APPENDICES

APPENDIX - I

SOURCE CODE

% M-File to estimate the channel and to develop the bit loading profile

clc;	
clearall;	
closeall;	
% Inputs from user	
pow = input('Enter the total power of the	signal to be transmitted: ');
l(1)=input(' enter the length to be conside	ered in Km : ');
margin= input('enter the margin to be giv	en in dB: ');
gapfactordb=9.8+margin;	
disp('taking a bit error tolerance of 10^(-6	5) and no coding gain the gap factor in dB is: ');
disp(gapfactordb);	
%Define the properties of the transmitter	
tsym=3.3333333333334e-05	%symbol duration to make
df=1875;	
rsym=1/tsym;	%symbol rate
N=16;	%Number of parallel symbols
tnew=N*tsym;	%new symbol duration = 800 micro
seconds	
rnew=1/tnew;	
df=rnew;	%frequency spacing
%frequency values for the sixteen tones	
fval=zeros([1 16]);	
fori=1:16	

```
fval(i)=1875+(i-1)*df;
fval(i)=floor(fval(i));
end
% QPSK Modulation of the bit pattern generated above
% 64 tones in total, inter-tone spacing of 500 Hz
seq = randi([0 1], 32, 1);
                                                    % Random 32 bit sequence - 16 tones in
the 20KHz channel
mod = modem.pskmod(4,pi/4);
mod.inputtype = 'bit';
mod.SymbolOrder = 'gray';
q = modulate(mod,seq);
%figure(3)
disp(q);
% IFFT Definition
Fs = 120000;
                                                         %so that the 20k component of freq
exists.
T = 1/Fs;
L = 120000;
t = (0:L-1)*T;
freqdom=zeros([1 120000]);
q = [q;flipud(conj(q))];
                                                   %complex conjugate of the signals added
to fft
input('Press Enter to view the inputs to the IFFT block');
disp('Input to the IFFT Block');
%figure(1)
disp(q);
fori = 1:16
freqdom(fval(i)) = q(i);
freqdom(120000-fval(i)) = q(33-i);
```

```
% Displaying the transmitted signal
%figure(2)
subplot(2,3,1);
plot(abs(freqdom(1:30000)));
                                                 %display only upto 30k
holdon;
xlabel('Frequency - Hz');
ylabel('Tone Amplitude in Watt');
title('Multi-Tone Structure Transmitted - Frequency Domain');
gridon;
freqdom1 = freqdom.*60000;
                                                  %according to ifftdefiniton
realsig = ifft(freqdom1);
% Channel and Noise estimation
                                                  % Spreading factor
k=1.5:
fmax = 30500;
                                                          % Maximum value of frequency
required for calculation
                                                   % Function call to calculate noise
noise = ambient_noise(fmax);
% Use any one of the three channel models and comment the other two
%prop_const=thorp_model(fmax);
%prop_const=fisher_simmons_model(fmax);
prop_const=anslie_mccolm_model(fmax);
% Calculation of attenuation
L=l(1);
AdB1 = (k*10*log10(L) + L*prop\_const);
C1 = AdB1./10;
A1=10.^{(C1)};
```

```
% Calculating the effect of channel on the signal in frequency domain
freqdom_2(1:30000)=freqdom(1:30000)./A1(1:30000);
subplot(2,3,2);
plot(abs(freqdom_2));
xlabel('Frequency - Hz');
ylabel('Tone Amplitude in Watt');
title('Multi-Tone Structure at Receiver - Frequency Domain');
gridon;
% Convertion of power to voltage and then to dBreuPa %
                                                        % Dividing the power in all tones
pow = pow/16;
equally
imp = 2870;
                                                       % Refer data sheets for impedance
value
volts= sqrt (pow * imp);
convert = 31.6e6; %1v=31.6e6 micropascal
                                                                           % standard for
sound projectors
gapfactor = 10^(gapfactordb/20);
noise1= zeros(1,16);
% SNR and CbyBcalculation
SNR=zeros(1,16);
CbyB=zeros(1,16);
for G=1:16
Signal_power_dbreupa= volts * convert;
fori=(G*1875)-500:1:(G*1875)+500;
       noise1(G) = noise1(G) + A1(i)*noise(i);
                                                  %Integrating the noise
end
SNR(G)=(Signal_power_dbreupa/(noise1(G)))^2;
CbyB(G)=(log2(1+(SNR(G)/gapfactor)));
SNR(G) = 10 * log10(SNR(G));
```

```
end
subplot(2,3,4);
stem(SNR);xlabel('TONE NUMBER');
ylabel('SNR in dB');
title('SNR Profile');
gridon;
subplot(2,3,5);
stem(CbyB);
xlabel('TONE NUMBER');
ylabel('C/B in bits/Hz');
title('C/B Profile');
gridon;
% Rounding off by calling the equalize function by using any one of the functions and
comment the other two
CbyB_round = equalize_2(CbyB');
                                                  % Mean Offset algorithm
subplot(2,3,6);
stem(CbyB_round);
xlabel('TONE NUMBER');
ylabel('C/B bits/Hz');
title(' C/B Profile after Rounding off');
gridon;
% Recalculation of Power and plotting it
pow_tones=pow*((2.^(CbyB_round))-1)./((2.^(CbyB'))-1);
subplot(2,3,3);
stem(pow_tones);
xlabel('TONE NUMBER');
ylabel('POWER in Watt');
title('PSD AFTER ROUNDING');
gridon;
```

Functions used:

Ambient noise function

Anslie model function

Fisher simmons model function

```
function [prop_const]=fisher_simmons_model(fmax)
ph = 4;
S = 35;
                                                   % Salinity of water
d = 1000;
                                                   % depth of measurement
T = 10;
                                                  % temperature in degresscelsius
c = 1412 + 3.21*T + 1.19*S + 0.0167*d;
                                                   % speed of sound in m/s
t = 273 + T;
A1 = (8.86/c)*10^{(0.78*ph - 5)};
                                                   % Coefficients for the Equation
A2 = 21.44*(S/c)*(1 + 0.025*T);
A3 = 4.937*10^{(-4)} - 2.59*10^{(-5)}*T + 9.11*10^{(-7)}*T*T - 1.5*10^{(-8)}*T*T*T;
f1 = 2.8* sqrt(S/35)*10^{(4-1245/t)};
f2 = (8.17*10^{(8-1990/t)})/(1 + 0.0018*(S-35));
P1 = 1;
P2 = 1 - 1.37*(10^{-4})*d + 6.2*(10^{-9})*d*d;
P3 = 1 - 3.84*(10^{5})*d + 4.9*(10^{1})*d*d;
f=(1:1:fmax)/1000;
prop\_const = (A1*P1*f1*f.*f)./(f1*f1 + f.*f) + (A2*P2*f2*f.*f)./(f2*f2 + f.*f) +
A3*P3*f.*f;
end
```

Thorp model function

```
function [prop_const]=thorp_model(fmax)
f=(1:1:fmax)/1000;
prop_const=(0.11*(f.*f)./(1+f.*f))+(44*(f.*f)./(4100+f.*f))+2.75*10.^(-4)*f.*f+0.003;
end
```

Round off algorithm function

```
function [q1] = equalize_2(q)

q=q';

diff_easy = 0;

fori=1:16

if ( diff_easy> 0 )

q1(i) = floor(q(i));

elseif ( diff_easy< 0 )

q1(i) = ceil(q(i));

else

q1(i) = round(q(i));

end

diff_easy = diff_easy + q1(i) - q(i);

end

q1=q1';

end
```