 0 1-> theta = 45
  3. 0 1 1-> theta = 90
% 4. 0 1 0-> theta = 135
% 5. 1 1 0-> theta = 180
  6. 1 1 1-> theta = 225
  7. 1 0 1-> theta = 270
  8. 1 0 0-> theta = 315
% Each symbol is represented by S(i) = 1* \exp(itheta)
 \texttt{M8PSK mapper} = [[0,0,0];[0,0,1];[0,1,1];[0,1,0];[1,1,0];[1,1,1];[1,0,1];[1,0,0]]; \\
M8PSK data = zeros(length(M8PSK reshaped binary data),1); %Vector to store the complex
value of symbols of 8PSK
%Loop over each row in the reshaped data
for M8PSK reshaped binary data row = 1 : length(M8PSK reshaped binary data)
    %Loop over rows of 8PSK mapper to map to the right angle
    for M8PSK_mapper_row = 1 : length(M8PSK_mapper)
        %Check if the row of the reshaped data equals to the mapper row
        if isequal (M8PSK reshaped binary data (M8PSK reshaped binary data row,:)...
                ,M8PSK_mapper(M8PSK_mapper_row,:))
            %Then assign the complex value correspondes to the corresponding angle
            %Calculate the corresponding angle
            M8PSK Theta = (M8PSK mapper row - 1) * (2*pi / M);
            %Store the complex value that represents the symbol
            M8PSK data(M8PSK reshaped binary data row) = exp(1i * M8PSK Theta);
            break; %Break on matching the rows
        end
   end
%Channel (AWGN)
M8PSK channelNoise real = zeros(size(M8PSK data,1),SNR max value); %Matrix to store the
real noise in each channel
M8PSK channelNoise complex = zeros(size(M8PSK data,1),SNR max value); %Matrix to store the
complex noise in each channel
M8PSK No = zeros(1, SNR max value + 1);
%Calculating No of the channel for different SNR
```

```
for SNR dB = 1 : (SNR_max_value + 1)
    %Generating different noise vector for different channels with different SNR
    M8PSK No(SNR dB) = M8PSK Eb/10.^((SNR dB-1)/10);
    noise I = randn(1, size(M8PSK data, 1));
   {\tt M8PSK\_channelNoise\_real(:,SNR\_dB) = noise\_I * sqrt (M8PSK\_No(SNR\_dB)/2); \$scale the}
    noise Q = randn(1, size(M8PSK data, 1));
    M8PSK_channelNoise_complex(:,SNR_dB) = noise_Q * sqrt (M8PSK_No(SNR_dB)/2); %scale the
noise.
end
%Demapper
M8PSK demappedBits = zeros(1,length(Bit Stream)); %Vector to store the demapped stream of
hits
M8PSK recieved Bits = zeros((Bits Number/3),3);
for SNR dB = 1: (SNR max value + 1)
    %Recieved Data
    M8PSK DataRx = ((real(M8PSK data) + M8PSK channelNoise real(:, SNR dB)) ...
    + 1i *(imag(M8PSK data)+ M8PSK channelNoise complex(:,SNR dB)));
    %Demapping the recieved symbol stream for each channel
        Demapping by computing the angle of the recieved bit and decide the
        %corresponding symbol based on the decision reagin.
        for counter = 1 : size(M8PSK DataRx,1) %Loop on each symbol of the recieved data.
            Rx symbol angle = angle(\overline{M8}PSK DataRx(counter)); %Calculate the angle of the
symbol in radian
            %Return the angle ot the positve value if it's negative
            if(Rx_symbol_angle < 0)</pre>
                Rx symbol angle = Rx symbol angle + 2*pi;
            %Compare the angle of the symbol with the angles of the descion regions
            %if 22.5^{\circ} = angle || 337.5^{\circ} <= angle
            if((Rx_symbol_angle <= pi/8) || (Rx_symbol_angle >= 15* pi/8))
                %which means 0 0 0 case
                M8PSK recieved Bits(counter, :) = M8PSK mapper (1,:);
            else
                for bounder = 1 : 2 : 13
                    if((Rx symbol angle > bounder * pi/8)&&(Rx symbol angle <= (bounder +</pre>
2) * pi/8))
                        M8PSK recieved Bits(counter, :) = M8PSK mapper (((bounder + 1)/2) +
1 ,:);
                        break;
                    end
                end
            end
        end
    %Reshape the recieved bits to one vector bit stream
    M8PSK demappedBits = reshape (M8PSK recieved Bits',1,[]);
    [N_BER_M8PSK , M8PSK_BER(SNR_dB)] = symerr(M8PSK_demappedBits,Bit Stream); %Calculated
BER
   M8PSK BER Theo(SNR dB) = erfc(sqrt(1/M8PSK No(SNR dB)) * sin(pi/8))/3; %Theoritical BER
end
%Plotting constillation of 8PSK
subplot(2,2,3,'LineWidth',3)
plot(real(M8PSK DataRx),imag(M8PSK DataRx),'g*')
hold on
% Plotting the symbols on the constalation
M8PSK sPoints = zeros(1, M);
for counter = 1:M
    Theta = (counter - 1) * (2*pi / M);
    M8PSK sPoints(counter) = exp(1i * Theta);
    plot(real(M8PSK sPoints(counter)),imag(M8PSK sPoints(counter)),'k.');
    hold on
end
hold off
title('8PSK Modulation for SNR = 10')
%Plotting BER
figure (2)
subplot(2,2,3,'LineWidth',3)
EbNO dB = 1:1:(SNR max value + 1);
%Plotting theoritical BER
semilogy((EbN0 dB - 1), M8PSK BER Theo, '--')
```

```
%Plotting calculated BER
hold on
semilogy((EbN0 dB - 1), M8PSK BER, '-')
arid on
ylabel('BER')
xlabel('E_b/N_0')
title('Bit Error Rate for 8PSK')
legend('Theoretical','Calculated');
%% 160AM
%initializations needed for Simulation of 16QAM
QAM16 Eb = 2.5;
QAM16 BER Theo = zeros(1, SNR max value);
                                                     %Vector to store the theortical BER
for different SNR channel
QAM16 BER = zeros(1,SNR_max_value);
                                                     %Vector to store the calculated BER
for different SNR channel
%Mapper
%Grouping the binary data into groups of 4 bits
QAM16 reshaped binary data = reshape(Bit Stream, 4, []) ';
%Mapping every four bits in one symbol coded in grey code
%Mapping table
QAM16_map = [-3-3i, -3-1i, -3+3i, -3+1i,... % 0000 -> -3 -3i | 0001 -> -3 - 1i | 0010 -> -3
+ 3i | 0011 -> -3 + 1i
             -1-3i, -1-1i, -1+3i, -1+1i,... % 0100 -> -1 -3i | 0101 -> -1 - 1i | 0110 -> -1
+ 3i | 0111 -> -1 + 1i
3-3i, 3-1i, 3+3i, 3+1i,... % 1000 -> 3 -3i | 1001 -> 3 - 1i | 1010 -> 3 + 3i | 1011 -> 3 + 1i
             1-3i, 1-1i, 1+3i, 1+1i]; % 1100 -> 1 -3i | 1101 -> 1 - 1i | 1110 -> 1
+ 3i | 1111 -> 1 + 1i
%Map the bits to the symbol
QAM16 data = QAM16 map(bi2de(QAM16 reshaped binary data, 'left-msb') + 1);
%Channel (AWGN)
QAM16_channelNoise_real = zeros(length(QAM16_data),SNR_max_value); %Matrix to store the
real noise in each channel
QAM16 channelNoise complex = zeros(length(QAM16 data),SNR max value); %Matrix to store the
complex noise in each channel
\mbox{\ensuremath{\mbox{$NC}}} alculating No of the channel for different SNR
QAM16 No = zeros(1, SNR max value + 1);
for SNR dB = 1: (SNR max value + 1)
    %Generating different noise vector for different channels with different SNR
    QAM16 No(SNR dB) = QAM16 Eb./10.^((SNR dB-1)/10);
    noise I = randn(1,length(QAM16 data));
    QAM16 channelNoise real(:,SNR dB) = noise I * sqrt (QAM16 No(SNR dB)/2); %scale the
noise.
    noise Q = randn(1,length(QAM16 data));
   QAM16 channelNoise_complex(:,SNR_dB) = noise_Q * sqrt (QAM16_No(SNR_dB)/2); %scale the
noise.
end
%Demapper
QAM16 demappedBits = zeros(1,length(Bit Stream)); %Vector to store the demapped bit
QAM16 recieved Bits = zeros(length(QAM16 data),4);
for SNR dB = 1: (SNR max value + 1)
    %Recieved Data
    QAM16 DataRx = (real(QAM16 data) + QAM16 channelNoise real(:,SNR dB)') ...
    + 1i * (imag(QAM16 data) + QAM16 channelNoise complex(:, SNR dB)');
    %Demapping the recieved symbol stream for each channel
    for counter = 1 : length(QAM16 DataRx)
        %Assigning the real part (b0b1xx)
        if(real(QAM16 DataRx(1,counter)) > 2)
                                                    Symbol = 10xx
            QAM16 recieved Bits(counter,1) = 1;
            QAM16 recieved Bits(counter,2) = 0;
        elseif(real(QAM16 DataRx(counter)) > 0)
                                                    Symbol = 11xx
            QAM16 recieved Bits (counter, 1) = 1;
            QAM16_recieved Bits(counter,2) = 1;
        elseif(real(QAM16 DataRx(counter)) > -2)
                                                  %Symbol = 01xx
```

```
QAM16 recieved Bits(counter,1) = 0;
            QAM16 recieved Bits(counter,2) = 1;
        else
                                                     Symbol = 00xx
            QAM16 recieved Bits(counter,1) = 0;
            QAM16 recieved Bits (counter, 2) = 0;
        end
        %Assigning the complex part (xxb2b3)
        if (imag(QAM16 DataRx(1,counter)) > 2)
                                                     Symbol = xx10
            QAM16 recieved Bits (counter, 3) = 1;
            QAM16_recieved_Bits(counter,4) = 0;
        elseif(imag(QAM16 DataRx(counter)) > 0)
                                                     Symbol = xx11
            QAM16 recieved Bits (counter, 3) = 1;
            QAM16 recieved Bits (counter, 4) = 1;
        elseif(imag(QAM16 DataRx(counter)) > -2)
                                                     %Svmbol = xx01
            QAM16_recieved_Bits(counter,3) = 0;
            QAM16 recieved Bits (counter, 4) = 1;
                                                     Symbol = xx00
            QAM16_recieved_Bits(counter,3) = 0;
QAM16_recieved_Bits(counter,4) = 0;
        end
    end
    %Reshape the recieved bits to vector
    QAM16_demappedBits = reshape(QAM16_recieved_Bits',1,[]);
    [N BER QAM16,QAM16 BER(SNR dB)] = symerr(QAM16 demappedBits,Bit Stream); %Calculated
BER
    QAM16 BER Theo(SNR dB) = 3/8 * erfc(sqrt(1./(1.*QAM16 No(SNR dB))));  %Theoritical BER
end
%Plotting constillation of 16QAM
figure(1)
subplot(2,2,4,'LineWidth',3)
plot(real(QAM16 DataRx),imag(QAM16 DataRx),'m*')
%Plotting the symbols on the constalation
for row = -3: 2: 3
    for col = -3: 2 : 3
        plot(real(row),imag(1i* col),'k.');
        hold on
    end
end
hold off
title('16QAM Modulation for SNR = 10')
%Plotting BER
figure(2)
subplot(2,2,4,'LineWidth',3)
EbNO dB = 1:1:(SNR max value + 1);
%Plotting theoritical BER
semilogy((EbN0 dB - 1),QAM16 BER Theo,'--')
%Plotting calculated BER
semilogy((EbN0 dB - 1),QAM16 BER,'-')
arid on
ylabel('BER')
xlabel('E b/N 0')
title('Bit Error Rate for 16QAM')
legend('Theoretical','Calculated');
%% Plotting all the BER, theoritcal and calculated, on the same graph
figure(3)
Ebn0 dB = 1:1:(SNR max_value + 1);
%Plotting theoritical BPSK_BER
semilogy((EbN0_dB - 1),BPSK_BER_Theo,'--')
%Plotting theoritical QPSK BER
semilogy((EbN0 dB - 1),QPSK BER Theo,'--')
hold on
%Plotting theoritical 8PSK BER
semilogy((EbN0 dB - 1), M8PSK BER Theo, '--')
hold on
%Plotting theoritical 16QAM BER
semilogy((EbN0 dB - 1),QAM16 BER Theo,'--')
hold on
%Plotting theoritical BFSK BER
%---- Plot the BFSK BER here ----
```

```
%hold on
%Plotting calculated BPSK BER
semilogy((EbN0 dB - 1), BPSK BER, '-o')
%Plotting calculated QPSK BER
semilogy((EbN0 dB - 1),QPSK BER,'-*')
hold on
%Plotting calculated 8PSK BER
semilogy((EbN0 dB - 1), M8PSK BER, '-x')
hold on
%Plotting calculated 16QAM BER
semilogy((EbN0 dB - 1),QAM16 BER,'-+')
hold off
arid on
legend('Theoretical BSPK BER','Theoretical QPSK BER','Theoretical 8PSK BER',...
    'Theoretical QAM16 BER', 'BSPK BER', 'QPSK BER', '8PSK BER', 'QAM16 BER');
xlabel('E {b}/N {o}');
ylabel('BER');
title("BER {theoritical and calculated} of some modulation techniques");
```

#### **BFSK Code**

```
* FILE DESCRIPTION
* File: BFSK.m
* Digital communication project-assignment

* Description: Simulation of BFSK modulation technique
* Author:Omar Muhammad Mustafa, Omar Muhammad Tolba
* Date: 6th May 2023
용 }
%% clear all WorkSpace Variables and Command Window
clc;
clear ;
close all;
%% initialization
Bits Number = 1.2e5+1;
                                                           % number of bits to be generated
SNR max value = 10;
N Samples = 10;
                   % Number of samples
N Realization = 10; % Number of realization
%% Generating the data bits
Bits Sent = randi([0 1], N Realization, Bits Number);
Bit Stream = Bits Sent(1,:);
%% Initialzie Parameters
BFSK Eb = ((1 +1)/2)/1;
                                                      %Calculating Eb of BFSK modulation
BFSK BER Theo = zeros(1,SNR_max_value);
                                                      %Vector to store the theortical BER for
different SNR channel
BFSK BER = zeros(1, SNR max value);
                                                     %Vector to store the calculated BER for
different SNR channel
\% Mapper f1 and f2 need to be orthogonal at each other
BFSK symbolStream = zeros(1,Bits Number);
for \overline{i} = 1: Bits Number
    if(Bit Stream(i) == 1) % Map to f1 (1,0i)
        BFSK symbolStream(i)=1;
    else
        BFSK symbolStream(i)=1i;% Map to f1 (0,i)
end
%% Channel (AWGN)
BFSK channelNoise real = zeros(length(Bit Stream), SNR max value); %Matrix to store the real
noise in each channel
BFSK channelNoise complex = zeros(length(Bit Stream), SNR max value); %Matrix to store the
complex noise in each channel
BFSK No = zeros(1, SNR max value + 1);
noise I = randn(1,length(Bit Stream)); % in-phase noise
noise_Q = randn(1,length(Bit_Stream)); % quadrature phase noise
%Calculating No of the channel for different SNR
for SNR dB = 1: (SNR max value + 1)
    %Generating different noise vector for different channels with different SNR
```

```
BFSK No(SNR dB) = BFSK Eb/10.^{(SNR dB-1)/10)};
    BFSK channelNoise real(:, SNR dB) = noise I * sqrt (BFSK No(SNR dB)/2); %scale the
noise. [TODO] (check variance)); %scale the noise.
   BFSK channelNoise complex(:,SNR dB) = noise Q * sqrt (BFSK No(SNR dB)/2); %scale the
noise. [TODO] (check variance)); %scale the noise.
end
%% DeMapper
BFSK ReceivedBits = zeros(1,length(Bit Stream));
                                                  %Vector to store the demapped stream
BFSK DemappedBits = zeros(1,length(Bit Stream));
                                                   %Vector to store the demapped stream
for SNR dB = 1: (SNR max value + 1)
    % Recieved Data for the last channel noise [10]
    BFSK DataRx = (real(BFSK symbolStream) + BFSK channelNoise real(:,SNR dB)')...
        +1i * (imag(BFSK symbolStream) +BFSK channelNoise complex(:,SNR dB)');
    % Demapping the recieved symbol stream for each channel
    for counter = 1 : Bits Number
        if((angle(BFSK DataRx(1,counter)) <= pi/4) && (angle(BFSK DataRx(1,counter)) >= -
(3*pi/4))
            BFSK ReceivedBits(1,counter) = 1+0i;
            BFSK_DemappedBits(1,counter) = 1;
        else
            BFSK ReceivedBits(1,counter) = 0+1i;
            BFSK DemappedBits(1,counter) = 0;
        end
    end
    [N BER BPSK, BFSK BER(SNR dB)] = symerr(BFSK DemappedBits, Bit Stream); %Calculated BER
    BFSK BER Theo(SNR dB) = 0.5 * erfc(sqrt(BFSK Eb/(2*BFSK No(SNR dB)))); %Theoritical BER
% Plotting constillation of BPSK
figure(1)
subplot(1,1,1,'LineWidth',3)
plot(real(BFSK DataRx),imag(BFSK DataRx),'r*',real(sqrt(BFSK Eb)),imag(0),'k.',real(0),imag
(1i*sqrt(BFSK Eb)), 'k.');
title('BFSK Modulation for SNR = 10')
%Plotting BER
figure (2)
subplot(1,1,1,'LineWidth',3)
EbN0 dB = 1:1:(SNR max value + 1);
%Plotting theoritical BER
semilogy((EbN0 dB - 1), BFSK BER Theo, '--')
%Plotting calculated BER
hold on
semilogy((EbN0 dB - 1), BFSK BER, '-')
arid on
-
ylabel('BER')
xlabel('E_b/N_0')
title('Bit Error Rate for BPSK')
legend('Theoretical','Calculated');
%% PSD
Tb = 1e-3;
f1 = 1e3;
f2 = f1 + (1/Tb);
t =0:(1/N Samples)*Tb:Tb*(1-1/N Samples); % sampling interval
s1bb = repelem(sqrt((2*BFSK Eb)/Tb), N Samples); %Mapping zero
s2bb = sqrt((2*BFSK Eb)/Tb) * exp(2*1i*pi*t*(1/Tb)); % Mapping 1
BFSK Mapped_Realization1 = zeros(1, N_Samples*Bits_Number);
BFSK_Mapped_Ensemble=zeros(N_Realization, N_Samples*Bits_Number);
mapped ensamble delayed = [];
for realiz counter = 1 : N Realization
    for bit count = 1 : Bits Number
        if(Bits Sent(realiz counter,bit count) == 0)
            BFSK Mapped Ensemble(realiz counter, (bit count-1) *N Samples + 1:
N Samples*bit count) = \overline{s1bb};
            BFSK Mapped Ensemble (realiz counter, (bit count-1) *N Samples + 1 :
N Samples*bit count) = \overline{s2bb};
        end
    end
    %%% adding random delay to the ensamble %%%
```

```
delay = randi([1,N Samples]) + 1; % delay = 1 means there is no delay, hence we add a
+1 offset
   mapped ensamble delayed = vertcat(mapped ensamble delayed,
BFSK Mapped Ensemble(realiz counter, delay : end - N Samples + (delay -1) ));
end
%% PSD calculation
for column = 1: size(mapped ensamble delayed,2)
   R tau(column) = (1/N Realization) * sum(mapped_ensamble_delayed(:, 1) .*
(mapped_ensamble_delayed(:, column)));
end
%% plottion PSD
%%Bandwidth
L = length(R_tau);
                      % Length of signal
n = 2^nextpow2(L);
T = 0.1 * Tb;
                        % Sampling period
Fs = 1/T;
                      % Sampling frequency
t = (0:L-1)*T;
                      % Time vector
%Compute the Fourier transform of the signal.
PSD = fft(R tau,n);
PSD = fftsh\overline{i}ft(PSD);
f = Fs * (-n/2:n/2-1)/n;
%Compute the two-sided spectrum P2
P2 = abs(PSD/n).^2;
Smoothed PSD = smooth(P2); %Smooth the curve from the noise, to be plotted
figure (\overline{4})
subplot(1,1,1,'LineWidth',5)
plot(f,Smoothed PSD)
ylim([0 1e3])
title("PSD")
xlabel("f(Hz)")
ylabel("S(f)")
% End of File [BFSK.m]
```