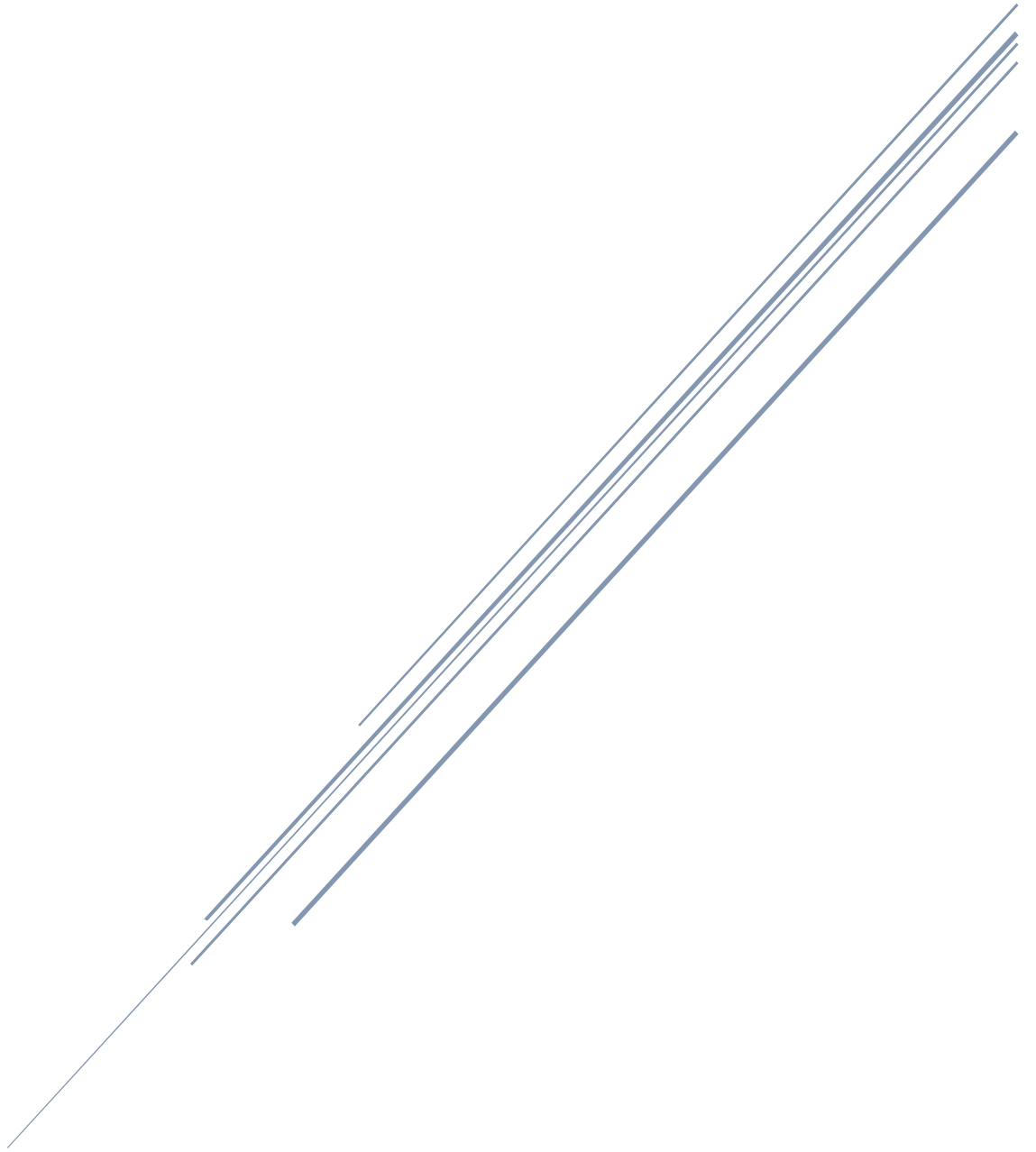
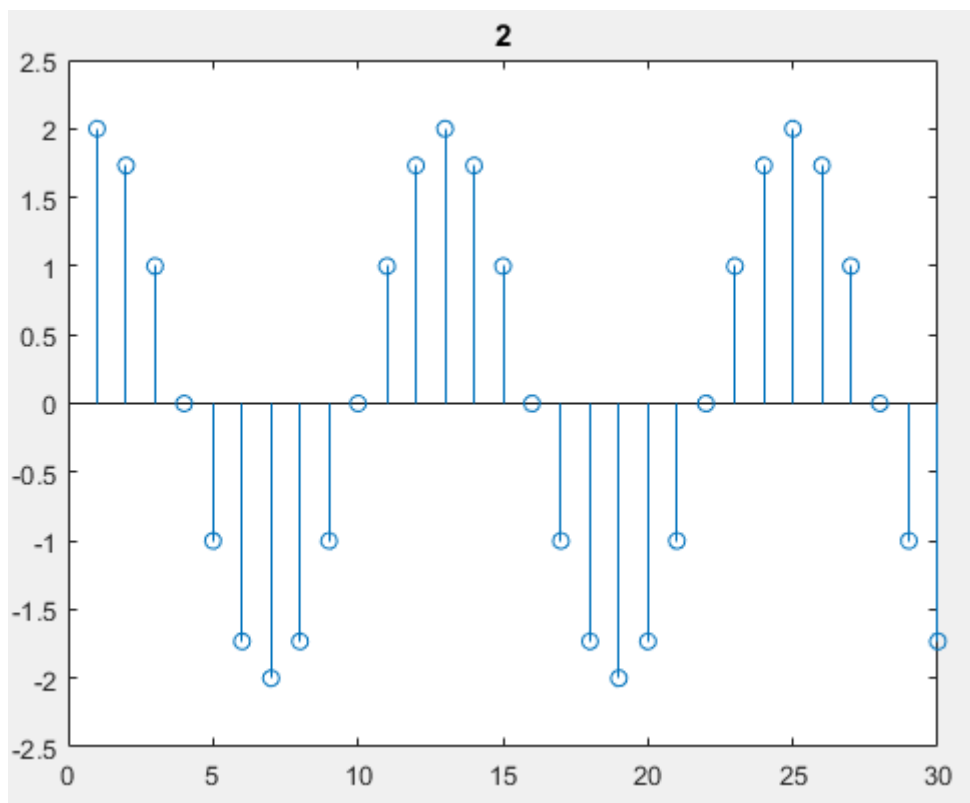
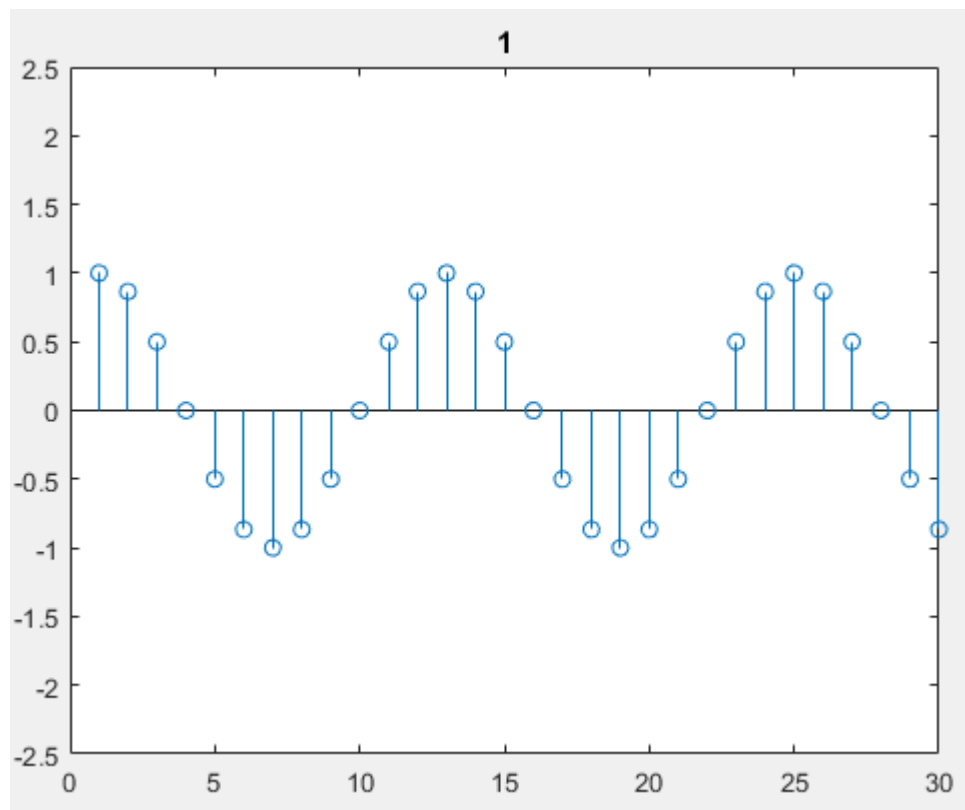


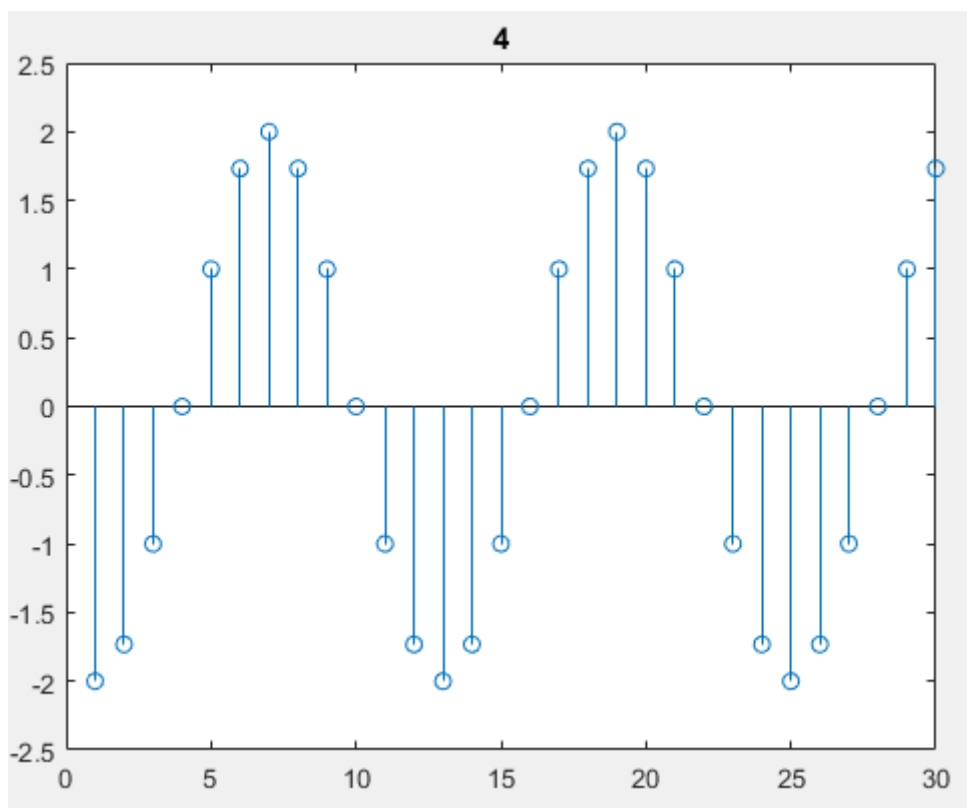
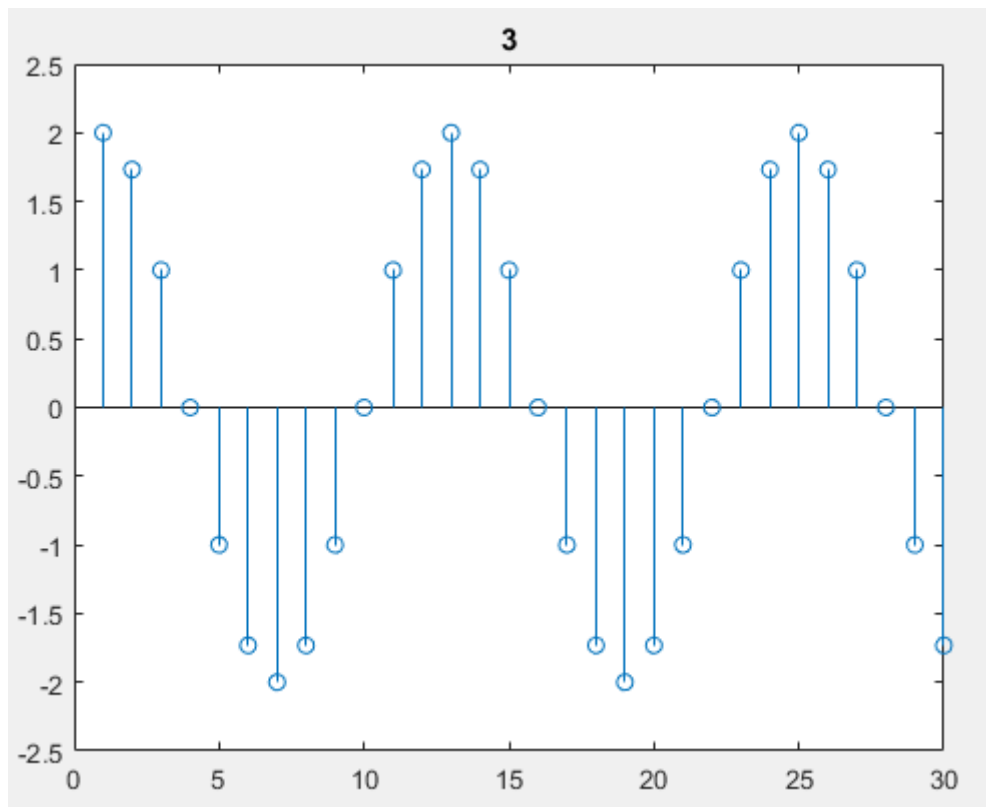
# BLG354E – HOMEWORK 1

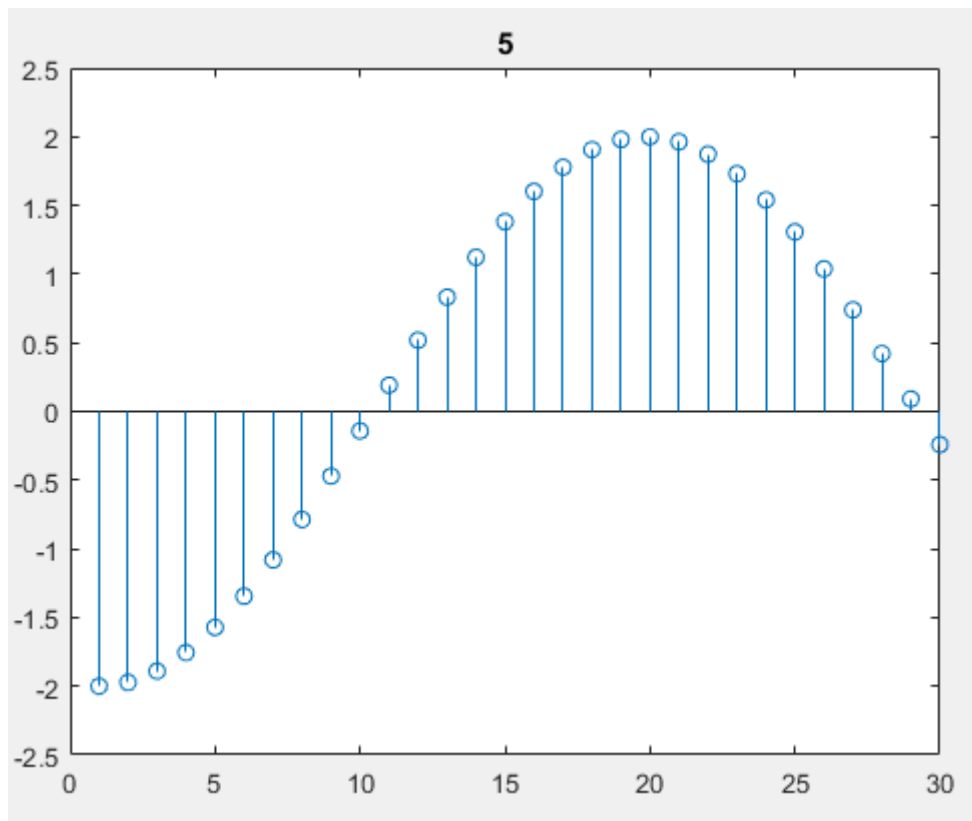


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1)







Matlab script SaS\_Q1.m is used for plotting the signals.

```
A=[1 2 2 2 2];
omega0=[pi/6 pi/6 pi/6 pi/6 1/6];
theta=[0 0 4*pi pi pi];
n=0:30;

for a = 1:5
    figure(a)
    stem(A(a)*cos(omega0(a)*n+theta(a)))
    title([num2str(a)])
    axis([0 30 -2.5 2.5])
end
```

$$\begin{aligned}
 x_1[n] &= 1 \cos\left(\frac{\pi}{6}n\right) && \text{half the magnitude} \\
 x_2[n] &= 2 \cos\left(\frac{\pi}{6}n\right) && \text{no time shift here} \\
 x_3[n] &= 2 \cos\left(\frac{\pi}{6}n + 4\pi\right) && \text{Bigger time shift} \\
 x_4[n] &= 2 \cos\left(\frac{\pi}{6}n + \pi\right) && \text{Smaller period} \\
 x_5[n] &= 2 \cos\left(\frac{1}{6}n + \pi\right)
 \end{aligned}$$

2)

$$a) x_1 = \cos(t)$$

$$\text{period } T_0 = \frac{2\pi m}{\omega_0}$$

$$\omega_0 = 1 \quad T_1 = \frac{2\pi m}{1} = 2\pi$$

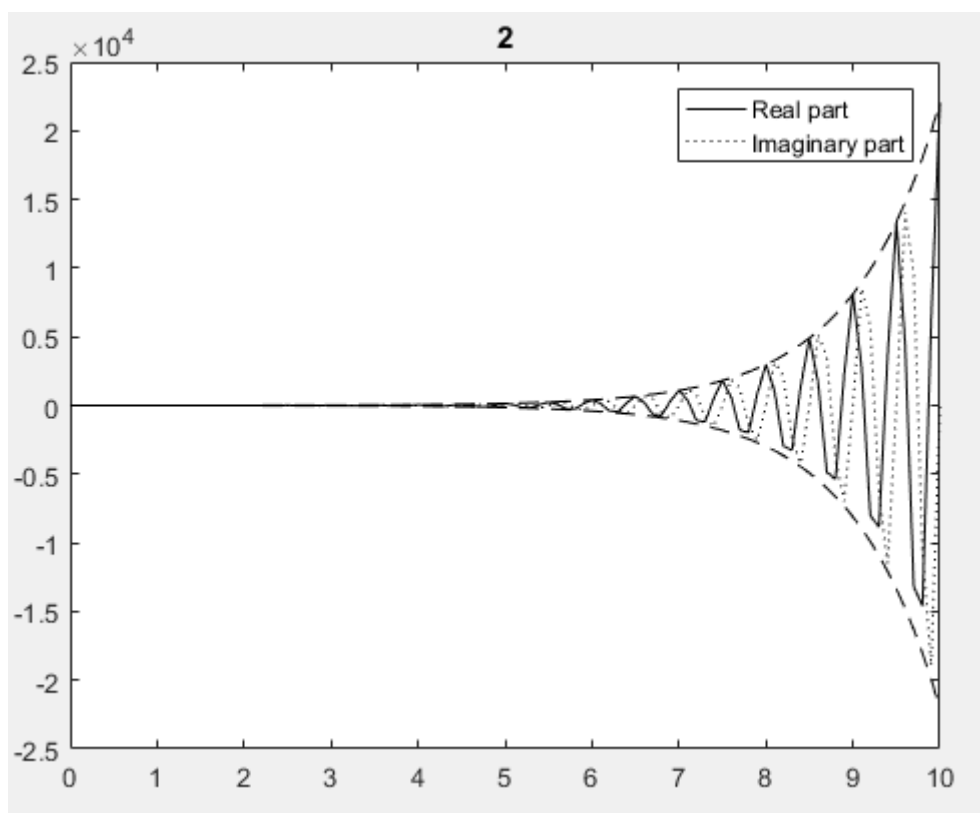
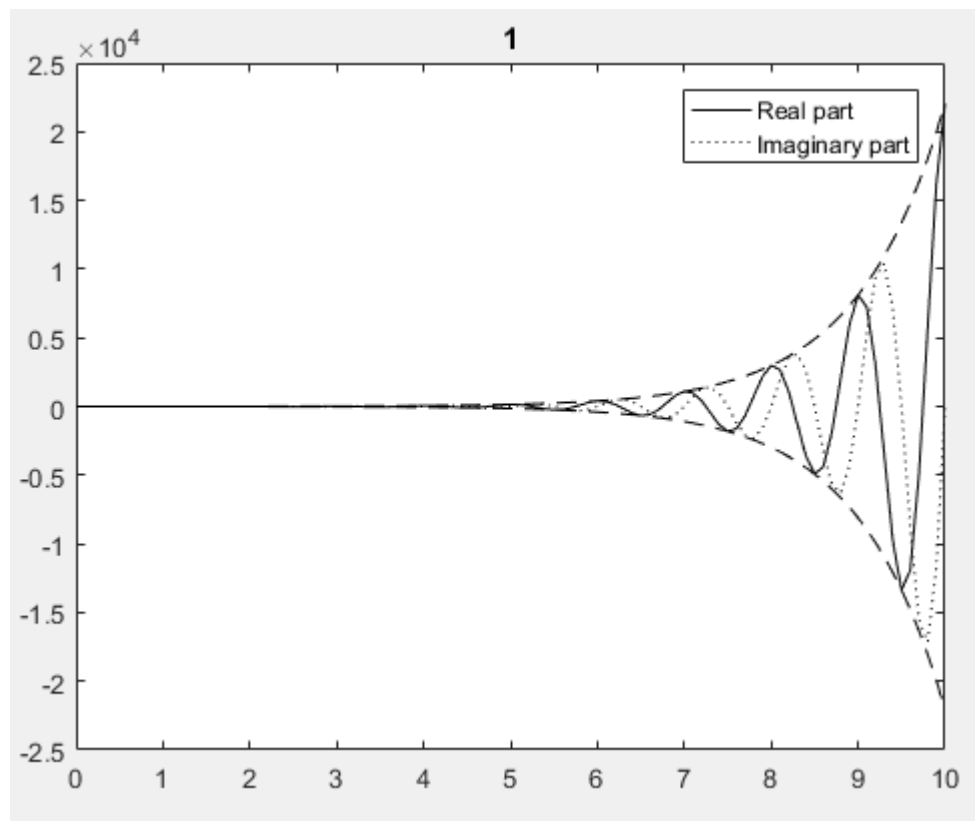
$$x_2 = \sin(\pi t)$$

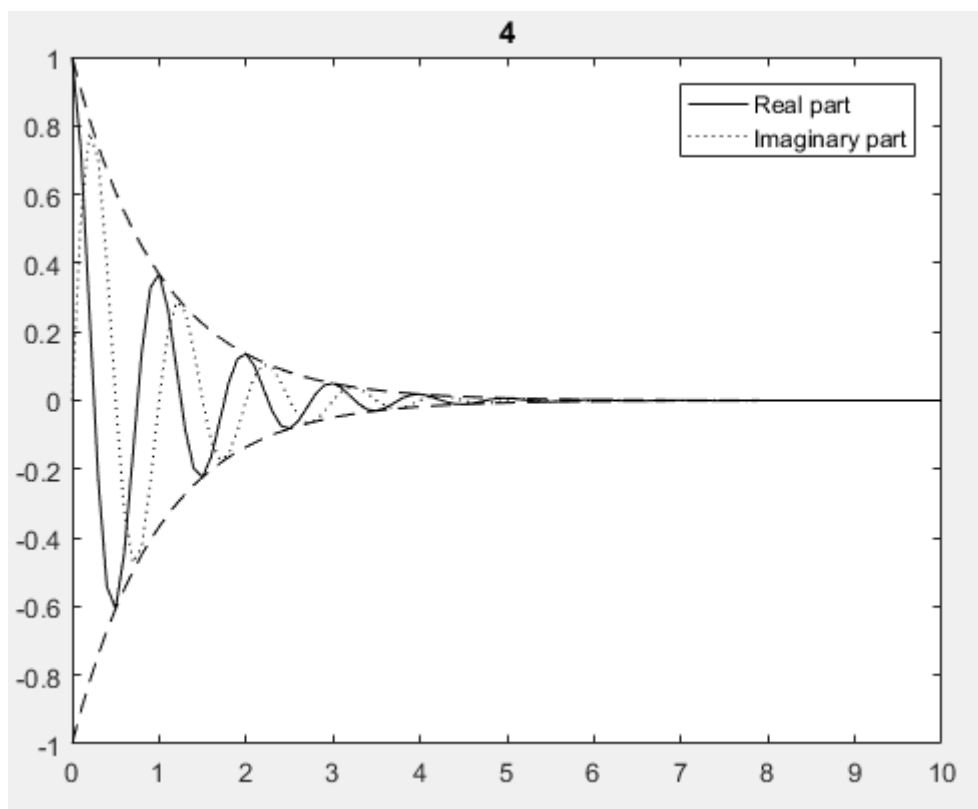
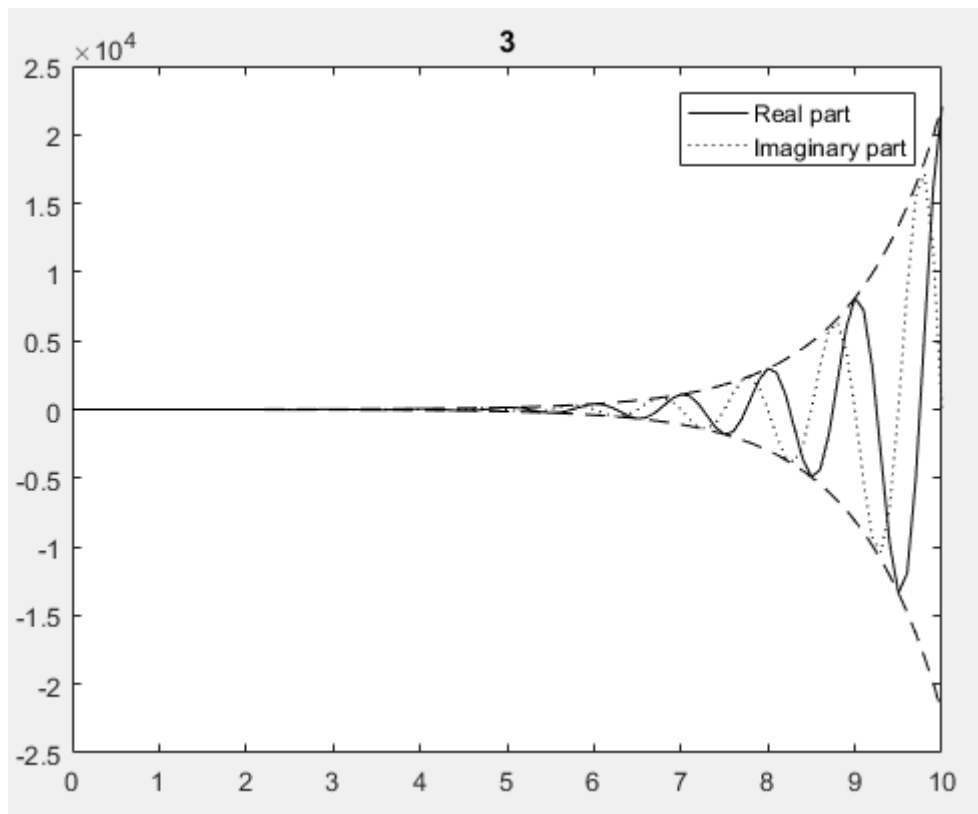
$$\omega_0 = \pi \quad T_2 = \frac{2\pi m}{\pi} = 2$$

$$b) x_3(t) = \underbrace{\cos(t)}_{\substack{\text{periodic} \\ T_1 = 2\pi}} + \underbrace{\sin(\pi t)}_{\substack{\text{periodic} \\ T_2 = 2}}$$

$$\text{ratio } \frac{T_1}{T_2} = \frac{2\pi}{2} = \pi \text{ not rational} \\ \text{not periodic}$$

3)





Since ID\_4 has negative  $r$  value it is a decreasing function, with positive  $r$  values ID\_1, ID\_2, ID\_3 are increasing functions.

ID\_3 having negative  $\omega_0$  value its real part is faster than imaginary part. With positive  $\omega_0$  values ID\_1, ID\_2, ID\_4 have their imaginary part faster than their real part.

Since ID\_1, ID\_3, ID\_4 has  $|\omega_0|$  values equal they share the same period  $T=1$ , and ID\_2 has  $T=1/2$ .

Matlab script SaS\_Q3.m is used for plotting the signals.

```
absC=1;
omega0=[2*pi 4*pi -2*pi 2*pi];
r=[1 1 1 -1];
t=0:.1:10;
theta=2*pi;
for z = 1:4
    C=absC*exp(j.*theta);
    a=r(z)+j*omega0(z);
    xenvu=abs(C)*exp(r(z)*t);
    xenvd=-abs(C)*exp(r(z)*t);
    func= C.*exp(a.*t);
    figure(z)
    plot(t,real(func),'k',t,imag(func),'k:',t,xenvu,'k--',t,xenvd,'k--')
    title([num2str(z)])
    legend('Real part','Imaginary part')

    title([num2str(z)])
end
```



4)

$$a) \quad x_1(t) \rightarrow t \cdot u(t) \cdot x(t) \rightarrow y_1(t)$$

$$x_2(t) \rightarrow t \cdot u(t) \cdot x_2(t) \rightarrow y_2(t)$$

$$x_3(t) = x_1(t) + x_2(t)$$

$$y_3(t) = t \cdot u(t) \cdot [x_1(t) + x_2(t)] = t \cdot u(t) \cdot x_1(t) + t \cdot u(t) \cdot x_2(t)$$

$x_3(t)$   
 $\downarrow$

$y_1(t)$

$y_2(t)$

linear

$$b) \quad y(t) = t \cdot u(t) \cdot x(t)$$

$y$  does not depend on

a previous value of  $t$ , ( $t - \dots$ )

so system is memoryless

$$c) \quad y(t) = t \cdot u(t) \cdot x(t)$$

for  $t = 1$

$$y(1) = 1 \cdot u(1) \cdot x(1)$$

output  $y$  depends on the

meantime of input  $x$

since they are on the same

time period system is casual

$$d) \quad y(t) = t \cdot u(t) \cdot x(t) = T[x(t)],$$

$$T[x(t - T_0)] = t \cdot u(t) \cdot x(t - T_0)$$

$$\neq y(t - T_0) = (t - T_0) \cdot u(t - T_0) \cdot x(t - T_0)$$

System is time varying.

5) Matlab script SaS\_Q5.m is used for this problem.

```
[y, Fs] = audioread('Frank_C_Stanley_-_08_-_When_The_Mockingbirds_Are_Singing_In_The_Wildwood.wav');

yx = zeros(size(y,1),2);

for i=2:size(y,1)
    yx(i,1) = exp(y(i,1)/2.5);
    yx(i,2) = exp(y(i,2)/2.5);
end

audiowrite('signal_div_a.wav',yx,Fs,'BitsPerSample',64)

[ww,Fs] = audioread('signal_div_a.wav');

y2 = zeros(size(ww,1),2);

for i=1:size(ww,1)
    y2(i,1) = log(ww(i,1))*2.5;
    y2(i,2) = log(ww(i,2))*2.5;
end

audiowrite('signal_reinverted_a.wav',y2,Fs,'BitsPerSample',64);

[y, Fs] = audioread('Frank_C_Stanley_-_08_-_When_The_Mockingbirds_Are_Singing_In_The_Wildwood.wav');

yx = zeros(size(y,1),2);

for i=2:size(y,1)
    yx(i,1) = y(i,1)+yx(i-1,1);
    yx(i,2) = y(i,2)+yx(i-1,1);
end

audiowrite('signal_div_b.wav',yx,Fs,'BitsPerSample',64)

[ww,Fs] = audioread('signal_div_b.wav');

y2 = zeros(size(ww,1),2);

for i=1:size(ww,1)
    y2(i,1) = ww(i,1)-y2(i+1,1);
    y2(i,2) = ww(i,2)-y2(i+1,2);
end

audiowrite('signal_reinverted_b.wav',y2,Fs,'BitsPerSample',64);
```