

Principles of Communication

Systems Lab

Lab 2, 20th August 2019

Pratyush Nandi (IMT2017518)

Answer to Q1

Note: For question 1 the signal $u(t) = 2I[1,3](t) - 3I[2,4](t)$ and $umf = u(-t)$,

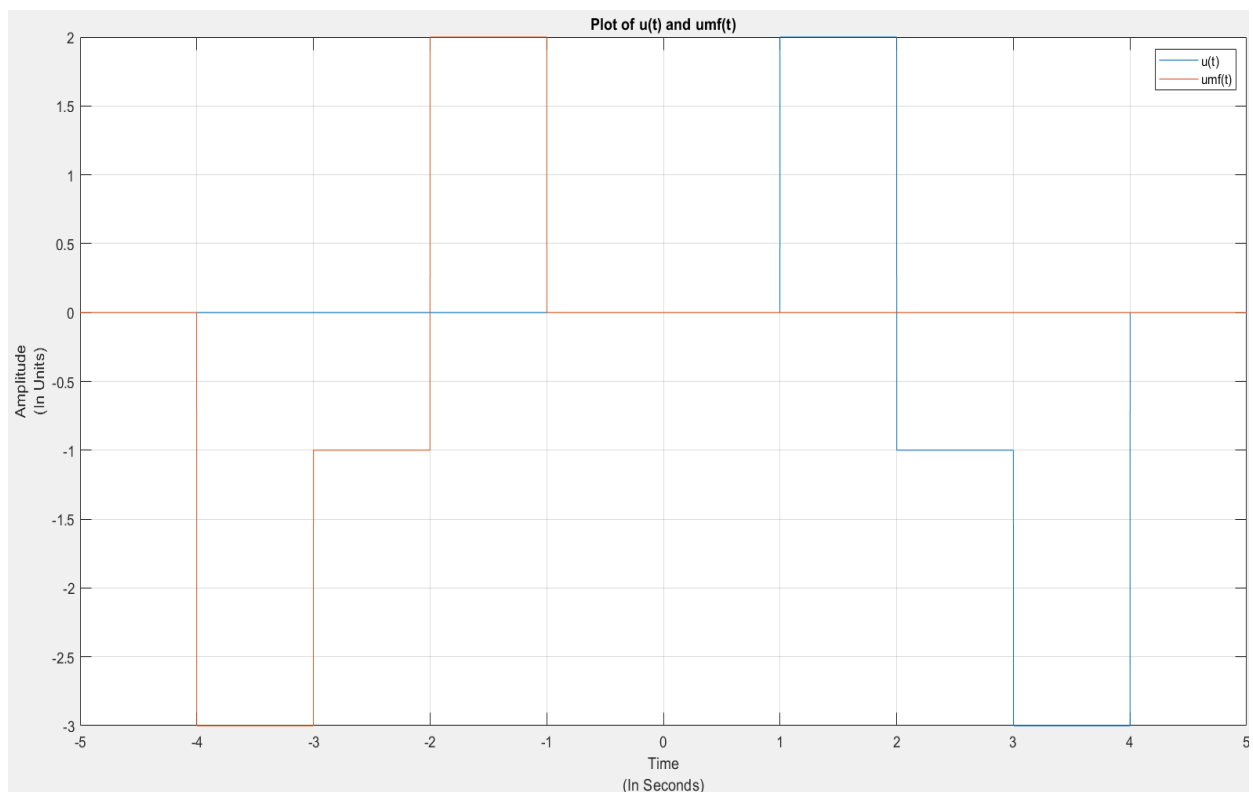
$s(t) = u(t) + j v(t)$ where $v(t) = I[-1,2](t) + 2I[0,1](t)$ and $smf = s^*(-t)$,

$s_1(t) = s(t - t_0)x(e^{j\theta})$ here $t_0 = 2$ and $\theta = (\pi/4)$,

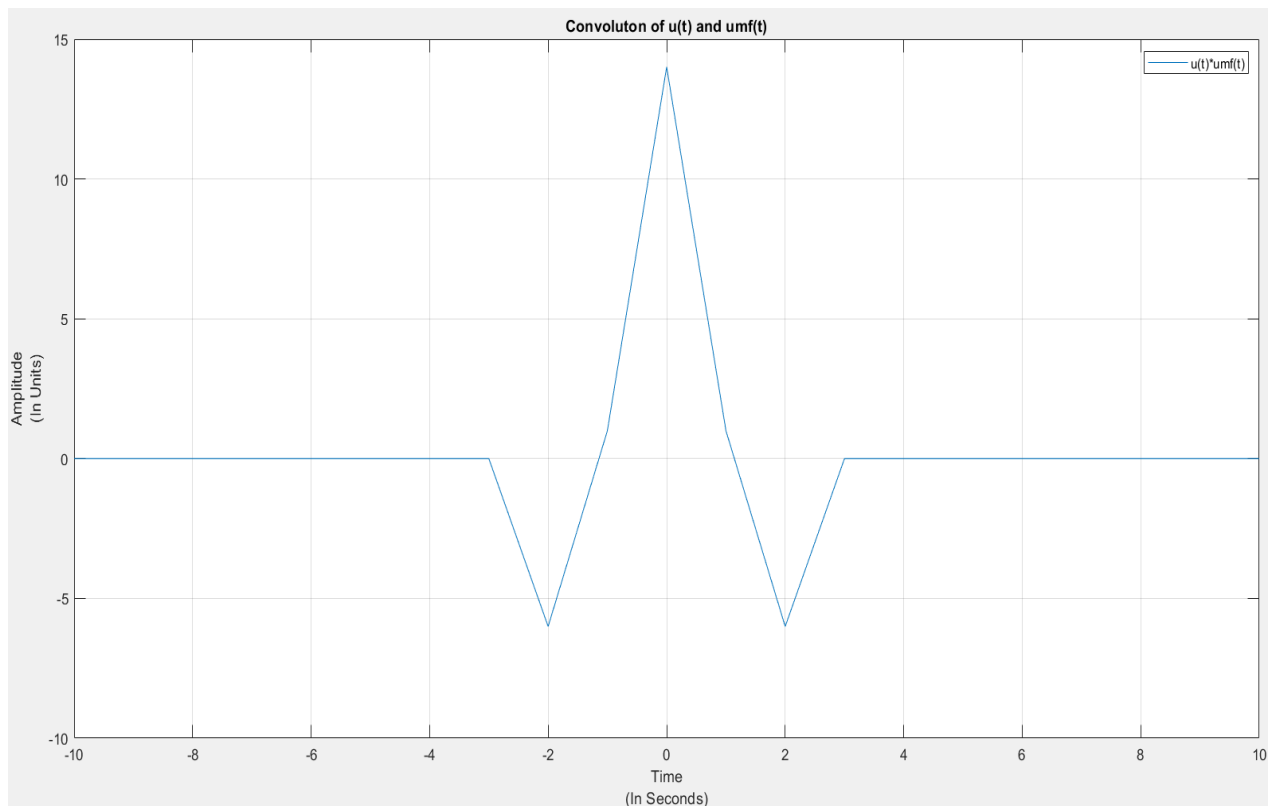
'*' here means conjugate.

Unit time is in seconds and sampled at 1KHz frequency.

1(a):

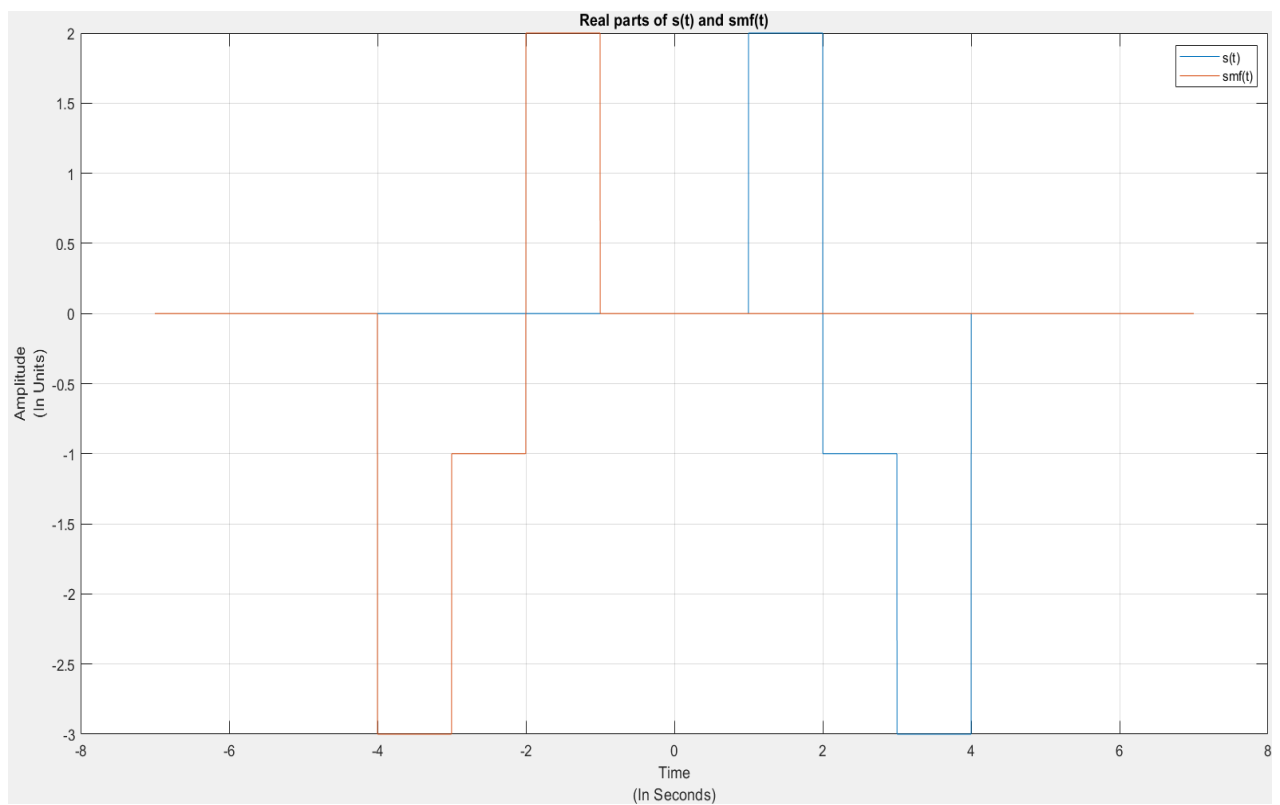


1(b):

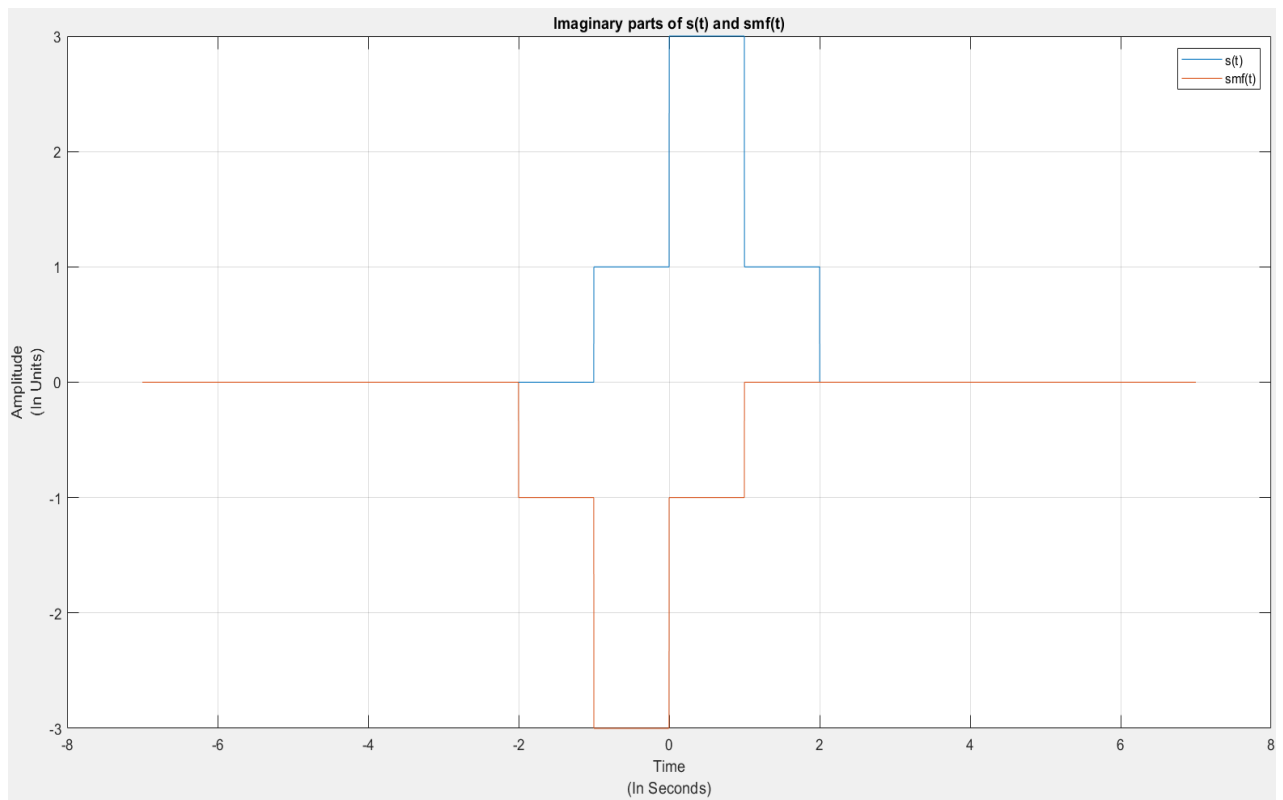


Ans: Peak can be seen at $t = 0$ s in the plot.

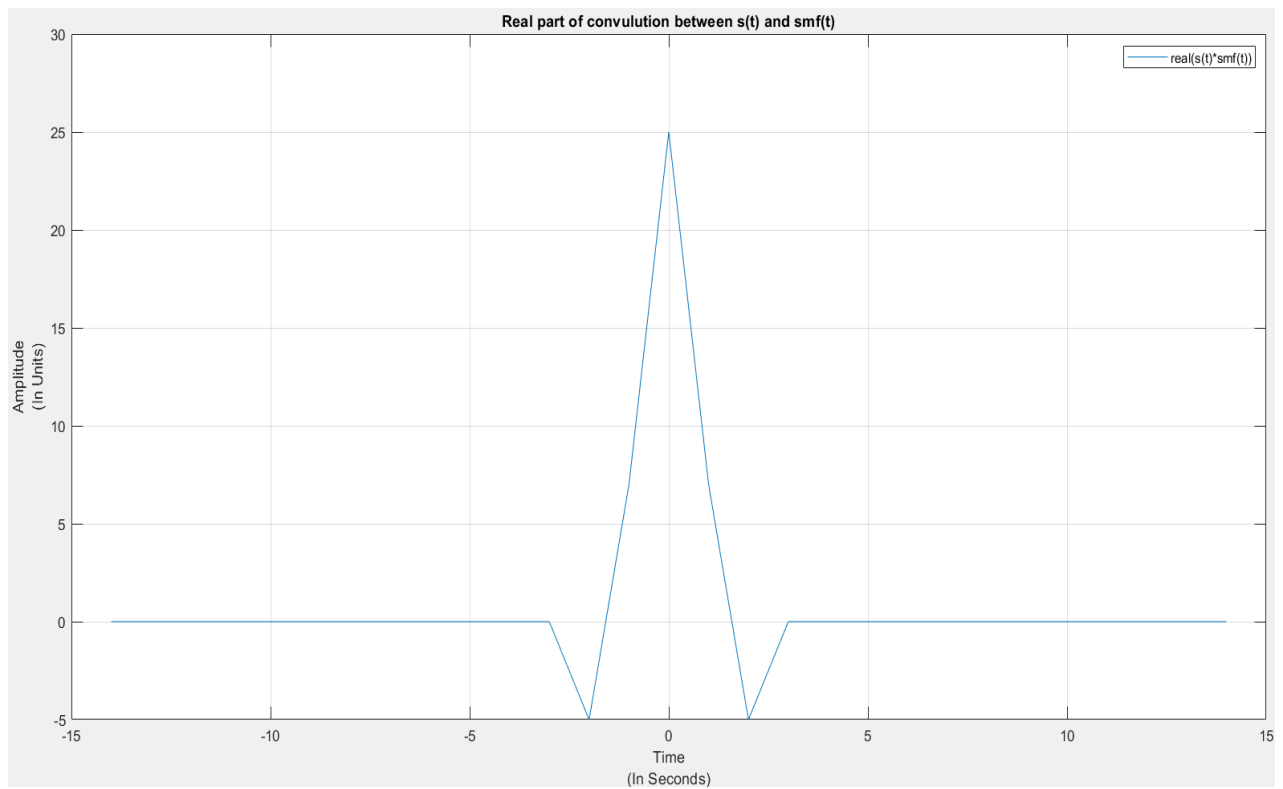
1(c): (i)



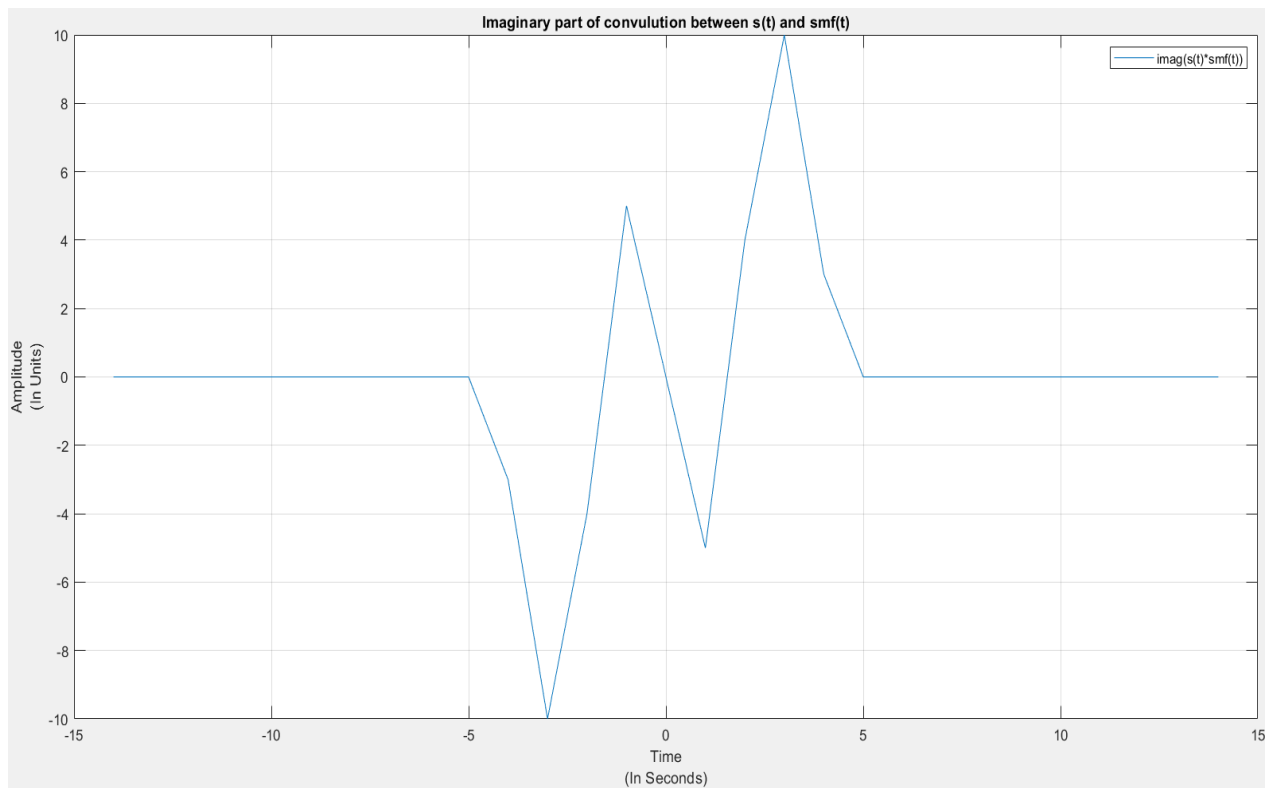
1(c): (ii)



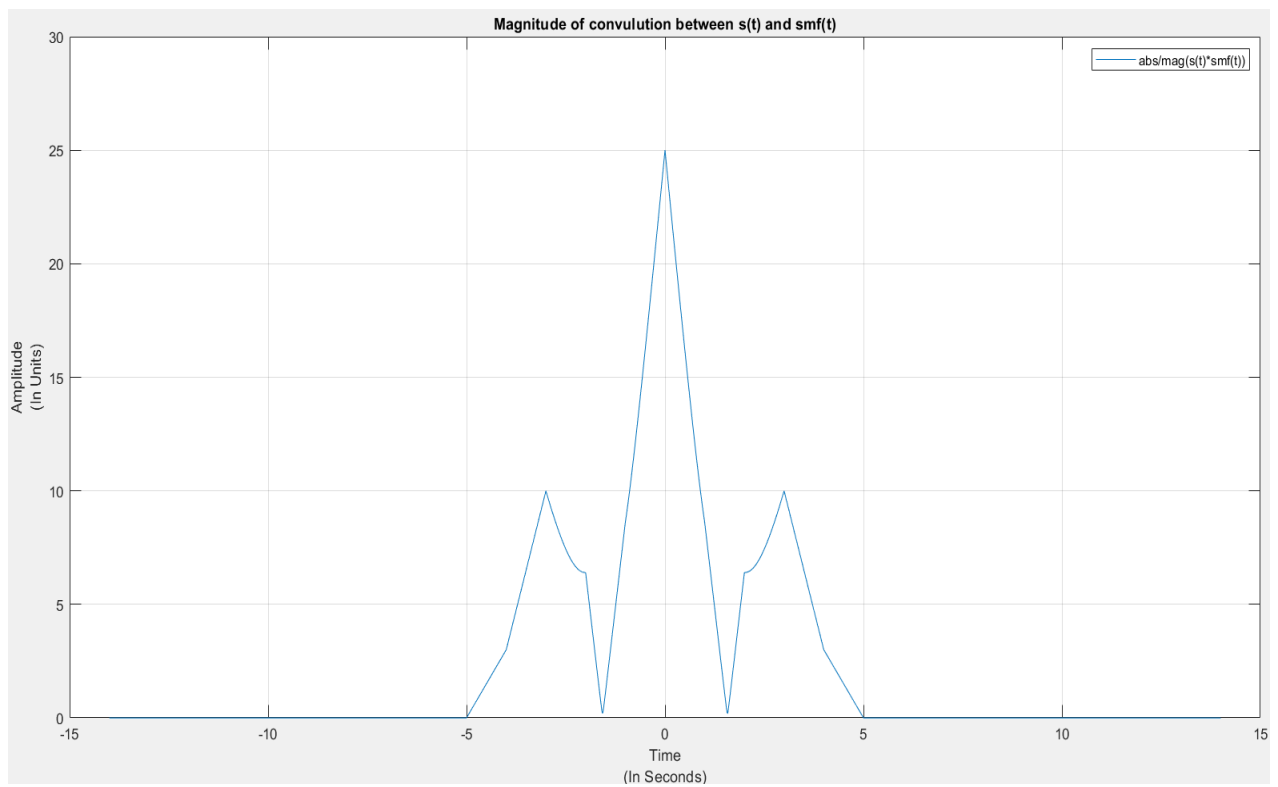
1(d): (i)



1(d): (ii)

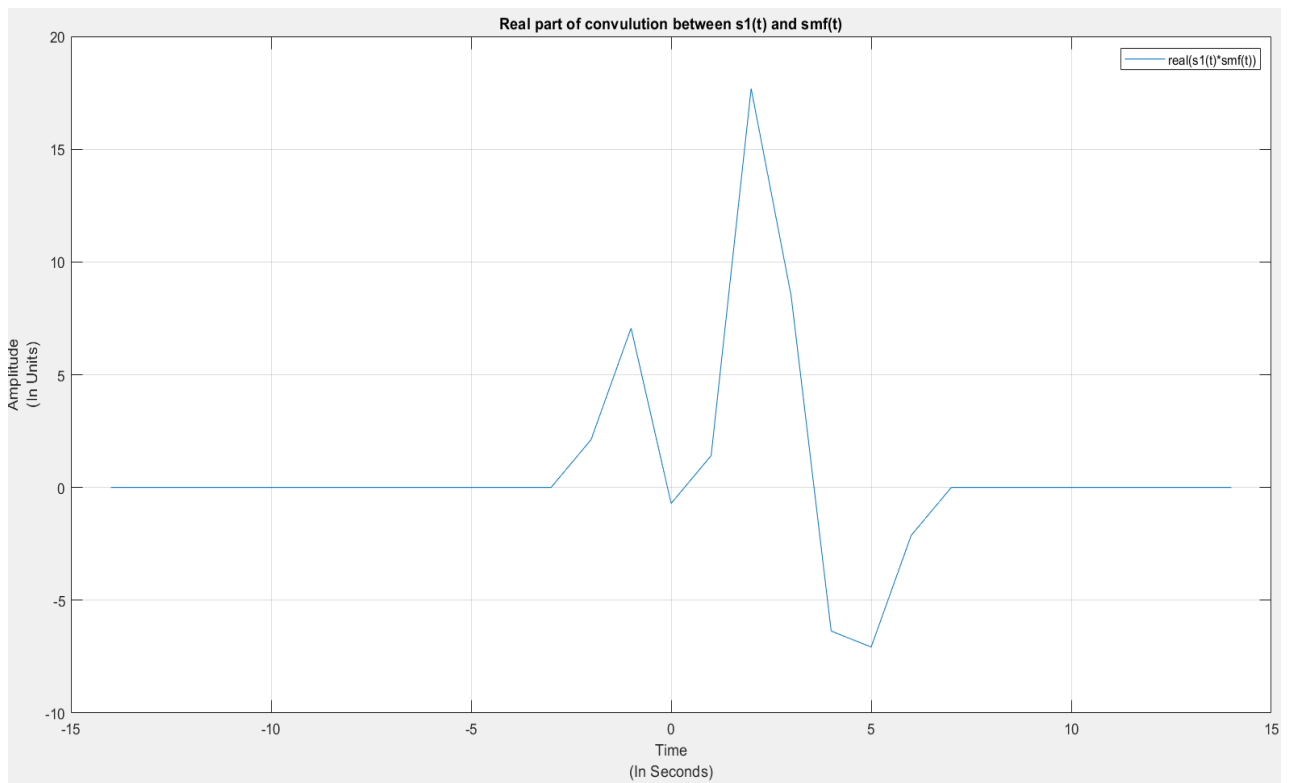


1(d): (iii)

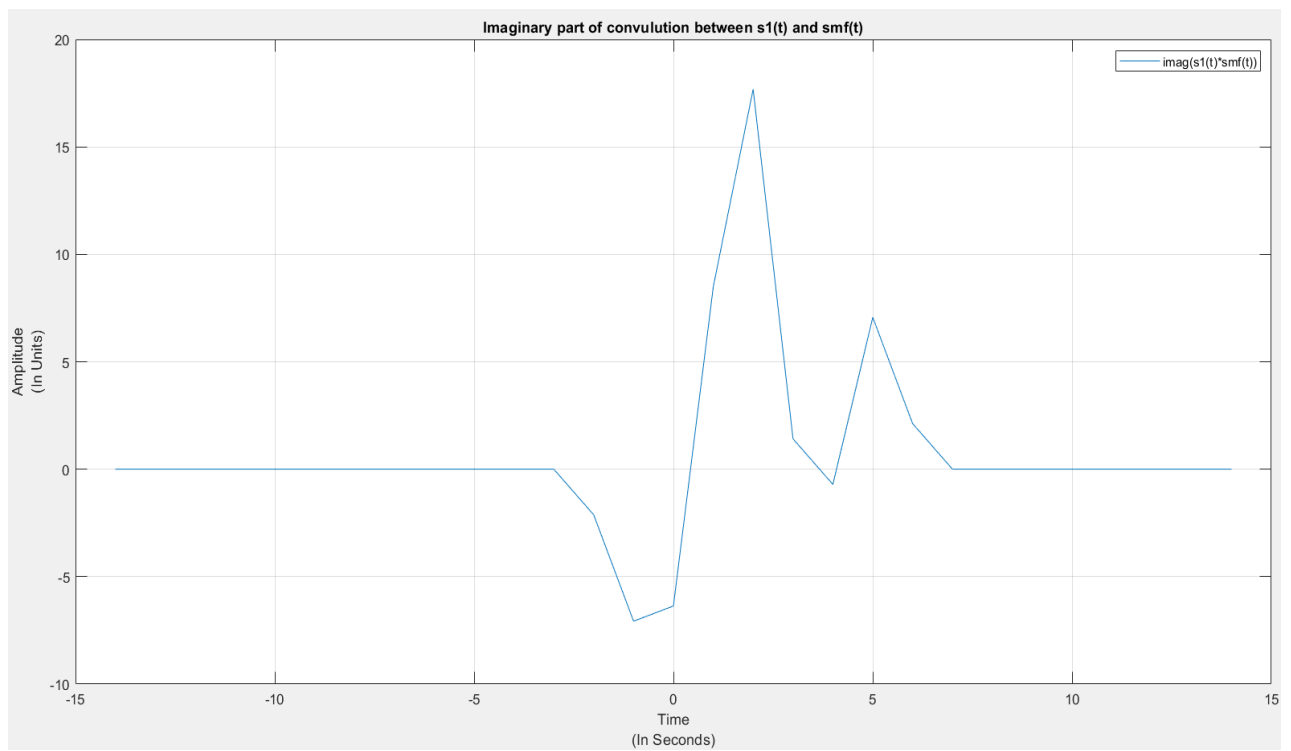


Ans: Peak can be seen in plot-7 at $t = 0$ s.

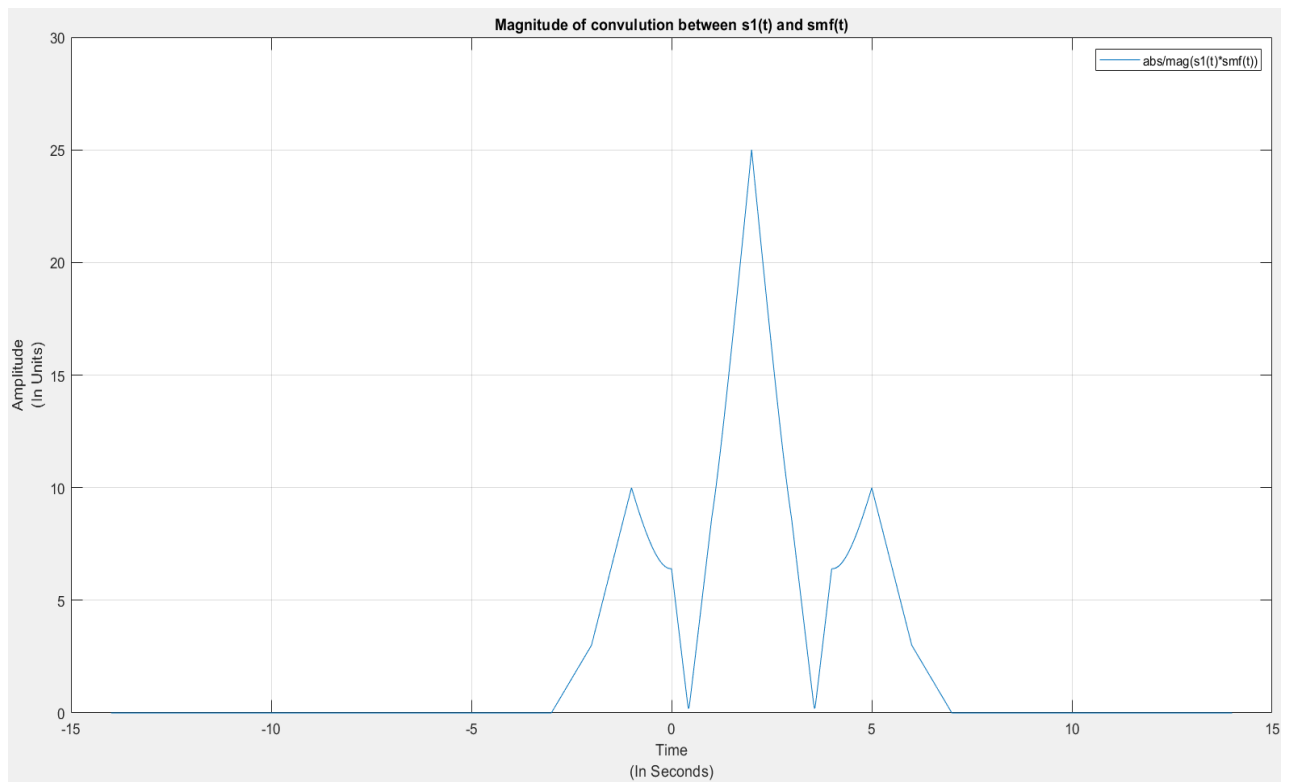
1(e): (i)



1(e): (ii)



1(e): (iii)



Ans: Peak can be seen in plot-10 at $t = 2s$.

1(f):

Let $y(t) = s(t) * \text{smf}(t)$. (Here $y(t)$ is output of convolution and '*' here means convolution)

$$s_1(t) * \text{smf}(t) = \{e^{i\theta} \{s(t - t_0)\} * \text{smf}(t)\} \quad \text{Eq(1)}$$

We know that,

There is a shift in the convolution of $s_1(t)$ and $\text{smf}(t)$ (Plot-10 shifted by 2 unit compared to Plot-7) that shift corresponds to the shift in $s_1(t-t_0)$. In our case $t_0 = 2$.

Eq(1):

$$= [y(t - t_0)]x[e^{i(\theta)}]$$

$$= [y(t - t_0)]x[\cos(\theta) + i \sin(\theta)] \quad (\text{here } e^{i(\theta)} = \cos(\theta) + i \sin(\theta))$$

$$= \{[\text{real}(y(t-t_0))]x[\cos(\theta)] - [\text{imag}(y(t-t_0))]x[\sin(\theta)]\} + i \{[\text{real}(y(t-t_0))]x[\sin(\theta)] + [\text{imag}(y(t-t_0))]x[\cos(\theta)]\}$$

Now, in the imaginary graph of $y(t)$ find t' where the $\text{imag}(y(t')) = 0$. Substitute $t = t' + t_0$ in above equation. And in the real graph of the equation get the value at $t = t' + t_0$. Divide this value with the value of the real graph of $y(t)$ at $t = t'$. The ratio will be equal to $\cos(\theta)$.

Similarly, we get the time t'' in the real graph of $y(t)$ where the value is 0. We substitute $t = t'' + t_0$ in the above equation and divide it with real graph of $y(t)$ at $t = t''$. This gives us the $-\sin(\theta)$.

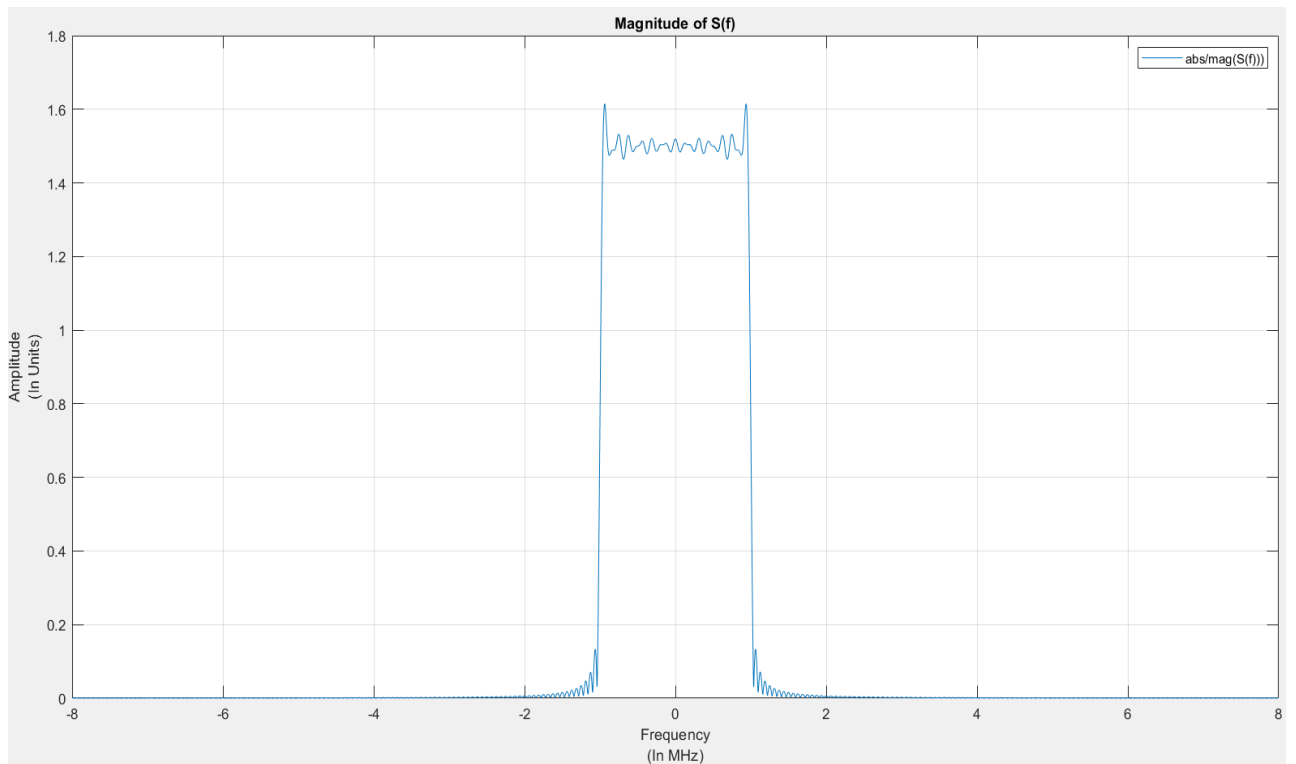
Final Conclusion:

$$\theta = \tan^{-1}(\sin(\theta)/\cos(\theta))$$

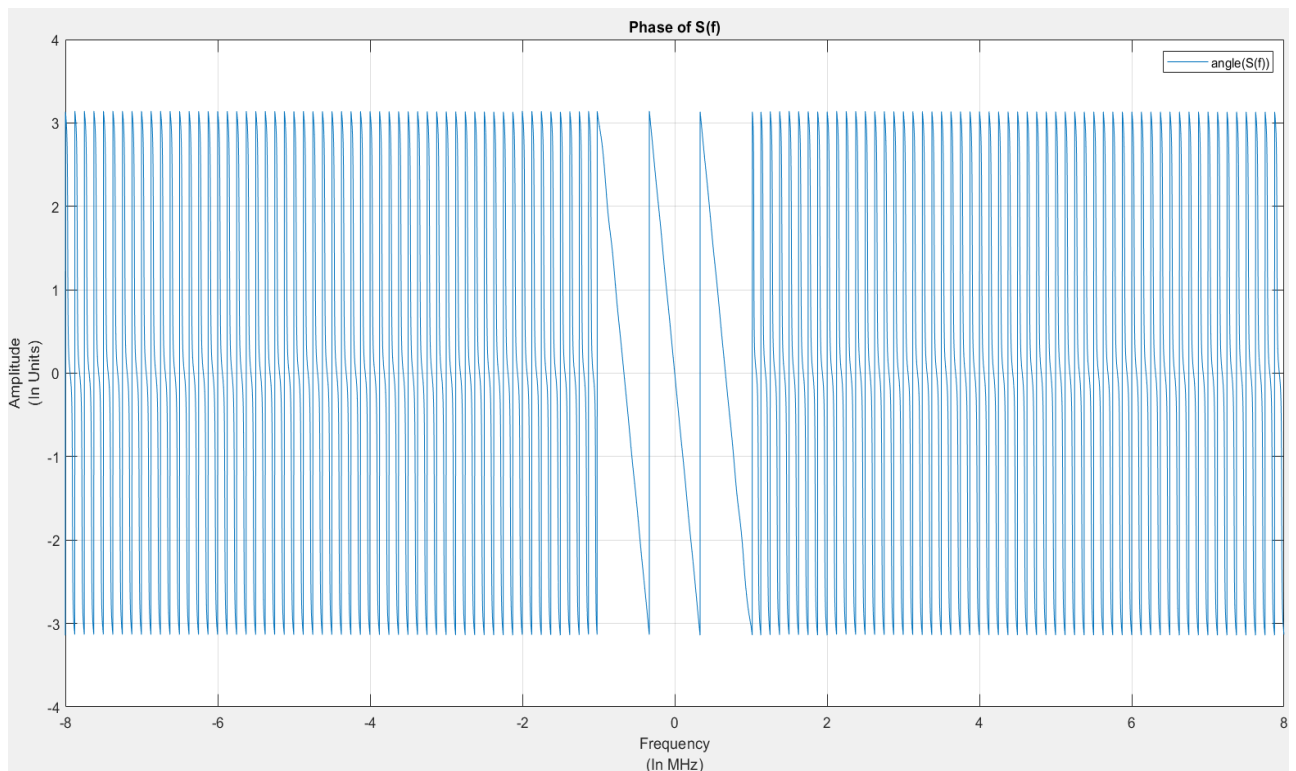
Answer to Q2

Note: For question 2 $s(t) = 3\text{sinc}(2t - 3)$, unit time is microsecond and signal is sampled at 16MHz and is truncated to the range $[-8,8]$ and desired frequency is 1 MHz. Here $S(f)$ is fourier transform of $s(t)$

2(b):



2(c):



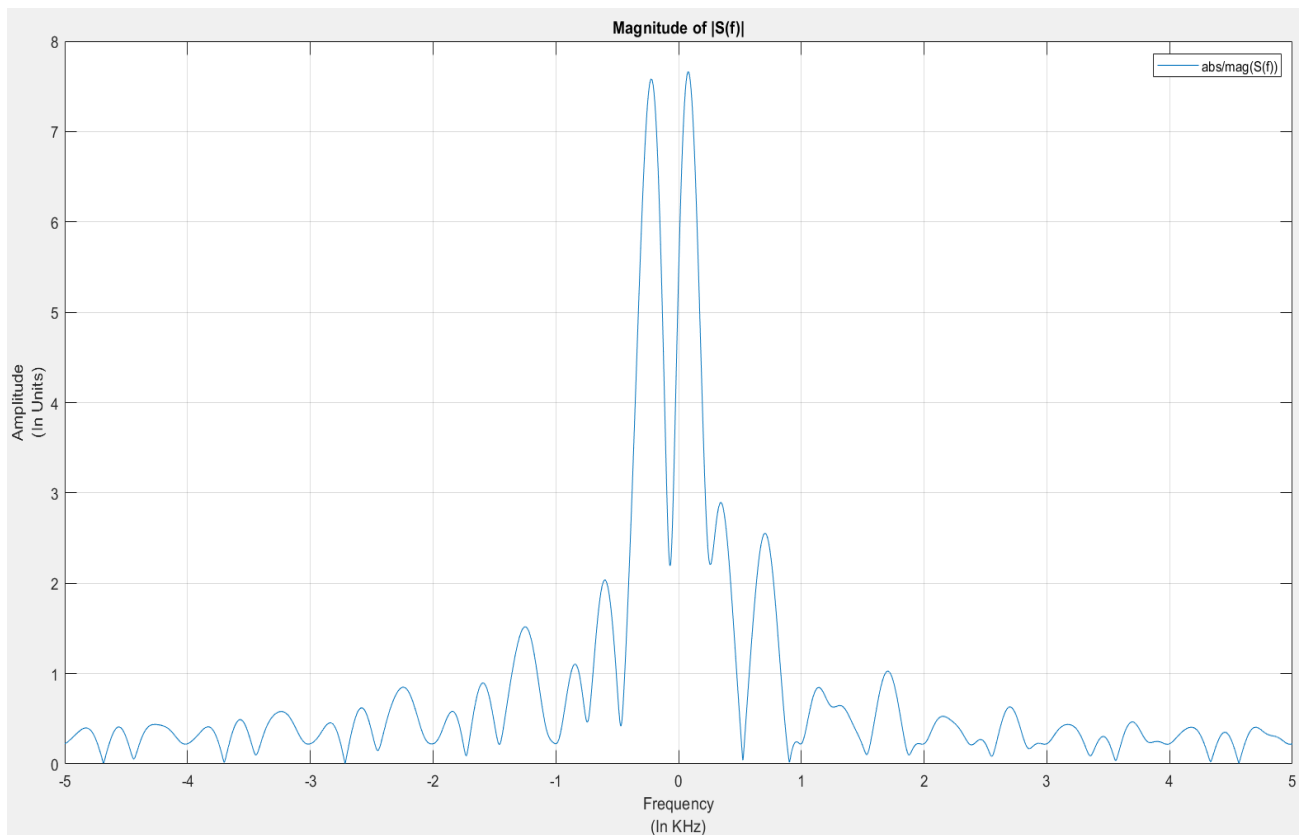
Ans: The range of frequencies over which phase plot has meaning is $[-1, 1]$ MHz.

Answer to Q3

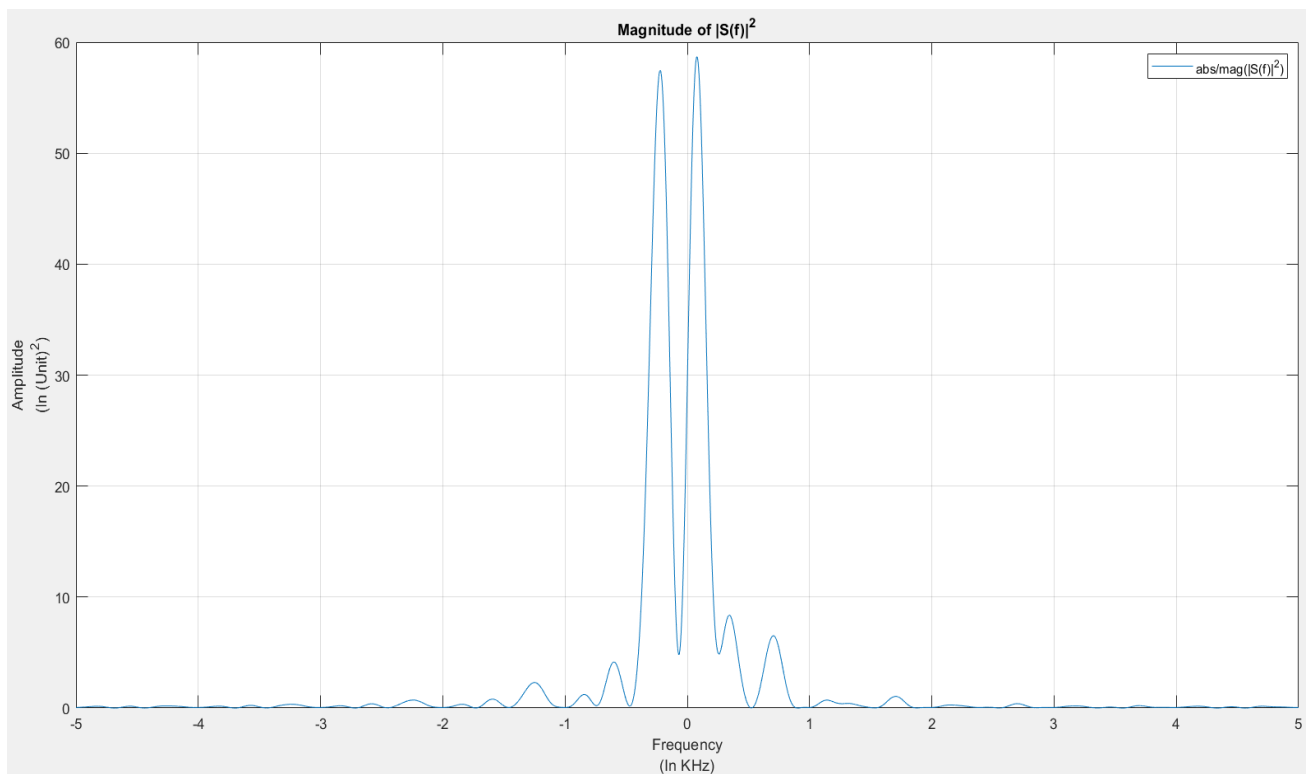
Note: For question 3 we will consider $s(t)$ from question 1 ie $s(t) = u(t) + j v(t)$ where $u(t) = 2I[1,3](t) - 3I[2,4](t)$, $v(t) = I[-1,2](t) + 2I[0,1](t)$ and $smf = s^*(-t)$,

‘*’ here means conjugate. Here unit time is millisecond and is sampled at 10MHz

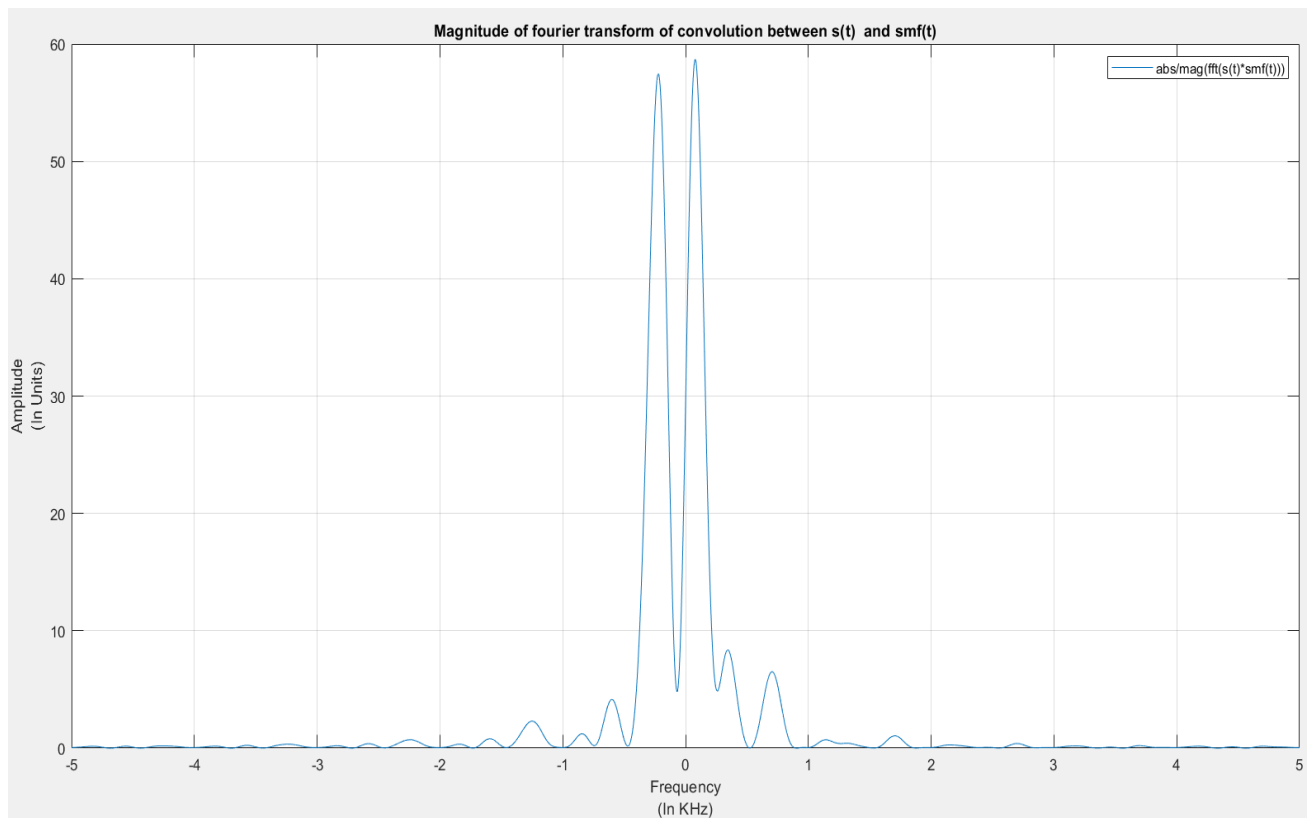
3(a):



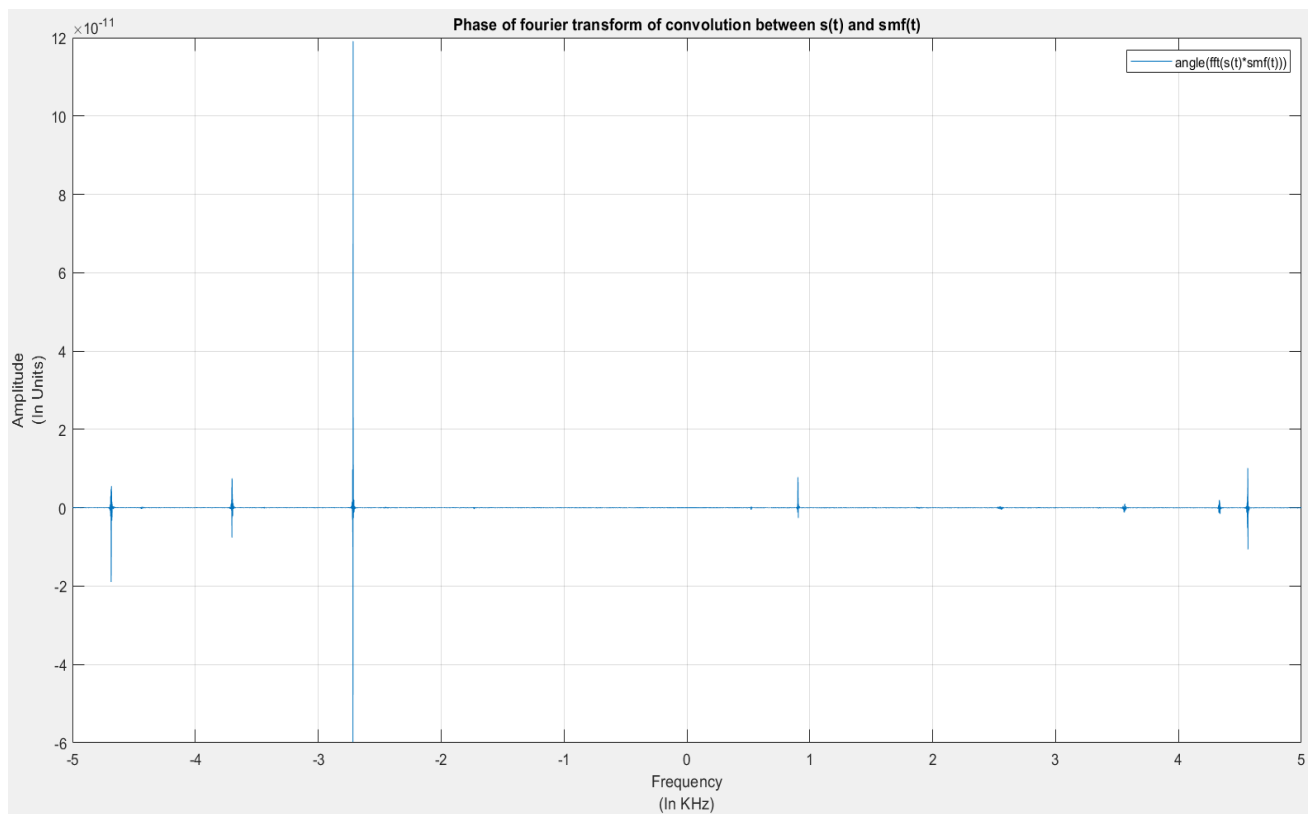
3(a2): The 3(b) Plot ie Plot-14 matches with this plot that means magnitude of convolution and $|S(f)^2|$ is same.



3(b):



3(c):



Ans: The convolution in time domain is nothing but multiplication in frequency domain so $s(t)*\text{smf}(t) \rightarrow (\text{fft})$ gives us $S(f) \times S(-f)$ (because $\text{smf}(t) = s^*(-t)$ and we know that $s^*(-t) \rightarrow (\text{fft})$ is $S(-f)$) so we can see that $S(f)$ and $S(-f)$ have opposite in sign and equal in magnitude phase so we get all the phase cancelled and phase plot is almost zero in amplitude.

Appendix:

Code for Q1 :

```
dt = 0.001;
t = -5:dt:5;
u = signalx(t);
umf = signalx(-t);
figure(1);
plot(t,u);
hold on;
plot(t,umf);
hold off;
xlabel({'Time', '(In Seconds)'}); ylabel({'Amplitude', '(In Units)'});
title('Plot of u(t) and umf(t)');
legend('u(t)', 'umf(t)');
grid on;
[y,t1] = contconv(double(u),double(umf),t(1),t(1),dt);
figure(2);
plot(t1,y);
xlabel({'Time', '(In Seconds)'}); ylabel({'Amplitude', '(In Units)'});
title('Convolution of u(t) and umf(t)');
legend('u(t)*umf(t)');
grid on;
function u = signalx(t)
syms x;
y = piecewise(1 <= x <= 2, 2, 2 <= x <= 3, -1, 3 <= x <= 4, -3, 0);
u = subs(y,x,t);
end
function [y,t] = contconv(x1,x2,s1,s2,dt)
y = conv(x1,x2)*dt;
s1_2 = s1 + (length(x1)-1)*dt;
s2_2 = s2 + (length(x2)-1)*dt;
t1 = s1+ s2;
t2 = s2_2 + s1_2;
t = t1:dt:t2;
end
```

```
dt = 0.001;
to = 2;
theta = pi/4;
t = -7:dt:7;
u = signalx(t);
v = signalx1(t);
s = u + 1i*v;
sc = signalx(-t) - 1i*signalx1(-t);
figure(3);
plot(t,double(real(s)));
hold on;
```

```

plot(t,double(real(sc)));
hold off;
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Real parts of s(t) and smf(t)');
legend('s(t)','smf(t)');
grid on;
figure(4);
plot(t,imag(s));
hold on;
plot(t,imag(sc));
hold off;
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Imaginary parts of s(t) and smf(t)');
legend('s(t)','smf(t)');
grid on;
[s_c,t1] = contconv(double(s),double(sc),t(1),t(1),dt);
figure(5);
plot(t1,real(s_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Real part of convolution between s(t) and smf(t)');
legend('real(s(t)*smf(t))');
grid on;
figure(6);
plot(t1,imag(s_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Imaginary part of convolution between s(t) and smf(t)');
legend('imag(s(t)*smf(t))');
grid on;
figure(7);
plot(t1,abs(s_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Magnitude of convolution between s(t) and smf(t)');
legend('abs/mag(s(t)*smf(t))');
grid on;
s1 = (signalx(t-2) + 1i*signalx1(t-2))*exp(1i*theta);
[s1_c,t3] = contconv(double(s1),double(sc),t(1),t(1),dt);
figure(8);
plot(t3,real(s1_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Real part of convolution between s1(t) and smf(t)');
legend('real(s1(t)*smf(t))');
grid on;
figure(9);
plot(t3,imag(s1_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Imaginary part of convolution between s1(t) and smf(t)');
legend('imag(s1(t)*smf(t))');
grid on;
figure(10);
plot(t3,abs(s1_c));
xlabel({'Time','(In Seconds)'});ylabel({'Amplitude','(In Units)'});
title('Magnitude of convolution between s1(t) and smf(t)');
legend('abs/mag(s1(t)*smf(t))');
xt = get(gca, 'XTick');
set(gca, 'XTick',xt, 'XTickLabel',xt/1)
grid on;
grid minor;
function u = signalx(t)
syms x;
y = piecewise(1 <= x <= 2, 2, 2 <= x <= 3, -1, 3 <= x <= 4, -3, 0);
u = subs(y,x,t);

```

```

end
function v = signalx1(t)
syms x;
y = piecewise(-1 <= x <= 0, 1, 0 <= x <= 1, 3, 1 <= x <= 2, 1, 0);
v = subs(y,x,t);
end
function [y,t] = contconv(x1,x2,s1,s2,dt)
y = conv(x1,x2)*dt;
s1_2 = s1 + (length(x1)-1)*dt;
s2_2 = s2 + (length(x2)-1)*dt;
t1 = s1+ s2;
t2 = s2_2 + s1_2;
t = t1:dt:t2;
end

```

Code for Q2 and Q3 :

```

function two
q_2();
q_3();
end
function q_2
dt = (1/16);
t = -8:dt:8;
s = 3*sinc(2*t - 3);
[Y,f,df] = contFT(s,t(1),dt,10^(-3));
figure(11);
plot(f,abs(Y));
xlabel({'Frequency','(In MHz)'});ylabel({'Amplitude','(In Units)'});
title('Magnitude of S(f)');
legend('abs/mag(S(f))');
grid on;
figure(12);
plot(f,angle(Y));
xlabel({'Frequency','(In MHz)'});ylabel({'Amplitude','(In Units)'});
title('Phase of S(f)');
legend('angle(S(f))');
grid on;
grid minor;
end
function q_3
dt = 0.1;
to = 2;
theta = pi/4;
t = -7:dt:7;
u = signalx(t);
v = signalx1(t);
s = u + 1i*v;
sc = signalx(-t) - 1i*signalx1(-t);
[s_c,t1] = contconv(double(s),double(sc),t(1),t(1),dt);
[S,f,df] = contFT(double(s),t(1),dt,10^(-3));
[S_C,F,DF] = contFT(double(s_c),t1(1),dt,10^(-3));
figure(13);
plot(f,abs(S));
xlabel({'Frequency','(In KHz)'});ylabel({'Amplitude','(In Units)'});
title('Magnitude of |S(f)|');
legend('abs/mag(S(f))');
grid on;
figure(14);

```

```

plot(F,abs(S_C));
xlabel({'Frequency','(In KHz)'});ylabel({'Amplitude','(In Units)'});
title('Magnitude of fourier transform of convolution between s(t) and
smf(t)');
legend('abs/mag(fft(s(t)*smf(t)))');
grid on;
figure(15);
plot(F,angle(S_C));
xlabel({'Frequency','(In KHz)'});ylabel({'Amplitude','(In Units)'});
title('Phase of fourier transform of convolution between s(t) and smf(t)');
legend('angle(fft(s(t)*smf(t)))');
grid on;
end
function [y,t] = contconv(x1,x2,s1,s2,dt)
y = conv(x1,x2)*dt;
s1_2 = s1 + (length(x1)-1)*dt;
s2_2 = s2 + (length(x2)-1)*dt;
t1 = s1+ s2;
t2 = s2_2 + s1_2;
t = t1:dt:t2;
end
function u = signalx(t)
syms x;
y = piecewise(1 <= x <= 2, 2, 2 <= x <= 3, -1, 3 <= x <= 4, -3, 0);
u = subs(y,x,t);
end
function v = signalx1(t)
syms x;
y = piecewise(-1 <= x <= 0, 1, 0 <= x <= 1, 3, 1 <= x <= 2, 1, 0);
v = subs(y,x,t);
end
function [X,f,df] = contFT(x,tstart,dt,df_desired)
%Use Matlab DFT for approximate computation of continuous time Fourier
transform
%INPUTS
%x = vector of time domain samples, assumed uniformly spaced %tstart= time
at which first sample is taken
%dt = spacing between samples
%df_desired = desired frequency resolution
%OUTPUTS
% X=vector of samples of Fourier transform
%f=corresponding vector of frequencies at which samples are obtained
%df=freq resolution attained (redundant--already available from %difference
of consecutive entries of f
%%%%%%%%%
%minimum FFT size determined by desired freq res or length of x
Nmin=max(ceil(1/(df_desired*dt)),length(x));
%choose FFT size to be the next power of 2
Nfft = 2^(nextpow2(Nmin));
%compute Fourier transform, centering around DC
X=dt*fftshift(fft(x,Nfft));
%achieved frequency resolution
df=1/(Nfft*dt);
%range of frequencies covered
f = ((0:Nfft-1)-Nfft/2)*df;
%same as f=-1/(2*dt):df:1/(2*dt) - df %phase shift associated with start
time
X=X.*exp(-1i*2*pi*f*tstart);
end

```

