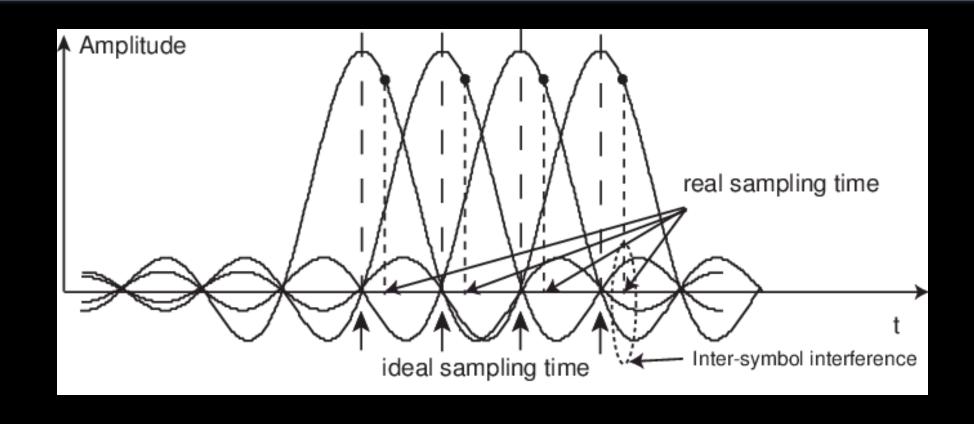
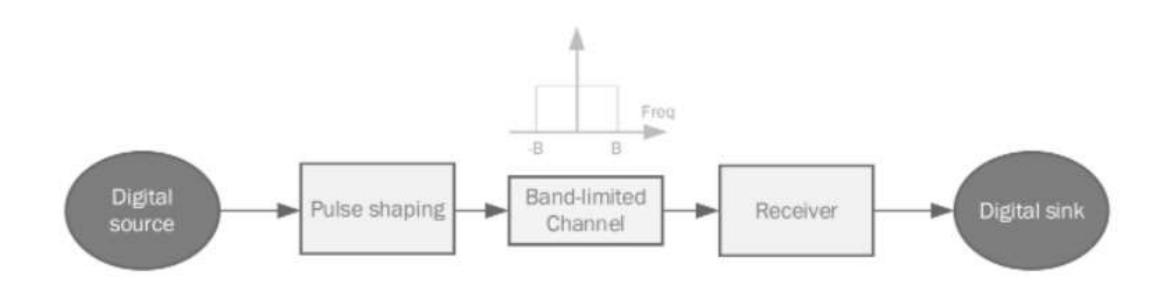
DIGITAL COMMUNICATIONS

LAB 3

WHAT IS INTER-SYMBOL INTERFERENCE (ISI)?

- Inter-symbol interference (ISI) is a form of distortion of a signal in which one symbol interferes with subsequent symbols.
- It's an unwanted phenomenon as the previous symbols have similar effect as noise.
- It makes the communication less reliable.



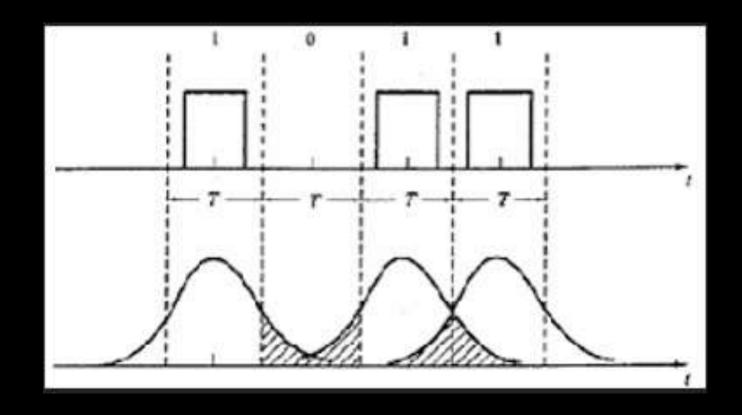


INTER-SYMBOL INTERFERENCE DUE TO BAND-LIMITED CHANNELS

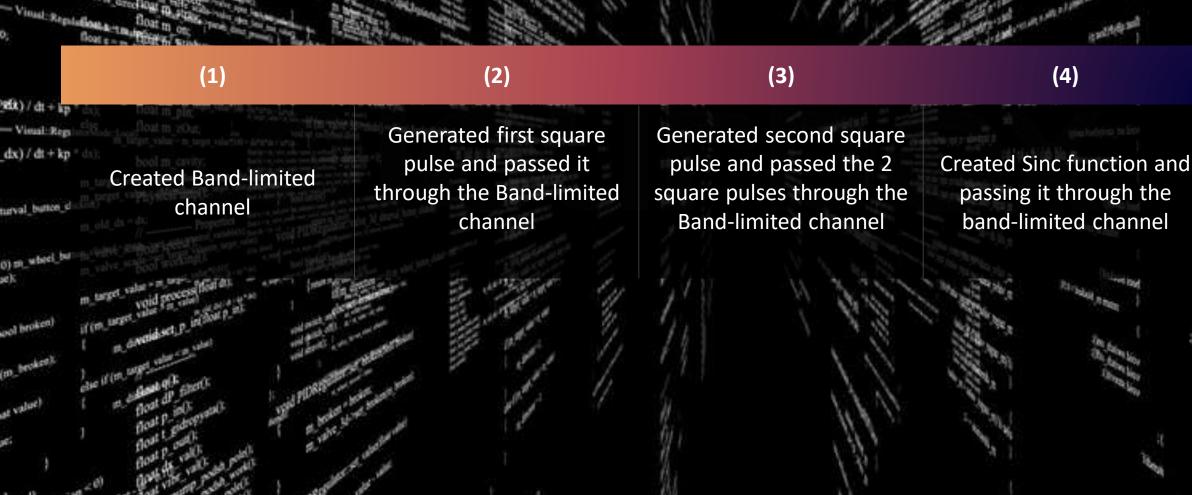
PART (1)

WHAT'S A BAND-LIMITED CHANNEL?

- The channel only allows a limited range of frequency
- components to pass.
- It blocks frequency components outside this range.
- The channel obviously limits the kind of signals that can pass unchanged through the channel.
- Most signals will have an output signal different from their input signal after passing through it.



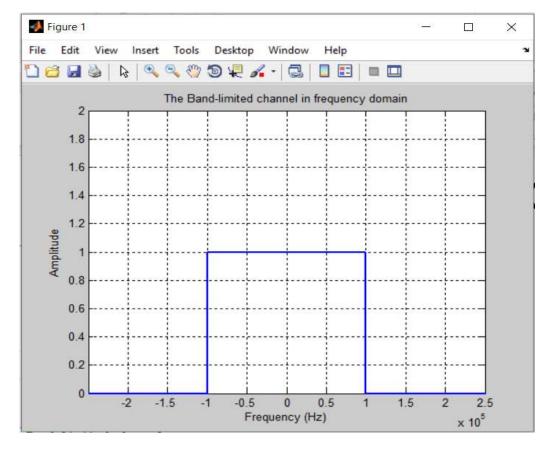




cavity;

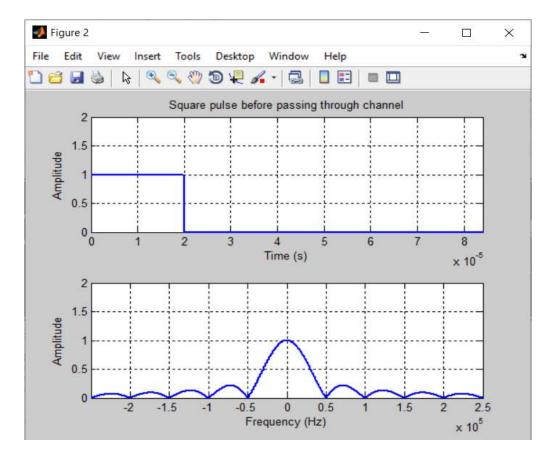
CREATED BAND-LIMITED CHANNEL

```
clc
close all
clear all
fs = le7;
Ts = 1/fs;
N = 1e7;
time axis = (0:N-1)*Ts;
freq axis = -fs/2:fs/N:fs/2-1/N;
B = 100e3;
T = 2/B;
%% Creating Band-limited channel
one square = ones(1,200e3);
zero me = zeros(1,9800e3/2);
Band limited channel= [zero me one square zero me];
figure
plot(freq_axis, Band_limited_channel, 'linewidth', 2)
grid on
ylim([0 2])
xlim([-1/T 1/T]*5)
xlabel ('Frequency (Hz)', 'linewidth', 2)
ylabel ('Amplitude', 'linewidth', 2)
title ('The Band-limited channel in frequency domain', 'linewidth', 10)
```



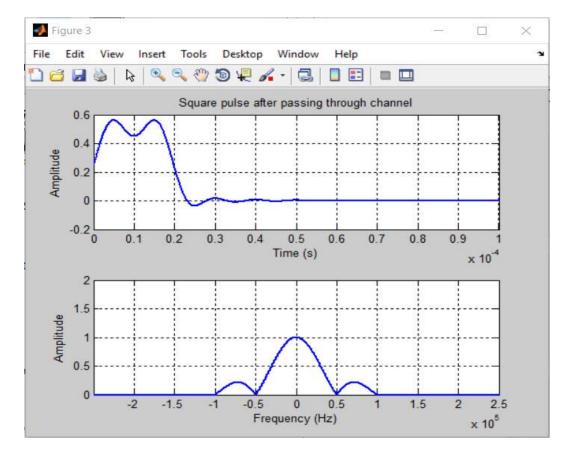
GENERATED FIRST SQUARE PULSE

```
%% Generating first square pulse
x bits = [1];
pulsel = rectpuls(time axis-1/B,T);
pulsel length = length(pulsel);
pulsel fft = (1/200) *fftshift(fft(pulsel));
freq axis = -fs/2:fs/pulsel length:fs/2-1/pulsel length;
%% Plotting first square pulse
figure
subplot (2, 1, 1)
plot(time axis, pulsel, 'b', 'linewidth', 2); hold on;
grid on
xlim([0 T*4.2])
ylim([0 2])
xlabel('Time (s)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
subplot(2,1,2)
plot(freq axis,abs(pulsel fft),'b','linewidth',2); hold on;
grid on
ylim([0 2])
xlim([-1/T 1/T]*5)
xlabel ('Frequency (Hz)', 'linewidth', 2)
ylabel ('Amplitude', 'linewidth', 2)
subplot(2,1,1)
title('Square pulse before passing through channel', 'linewidth', 10)
```



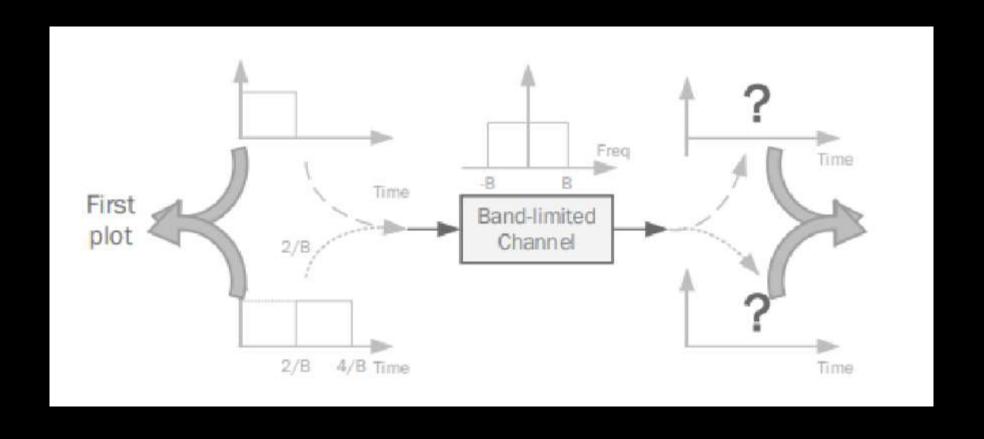
PASSED FIRST SQUARE PULSE THROUGH THE BAND-LIMITED CHANNEL

```
%% Passing first square pulse through the Band-limited channel
pulsel after chann = pulsel fft .* Band limited channel;
pulsel after chann T =100* ifft(ifftshift(pulsel after chann));
figure
subplot(2,1,1)
plot(time axis, pulsel after chann T, 'b', 'linewidth', 2); hold on;
grid on
xlim([0 T*5])
xlabel('Time (s)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
subplot(2,1,2)
plot(freq axis, abs(pulsel after chann), 'b', 'linewidth', 2); hold on;
grid on
xlim([-1/T 1/T]*5)
ylim([0 2])
xlabel('Frequency (Hz)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
subplot(2,1,1)
title ('Square pulse after passing through channel', 'linewidth', 10)
```



PASSING 2 SQUARE PULSES THROUGH THE CHANNEL

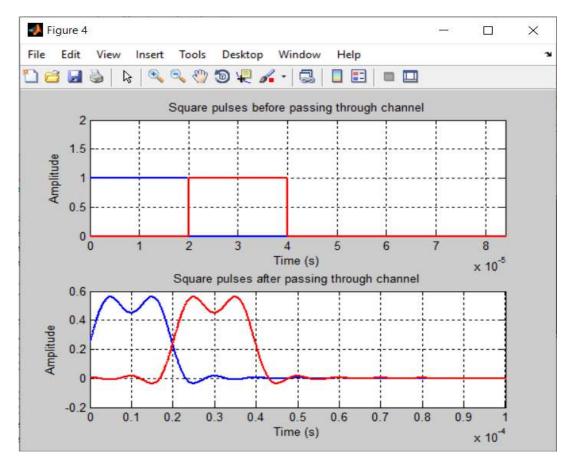
- Our expectation is that inter-symbol interference is going to be clear.
- Each one of the square pulses change shape so they will interfere with one another.
- The output signal will differ from the original signal.



GENERATED SECOND SQUARE PULSE AND

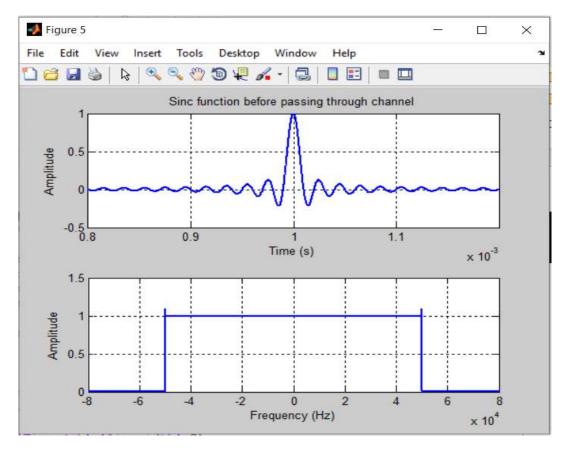
PASSED THE 2 SQUARE PULSES THROUGH THE BAND-LIMITED CHANNEL

```
%% Generating second square pulse
x bits = [0 1];
pulse2 = rectpuls(time axis-3/B,T);
pulse2 length = length(pulse2);
pulse2 fft =(1/200) * fftshift(fft(pulse2));
%% Plotting the 2 square pulses
figure
subplot (2,1,1)
plot(time axis, pulsel, 'b', 'linewidth', 2); hold on;
plot(time axis, pulse2, 'r', 'linewidth', 2); hold on;
grid on
xlim([0 T*4.2])
ylim([0 2])
xlabel('Time (s)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
title ('Square pulses before passing through channel', 'linewidth', 10)
%% Passing the 2 square pulses through the Band-limited channel
pulse2 after chann = pulse2 fft .* Band limited channel;
pulse2 after chann T =100* ifft(ifftshift(pulse2 after chann));
subplot (2,1,2)
plot(time_axis, pulsel_after_chann_T, 'b', 'linewidth', 2); hold on;
plot(time axis, pulse2 after chann T, 'r', 'linewidth', 2); hold on;
grid on
xlim([0 T*5])
xlabel('Time (s)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
title ('Square pulses after passing through channel', 'linewidth', 10)
```



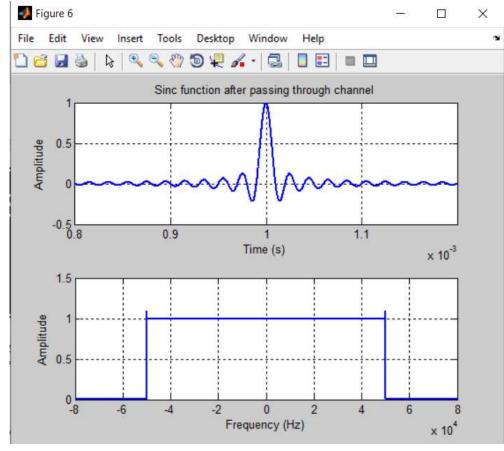
CREATED SINC FUNCTION

```
%% creating Sinc function and passing it through the band-limited channel
time axis = (-N/2:N/2-1)*Ts;
y = sinc(time axis*B);
yl = [zeros(1,10000) y(1:9990000)];
yl length = length(yl);
yl f = (1/100) * fftshift(fft(yl));
freq axis = -fs/2:fs/yl length:fs/2-1/yl length;
figure
subplot (2,1,1)
plot(time_axis,yl,'b','linewidth',2); hold on;
xlim([0.0008 0.0012])
grid on
xlabel('Time (s)', 'linewidth', 2)
ylabel('Amplitude', 'linewidth', 2)
subplot (2,1,2)
plot(freq_axis,abs(yl_f),'b','linewidth',2); hold on;
xlim([-80000 80000])
grid on
xlabel('Frequency (Hz)','linewidth',2)
ylabel ('Amplitude', 'linewidth', 2)
subplot (2, 1, 1)
title ('Sinc function before passing through channel', 'linewidth', 10)
```



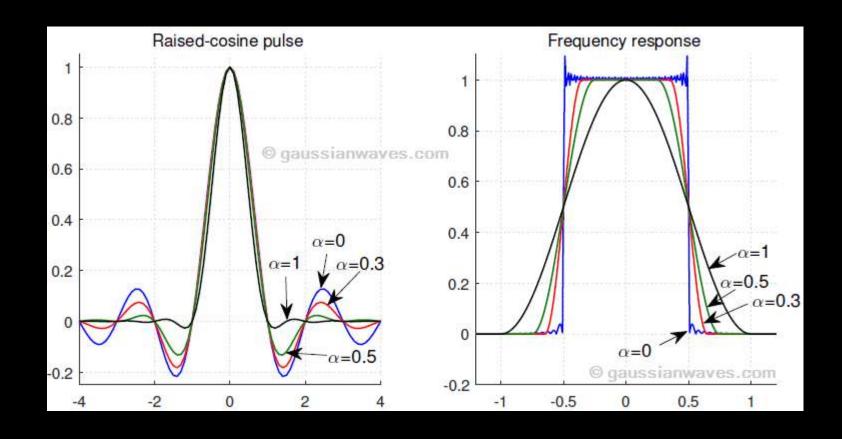
PASSED SINC FUNCTION THROUGH THE BAND-LIMITED CHANNEL

```
yl after ch = yl f .* Band limited channel;
yl after ch T = ifft(ifftshift(yl after ch));
figure
subplot (2,1,1)
plot(time axis, yl after ch T, 'b', 'linewidth', 2); hold on;
xlim([0.0008 0.0012])
grid on
xlabel('Time (s)','linewidth',2)
ylabel('Amplitude', 'linewidth', 2)
subplot (2,1,2)
plot(freq axis, abs(yl after ch), 'b', 'linewidth', 2); hold on;
xlim([-80000 80000])
grid on
xlabel('Frequency (Hz)','linewidth',2)
ylabel('Amplitude', 'linewidth', 2)
subplot(2,1,1)
title ('Sinc function after passing through channel', 'linewidth', 10)
```



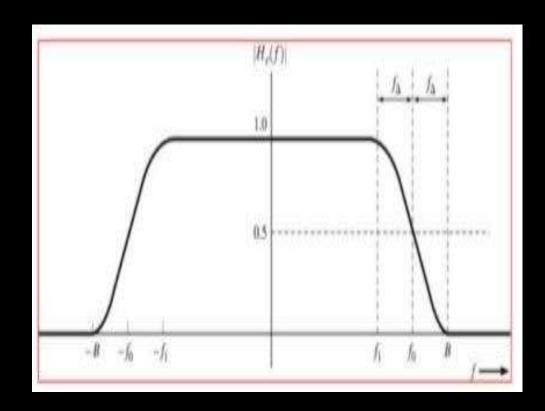
OVERCOME SQUARE PULSE PROBLEM WITH BAND-LIMITED CHANNELS

- According to previous outputs, we concluded that we can't use square pulse signals with band-limited channels.
- Other pulse shapes are better suited for band-limited channels.
- We used raised cosine to solve the problem of ISI.
- The raised-cosine filter is an implementation of a low-pass Nyquist filter.



RAISED-COSINE FILTER

NYQUIST'S FIRST METHOD (ZERO ISI)



$$h_e(t) = \frac{\sin \pi f_s t}{\pi f_s t} \longleftrightarrow H_e(f) = \frac{1}{f_s} \Pi \left(\frac{f}{f_s}\right)$$

RAISED-COSINE FILTER

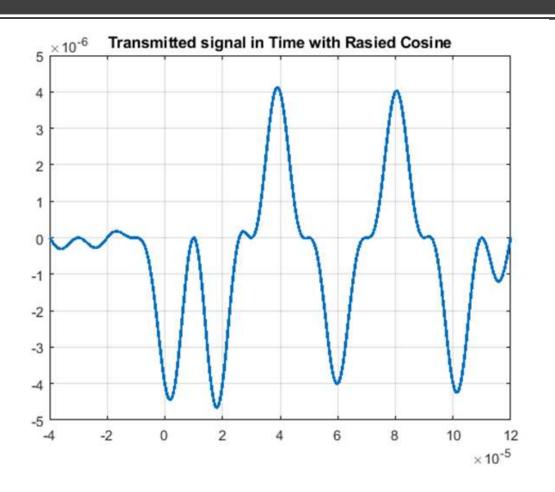
TIME DOMAIN

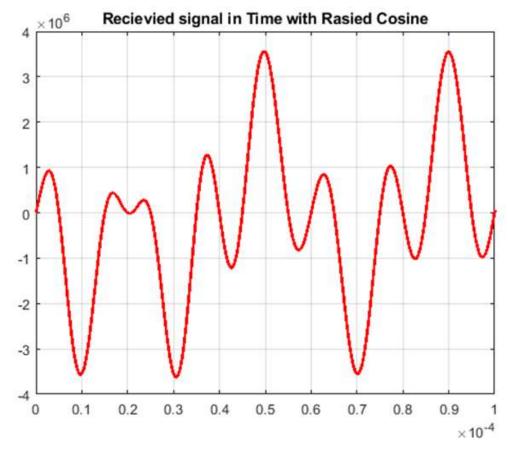
$$p(t) = \operatorname{sinc}(2Wt) \frac{\cos(2\pi\alpha Wt)}{1 - 16\alpha^2 W^2 t^2}$$

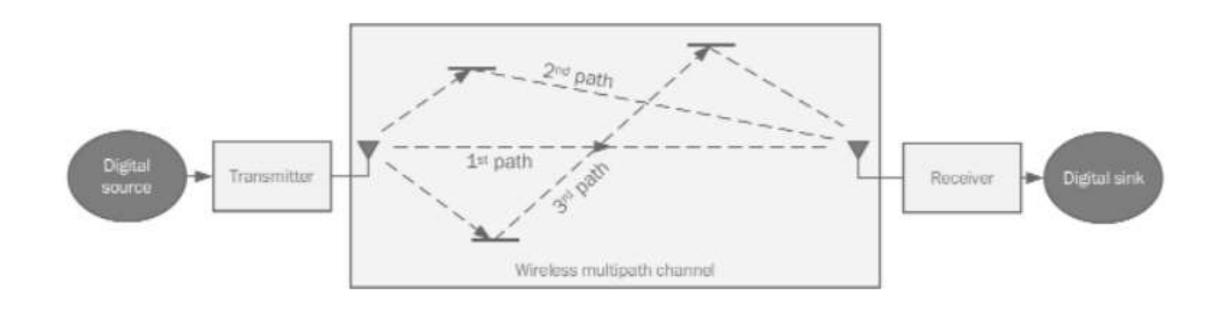
FREQUENCY DOMAIN

$$P(f) = \begin{cases} \frac{1}{2W}, & 0 \le |f| < f_1 \\ \frac{1}{4W} \left\{ 1 + \cos \left[\frac{\pi}{2W\alpha} (|f| - f_1) \right] \right\}, & f_1 \le |f| < 2W - f_1 \\ 0, & |f| \ge 2W - f_1 \end{cases}$$

RAISED-COSINE FILTER BEFORE AND AFTER PASSING THROUGH BAND-LIMITED CHANNEL





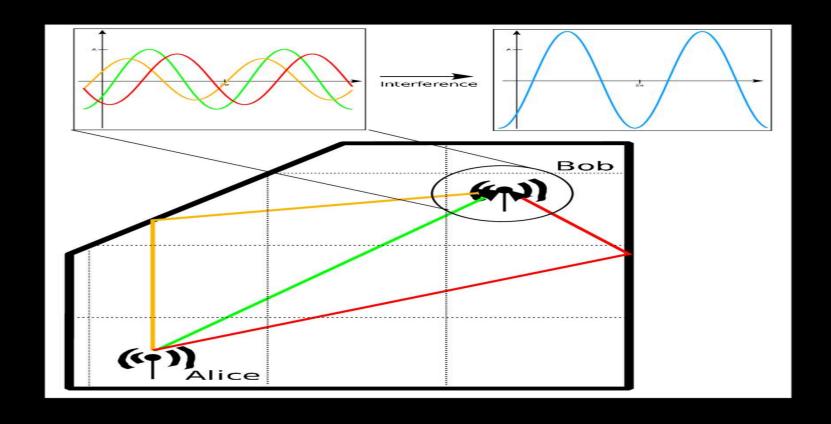


INTER-SYMBOL INTERFERENCE DUE TO MULTI-PATH CHANNELS

PART (2

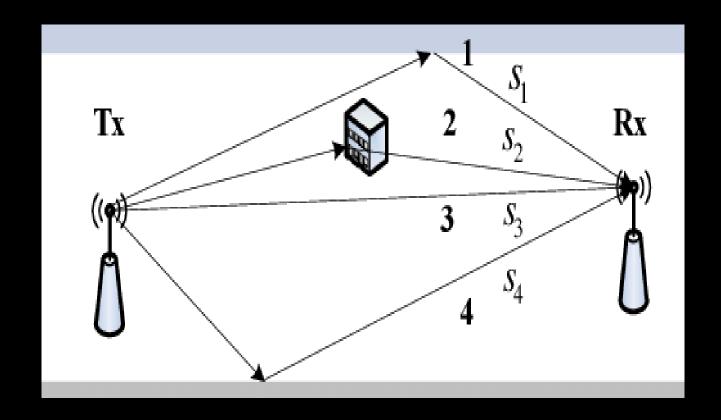
WHAT ARE MULTI-PATH CHANNELS?

- In wireless channels, signals are transmitted via electromagnetic waves which propagate through the air until it reaches the receiver.
- The nature of electromagnetic waves allow that multiple copies of the signal would travel around and reach the receiver at different times.
- A symbol transmitted by the transmitter would traverse multiple paths until it reaches the receiver.



WHAT ARE MULTI-PATH CHANNELS?

- Therefore, the receiver is expected to receiver multiple copies of the same transmitted signal.
- Each of these copies would arrive at a different time and with a different magnitude.
- The time is determined by how long the path is.
- The magnitude is determined by how much attenuation that the signal suffered from during the transmission across the path.



MATHEMATICAL POINT OF VIEW

- Let the first symbol transmitted by the transmitter be labelled as x[0].
- Let y[0] be the received signal.
- Let h0 be the channel effect of the first path on the transmitted signal x[0].
- Let n[0] be the noise component.

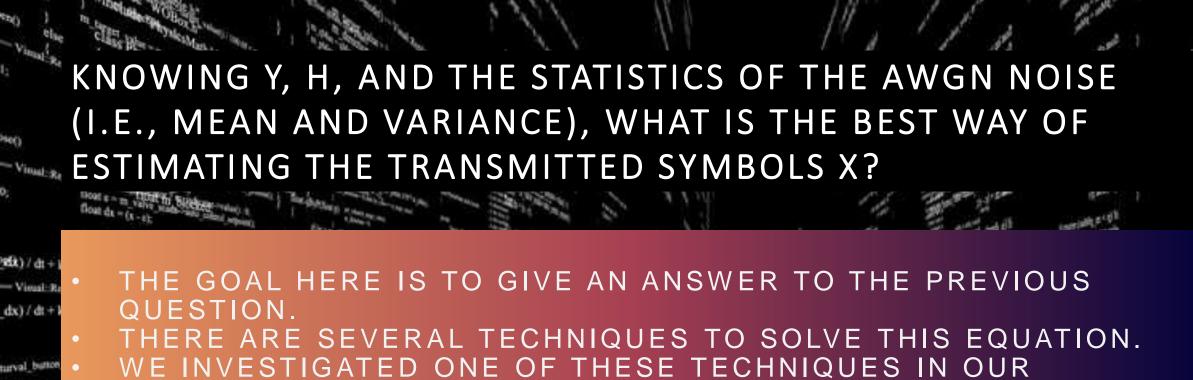
$$y[0] = h_0 x[0] + n[0]$$

$$y[1] = h_0x[1] + h_1x[0] + n[1]$$

$$y[L-1] = h_0 x[L-1] + h_1 x[L-2] + h_2 x[L-3] + \dots + h_{L-2} x[1] + h_{L-1} x[0] + n[L-1]$$

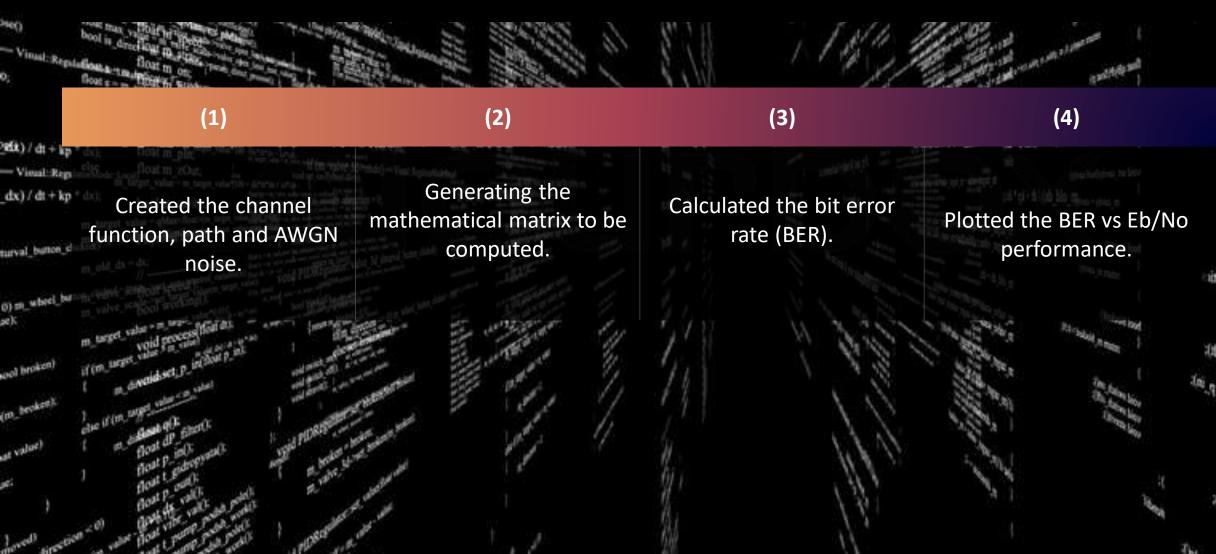
$$\begin{split} y[0] &= h_0 x[0] + n[0] \\ y[1] &= h_0 x[1] + h_1 x[0] + n[1] \\ y[2] &= h_0 x[2] + h_1 x[1] + h_2 x[0] + n[2] \\ \vdots \\ y[L-1] &= h_0 x[L-1] + h_1 x[L-2] + h_2 x[L-3] + \dots + h_{L-2} x[1] + h_{L-1} x[0] + n[L-1] \end{split}$$

$$\underbrace{ \begin{bmatrix} y[0] \\ y[1] \\ y[2] \\ \vdots \\ y[L-1] \end{bmatrix} }_{\hat{Y}} = \underbrace{ \begin{bmatrix} h_0 & & & & & \\ h_1 & h_0 & & & \\ h_2 & h_1 & h_0 & & \\ \vdots & & \ddots & h_1 & h_0 \\ \vdots & & \ddots & h_1 & h_0 \\ \vdots & & \ddots & h_2 & h_1 & h_0 \end{bmatrix} }_{\hat{X}} \underbrace{ \begin{bmatrix} x[0] \\ x[1] \\ x[2] \\ \vdots \\ x[L-3] \\ x[L-2] \\ x[L-1] \end{bmatrix} }_{\hat{X}} + \underbrace{ \begin{bmatrix} n[0] \\ n[1] \\ n[2] \\ \vdots \\ n[L-1] \end{bmatrix} }_{\hat{N}}$$



FOLLOWING TRIAL.





cavity;

CREATED THE CHANNEL FUNCTION, PATH AND AWGN NOISE

```
%Transmitted signal is x
% received signal is y
% N is the awgn noise that corrupts y
% our function is Y = HX+N where H(i) is the channel effect of the i+l
% channel and is represented by a 2D matrix of dimension L*L
%initializing variables
Eb = 1;
%dimension of matrix, could be changed, I used 5 for debugging
L = 50;
H = zeros(L,L);
$channel function that we're going to use $ e^-0.5*x^2
channel function = exp(-0.5*[0:L-1].^2);
%channel function = repmat(channel function, 1, 1);
%creating an array of normally distributed numbers, mean =0, variance =1
h = randn(L, 1);
%i added the other randn as the graph looked weird
% awgn noise 11 values
No = [0 0.001 0.005 0.01 0.05 0.1 0.25 0.5 0.7 0.9 1];
% initializing empty array to store the BER values inside the loop for plotting
temporary BER = [];
% initializing empty array for BER/noise
BER = [];
h = abs(h).*channel function;
```

GENERATING THE MATHEMATICAL MATRIX TO BE COMPUTED

```
% initializing empty array for BER/noise
 BER = [];
 h = abs(h).*channel_function;
 %filling matrix
 i=1;
 j = 1;
- for k = 1:L
    for m = i:-1:1
         H(k,j) = h(m);
         j = j + 1;
     end
     j = 1;
     i = i + 1;
 end
 %disp(h)
 %disp(H)
 %inverse H = inv(H); %inversing the matrix
```

CALCULATED THE BIT ERROR RATE (BER).

```
- for a = No
     %calculating noise power
     N = \operatorname{sqrt}(a/2) \operatorname{*randn}(L, 1);
     %calculating the BER 11 times for each noise
     for i = 1:11
         %returning the non zero values with no repitition
          non zero values = setdiff(-1:1, 0);
         x = non_zero_values( randi(length(non_zero_values), L, 1) );
          %recieved bits/ signal
         v = H*x' + N;
          %recieved bits plus noise
          x received =inv(H) *y;
          D = zeros(size(x received));
         for k = 1:L
             if x received(k) <= 0
          %polar type, if it's less than zero it's -1, if it's more it's -1
                  D(k) = -1;
              else
                  D(k) = 1;
              end
          end
          n = 0;
          for k = 1:L
              %checking with initial x to calculate BER
              if D(k) ~= x(k)
                  n = n + 1;
              end
          temporary BER = [temporary BER n/L];
      end
     %calculating the mean BER for each noise
     BER = [BER mean(temporary BER)];
     %reseting the array for the new noise
     temporary BER = [];
                                                                               25
```

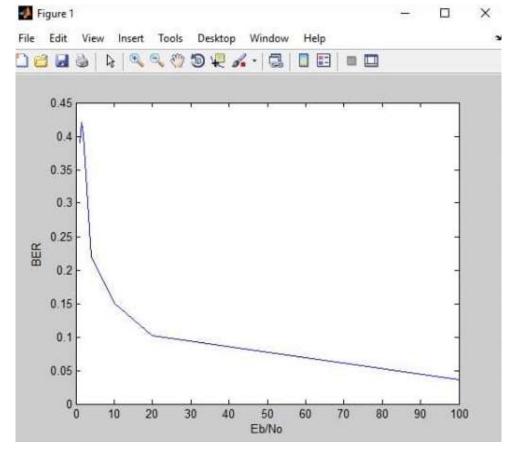
PLOTTED THE BER VS EB/NO PERFORMANCE.

```
plot(Eb./No, BER)

xlabel('Eb/No')

ylabel('BER')

xlim([0,100])
```



THANK YOU

Team Members

Mennatallah Moustafa 6234

Farah Ahmed 6274

Alaa Mohamed Abdel Hamid 6473

Rawan Hindawy 6491

Nouran Hisham 6532

Kareem Sabra 6594

Esraa Mahmoud 6597

Nourhan Waleed 6609

Seif Mohamed 6624

Fatema Moharam 6655