

DFT Octave Codes (0A)

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Based on
M.J. Roberts, Fundamentals of Signals and Systems

B.D Storey, Computing Fourier Series and Power Spectrum with MATLAB
B Ninness, Spectral Analysis using the FFT
U of Rhode Island, ELE 436, FFT Tutorial

Normalized ω_s and ω_0

```
N = 2000;  
x = [0:100]/100;  
f = ones(1,101)*1/2;  
for i = 1:2:N  
    a = 2/pi/i;  
    f = f + a*sin(2*pi*i*x);  
end  
plot(x, f);
```

Normalized ω_s and ω_0

```
N = 8;  
t = [0:N-1]'/N;  
f = sin(2*pi*t);  
p = abs(fft(f))/(N/2);  
p = p(1:N/2).^2
```

Normalized ω_s and ω_0

```
N = 10000;  
T = 3.4;  
t = [0:N-1]/N;  
t = t*T;  
f = sin(2*pi*10*t);  
p = abs(fft(f))/(N/2);  
p = p(1:N/2).^2;  
freq = [0:N/2-1]/T;  
seilogy(freq,p);  
axis([0 20 0 1]);
```

Normalized ω_s and ω_0

```
Fs = 44100;  
y = wavrecord(5*Fs, Fs);  
wavplay(y, Fs);
```

Normalized ω_s and ω_0

```
fs= 10;  
t= (0:1:length(x)-1)/fs;  
plot(t,x);
```

```
X= fft(x);  
plot(abs(X));
```


Normalized ω_s and ω_0

```
N= length(x);  
ws= 2*pi/N;  
wnorm= -pi:ws:pi;  
wnorm= wnorm(1:length(x));  
w= wnorm*fs;  
plot(w,abs(fftshift(X)));  
axis([-30,30,0,160]);
```

Normalized ω_s and ω_0

```
[freq,amp]= ginput;  
amp*2/N;  
fil= [zeros(1,160), ones(1,15), zeros(1,50),  
ones(1,15), zeros(1,160)];  
plot(w,abs(fftshift(X)), 'b',w,160*fil, '-.r');  
  
Xf= fil.*fftshift(X);  
xf= ifft(fftshift(Xf));  
plot(t, real(xf));
```

Normalized ω_s and ω_0

```
N= length(x);  
ws= 2*pi/N;  
wnorm= -pi:ws:pi;  
wnorm= wnorm(1:length(x));  
w= wnorm*fs;  
X= fft(x);  
plot(w,abs(fftshift(X)));  
axis([-30,30,0,500]);
```

Normalized ω_s and ω_0

```
n= [0:29];  
x= cos(2*pi*n/10);  
  
N1= 64;  
N2= 128;  
N3= 256;  
X1= abs(fft(x,N1));  
X2= abs(fft(x,N2));  
X3= abs(fft(x,N3));  
  
F1= [0: N1-1]/N1;  
F2= [0: N2-1]/N2;  
F3= [0: N3-1]/N3;
```

Normalized ω_s and ω_0

```
subplot(3,1,1);  
plot(F1,X1,'-x'), title('N=64'), axis([0 1 0 20]);  
subplot(3,1,2);  
plot(F2,X2,'-x'), title('N=128'), axis([0 1 0 20]);  
subplot(3,1,3);  
plot(F3,X3,'-x'), title('N=256'), axis([0 1 0 20]);
```

Normalized ω_s and ω_0

```
n= [0:29];  
x1= cos(2*pi*n/10); % 3 periods  
x2= [x1 x1]; % 6 periods  
x3= [x1 x1 x1] % 9 periods
```

```
N= 2048;
```

```
X1= abs(fft(x1,N));  
X2= abs(fft(x2,N));  
X3= abs(fft(x3,N));
```

```
F= [0:N-1]/N;
```

Normalized ω_s and ω_0

```
subplot(3,1,1);  
plot(F,X1), title('3 periods'), axis([0 1 0 50]);  
subplot(3,1,2);  
plot(F,X2), title('6 periods'), axis([0 1 0 50]);  
subplot(3,1,3);  
plot(F,X3), title('9 periods'), axis([0 1 0 50]);
```

Normalized ω_s and ω_0

```
n= [0: 140];  
x1= cos(2*pi*n/10);  
  
N= 2048;  
  
X= abs(fft(x1,N));  
X= fftshift(X);  
  
F= [-N/2:N/2-1]/N;  
  
plot(F, X);  
xlabel('frequency / f_s');
```


Normalized ω_s and ω_0

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```

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References

- [1] <http://en.wikipedia.org/>
- [2] J.H. McClellan, et al., Signal Processing First, Pearson Prentice Hall, 2003
- [3] M.J. Roberts, Fundamentals of Signals and Systems
- [4] S.J. Orfanidis, Introduction to Signal Processing
- [5] K. Shin, et al., Fundamentals of Signal Processing for Sound and Vibration Engineerings

- [6] A “graphical interpretation” of the DFT and FFT, by Steve Mann