

Lab-6

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1)

a)

Here $p[n] = \cos((2\pi n f_0)/f_s)$

The DTFT of $p[n]$ consists of impulses at frequencies $+2\pi f_0/f_s$, $-2\pi f_0/f_s$ (due to the cos terms), these impulses are centered around the origin ($\omega=0$). The magnitudes of the impulses are determined by the amplitude of the cos wave.

Lab-6

1.
a) Given

$$P[n] = \cos\left(\frac{2\pi n f_0}{f_s}\right)$$
 DTFT of $P[n] = P(e^{j\omega})$

$$P(e^{j\omega}) = \sum_{n=-\infty}^{\infty} P[n] e^{-j\omega n}$$
 let $\frac{2\pi f_0}{f_s} = \alpha$

$$= \sum_{n=-\infty}^{\infty} \cos\left(\frac{2\pi n f_0}{f_s}\right) e^{-j\omega n}$$

$$= \sum_{n=-\infty}^{\infty} \cos(n\alpha) e^{-j\omega n} = \sum_{n=-\infty}^{\infty} \frac{e^{jn\alpha} + e^{-jn\alpha}}{2} e^{-j\omega n}$$

$$= \frac{1}{2} \left(\sum_{n=-\infty}^{\infty} e^{j(\alpha-\omega)n} + \sum_{n=-\infty}^{\infty} e^{-j(\alpha+\omega)n} \right)$$

$$= \frac{1}{2} \left(\sum_{n=0}^{\infty} e^{j(\alpha-\omega)n} + \sum_{n=1}^{\infty} e^{j(\omega-\alpha)n} \right) + \frac{1}{2} \left(\sum_{n=0}^{\infty} e^{-j(\alpha+\omega)n} + \sum_{n=1}^{\infty} e^{j(\alpha+\omega)n} \right)$$

$$= \frac{1}{2} \left(\frac{1}{1 - e^{j(\alpha-\omega)}} + \frac{e^{j(\omega-\alpha)}}{1 - e^{j(\omega-\alpha)}} \right) + \frac{1}{2} \left(\frac{1}{1 - e^{-j(\alpha+\omega)}} + \frac{e^{-j(\alpha+\omega)}}{1 - e^{-j(\alpha+\omega)}} \right)$$

$$= \frac{1}{2} \left(\frac{1 + e^{j(\omega-\alpha)}}{1 - e^{j(\omega-\alpha)}} \right) + \frac{1}{2} \left(\frac{-e^{j(\omega-\alpha)}}{1 - e^{j(\omega-\alpha)}} + \frac{e^{-j(\alpha+\omega)}}{1 - e^{-j(\alpha+\omega)}} \right)$$

$$= \frac{1}{2} \left(\frac{1 + e^{j(\omega-\alpha)}}{1 - e^{j(\omega-\alpha)}} \right) = \frac{1}{2} \left(\frac{e^{j\alpha} + e^{j\omega}}{e^{j\alpha} - e^{j\omega}} \right)$$

$$P(e^{j\omega}) = \frac{1}{2} \left(\frac{\frac{2\pi f_0}{f_s} + j\omega}{\frac{2\pi f_0}{f_s} - j\omega} \right)$$

*Here if we equate the $P(e^{j\omega})$ to zero then we get the $\omega = 2\pi f_0 / f_s$.

b) Relationship between impulses:

In the DTFT of $p[n]$, the impulses at $+2\pi f_0 / f_s$, $-2\pi f_0 / f_s$ are symmetric with respect to the origin this symmetry arises from the real values nature of the cos function. Mathematically, it means that $P(e^{j\omega})$ is conjugate symmetric, i.e.,

$$P(e^{j\omega}) = P^*(e^{-j\omega})$$

Where $P^*(e^{-j\omega})$ is the conjugate complex of the $P(e^{j\omega})$. This implies that the magnitude spectrum is even, and the phase spectrum is odd.

Here $w[n] = 1$ ($0 \leq n \leq L - 1$)

$w[n] = 0$ for all other values of n

$$x[n] = p[n] \cdot w[n]$$

c)

c) Given $x[n] = p[n] \times w[n]$ where $p[n] = \cos\left(\frac{2\pi n f_0}{f_s}\right)$ and $w[n] = \begin{cases} 1 & 0 \leq n \leq L-1 \\ 0 & \text{o.w} \end{cases}$

DTFT of $x[n]$ is $X(e^{j\omega})$

$$X(e^{j\omega}) = \text{DTFT}(p[n] \times w[n])$$

$$X(e^{j\omega}) = p(e^{j\omega}) * W(e^{j\omega})$$

$$\Rightarrow X(e^{j\omega}) = p(e^{j\omega}) * W(e^{j\omega})$$

$$\Rightarrow X(\omega) = p(\omega) * W(\omega)$$

$$= \sum_{k=-\infty}^{\infty} p(k) \cdot W(\omega - k)$$

$$= \sum_{k=-\infty}^{\infty} \frac{1}{2} \left(\frac{e^{\frac{2\pi j f_0}{f_s} k} + e^{-\frac{2\pi j f_0}{f_s} k}}{e^{\frac{2\pi j f_0}{f_s} k} - e^{-\frac{2\pi j f_0}{f_s} k}} \right) \left(\frac{1 - e^{-jL(\omega - k)}}{1 - e^{-j(\omega - k)}} \right)$$

$$= \sum_{n=0}^{L-1} 1 \cdot e^{-j\omega n}$$

$$= \frac{1 - e^{-j\omega L}}{1 - e^{-j\omega}}$$

$$X(\omega) = \sum_{k=-\infty}^{\infty} \frac{1}{2} \left(\frac{e^{\frac{2\pi j f_0}{f_s} k} + e^{-\frac{2\pi j f_0}{f_s} k}}{e^{\frac{2\pi j f_0}{f_s} k} - e^{-\frac{2\pi j f_0}{f_s} k}} \right) \left(\frac{1 - e^{-jL(\omega - k)}}{1 - e^{-j(\omega - k)}} \right)$$

If we window the signal $p[n]$ using a rectangular window $w[n]$, given as

$$x[n] = p[n] \cdot w[n]$$

here we are using the multiplication property of DTFT:

$$X(e^{j\omega}) = P(e^{j\omega}) * W(e^{j\omega})$$

Where

$$p[n] = \cos\left(\frac{2\pi n f_0}{f_s}\right)$$

and

$W(e^{j\omega})$ is the DTFT of the rectangular window $w[n]$.

The effect of the window on the spectrum is to convolve $P(e^{j\omega})$ with $W(e^{j\omega})$, resulting in a change in the spectral shape and sidelobe levels. The mainlobe width will also be affected.

d)

here $f_0=12\text{Hz}$

$f_s=64\text{Hz}$

$L=16$

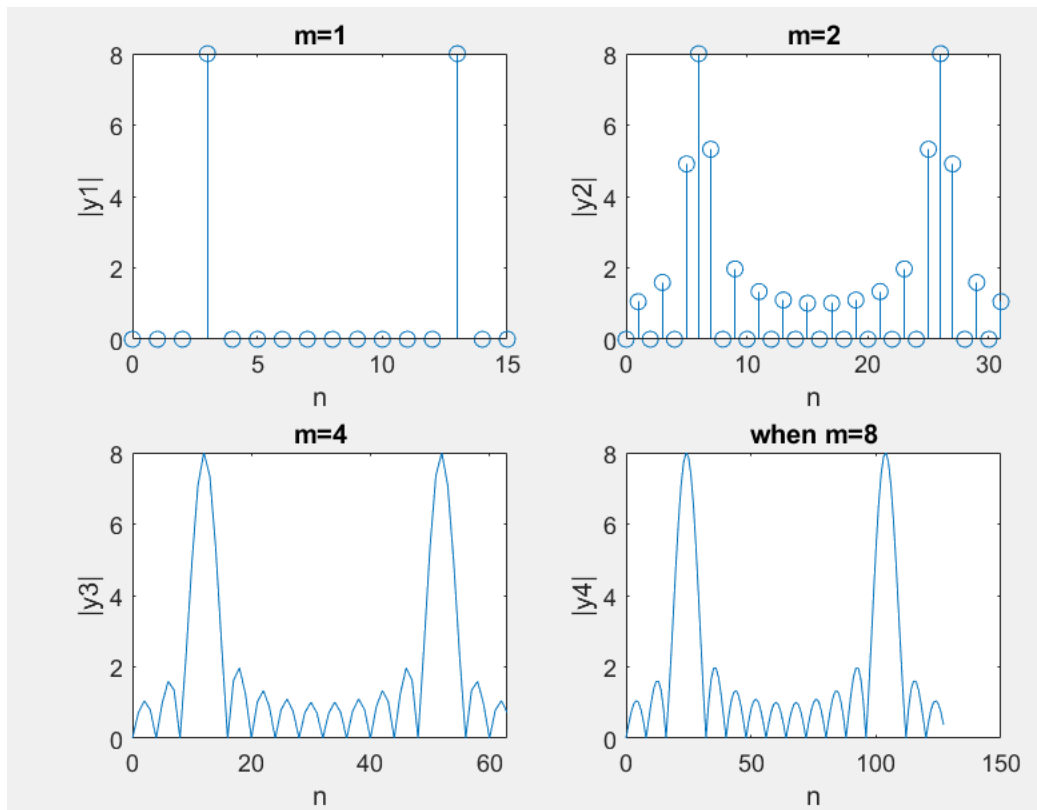
We compute the DFT $X[k]$ for the different values of $N=m*L$ where $m=\{1,2,4,8\}$

The `fft` automatically performs zero padding if $L < N$ so here we are using the `fft` command .

CODE:

```
1  f0=12;
2  fs=64;
3  L=16;
4  n1=0:15;
5  n2=0:31;
6  n3=0:63;
7  n4=0:127;
8  pn1=cos((2*pi*f0*n1)/fs);
9  pn2=cos((2*pi*f0*n2)/fs);
10 pn3=cos((2*pi*f0*n3)/fs);
11 pn4=cos((2*pi*f0*n4)/fs);
12 wn1=(0<=n1 & n1<=15);
13 wn2=(0<=n2 & n2<=15);
14 wn3=(0<=n3 & n3<=15);
15 wn4=(0<=n4 & n4<=15);
16 xn1=pn1.*wn1;
17 xn2=pn2.*wn2;
18 xn3=pn3.*wn3;
19 xn4=pn4.*wn4;
20 y1=fft(xn1,16);
21 y2=fft(xn2,32);
22 y3=fft(xn3,64);
23 y4=fft(xn4,128);
24 figure
25 subplot(2,2,1)
26 stem(n1,abs(y1));
27 title('m=1')
28 xlabel('n');
29 ylabel('|y1|');
30 subplot(2,2,2)
31 stem(n2,abs(y2));
32 title('m=2')
33 xlabel('n');
34 ylabel('|y2|');
35 subplot(2,2,3)
36 plot(n3,abs(y3));
37 title('m=4')
38 xlabel('n');
39 ylabel('|y3|');
40 subplot(2,2,4)
41 plot(n4,abs(y4));
42 title('when m=8')
43 xlabel('n');
44 ylabel('|y4|')
```

Plot:



Mainlobe width:

Compare the width of the mainlobe in the DFT plots here and with the part c .The main lobe width depends on the DFT length N.

Spectral Leakage:

Spectral leakage occurs when energy from one frequency component spreads to neighbouring bins .

So the plots are not consistent with the part c answers (we can say this from the Mainlobe width and spectral leakage concepts).

e)

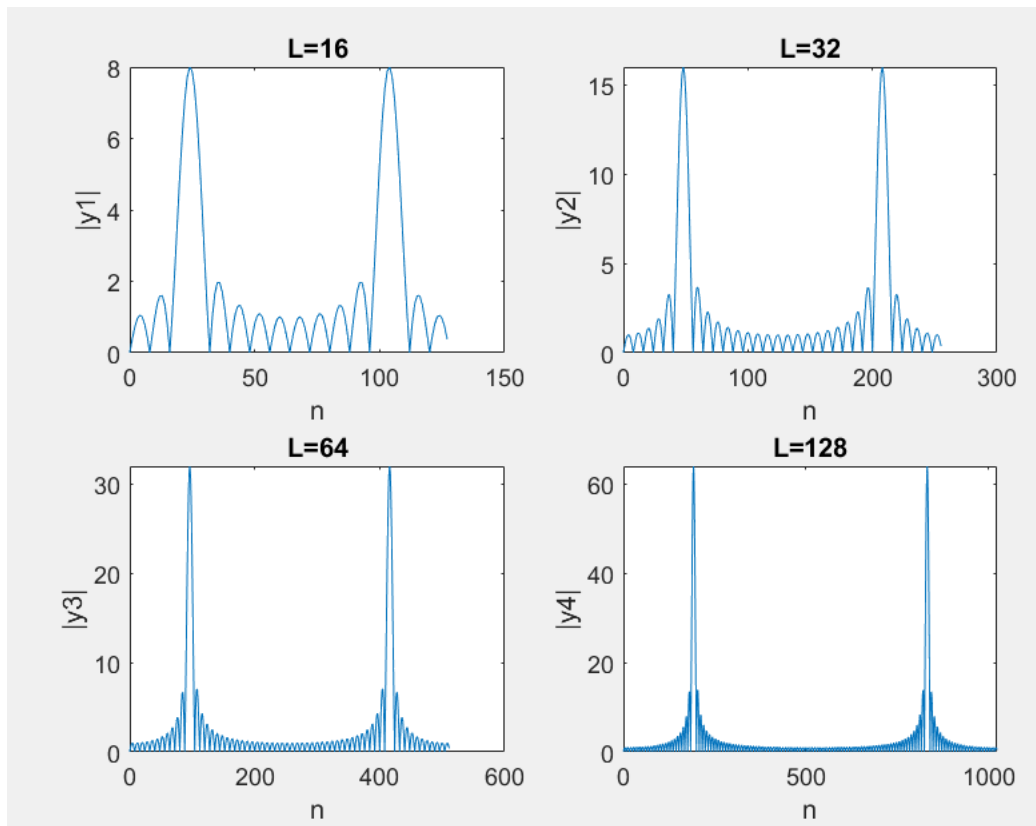
CODE:

```

1      fo=12;
2      fs=64;
3      n1=0:127;
4      n2=0:255;
5      n3=0:511;
6      n4=0:1023;
7      pn1=cos((2*pi*fo*n1)/fs);
8      pn2=cos((2*pi*fo*n2)/fs);
9      pn3=cos((2*pi*fo*n3)/fs);
10     pn4=cos((2*pi*fo*n4)/fs);
11     wn1=(0<=n1 & n1<=15);
12     wn2=(0<=n2 & n2<=31);
13     wn3=(0<=n3 & n3<=63);
14     wn4=(0<=n4 & n4<=127);
15     xn1=pn1.*wn1;
16     xn2=pn2.*wn2;
17     xn3=pn3.*wn3;
18     xn4=pn4.*wn4;
19     y1=fft(xn1,128);
20     y2=fft(xn2,256);
21     y3=fft(xn3,512);
22     y4=fft(xn4,1024);
23     figure
24     subplot(2,2,1)
25     plot(n1,abs(y1));
26     title('L=16')
27     xlabel('n');
28     ylabel('|y1|');
29     subplot(2,2,2)
30     plot(n2,abs(y2));
31     title('L=32')
32     xlabel('n');
33     ylabel('|y2|');
34     subplot(2,2,3)
35     plot(n3,abs(y3));
36     title('L=64')
37     xlabel('n');
38     ylabel('|y3|');
39     subplot(2,2,4)
40     plot(n4,abs(y4));
41     title('L=128')
42     xlabel('n');
43     ylabel('|y4|');

```

PLOT:



Observation:

From the plots we can say that the Mainlobe width decreases and leading to better Frequency Resolution with the increase of L .

f)

CODE:

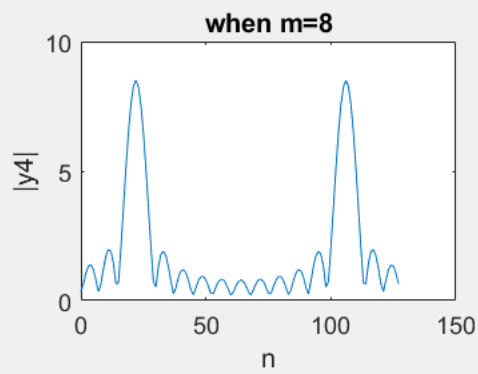
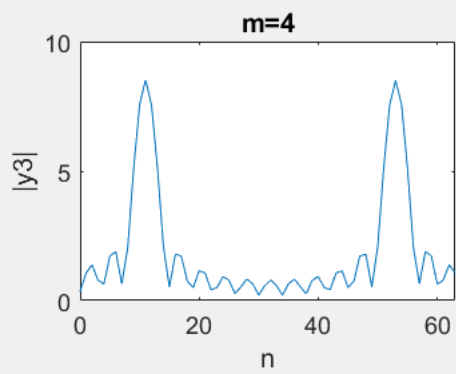
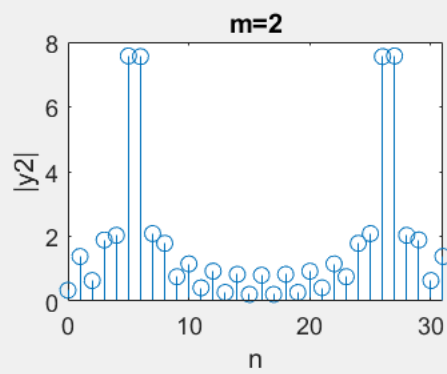
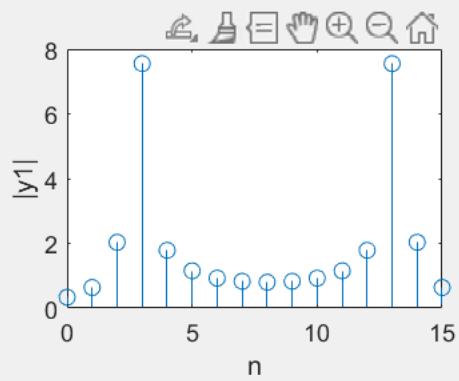
```

1     fo=11;|
2     fs=64;
3     L=16;
4     n1=0:15;
5     n2=0:31;
6     n3=0:63;
7     n4=0:127;
8     pn1=cos((2*pi*fo*n1)/fs);
9     pn2=cos((2*pi*fo*n2)/fs);
10    pn3=cos((2*pi*fo*n3)/fs);
11    pn4=cos((2*pi*fo*n4)/fs);
12    wn1=(0<=n1 & n1<=15);
13    wn2=(0<=n2 & n2<=15);
14    wn3=(0<=n3 & n3<=15);
15    wn4=(0<=n4 & n4<=15);
16    xn1=pn1.*wn1;
17    xn2=pn2.*wn2;
18    xn3=pn3.*wn3;
19    xn4=pn4.*wn4;
20    y1=fft(xn1,16);
21    y2=fft(xn2,32);
22    y3=fft(xn3,64);
23    y4=fft(xn4,128);
24    figure
25    subplot(2,2,1)
26    stem(n1,abs(y1));
27    title('m=1')
28    xlabel('n');
29    ylabel('|y1|');
30    subplot(2,2,2)
31    stem(n2,abs(y2));
32    title('m=2')
33    xlabel('n');
34    ylabel('|y2|');
35    subplot(2,2,3)
36    plot(n3,abs(y3));
37    title('m=4')
38    xlabel('n');
39    ylabel('|y3|');

```

PLOT:

when $f_0=11\text{Hz}$



g)

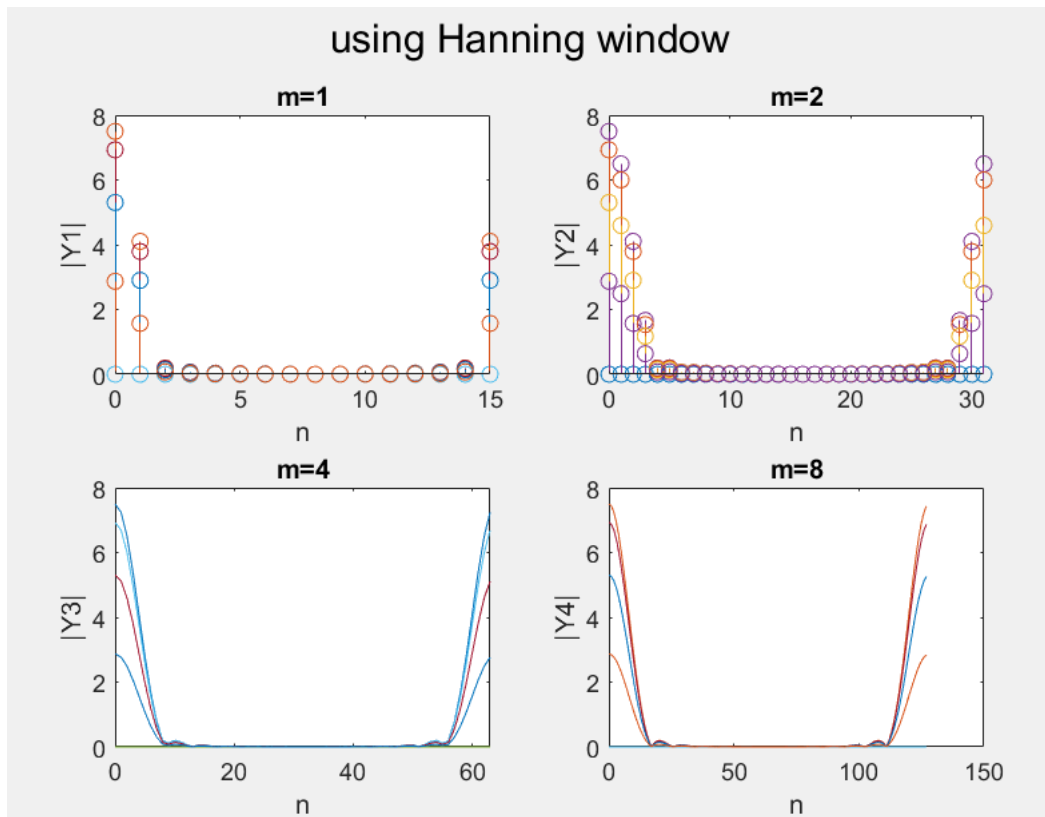
code:

```

1    fo=12;
2    fs=64;
3    L=16;
4    n1=0:15;
5    n2=0:31;
6    n3=0:63;
7    n4=0:127;
8    pn1=cos((2*pi*fo*n1)/fs);
9    pn2=cos((2*pi*fo*n2)/fs);
10   pn3=cos((2*pi*fo*n3)/fs);
11   pn4=cos((2*pi*fo*n4)/fs);
12   wn1=hann(L);
13   wn2=hann(L);
14   wn3=hann(L);
15   wn4=hann(L);
16   xn1=pn1.*wn1;
17   xn2=pn2.*wn2;
18   xn3=pn3.*wn3;
19   xn4=pn4.*wn4;
20
21   y1=fft(xn1, 16);
22   y2=fft(xn2, 32);
23   y3=fft(xn3, 64);
24   y4=fft(xn4, 128);
25
26   figure;
27
28   subplot(2, 2, 1)
29   stem(n1, abs(y1));
30   title("m=1");
31   xlabel('n');
32   ylabel(' |Y1| ');
33
34   subplot(2, 2, 2)
35   stem(n2, abs(y2));
36   title("m=2");
37   xlabel('n');
38   ylabel(' |Y2| ');
39
40   subplot(2, 2, 3)
41   plot(n3, abs(y3));
42   title("m=4");
43   xlabel('n');
44   ylabel(' |Y3| ');
45
46   subplot(2, 2, 4)
47   plot(n4, abs(y4));
48   title("m=8");
49   xlabel('n');
50   ylabel(' |Y4| ');
51
52   sgtitle("using Hanning window")

```

plot:



Observation:

The Hanning window has a different spectral response compared to the rectangular window , affecting the mainlobe width and sidelobe levels .

Mainlobe width:

When we are using the Hanning window , the main lobe of the DFT magnitude spectrum will be narrower compared to the main lobe width observed in the part d with a rectangular window.

* The Hanning window results in better frequency localization , and the main lobe is more concentrated around the true frequency f_0

Spectral Leakage:

*spectral leakage is reduced when a Hanning window is applied . The sidelobes in the DFT magnitude spectrum are lower in magnitude and spread over a narrower range of frequencies.

*compared to the rectangular window in part(d) , the Hanning window minimizes the spreading of energy from the mainlobe to adjacent frequency bins.

=> The Hanning window improves the spectral analysis by providing the better frequency resolution and reduced spectral leakage .

=> The main-lobe width becomes narrower , which may limit the ability to distinguish closely spaced frequency components.

h)

part d:

code:

```
1  f0 = 12;
2  fs = 64;
3  L = 16; |
4  m_values = [1, 2, 4, 8];
5  f0_estimates = zeros(1, length(m_values));
6  n = 0:15;
7
8  for i = 1:length(m_values)
9      m = m_values(i);
10     N = m * L;
11     pn = cos((2 * pi * f0 * n)/fs );
12     wn=(0<=n & n<=15);
13     xn=pn.*wn;
14     X = fft(xn, N);
15     f = (0:N-1) * fs / N;
16     [~, index] = max(abs(X));
17     f0_estimates(i) = f(index);
18 end
19
20 f0_estimates
```

Output:

```
12    12    12    12
```

*In this case , the choice of N does not significantly affect the f_0 because there is minimal spectral leakage due to the rectangular window.

The peaks are appeared to be around $f_0=12\text{Hz}$ for all N values.

Part-f:

Code:

```

1   f0 = 11;
2   fs = 64;
3   L = 16;
4   m_values = [1, 2, 4, 8];
5   f0_estimates = zeros(1, length(m_values));
6   n = 0:15;
7
8   for i = 1:length(m_values)
9       m = m_values(i);
10      N = m * L;
11      pn = cos((2 * pi * f0 * n)/fs );
12      wn=(0<=n & n<=15);
13      xn=pn.*wn;
14      X = fft(xn, N);
15      f = (0:N-1) * fs / N;
16      [~, index] = max(abs(X));
17      f0_estimates(i) = f(index);
18   end
19
20   f0_estimates

```

Output:

```

12    10    11    11

```

Here the choice of N does not affect f_0 significantly due to minimal spectral leakage.

So the peaks appeared are around 11Hz.

Part-g:

```

1   f0 = 12;
2   fs = 64;
3   L = 16;
4   m_values = [1, 2, 4, 8];
5   f0_estimates = cell(1, length(m_values));
6
7   n = 0:15;
8
9   for i = 1:length(m_values)
10      m = m_values(i);
11      N = m * L;
12      pn = cos((2 * pi * f0 * n) / fs);
13      wn = hann(L);
14      xn = pn .* wn;
15      X = fft(xn, N);
16      f = (0:N-1) * fs / N;
17      [~, index] = max(abs(X));
18      f0_estimates{i} = f(index);
19   end
20   for i = 1:length(m_values)
21       disp(['m = ' num2str(m_values(i)) ', Estimated f0 = ' num2str(f0_estimates{i}) ' Hz']);
22   end

```

The choice of N does not affect fo significantly due to reduced spectral leakage with the Hanning window.

Observation:

In parts d,f,g of this case are not affected fo significantly by the N because the spectral leakage is minimal or reduced by using the Hanning window , resultind in more accurate frequency estimation.

1)

i)

here I am loading the audio file numbered '1' because the modulo of my roll no is '1'(ie modulo(10) of 2022102021 is '1').so by using the below code

```
1  audioFileName = '1.wav';
2  [y, fs] = audioread(audioFileName);
3  L = length(y);
4  N = 8 * L;
5  Y = fft(y, N);
6  f = (0:N-1) * fs / N;
7  [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8  strongestFrequencies = f(sortedIndices(1:3));
9  disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);
11
```

I am getting the first three strongest frequencies are

```
Three Strongest Frequencies (in Hz):
    10.0000    990.0000    10.0125
```

FOR OTHER PARTS:

For file named '0'

Code:

```

1 audioFileName = '0.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);
11

```

Output:

```

Three Strongest Frequencies (in Hz):
    5.0000   995.0000    5.0125

```

*For file named '2'

Code:

```

1 audioFileName = '2.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

```

Three Strongest Frequencies (in Hz):
   15.0000   985.0000   15.0125

```

*For file named '3'

Code:

```

1 audioFileName = '3.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

```

Three Strongest Frequencies (in Hz):
   20.0000   980.0000   20.0125

```

*For file named '4'

Code:

```
1 audioFileName = '4.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);
```

Output:

```
Three Strongest Frequencies (in Hz):
    25.0000    975.0000    25.0125
```

*For file named '5'

Code:

```
1 audioFileName = '5.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);
```

Output:

```
Three Strongest Frequencies (in Hz):
    30.0000    970.0000    30.0125
```

*For file named '6'

Code:


```

1 audioFileName = '6.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

```

Three Strongest Frequencies (in Hz):
    35.0000    965.0000    34.9875

```

*For file named '7'

Code:

```

1 audioFileName = '7.wav';
2 [y, fs] = audioread(audioFileName);
3 L = length(y);
4 N = 8 * L;
5 Y = fft(y, N);
6 f = (0:N-1) * fs / N;
7 [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8 strongestFrequencies = f(sortedIndices(1:3));
9 disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

```

Three Strongest Frequencies (in Hz):
    40.0000    960.0000    40.0125

```

*For file named '8'

Code:

```

1  audioFileName = '8.wav';
2  [y, fs] = audioread(audioFileName);
3  L = length(y);
4  N = 8 * L;
5  Y = fft(y, N);
6  f = (0:N-1) * fs / N;
7  [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8  strongestFrequencies = f(sortedIndices(1:3));
9  disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

```

Three Strongest Frequencies (in Hz):
    45.0000    955.0000    45.0125

```

*For file named '9'

Code:

```

1  audioFileName = '9.wav';
2  [y, fs] = audioread(audioFileName);
3  L = length(y);
4  N = 8 * L;
5  Y = fft(y, N);
6  f = (0:N-1) * fs / N;
7  [sortedValues, sortedIndices] = sort(abs(Y), 'descend');
8  strongestFrequencies = f(sortedIndices(1:3));
9  disp('Three Strongest Frequencies (in Hz):');
10 disp(strongestFrequencies);

```

Output:

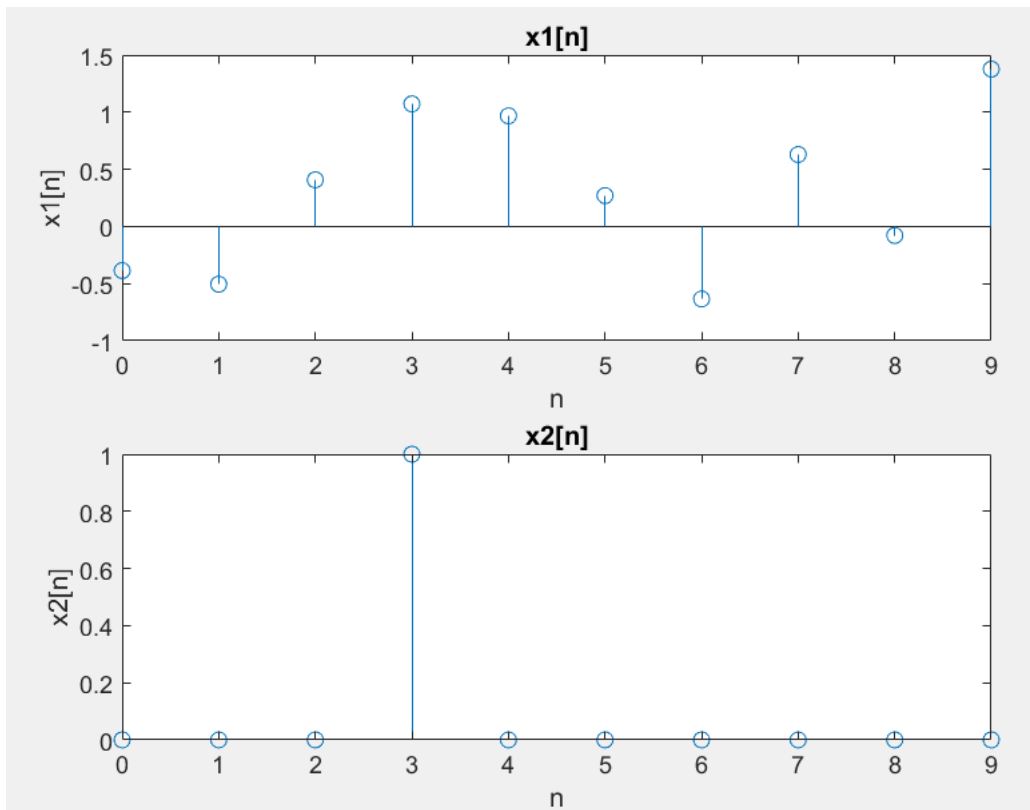
```

Three Strongest Frequencies (in Hz):
    50.0000    950.0000    49.9875

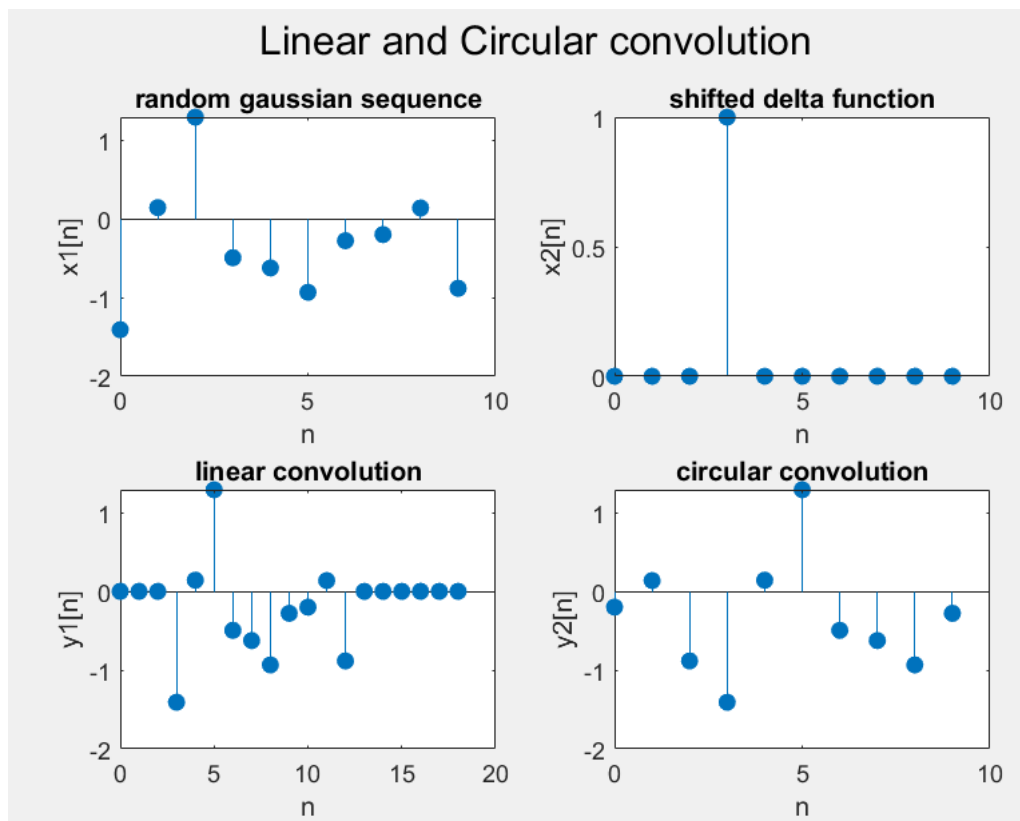
```

2)

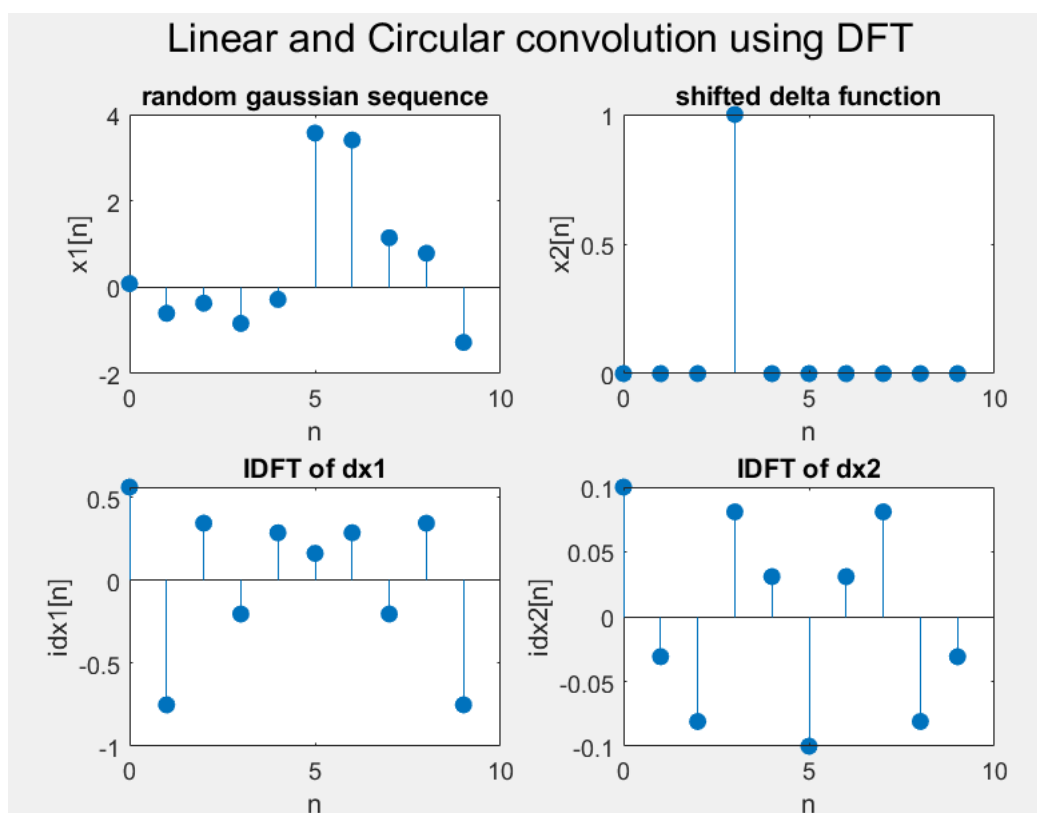
a)



b)



c)



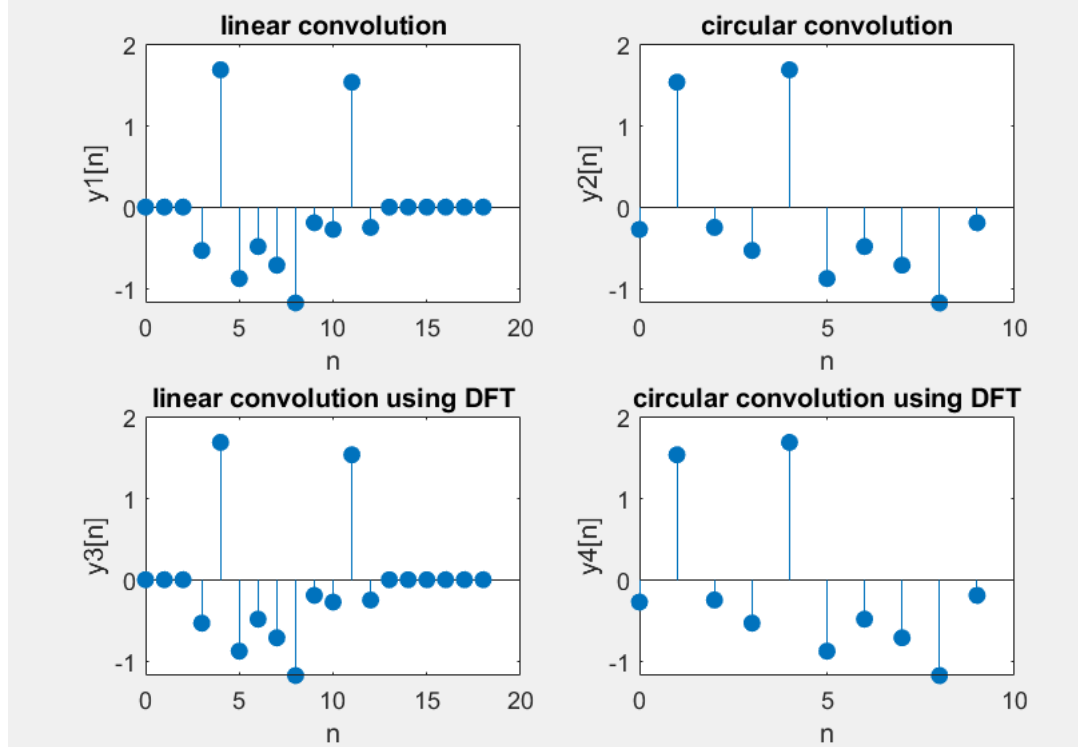
d)

CODE:

```
1  n=0:9;
2  x1=randn(1,10);
3  % x2=[0,0,0,1,0,0,0,0,0,0];
4  x2=(n==3);
5
6  dx1=fft(x1);
7  dx2 =fft(x2);
8  idx1=ifft(dx1);
9  idx2=ifft(dx2);
10 y1=conv(x1,x2);
11 y2=cconv(x1,x2);
12 y3=conv(idx1,idx2);
13 y4=cconv(idx1,idx2);
14 c_n = 0:length(y2)-1;
15
16 figure
17
18 subplot(2,2,1);
19 stem(0:18, y1, "filled");
20 title('linear convolution');
21 xlabel('n');
22 ylabel('y1[n]');
23
24 subplot(2,2,2);
25 stem(c_n, y2, "filled");
26 title('circular convolution');
27 xlabel('n');
28 ylabel('y2[n]');
29
30
31 subplot(2,2,3);
32 stem(0:18, y3, "filled");
33 title('linear convolution using DFT');
34 xlabel('n');
35 ylabel('y3[n]');
36
37 subplot(2,2,4);
38 stem(c_n, y4, "filled");
39 title('circular convolution using DFT');
40 xlabel('n');
41 ylabel('y4[n]');
42
43 sgtitle('Linear and Circular convolution with and without using DFT');
```

PLOT:

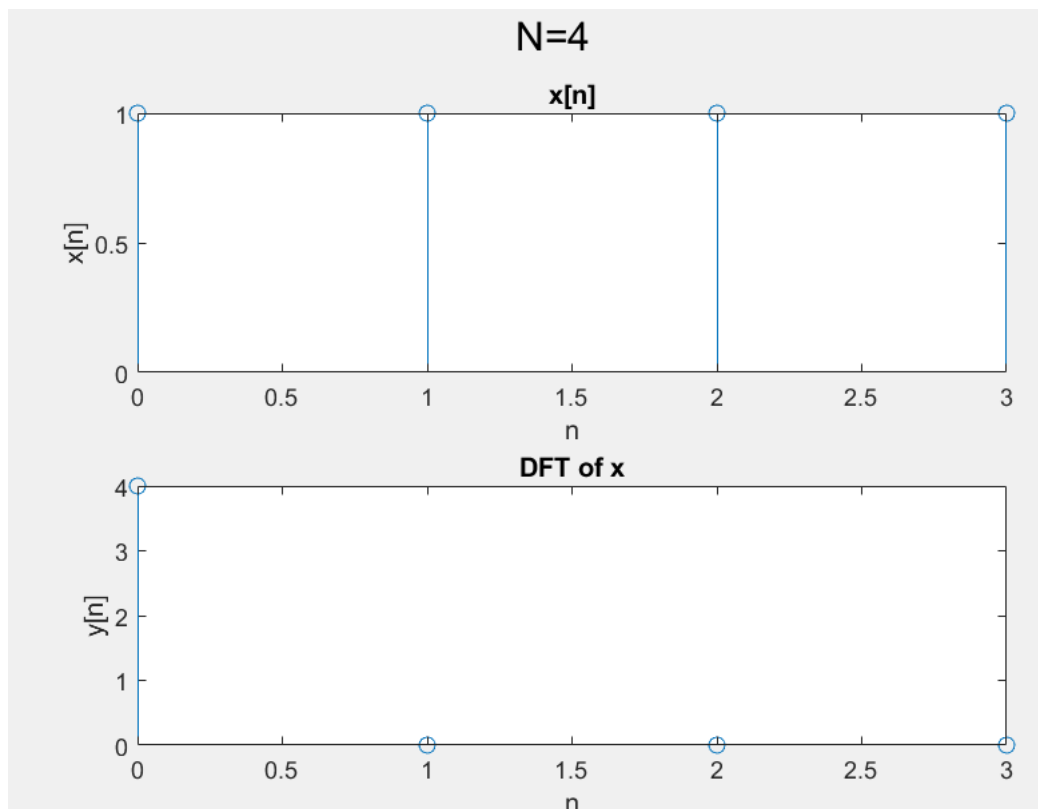
Linear and Circular convolution with and without using DFT

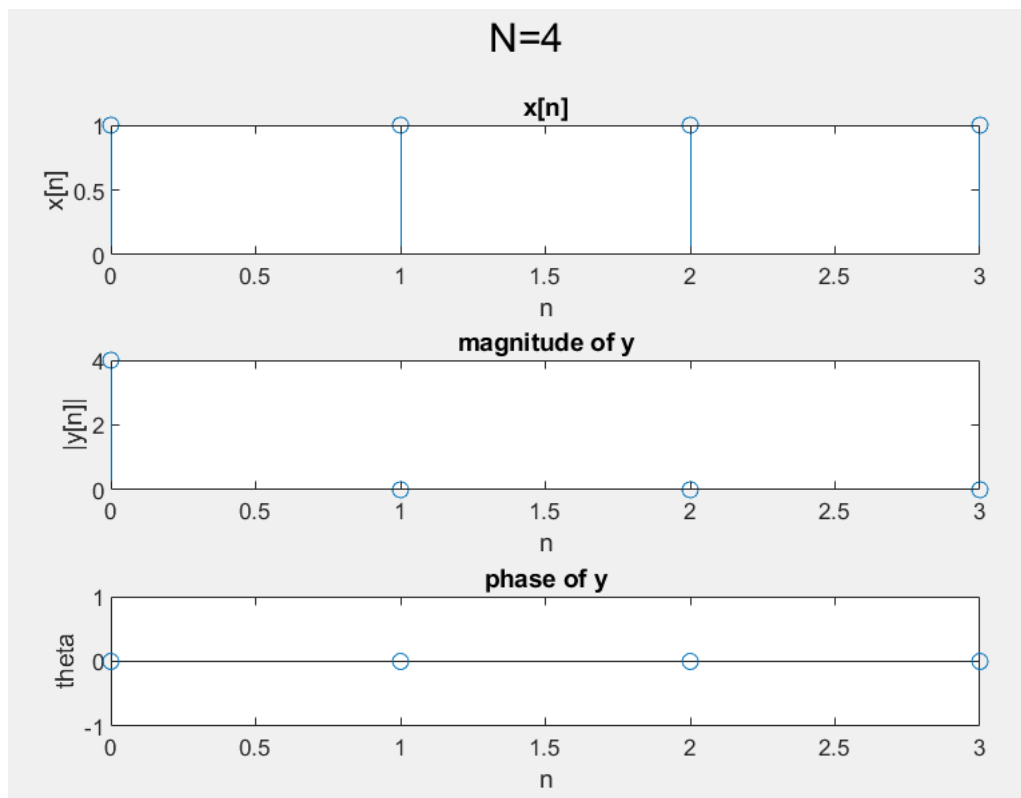


3)

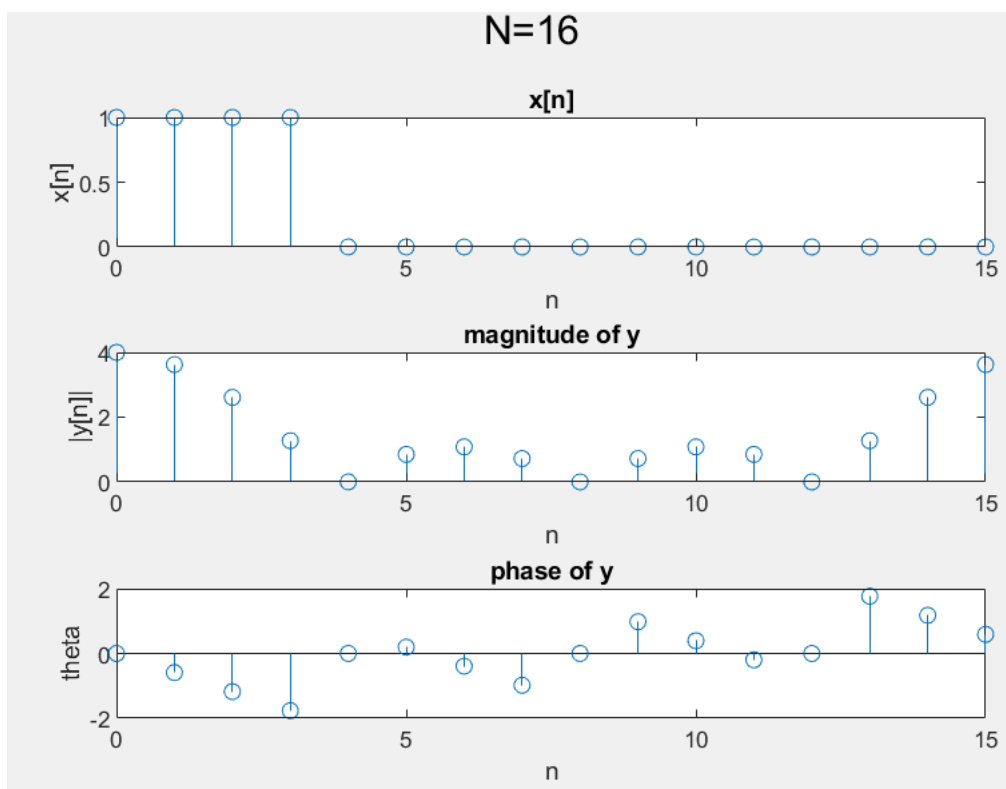
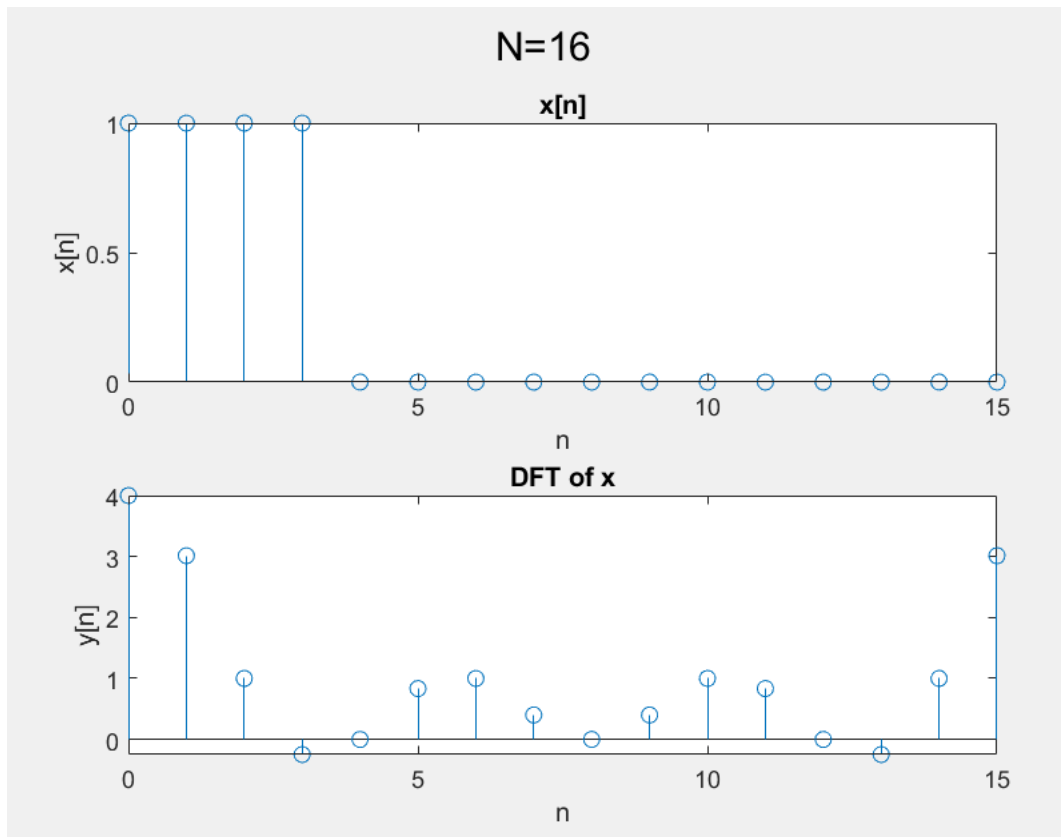
a)

when $N=4$



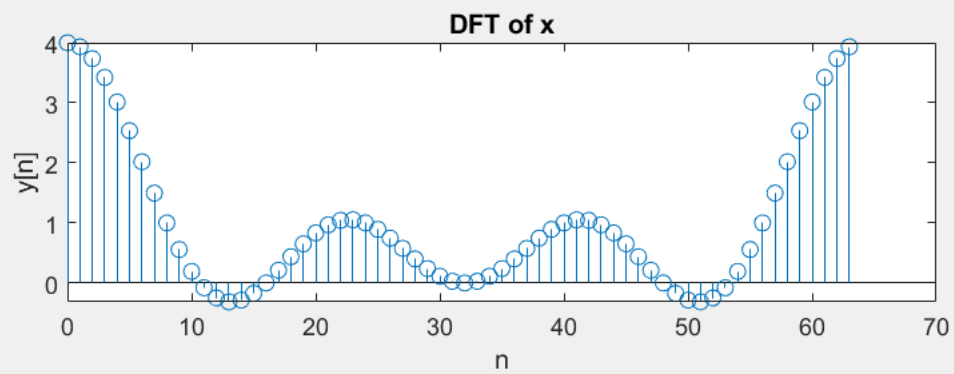
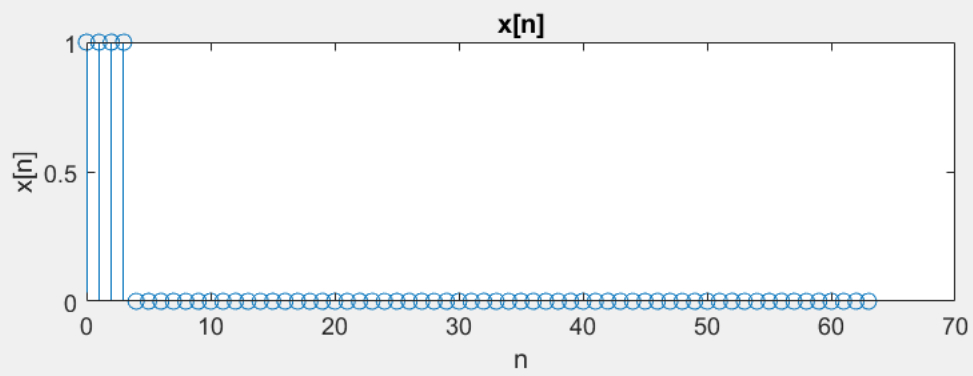


When $N=16$

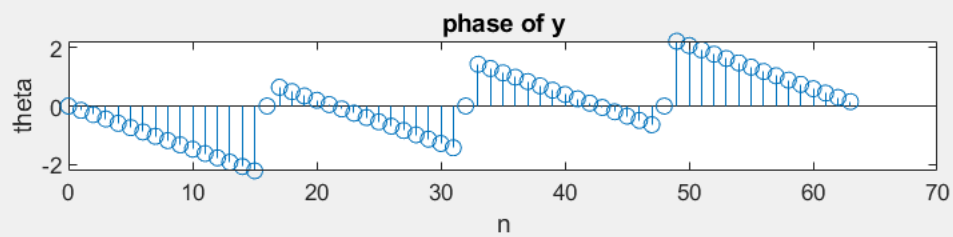
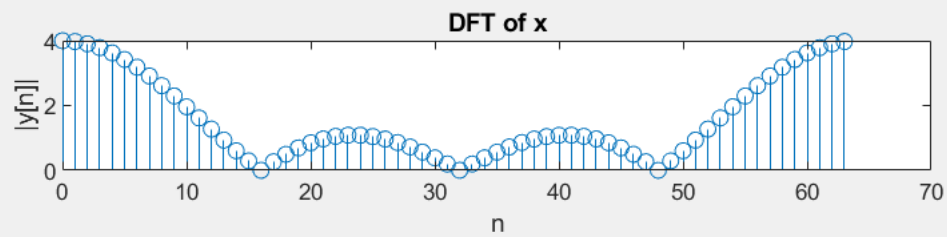
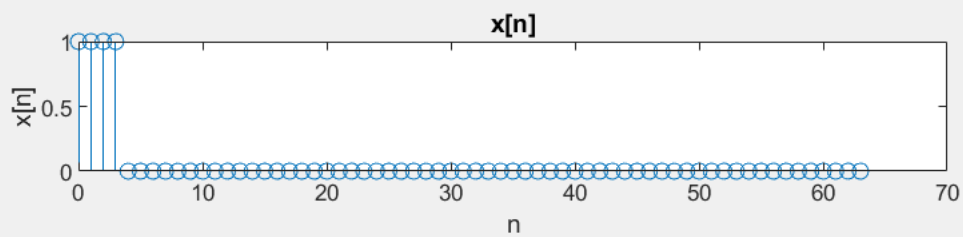


When N=64

N=64



N=64

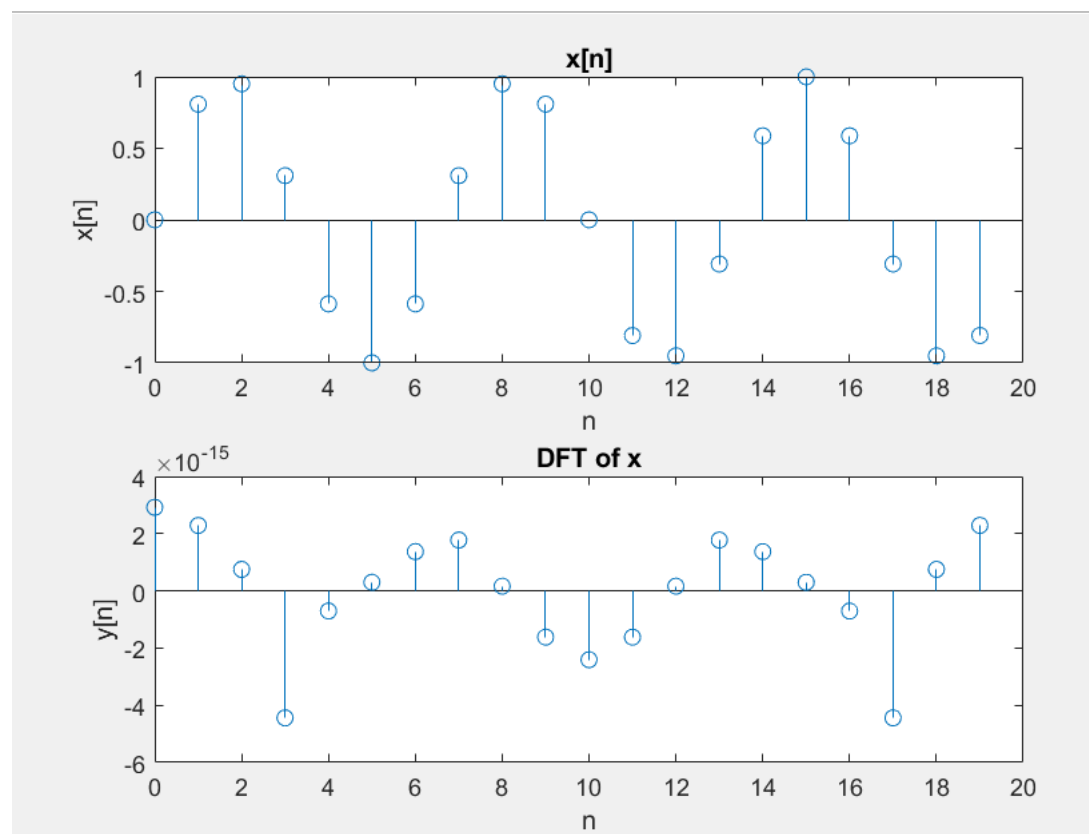


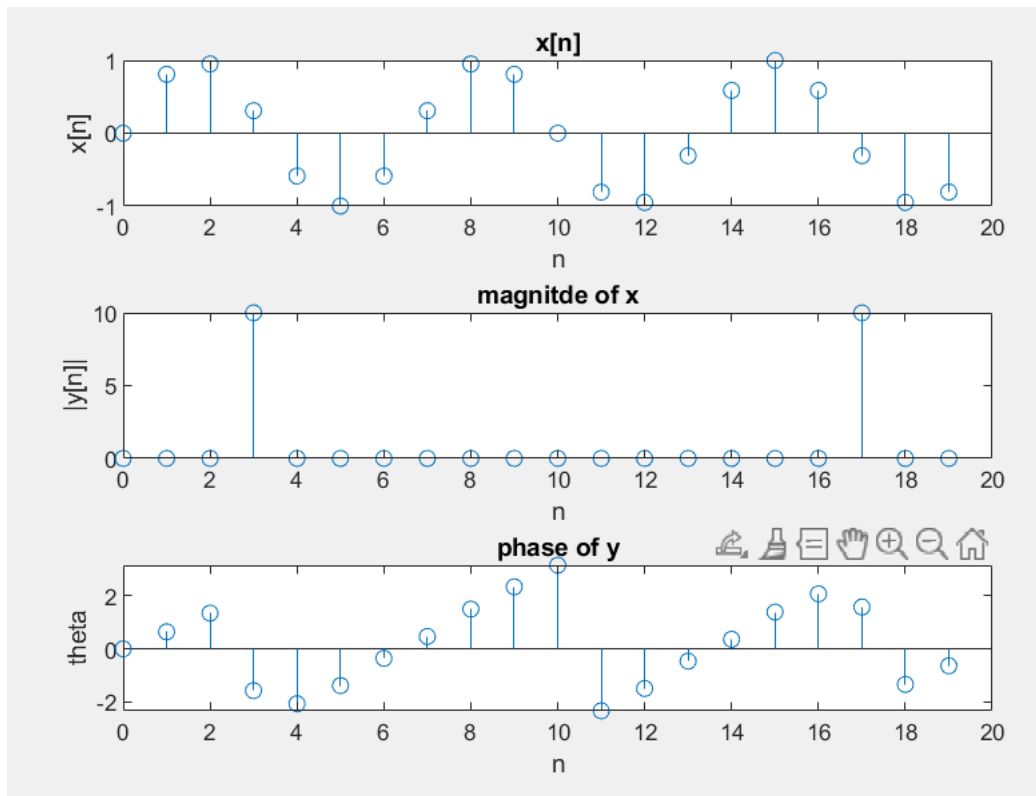
b)

CODE:

```
1 N=20;
2 n=0:N-1;
3 wo=(3*pi)/10;
4 x=sin(wo*n);
5 y=fft(x,N);
6 figure
7 subplot(3,1,1)
8 stem(n,x);
9 title('x[n]')
10 xlabel('n');
11 ylabel('x[n]');
12 subplot(3,1,2)
13 stem(n,abs(y));
14 title('absolute of DFT of x')
15 xlabel('n');
16 ylabel('|y[n]|');
17 subplot(3,1,3)
18 stem(n,angle(y));
19 title('phase of y')
20 xlabel('n');
21 ylabel('theta');
```

Plots:





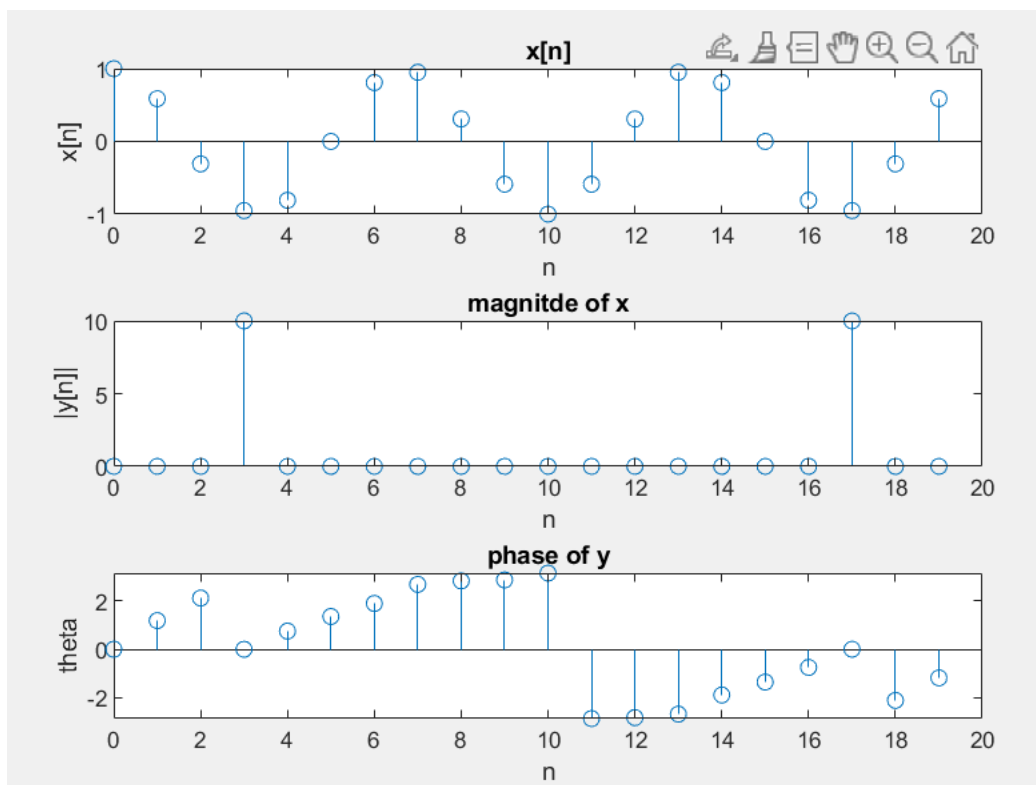
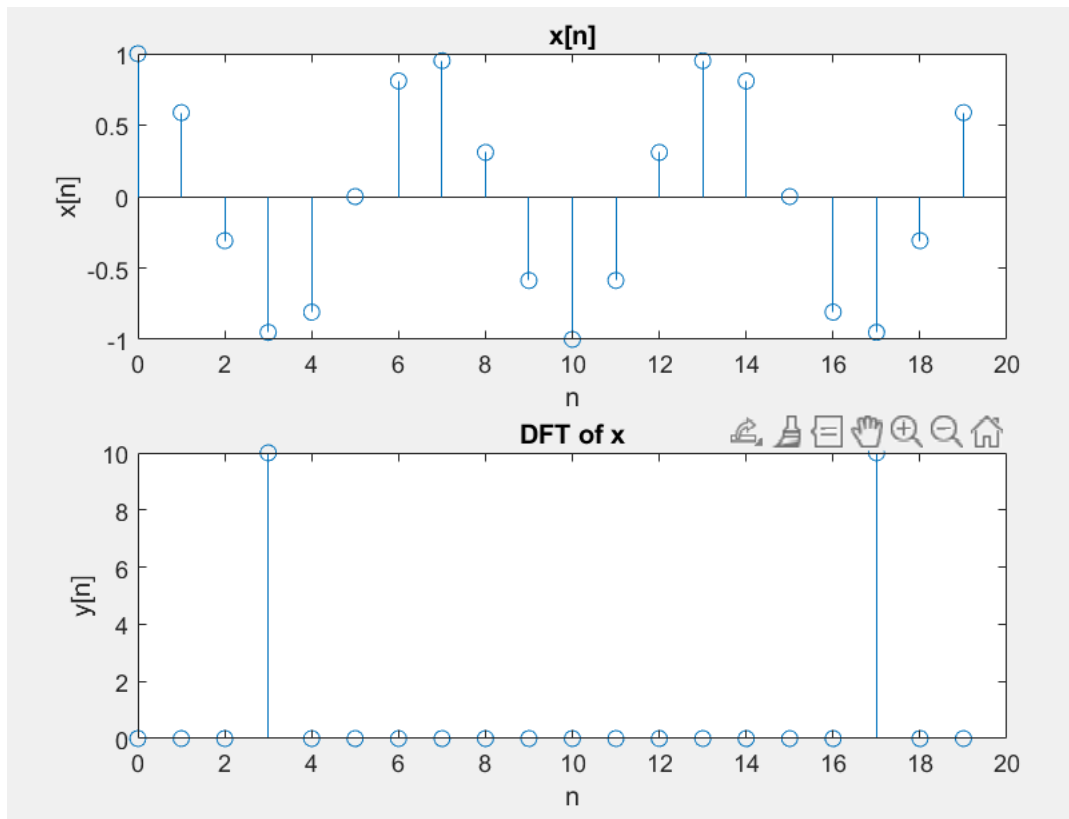
c)

CODE:

```

1  N=20;
2  n=0:N-1;
3  wo=(3*pi)/10;
4  x=cos(wo*n);
5  y=fft(x,N);
6  figure
7  subplot(3,1,1)
8  stem(n,x);
9  title('x[n]')
10 xlabel('n');
11 ylabel('x[n]');
12 subplot(3,1,2)
13 stem(n,abs(y));
14 title('absolute of DFT of x')
15 xlabel('n');
16 ylabel('|y[n]|');
17 subplot(3,1,3)
18 stem(n,angle(y));
19 title('phase of y')
20 xlabel('n');
21 ylabel('theta');
```

Plots:

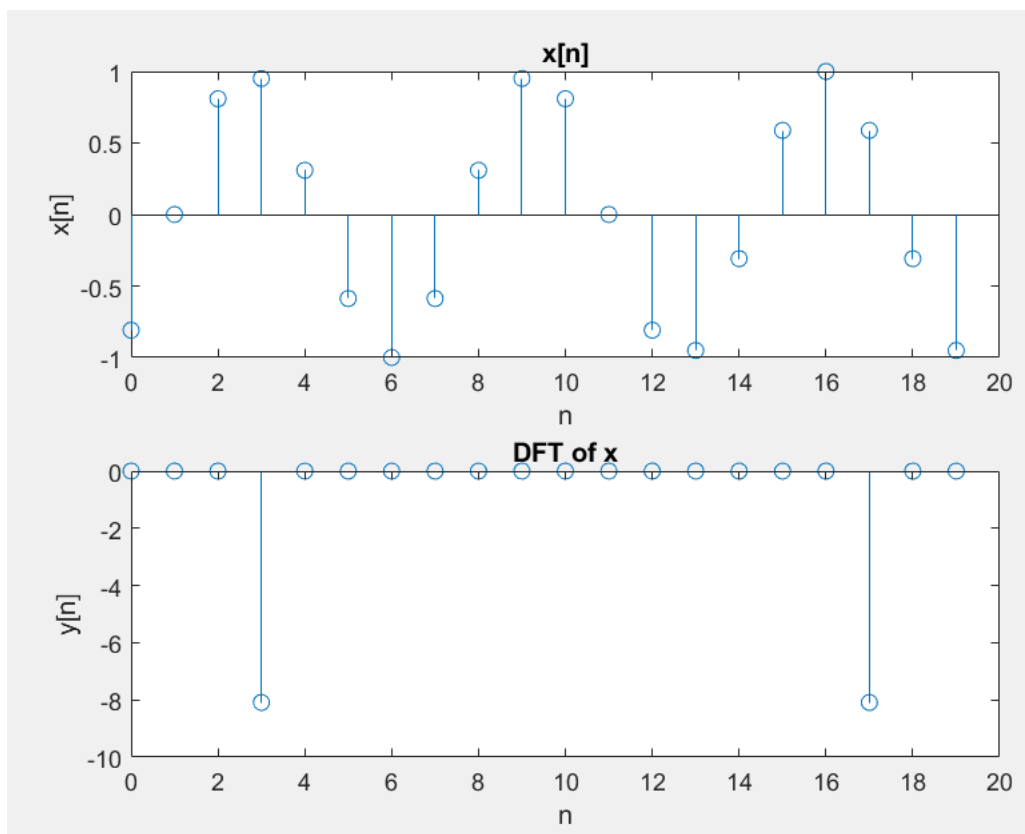


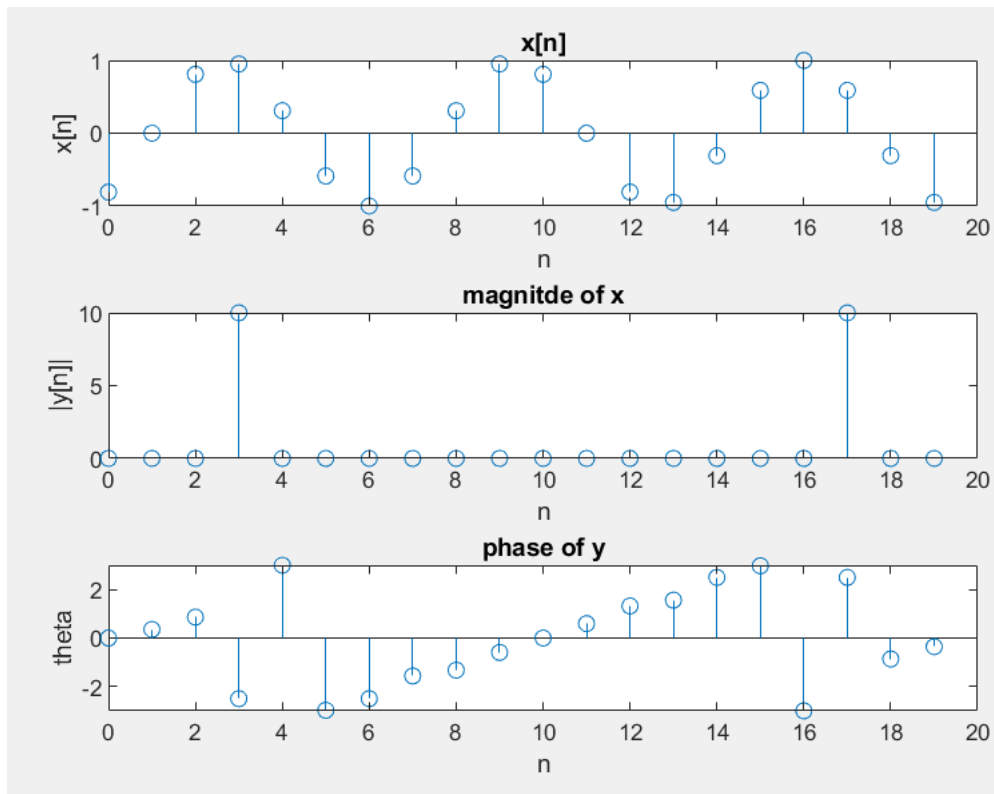
d)

CODE:

```
1 N=20;
2 n=0:N-1;
3 wo=(3*pi)/10;
4 x=sin(wo*(n-1));
5 y=fft(x,N);
6 figure
7 subplot(3,1,1)
8 stem(n,x);
9 title('x[n]')
10 xlabel('n');
11 ylabel('x[n]');
12 subplot(3,1,2)
13 stem(n,abs(y));
14 title('absolute of DFT of x')
15 xlabel('n');
16 ylabel('|y[n]|');
17 subplot(3,1,3)
18 stem(n,angle(y));
19 title('phase of y')
20 xlabel('n');
21 ylabel('theta');
```

Plots:





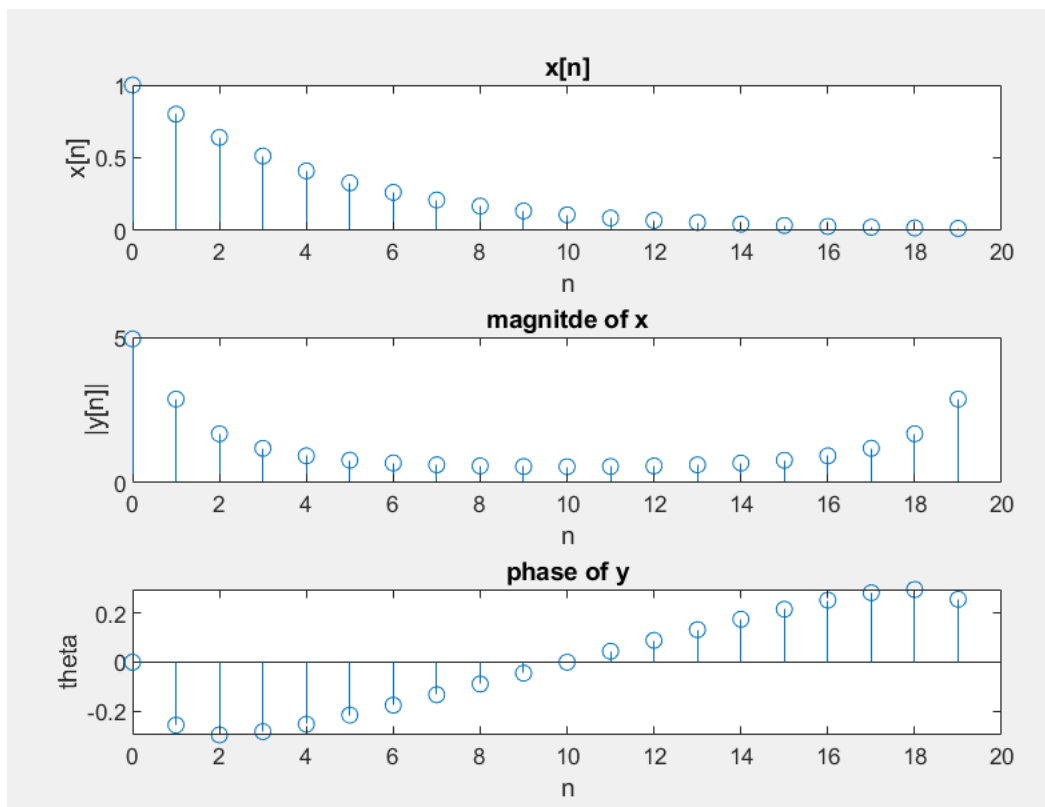
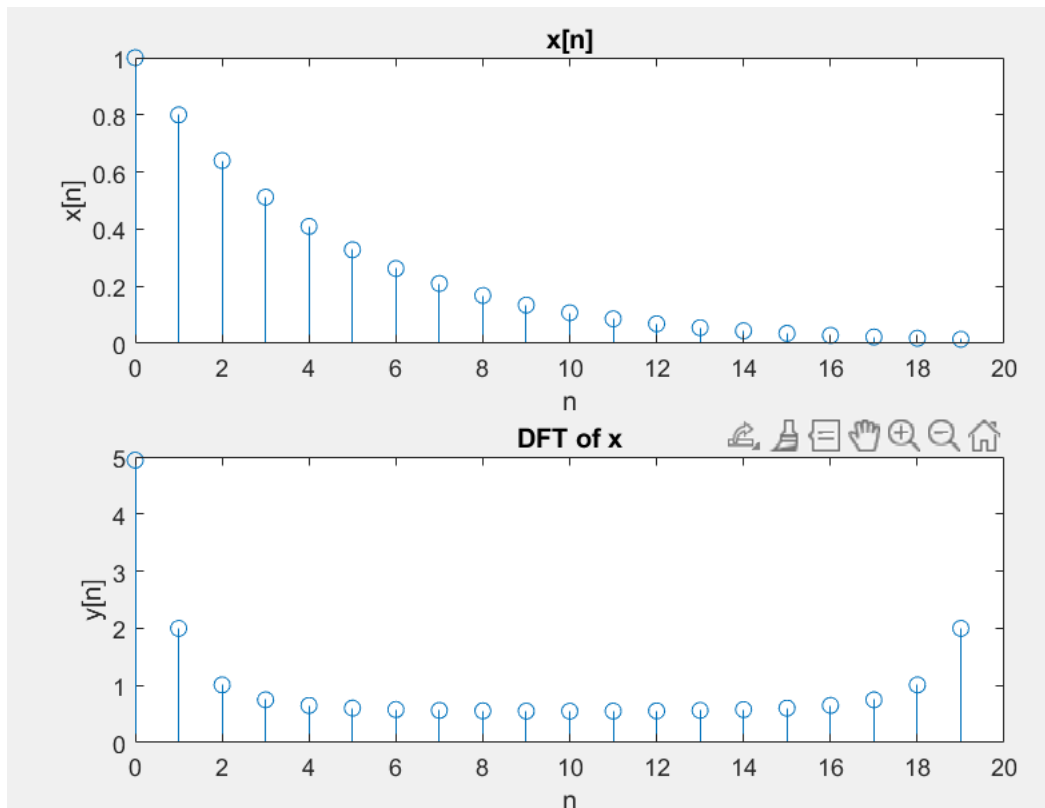
e)

CODES:

```

1  N=20;
2  n=0:N-1;
3  wo=(3*pi)/10;
4  x=(0.8).^n;
5  y=fft(x,N);
6  figure
7  subplot(3,1,1)
8  stem(n,x);
9  title('x[n]')
10 xlabel('n');
11 ylabel('x[n]');
12 subplot(3,1,2)
13 stem(n,abs(y));
14 title('absolute of DFT of x')
15 xlabel('n');
16 ylabel('|y[n]|');
17 subplot(3,1,3)
18 stem(n,angle(y));
19 title('phase of y')
20 xlabel('n');
21 ylabel('theta');
```

Plots:

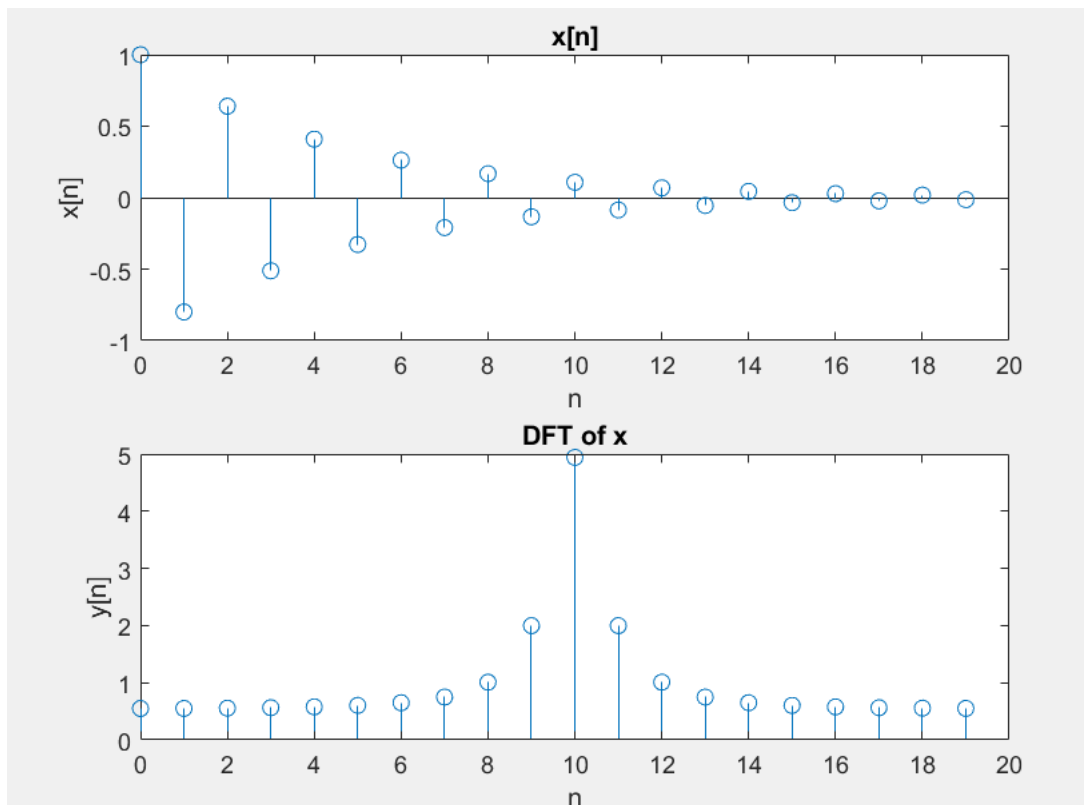


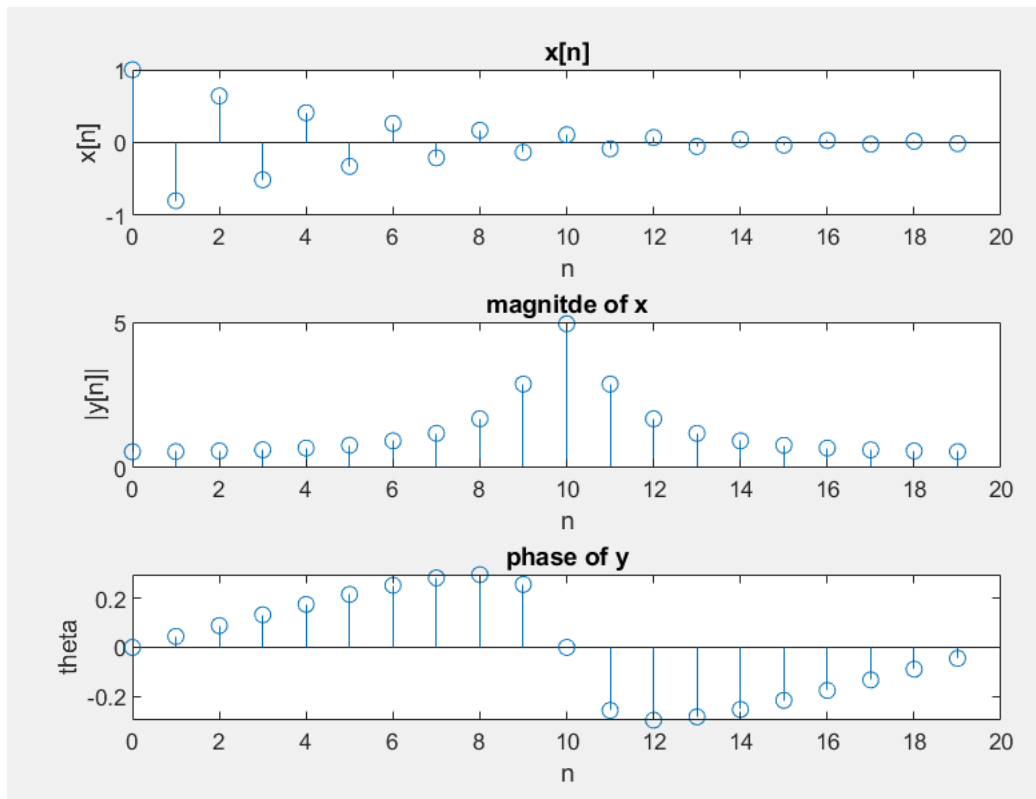
f)

CODE:

```
1 N=20;
2 n=0:N-1;
3 wo=(3*pi)/10;
4 x=(-1*0.8).^n;
5 y=fft(x,N);
6 figure
7 subplot(3,1,1)
8 stem(n,x);
9 title('x[n]')
10 xlabel('n');
11 ylabel('x[n]');
12 subplot(3,1,2)
13 stem(n,abs(y));
14 title('absolute of DFT of x')
15 xlabel('n');
16 ylabel('|y[n]|');
17 subplot(3,1,3)
18 stem(n,angle(y));
19 title('phase of y')
20 xlabel('n');
21 ylabel('theta');
```

Plots:





We can find the low-frequency and high-frequency components that contribute most significantly to the signal by analysing the magnitude spectrum .

The peaks or significant values that are closer to the zero frequency are associated with low – frequency components , while those farther from the zero frequency are associated with high-frequency components.