<u>DIGITAL COMMUNICATIONS LAB</u> <u>LAB 5.2</u> GENERATE BER vs. SNR PLOTS FOR (15,7), (15,11) CODER OVER AWGN CHANNEL

FREQUENCY SHIFT KEYING:

The tones used are: f1 = 1000Hz and f2 = 2000Hz The bit rate is 1000 bps. Therefore, the duration of each bit is 1/1000 = 1 millisecond Number of samples = Sampling frequency * duration of each bit = 8 samples The bandwidth is 2 * (bit rate) + (f2-f1) = 3000 Hz

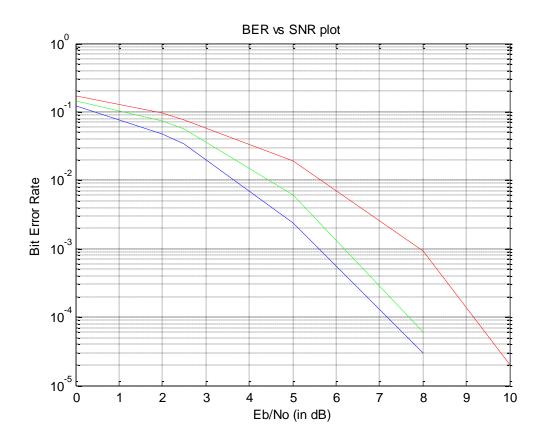
The two waveforms used are:

$$0: \sqrt{\frac{2}{T}}\cos(2000\pi t)$$

1:
$$\sqrt{\frac{2}{T}}\cos(4000\pi t)$$

The BER vs. SNR plot for the following three cases are plotted in the graph shown below:

- 1. No Coder
- 2. (15,11) coder
- 3. (15,7) coder



Legend:

Red - No Coder Green - (15,11) Coder Blue - (15,7) Coder

Note: The graph is plotted for Eb/No values : [-5 -2.5 0 2.5 5 10] dB

We can observe that the (15, 7) coder performs better than the (15, 11) coder and the no coder case as the BER values for the (15,7) coder is lower than those of the (15, 11) coder and the no coder case for the same Eb/No ratio. Similarly, the one-bit error correcting code shows a lower BER as compared to the no coder case.

APPENDIX-1

1. Function to modulate a given sequence of bits using Frequency Shift Keying

```
function symbols = bits2symbolsfsk(bits vector)
%This function returns a vector of symbols corresponding to the given
%vector of bits
%The number of samples taken for each time interval T(0.001s) is 8 samples
%The waveforms are orthogonal
%Bit 0 is mapped to Acos(2*pi*1000*t) and bit 1 is mapped to
A\cos(2\pi^2). Here, A is (2/T)^0.5
num samples = 8;
T = 0.001;
A = (2/T)^{(0.5)};
t = 0:T/num samples:T-0.00001;
x = A.*cos(2000*pi*t);
y = A.*cos(4000*pi*t);
output matrix = zeros(length(bits vector), num samples);
for i = 1:1:length(bits_vector)
    if bits vector(i) == 0
        output matrix(i,:) = x;
    else
        output matrix(i,:) = y;
    end
end
symbols = reshape(output matrix', 1, length(bits vector)*num samples);
```

2. Function to add white Gaussian noise to the waveform

```
function output vector = addwhitegaussiannoisefsk(input vector, SNR)
%The function takes a signal as an input
%The output is a noisy signal, with white gaussian noise added to the input
signal
%The input SNR is in dB
%The standard deviation(sigma) is computed using the given SNR value
SNR = db2pow(SNR);
%Eb = power * time
Tb = 0.001;
Eb = (sum(input vector.^2)/length(input vector))*Tb;
No = Eb/SNR;
%noise_power = (No/2) * Bandwidth
bandwidth = 3000;
noise power = (No/2)*bandwidth;
sigma = noise power^0.5;
s = size(input vector);
noise = sigma.*randn(s(1),s(2));
output vector = input vector + noise;
end
```

3. Function to convert the signal to corresponding bit sequence using a matched filter

```
function bit vector = symbols2bitsfsk(symbol vector)
num_samples = 8;
T = 0.001;
A = (2/T)^{(0.5)};
t = 0:T/num samples:T-0.00001;
%The matched filer corresponding to bit 1
matchedfilters1 = A.*cos(2000*pi*(T-t));
%The matched filter corresponding to bit 0
matchedfilters2 = A.*cos(4000*pi*(T-t));
%The number of samples in each time interval T is 8
%The number of symbols is equal to (length of signal)/(number of samples)
numsymbols = length(symbol vector)/num samples;
bit_vector = zeros(1, length(symbol_vector)/num samples);
symbol matrix = vec2mat(symbol vector, num samples);
%Convert each symbol to the corresponding bit
for i = 1:1:numsymbols
    current symbol = symbol matrix(i,:);
    %Convolve with matched filter corresponding to symbol 1 and get the
peak value
    z1 = max(conv(current symbol, matchedfilters1));
    %Convolve with matched filter corresponding to symbol 2 and get the
peak value
    z2 = max(conv(current_symbol, matchedfilters2));
    % If z1>z2, the symbol is s1(i.e. 1), otherwise symbol is s2(i.e. 0)
    if z1 > z2
        bit_vector(:,i) = 0;
    else
        bit vector(:,i) = 1;
    end
end
end
```

4. Script to plot BER vs. SNR when no coder is used

```
message = randi(2,1,10^5)-1;
SNR = [0,2,2.5,5,8,10,15];
ber = zeros(1,length(SNR));
block length = 7;
for j = 1: 1:length(SNR)
    x vector = [];
    x final = [];
    disp(SNR(j));
    x = message;
    signal = bits2symbolsfsk(x);
    noisy signal = addwhitegaussiannoisefsk(signal, SNR(j));
    recv signal = symbols2bitsfsk(noisy signal);
    x final = [x final, recv signal];
    disp(size(message));
    disp(size(x final));
    error matrix = mod((message-x final),2);
    disp(size(error_matrix));
    count = numel(find (error matrix == 1));
    bit error rate = count/length(message);
    ber(:,j) = bit error rate;
end
semilogy(SNR, ber, 'r');
grid;
xlabel('Eb/No (in dB)');
```

```
ylabel('Bit Error Rate');
title('BER vs SNR plot');
```

5. Function to plot BER vs. SNR graph using (15, 11) and (15, 7) coder

```
function ber = simulatechannelfsk(encoderType,color)
message = randi(2,1,10^5)-1;
if encoderType == 1
    g = [1, 0, 0, 1, 1];
    block length = 11;
elseif encoderType == 2
    g = [1, 1, 1, 0, 1, 0, 0, 0, 1];
    block length = 7;
end
message = vec2mat(message, block length);
SNR = [0,2,2.5,5,8,10,15];
ber = zeros(1,length(SNR));
for j = 1: 1:length(SNR)
    message final = zeros(size(message,1),block length);
    disp(SNR(j));
    for i = 1 : 1: size(message, 1)
        x = message(i,:);
        u = BCHEncoder(x,encoderType);
        signal = bits2symbolsfsk(u);
       noisy signal = addwhitegaussiannoisefsk(signal, SNR(j));
       recv signal = symbols2bitsfsk(noisy signal);
       msg = BCHDecoder(recv signal, encoderType);
       msq = deconv(msq, q);
       msg = mod(msg, 2);
       message final(i,:) = msg;
error matrix = mod((message-message final),2);
disp(size(error matrix));
count = numel(find (error matrix == 1));
bit_error_rate = count/(2 \times 10^5);
ber(:,j) = bit error rate;
semilogy(SNR, ber, color);
end
```