

Utility Functions Octave Codes (0B)

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This document was produced by using LibreOffice and Octave.

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

Time and Frequency Domain Plot

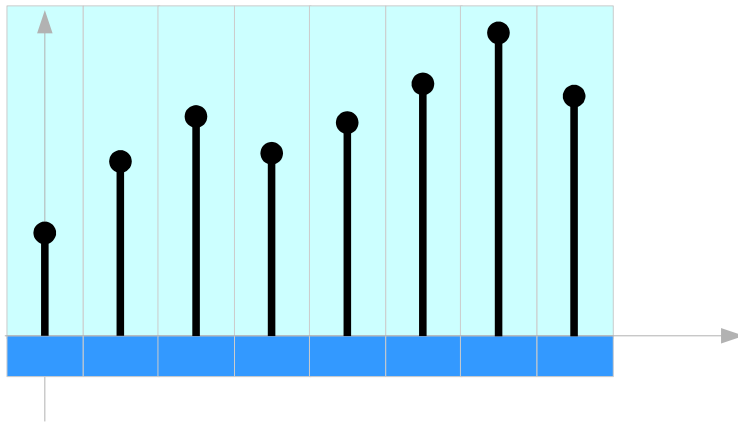
`x(1:Nf)` Time domain signal

`n(1:Nf)` Time index vector

`X(1:Nf)` Frequency domain spectrum

`k(1:Nf)` Frequency index vector

Time and Frequency Domain Plot

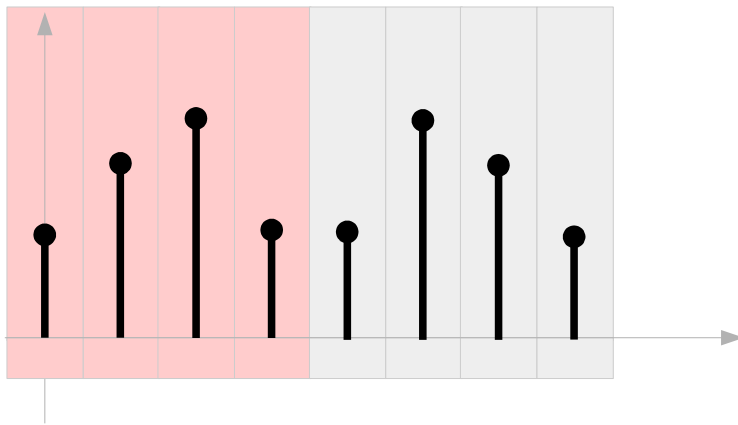


$x(1:NF)$

$x(1:NF)$

$n(1:NF)$

$n(1:NF)$



$X(1:NF)$

$X(1:NF/2)$

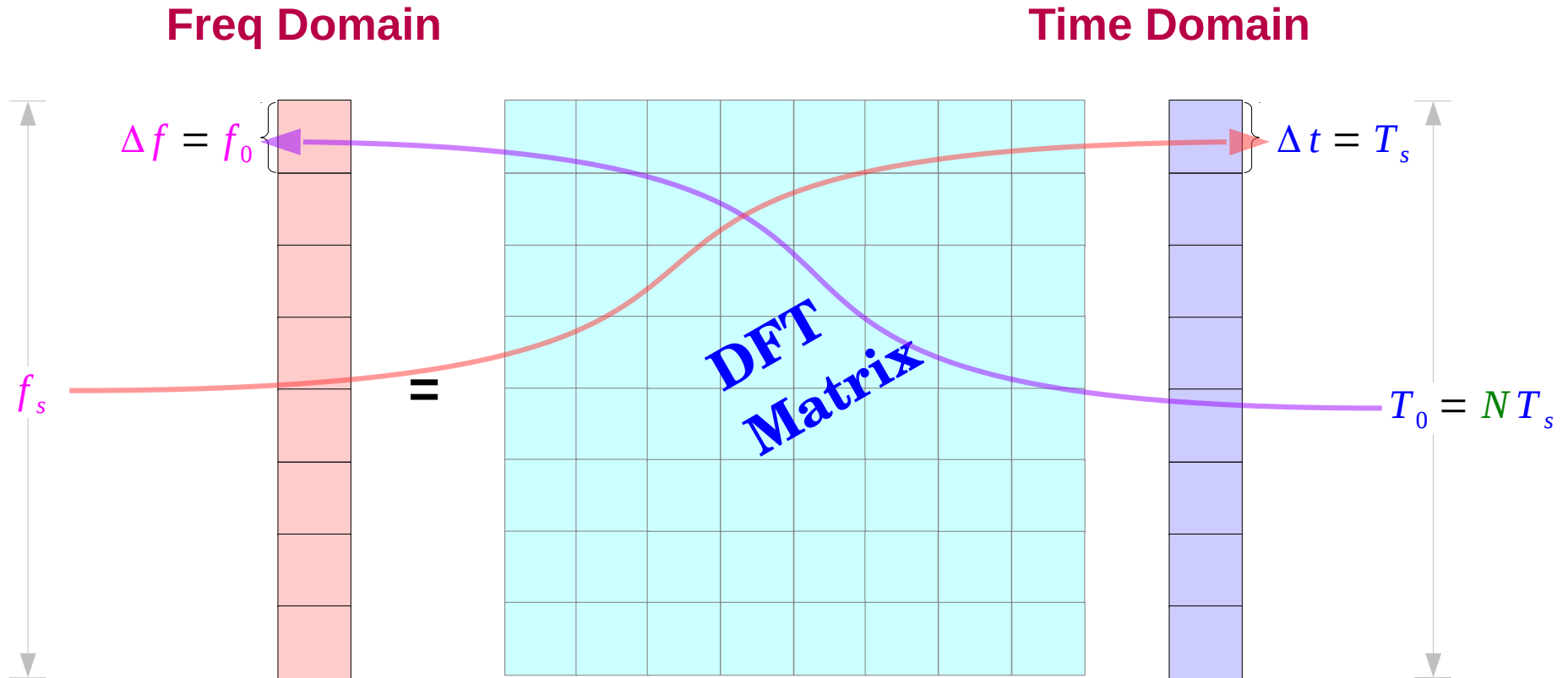
$k(1:NF)$

$k(1:NF/2)$

Time and Frequency Domain Plot

	mode	a:b	'full'	'half'
<code>x(1:Nf)</code>	Time domain signal	<code>x(a:b)</code>	<code>x(1:Nf)</code>	<code>x(1:Nf)</code>
<code>n(1:Nf)</code>	Time index vector	<code>n(a:b)</code>	<code>n(1:Nf)</code>	<code>n(1:Nf)</code>
<code>X(1:Nf)</code>	Frequency domain spectrum	<code>X(a:b)</code>	<code>X(1:Nf)</code>	<code>X(1:Nf/2)</code>
<code>k(1:Nf)</code>	Frequency index vector	<code>k(a:b)</code>	<code>k(1:Nf)</code>	<code>k(1:Nf/2)</code>

Using Sampling Frequency and Time



$$f_0 = \frac{1}{T_0} = \frac{1}{NT_s}$$

$$f_s \geq \frac{2}{T_0}$$

$$f_s \geq 2f_0$$

$$T_s = \frac{1}{f_s}$$

Time and Frequency Domain Plot

```
subplot(3,1,1);  
p = stem(n(trange), x(trange), 'k'); grid on;  
xlabel('Time, t (s)'); ylabel('x(t)');  
  
subplot(3,1,2);  
p = stem(k(frange), abs(X(frange)), 'k');  
set(p,'LineWidth',2,'MarkerSize',4); grid on;  
xlabel('Harmonic number, k');  
ylabel('|X(k)|');  
  
subplot(3,1,3);  
p = stem(k(frange), angle(X(frange)), 'k');  
set(p,'LineWidth',2,'MarkerSize',4); grid on;  
xlabel('Harmonic Number, k');  
ylabel('Ang{X[k]}');  
  
endfunction
```


Time and Frequency Domain Plot

```
# Prevent Octave from thinking that this
# is a function file:
1;

#-----
# n : time index vector
# x : time domain signal vector
# k : frequency index vector
# X : frequency domain signal vector
#-----
function TimeFreqPlot(n, x, k, X, mode)
```

Time and Frequency Domain Plot

```
NF = length(n);

if (strmatch(typeinfo(mode), "range"))
    trange = mode; frange = mode;
else
    if (strmatch(mode, "half"))
        trange = 1:NF; frange = 1:NF/2;
    elseif (strmatch(mode, "full"))
        trange = 1:NF; frange = 1:NF;
        fs = k(2)+k(N)
        X = fftshift(X); k = fftshift(k);
        k(1:NF/2) = k(1:NF/2) - fs;
    else
        trange = 1:NF; frange = 1:NF;
    endif
endif
```

Time and Frequency Domain Plot

```
subplot(3,1,1);  
p = stem(n(trange), x(trange), 'k'); grid on;  
xlabel('Time, t (s)'); ylabel('x(t)');  
  
subplot(3,1,2);  
p = stem(k(frange), abs(X(frange)), 'k');  
set(p,'LineWidth',2,'MarkerSize',4); grid on;  
xlabel('Harmonic number, k');  
ylabel('|X(k)|');  
  
subplot(3,1,3);  
p = stem(k(frange), angle(X(frange)), 'k');  
set(p,'LineWidth',2,'MarkerSize',4); grid on;  
xlabel('Harmonic Number, k');  
ylabel('Ang{X[k]}');  
  
endfunction
```

wavplot

```
function wavplot(fname)

[x, fs, nbits] = wavread(fname);
nn = length(x);
printf('fs = %f \n', fs);
printf('nn = %d \n', nn);

n = (0 : nn-1) / fs ;
k = fs * (0 : nn-1) / nn ;
X = fft( x );

source "../0.util.octave/util.m"
#TimeFreqPlot(n, x, k, X, "half")
#TimeFreqPlot(n, x, k, X, (1:1024)+1024*0.5)
#TimeFreqPlot(n, x, k, X, (1*1024:2*1024))
#TimeFreqPlot(n, x, k, X, (35200-0.5*1024:35200+0.5*1024))
TimeFreqPlot(n, x, k, X, (2*1024:3*1024))

endfunction
```

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