

Part 1.1 Matlab Code:

```
%1.1 DTMF Transmitter
phonenumber = [0 5 3 1 6 7 3 8 4 1 9];
x = DTMFTRA(phonenumber);
soundsc(x,8192);

function [x] = DTMFTRA (number)
    %matrix of frequencies 1,2,3...9,0
    freqs =
    [697,1209;697,1336;697,1477;770,1209;770,1336;770,1477;852,1209;
    852,1336;852,1477;941,1336];
    x = [] %creating an empty array
    for index = 1:length(number)
        starttime = (index-1)*0.25;
        t = [starttime:1/8192:starttime+0.25]%time interval with
1/8192 sample freq
        digit = number(index); %digit needs to be transmitted
        if digit == 0
            a=freqs(10,1);b=freqs(10,2);
        else
            a=freqs(digit,1);b=freqs(digit,2);
        end
        xc = cos(2*pi*a*t) + cos(2*pi*b*t); %create array in the
given time interval
        x = cat(2,x,xc(2:length(xc)));%add the array for given
time interval to final array
    end
end
```

LAB 03Mehmet Berk Şahin
21703190Part 11.1 DTMF Transmitter:

Yes, it is what I hear ^{when} I dial my phone number in analog phones. Since every number is transmitted in different frequencies, as the transmitted digit changes, the sound I hear changes as well.

1.2 DTMF Receiver:

$$a) x(t) = e^{j(2\pi f_0 t + \omega_0 t)}$$

$$X(j\omega) = \int_{-\infty}^{\infty} e^{j(2\pi f_0 t)} e^{j\omega_0 t} e^{-j\omega t} dt = e^{j\omega_0 t} \int_{-\infty}^{\infty} e^{j(2\pi f_0 - \omega)t} dt = \boxed{e^{j\omega_0 t} 2\pi \delta(\omega - \omega_0)}$$

$$b) x(t) = \cos(2\pi f_0 t) + \sin(2\pi f_0 t) \xrightarrow{FS} 2\pi \delta(\omega - \omega_0)$$

$$= \frac{e^{j\omega_0 t} + e^{-j\omega_0 t}}{2} + \frac{e^{j\omega_0 t} - e^{-j\omega_0 t}}{j2} = \frac{e^{j\omega_0 t} + e^{-j\omega_0 t}}{2} + j \left(\frac{e^{-j\omega_0 t} - e^{j\omega_0 t}}{2} \right)$$

$$= \left(\frac{1-j}{2} \right) e^{j\omega_0 t} + \left(\frac{1+j}{2} \right) e^{-j\omega_0 t} \xrightarrow{FS} \frac{(1-j)}{2} \delta(\omega - \omega_0) + \frac{(1+j)}{2} \delta(\omega + \omega_0)$$

$$= \boxed{\pi (\delta(\omega - \omega_0) - \delta(\omega + \omega_0)) + j\pi (\delta(\omega + \omega_0) - \delta(\omega - \omega_0))}$$

$$c) x(t) = \text{rect}\left(\frac{t}{T_0}\right) \rightarrow \begin{array}{c} \uparrow \\ \text{rect} \\ \downarrow \\ -T_0/2 \quad T_0/2 \end{array}$$

$$X(j\omega) = \int_{-T_0/2}^{T_0/2} 1 \cdot e^{-j\omega t} dt = \frac{e^{-j\omega t}}{-j\omega} \Big|_{-T_0/2}^{T_0/2} = \frac{1}{-j\omega} \left(e^{-j\omega T_0/2} - e^{j\omega T_0/2} \right) = \frac{-j2\sin\left(\frac{\omega T_0}{2}\right)}{-j\omega} = \boxed{\frac{2}{\omega} \sin\left(\frac{\omega T_0}{2}\right)}$$

$$d) x(t) = e^{j\omega_0 t} \text{rect}\left(\frac{t}{T_0}\right) \Rightarrow \boxed{X(j\omega) = \frac{2}{\omega - \omega_0} \sin\left(\frac{(\omega - \omega_0)T_0}{2}\right)}$$

$\left[2\pi f_0 = \omega_0 \right]$ $\frac{2}{\omega} \sin\left(\frac{\omega T_0}{2}\right)$ \uparrow Frequency shifting property

$$\text{we know: } \text{rect}\left(\frac{t}{T_0}\right) \xrightarrow{FT} \frac{2}{\omega} \sin\left(\frac{\omega T_0}{2}\right) \quad \frac{1}{2} \left(\frac{2}{\omega_{\text{mod}}}\right) \sin\left(\frac{(\omega_0 + \omega_{\text{mod}}) T_0}{2}\right) \quad \frac{1}{2} \left(\frac{2}{\omega_0 + \omega_{\text{mod}}}\right) \sin\left(\frac{(\omega_0 - \omega_{\text{mod}}) T_0}{2}\right)$$

Note: Frequency Shifting Property is used.

Time
switch

Time Shifting
and
Frequency Shifting!

$$\sin(2\omega t) = \frac{e^{j\omega t} - e^{-j\omega t}}{j2}$$

$$X(j\omega) = \frac{e^{-j\omega t_0}}{j} \left[\frac{\sin\left(\frac{(\omega - \omega_0)T_0}{2}\right)}{(\omega - \omega_0)} - \frac{\sin\left(\frac{(\omega + \omega_0)T_0}{2}\right)}{(\omega + \omega_0)} \right]$$

Peaks: $\frac{2\pi}{T} = \omega \Rightarrow f = \frac{\omega}{2\pi}$

x	Conversion to Hz	$f(Hz)$
± 9290	\rightarrow	1477
± 8396	\rightarrow	1336
± 7596	\rightarrow	1209
± 5912	\rightarrow	941
± 5350	\rightarrow	852
± 6377	\rightarrow	677

• These are the frequencies of the number used by the DTMF transmitters.

• Only from this figure, it cannot be understood which number is dialed, because they are not in order, just their frequencies are seen.

Let $r(t)$ be the rectangular signal such that

$$r(t) = \begin{cases} 1 & \text{for } 0 \leq t < 0.5 \\ 0 & \text{otherwise} \end{cases}$$

so $r(t) \cdot x(t) = x_1(t)$

Peaks for x_1 :

x	$f(Hz)$
± 8396	\rightarrow 1336
± 5912	\rightarrow 941

} \Rightarrow First Digit is 0

Peaks for x_2 :

x	$f(Hz)$
± 9290	\rightarrow 1477
± 5350	\rightarrow 852

} \Rightarrow Second Digit is 9

Peaks for x_3 :

x	$f(Hz)$
± 7596	\rightarrow 1209
± 6377	\rightarrow 677

} \Rightarrow Third Digit is 1

Peaks for x_4 :

x	$f(Hz)$
± 9290	\rightarrow 1477
± 6377	\rightarrow 677

} \Rightarrow Fourth Digit is 3

• This time I can understand which digit is dialed first, second, third and the last.

\rightarrow I shift the square wave $r(t)$ right at each step so that I only take one digit of the Fourier Transform $X(f)$ of the whole signal.

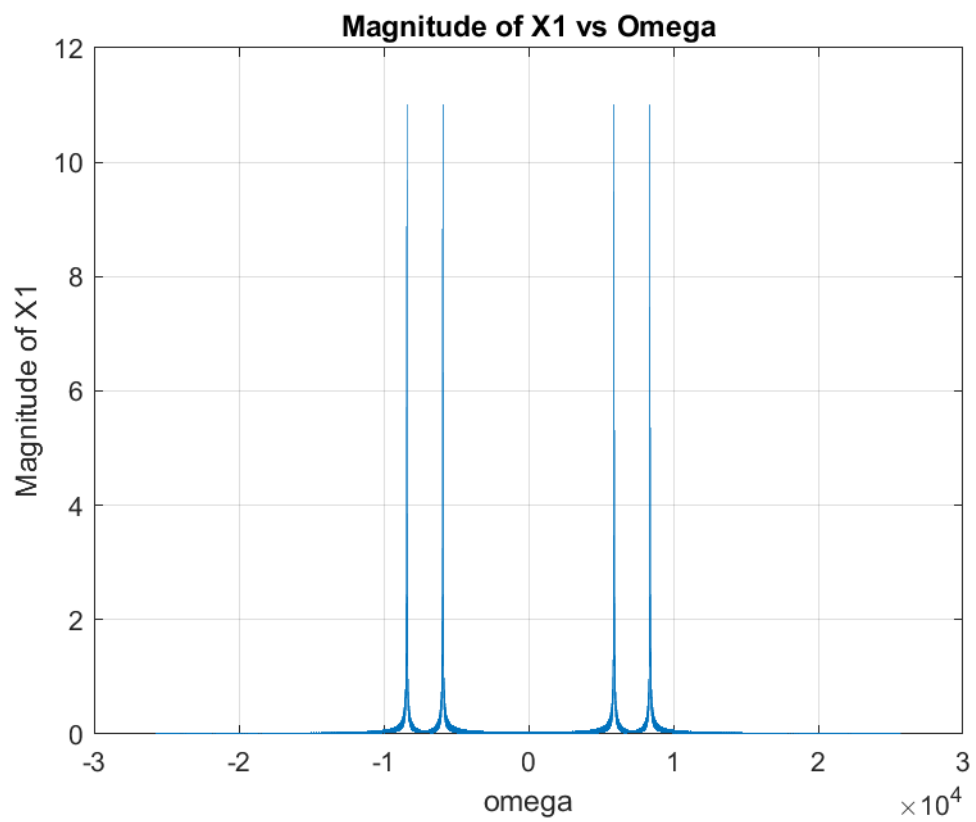
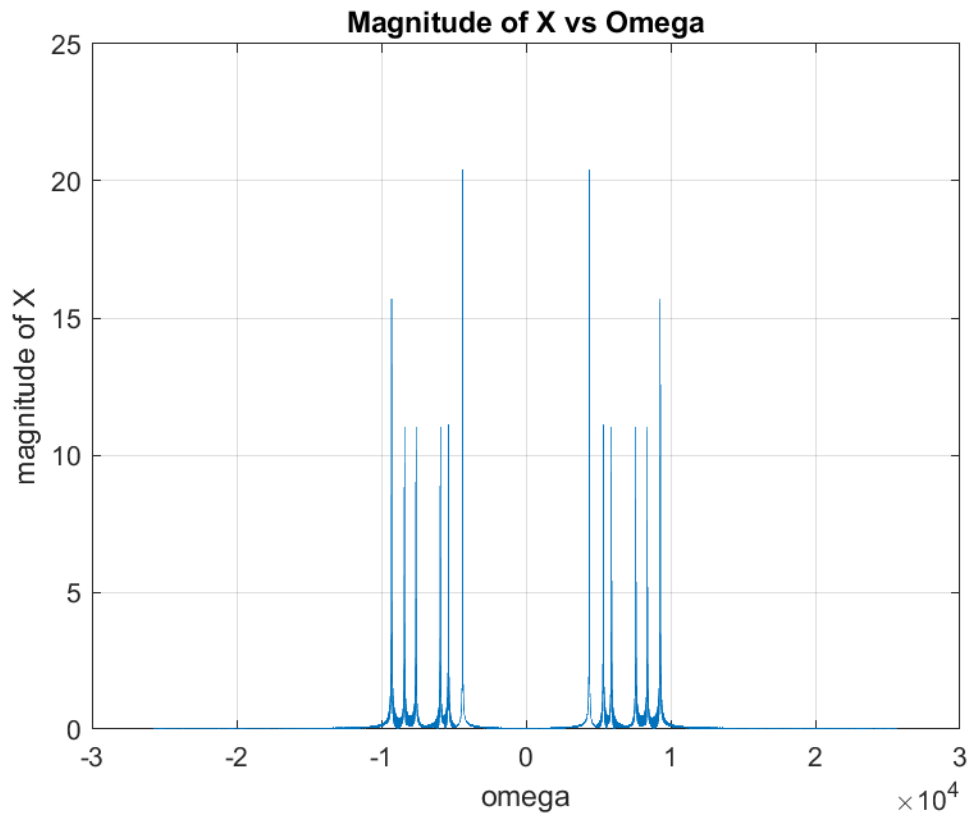
\rightarrow Since $X(\omega) = x_1(\omega) + x_2(\omega) + x_3(\omega) + x_4(\omega)$, when I looked at $X(\omega)$, all of the Fourier Transform of digit signals are inside X so they can not be distinguished as a digit. However, since x_1, x_2, x_3 and x_4 includes only two different frequencies, I was able to see the frequencies and decide which digit is dialed at the ^{sample} given t interval which is determined by $r(t)$.

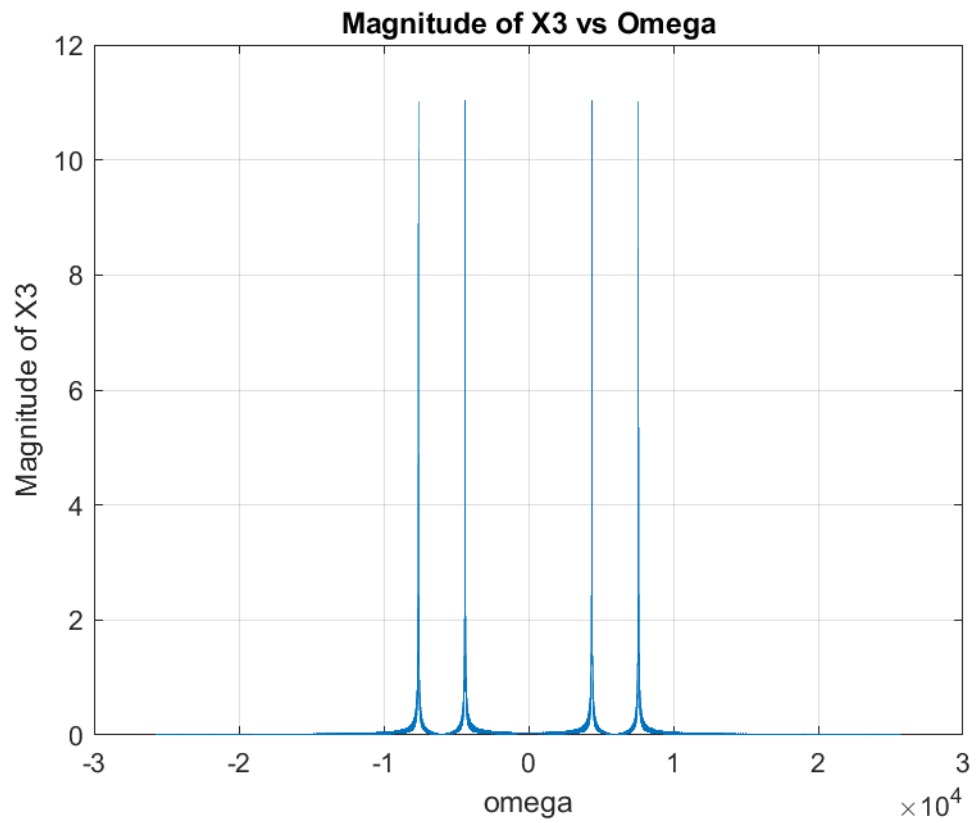
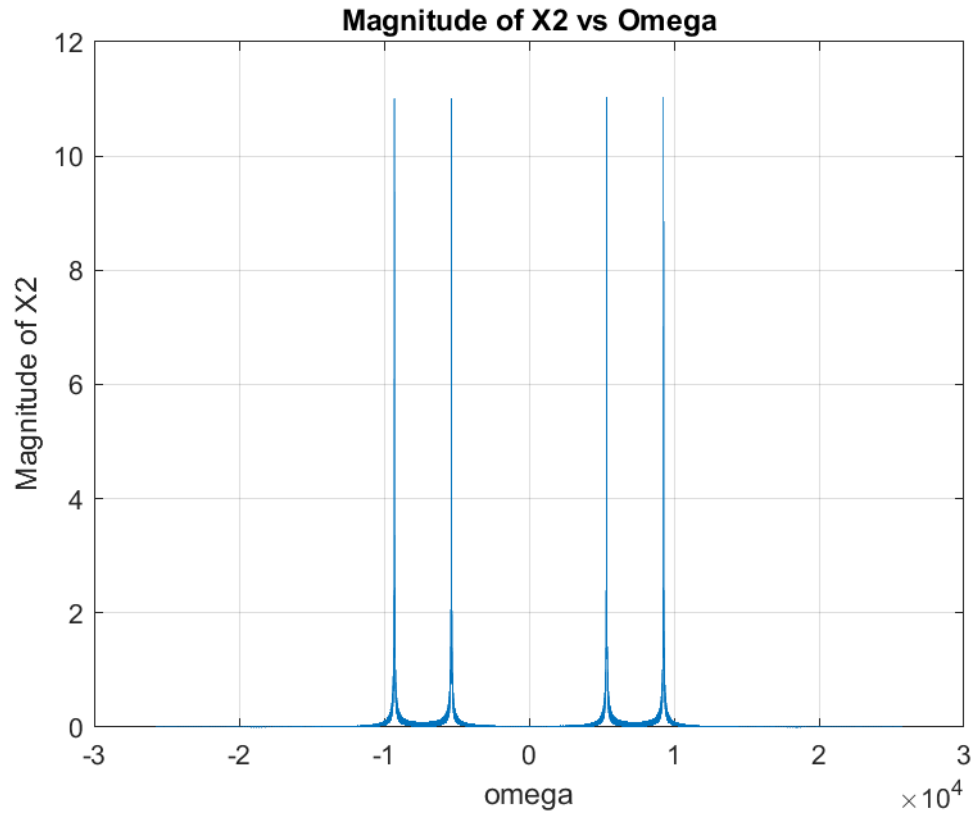
Part 1.2:

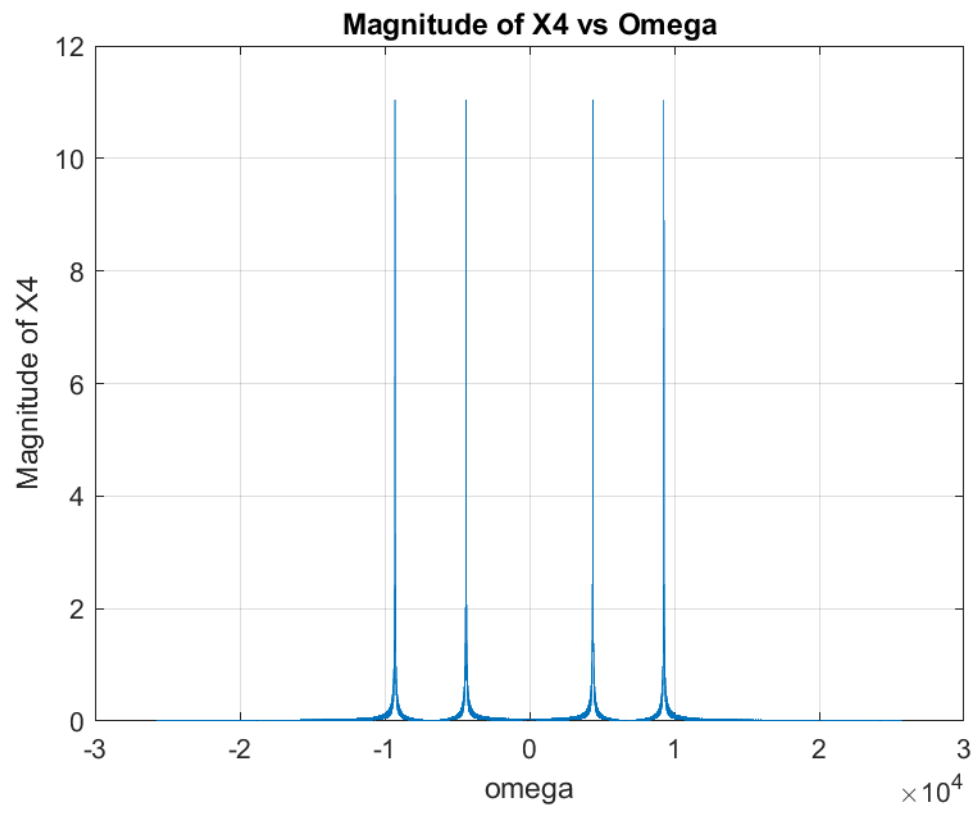
```
% %1.2 DTMF Receiver
Number = [0 9 1 3]; %ID: 21703190
x = DTMFTRA(Number);
orgx = x;
soundsc(x,8192);

%-----first part of 1.2 DTMF Receiver-----
X = FT(x);
omega=linspace(-8192*pi,8192*pi,8193);
omega=omega(1:8192);
figure;
plot(omega,abs(X));
xlabel('omega');
ylabel('Magnitude of X');
title('Magnitude of X vs Omega');
grid ON;

%-----second part of 1.2 DTMF Receiver-----
%----First Digit-----
figure
r = [ones(1,2048),zeros(1,6144)];
x=x.*r;
X1 = FT(x);
plot(omega,abs(X1));
xlabel('omega');
ylabel('Magnitude of X1');
title('Magnitude of X1 vs Omega');
grid ON;
% %----Second Digit-----
x = orgx;
figure
r = [zeros(1,2048),ones(1,2048),zeros(1,4096)];
x=x.*r;
X2 = FT(x);
plot(omega,abs(X2));
xlabel('omega');
ylabel('Magnitude of X2');
title('Magnitude of X2 vs Omega');
grid ON;
% %----Third Digit-----
x = orgx;
figure
r = [zeros(1,4096),ones(1,2048),zeros(1,2048)];
x=x.*r;
X3 = FT(x);
plot(omega,abs(X3));
xlabel('omega');
ylabel('Magnitude of X3');
title('Magnitude of X3 vs Omega');
grid ON;
% %----Fourth Digit-----
x = orgx;
figure
r = [zeros(1,6144),ones(1,2048)];
x=x.*r;
X4 = FT(x);
plot(omega,abs(X4));
xlabel('omega');
ylabel('Magnitude of X4');
title('Magnitude of X4 vs Omega');
grid ON;
```







Part 2:

my recording is "Hello guys, my name is Berk, my ID is 21703190 and this is signals and systems course, good luck!"

Matlab Code:

```
recObj = audiorecorder(8192, 8, 1);
disp("Start Recording...");
recordblocking(recObj, 10);
disp("Stop Recording...");
x = transpose(getaudiodata(recObj));
soundsc(x, 8192);
```

$$y(t) = x(t) + \sum_{i=1}^M A_i x(t - t_i)$$

$$a) \quad y(t) = x(t) * h(t) \xrightarrow{\mathcal{F}} Y(j\omega) = X(j\omega) H(j\omega) \Rightarrow H(j\omega) = \frac{Y(j\omega)}{X(j\omega)}$$

$$\Rightarrow y(t) \xrightarrow{\mathcal{F}} Y(j\omega) = X(j\omega) + \sum_{i=1}^M A_i e^{-j\omega t_i} X(j\omega)$$

$$\Rightarrow H(j\omega) = 1 + \sum_{i=1}^M A_i e^{-j\omega t_i} \quad \Rightarrow \quad \left. \begin{array}{l} \downarrow \mathcal{F}^{-1} \\ h(t) \end{array} \right\} \quad \left. \begin{array}{l} \downarrow \\ f(t - t_i) \end{array} \right\} \quad h(t) = f(t) + \sum_{i=1}^M A_i f(t - t_i)$$

$$b) \text{ for found in part a, } \boxed{H(j\omega) = 1 + \sum_{i=1}^M A_i e^{-j\omega t_i}}$$

$$c) \text{ Relation between } X(\omega), Y(\omega) \text{ and } H(\omega) \text{ is } \boxed{Y(\omega) = X(\omega) H(\omega)} \quad \text{+ convolution property}$$

$$d) \quad X(\omega) = \frac{Y(\omega)}{H(\omega)} \quad \text{by the convolution property!}$$

• The sound I hear is the sound I originally heard but with additional delayed echoes. There were 5 delayed signals with scaled amplitudes A_i . Sound was so crowded that after a while, I couldn't understand what I said.

• When I listen $x_c(t)$, there were still some delays but it was closer to the original record $x(t)$ than $y(t)$. I was able to understand what I said. $x_c(t)$ is different than $x(t)$, which is expected because $x_c(t)$ is estimation of $x(t)$. In fact, from the plots, how $x_c(t)$ reduce some of the noises (additional echoes) can be seen clearly.

Part 2 Matlab Code:

```
recObj = audiorecorder(8192,8,1);
disp("Start Recording...");
recordblocking(recObj,10);
disp("Stop Recording...");
x = transpose(getaudiodata(recObj));
%soundsc(x,8192); %listen the original sound

M=5;
Ai=[0.35 0.5 0.65 0.05 0.15];
time = [0:1/8192:10-1/8192];
ti=[0.5 1.25 2.5 3 2.75];
echo = zeros(1,81920);

for i = 1:M
    for t = 0:1/8192:10-1/8192
        if t>=ti(i) %cropping the signal
            echo(t*8192+1) = echo(t*8192+1) + Ai(i)*x((t-ti(i))*8192 +1); % summation operation
        end
    end
end
y = x + echo; %creating the final y(t) signal
%soundsc(y,8192); %listen echoed sound

%plots
figure
plot(time,x);
xlabel("time (second)");
ylabel("x(t)");
title("x(t) versus time");
figure
plot(time,y);
xlabel("time (seconds)");
ylabel("y(t)");
title("y(t) versus time");

Y=FT(y);
omega=linspace(-8192*pi,8192*pi,81921);
omega=omega(1:81920);
H=ones(1,81920);
for i = 1:M
    H = H + Ai(i)*exp(-1.0i*omega*ti(i)); %formula found in part b
end

h = IFT(H);
figure
plot(time,h);
title("h(t) versus time");
xlabel("time (second)");
ylabel("h(t)");
figure
plot(omega,abs(H));
title("|H(jw)| versus omega");
xlabel("omega (rad/s)");
ylabel("|H(jw)|");

Xe = Y./H; %estimated X(jw)
xe = IFT(Xe);
soundsc(xe,8192);%listen estimated x(t)

figure
plot(time,xe);
title("xe(t) versus time");
xlabel("time (second)");
ylabel("xe(t)");
```

