
Lab 4. Discrete Time Fourier Transform

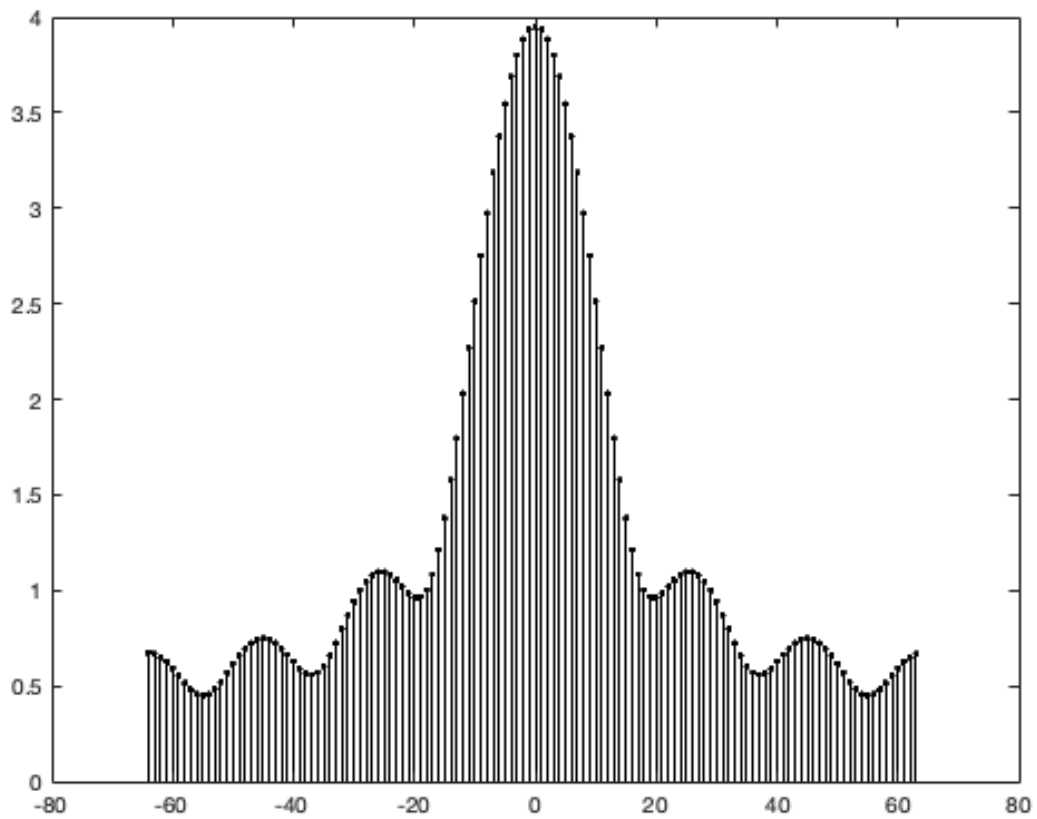
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A. Discrete-TimeFourierTransform(DTFT)

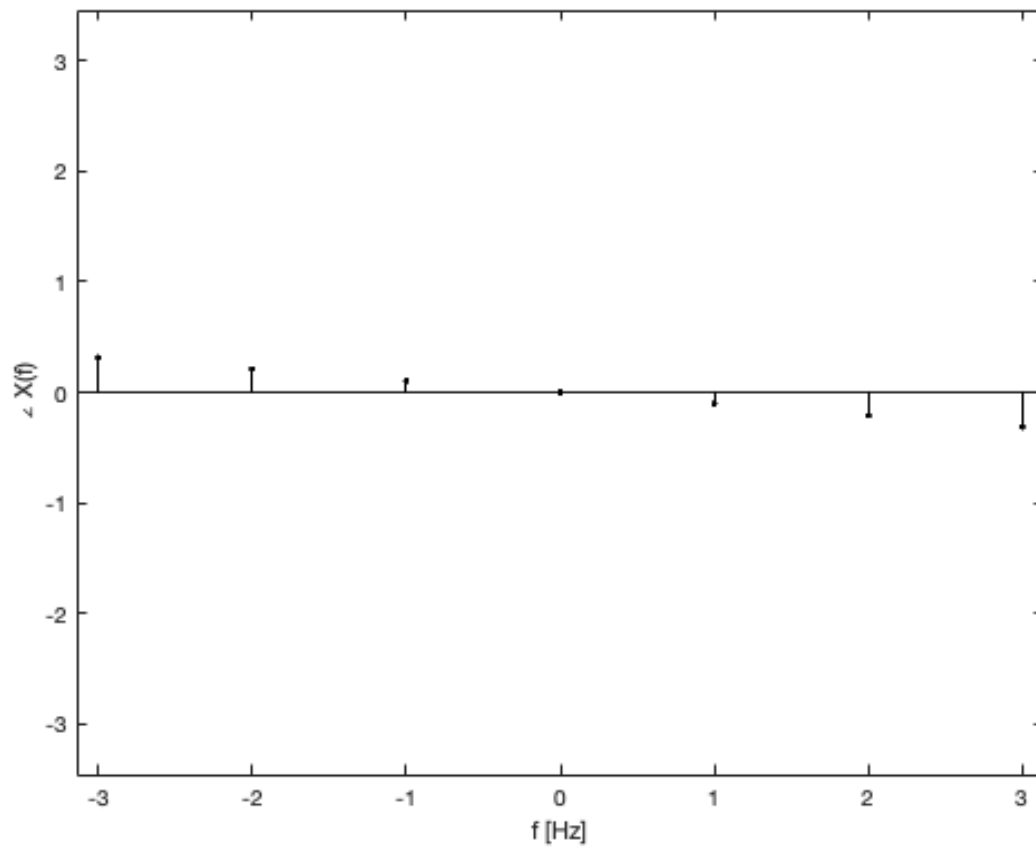
1.

```
N_0 = 128; T = 1/128; n = (0:N_0-1);  
x = (0.8.^n).*(1.0.*(n>=0 & n < 7));clc  
X = fft(x);  
  
f = (0:N_0-1)/(T*N_0); stem(f,abs(X),'k. ');  
axis([0 50 0 5]); xlabel('f [Hz]'); ylabel('|X(f)|');  
  
stem(f-1/(T*2),fftshift(abs(X)),'k.');
```



2.

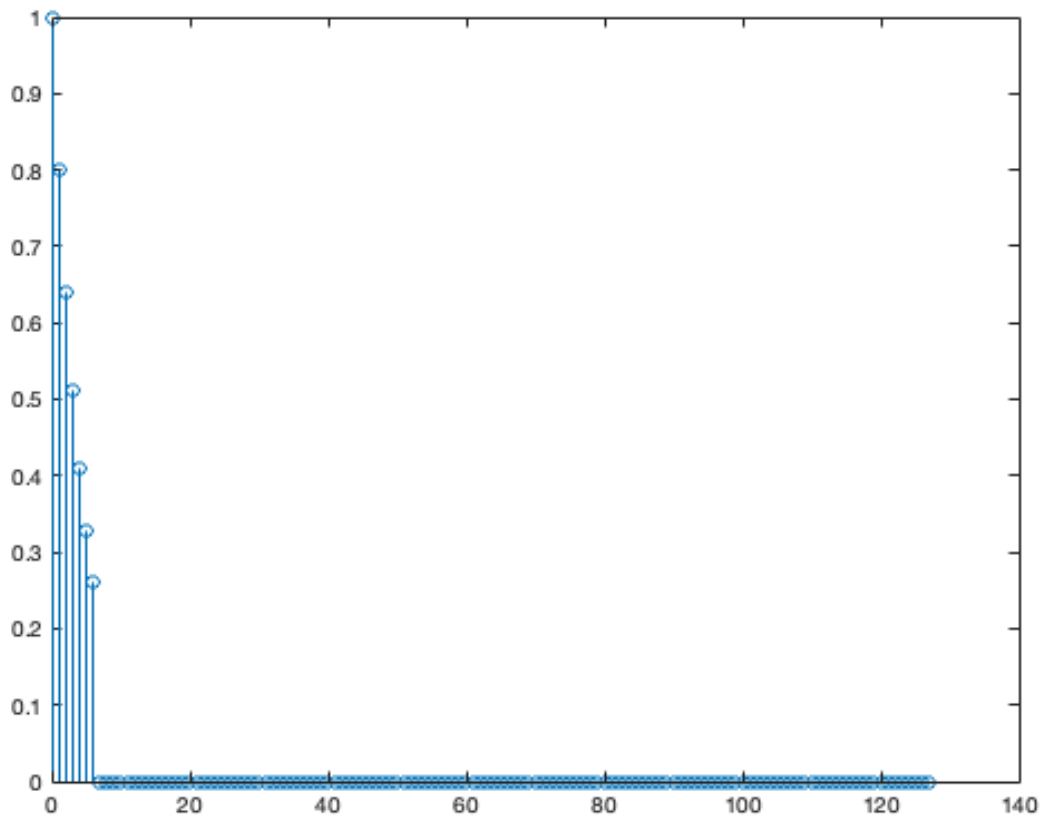
```
stem(f-1/(T*2),fftshift(angle(X)), 'k. ');  
axis([-pi pi -1.1*pi 1.1*pi]); xlabel('f [Hz]'); ylabel('\angle X(f)');
```



3.

```
x_2 = ifft(X);  
stem(n,x_2);
```

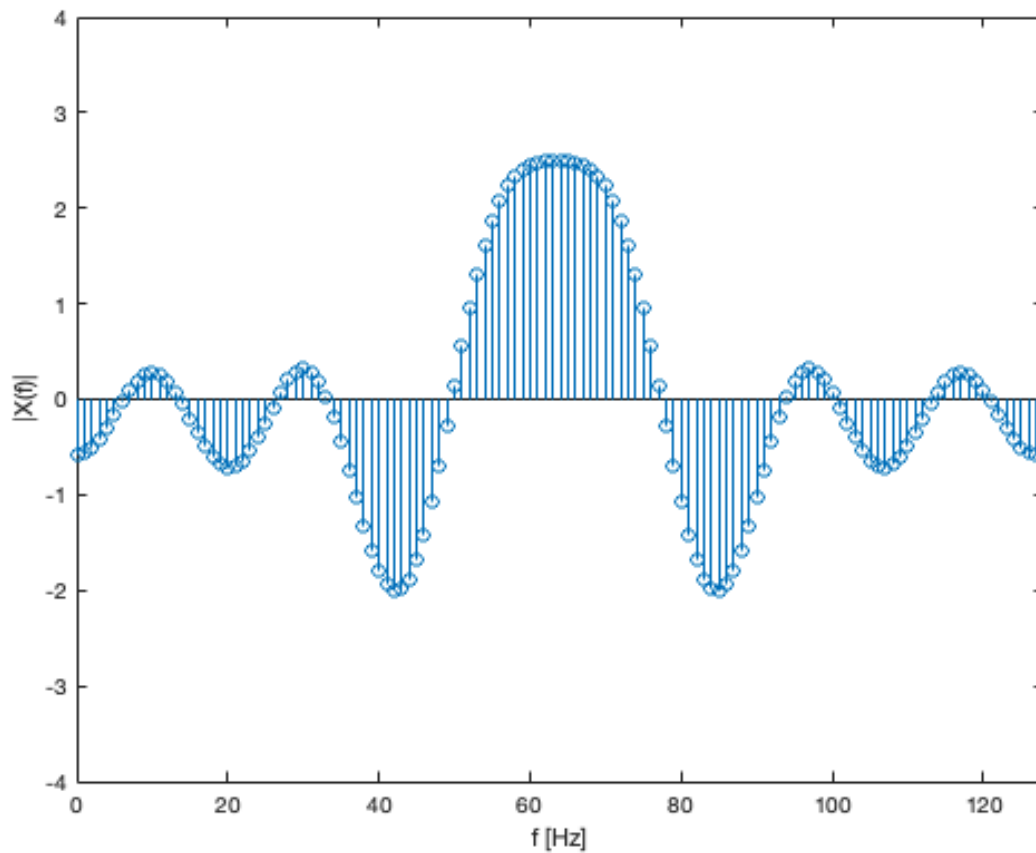
```
% Yes the signal is the same because we are taking the inverse fourier  
% transform
```



B. TimeConvolution

1.

```
x_1 = sin(2*pi*n/10).*(1.0.*(n>=0 & n < 7));  
  
omega= linspace(-pi,pi,128);  
W_omega = exp(-1i).^((0:length(x_1)-1)'*omega);  
CONV = (x_1*W_omega);  
  
stem(n,CONV);  
axis([0 128 -4 4]); xlabel('f [Hz]'); ylabel('|X(f)|');  
  
Warning: Using only the real component of complex data.
```



2.

```
h = 1.0.*(n>=0 & n<10);
```

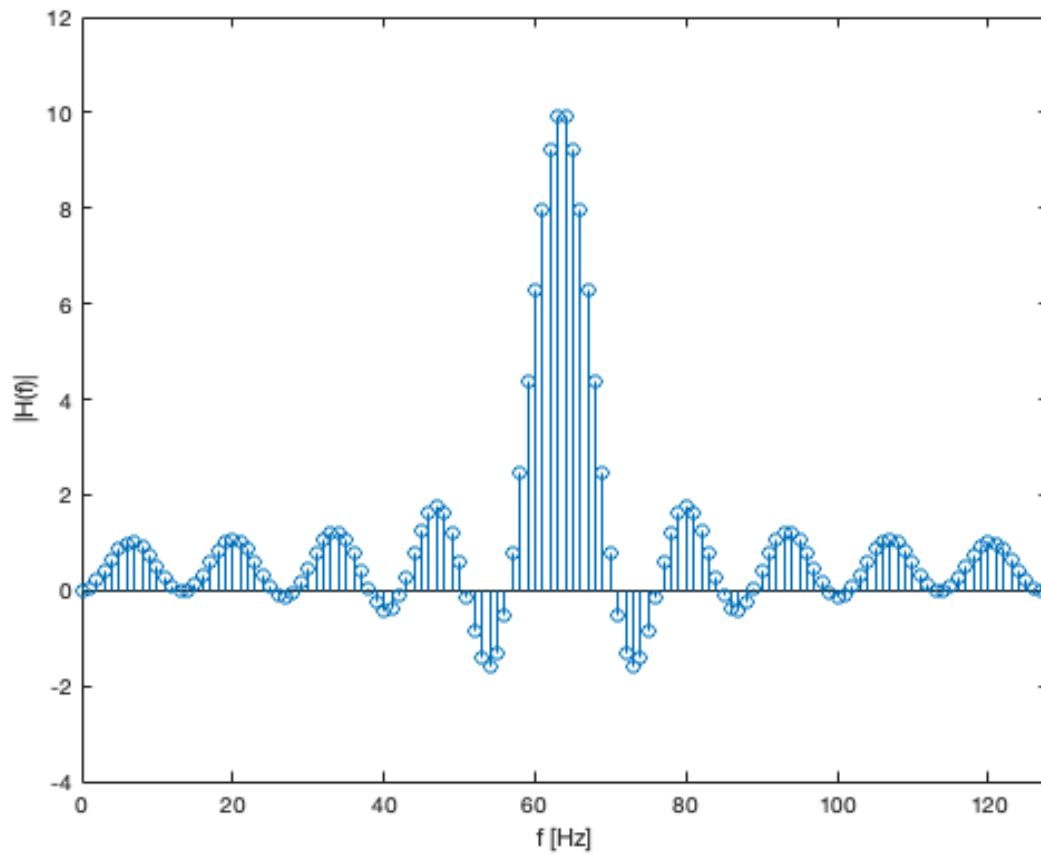
```
W_omega2 = exp(-1i).^((0:length(h)-1)*omega);
```

```
CONV_2 = (h*W_omega2);
```

```
stem(n,CONV_2);
```

```
axis([0 128 -4 12]); xlabel('f [Hz]'); ylabel('|H(f)|');
```

Warning: Using only the real component of complex data.



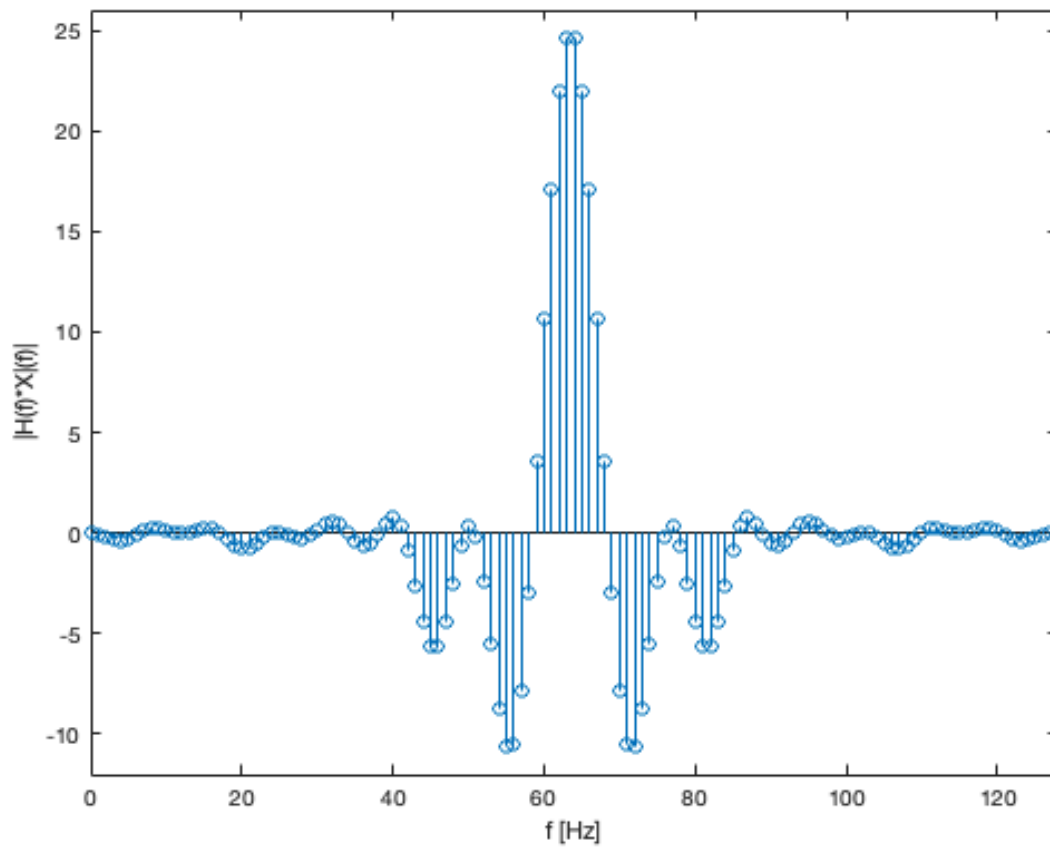
3.

```
CONV_3 = (CONV.*CONV_2);
```

```
stem(n,CONV_3);
```

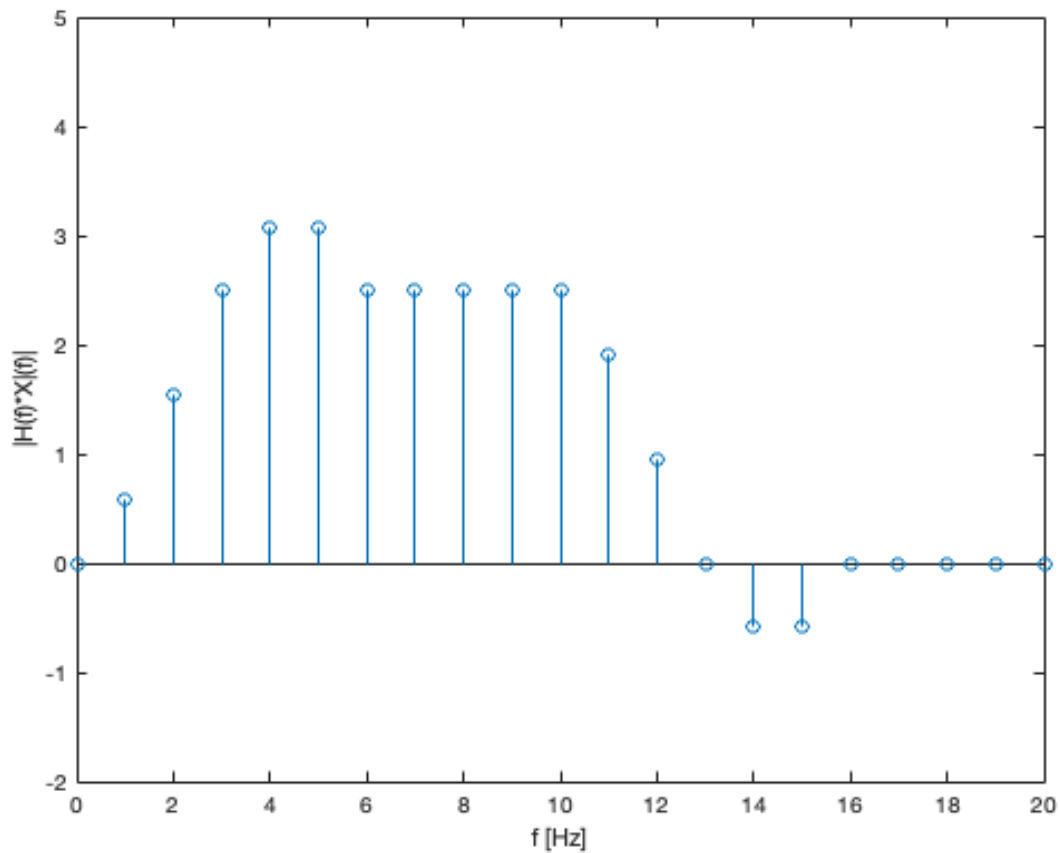
```
axis([0 128 -12 26]); xlabel('f [Hz]'); ylabel('|H(f)*X(f)|');
```

Warning: Using only the real component of complex data.



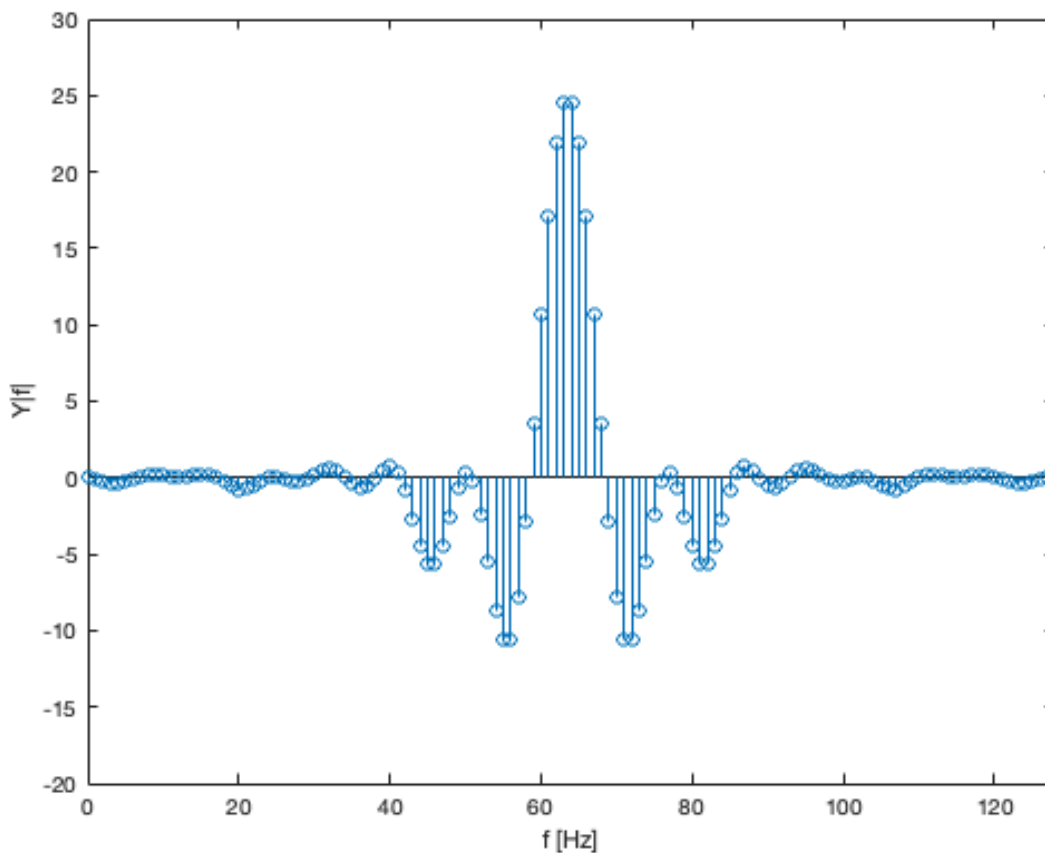
4.

```
n = (0:254);  
y = conv(x_1,h);  
  
stem(n,y);  
axis([0 20 -2 5]); xlabel('f [Hz]'); ylabel('|H(f)*X(f)|');
```



5.

```
W_omega3 = exp(-1i).^((0:length(y)-1)*omega);  
y_dtft = (y*W_omega3);  
n = 0:127;  
  
stem(n,y_dtft);  
axis([0 128 -20 30]); xlabel('f [Hz]'); ylabel('Y|f|');  
  
Warning: Using only the real component of complex data.
```

6. Yes the results of 3 and 5 are the same. This is because in 3 we first turned the x and h from time to frequency domain and did multiplication and in 5 we did convolution of the result y .

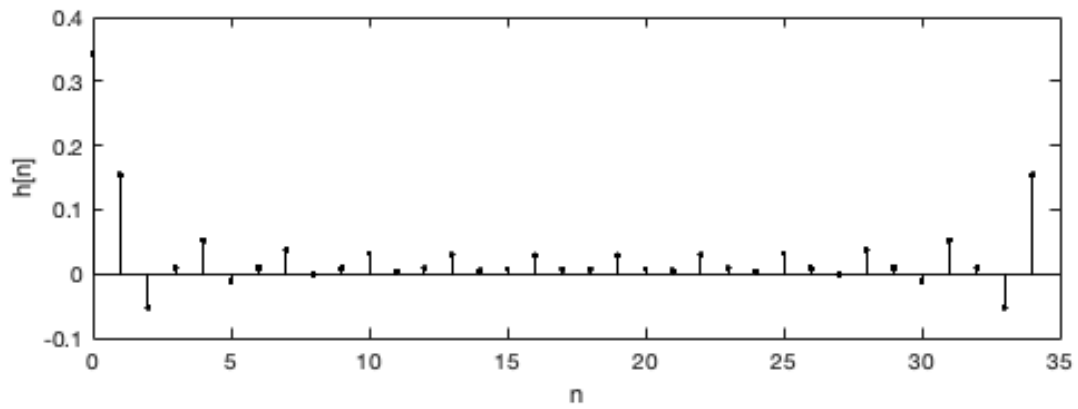
C. FIR Filter Design by Frequency Sampling

1.

```
H_d = @(Omega) (mod(Omega,2*pi)<2*pi/3)+(mod(Omega,2*pi/3)>2*pi-2*pi/3);  
N = 35;
```

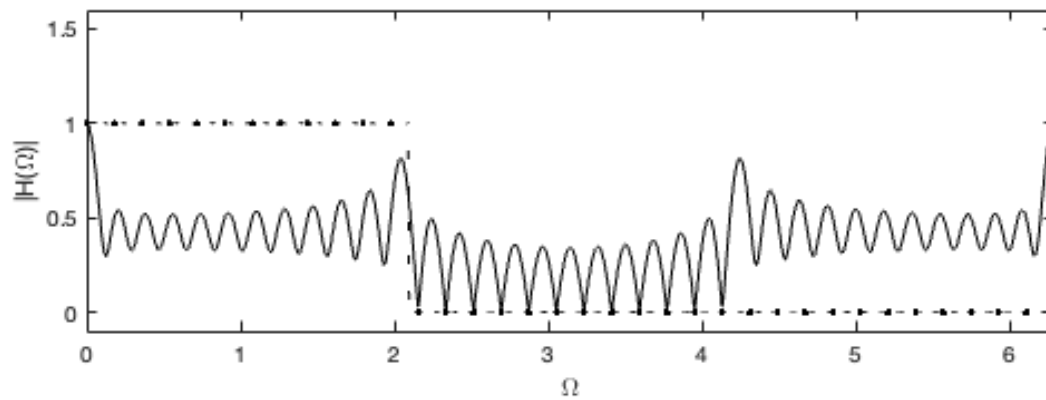
```
h = CH9MP1(N,H_d);  
Omega = linspace(0,2*pi,1000); samples = linspace(0,2*pi*(1-1/N),N);  
H = CH5MP1(h,1,Omega);
```

```
subplot(2,1,1); stem([0:N-1],h,'k. '); xlabel('n'); ylabel('h[n]');
```



2.

```
plot(samples,H_d(samples),'k.',Omega,H_d(Omega),'k:',Omega,abs(H),'k');  
axis([0 2*pi -0.1 1.6]); xlabel('\Omega'); ylabel('|H(\Omega)|');
```



3.

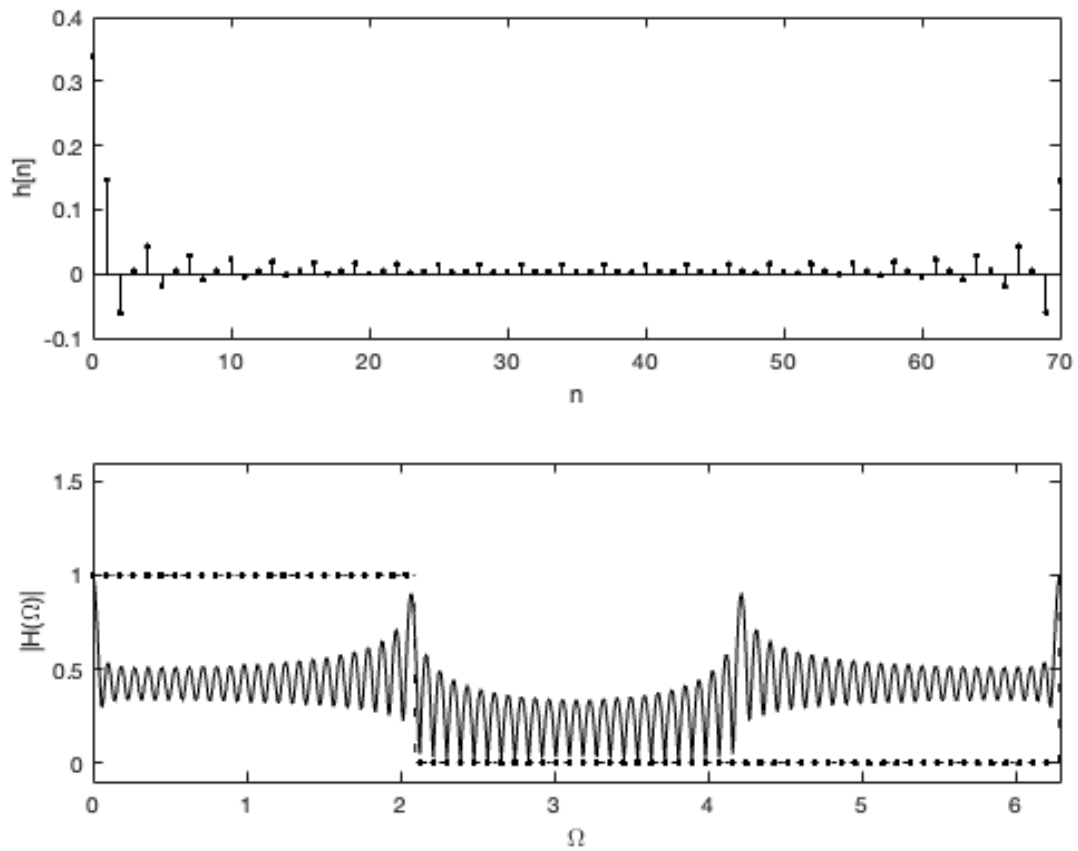
`% The results are pretty similar`

4.

```
H_d = @(Omega) (mod(Omega,2*pi)<2*pi/3)+(mod(Omega,2*pi/3)>2*pi-2*pi/3);
N = 71;
```

```
h = CH9MP1(N,H_d);
Omega = linspace(0,2*pi,1000); samples = linspace(0,2*pi*(1-1/N),N);
H = CH5MP1(h,1,Omega);
```

```
subplot(2,1,1); stem([0:N-1],h,'k. '); xlabel('n'); ylabel('h[n]');
subplot(2,1,2);
plot(samples,H_d(samples),'k.',Omega,H_d(Omega),'k:',Omega,abs(H),'k');
axis([0 2*pi -0.1 1.6]); xlabel('\Omega'); ylabel('|H(\Omega)|');
```



5.

`% With the number increase the length of filter increases too`

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