

Why use fftshift(fft(fftshift(x))) instead of fft(x) in Matlab?

Wu, Kan

Oct. 1 2009

fft(x) has some intrinsic problem due to its realization. Here is an example.

Consider a Gaussian pulse in time domain

$$x(t) = e^{-\frac{t^2}{2A^2}}$$

Its Fourier transform is

$$\begin{aligned} X(f) &= \int_{-\infty}^{+\infty} e^{-\frac{t^2}{2A^2} - j2\pi ft} dt \\ &= \int_{-\infty}^{+\infty} e^{-\left(\frac{t}{\sqrt{2}A} + j\sqrt{2}\pi Af\right)^2 - 2\pi^2 A^2 f^2} dt \\ &= \sqrt{2\pi} A e^{-2\pi^2 A^2 f^2} \end{aligned}$$

It is real and Gaussian.

Codes in Matlab are:

```
%-----  
Bx = 10;  
A = sqrt(log(2))/(2*pi*Bx);  
fs = 500;      %sampling frequency  
dt = 1/fs;     %time step  
T=1;          %total time window  
t = -T/2:dt:T/2-dt; %time grids  
df = 1/T;      %freq step  
Fmax = 1/2/dt; %freq window  
f=-Fmax:df:Fmax-df; %freq grids, not used in our examples, could be used by plot(f, X)  
%-----  
% Numerical evaluations  
%-----
```

```

x = exp(-t.^2/2/A^2);
Xan = A*sqrt(2*pi)*exp(-2*pi^2*f.^2*A^2);    %X(f), analytical Fourier transform of x(t), real
Xfft = dt * fft(x);          %directly using fft()
Xfftshift = dt * fft(fftshift(x));    %using fftshift() before fft()
Xfinal = dt * fftshift(fft(fftshift(x)));    %identical with analytical X(f), also note dt
%-----
hold on
plot(...)
plot(...)
%-----

```

- `fftshift()` is a function to swap the left half and right half of a vector. e.g., `x=[1 2 3 4 5 6]` then `fftshift(x)=[4 5 6 1 2 3]`
- `dt` is to scale the amplitude of `X` (e.g. `Xfft`, `Xfftshift`, `Xfinal`) equal to its analytical Fourier transform `X(f)`

Real part of `Xfft=dt * fft(x)` is plotted in Fig.1(a). There exists obvious oscillation. But if we swap the left half and right half of `x(t)` before `fft()`, then oscillation disappears, as `real(Xfftshift)` in Fig.1(a). A zoomed region of left side of Fig.1(a) is shown in Fig.1(b). It is clear to show that the real part of `Xfft` is periodically multiplied by `1,-1,1,-1...` compared with `Xfftshift`. Also note that this oscillation could not be found if we merely observe its absolute value as shown in Fig.2. Compared with analytical Fourier transform `X(f)`, a smooth curve `real(fft(fftshift(x)))` is correct.

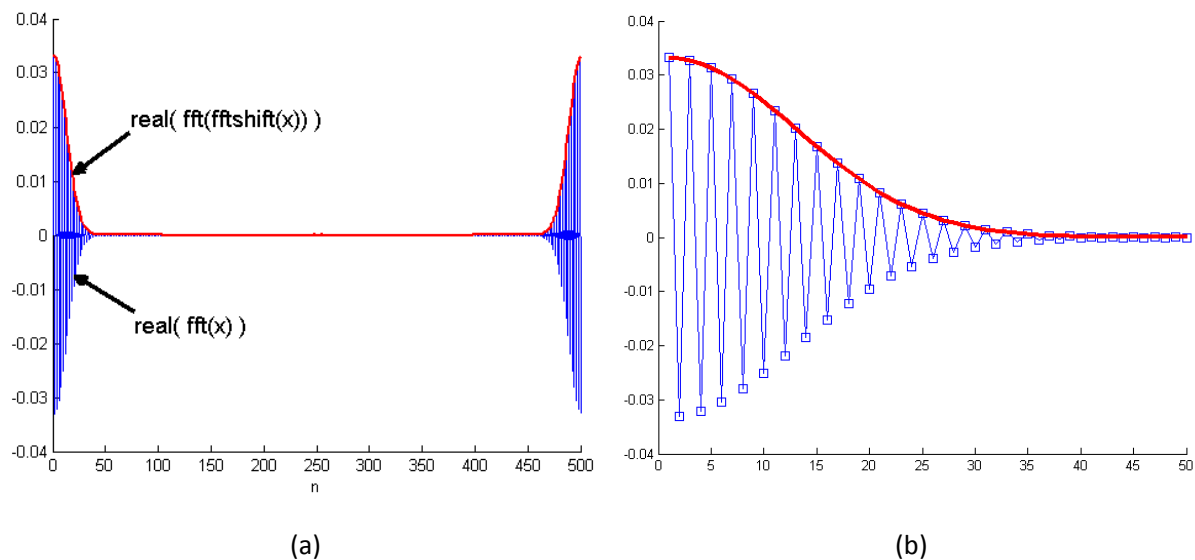


Fig. 1 Real part of `fft(x)` and `fft(fftshift(x))`

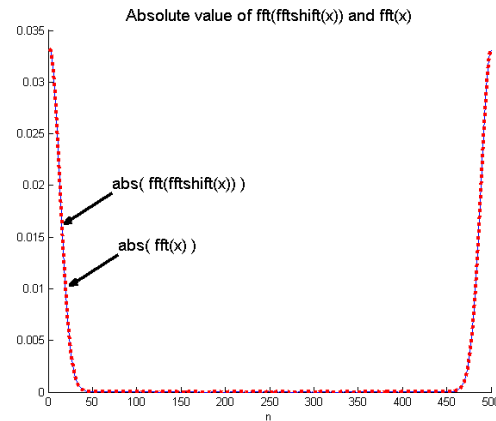


Fig. 2 Absolute value of $\text{fft}(x)$ and $\text{fft}(\text{fftshift}(x))$

Then we consider second step: since $\text{fft}()$ is intrinsically single sided, the $f=0$ point is at left side of horizontal axis and the right half is just the mirror of left half. For double side case, we need to move the $f=0$ point to the center of horizontal axis by swapping the right half and left half of the data. This leads to:

$X_{\text{final}} = \text{fftshift}(X_{\text{fftshift}}) = dt * \text{fftshift}(\text{fft}(\text{fftshift}(x)))$; %don't forget to multiply dt!

As plotted in Fig.3, both real and imaginary part of X_{final} well fit the analytical Fourier transform $X(f)$. The error shown in imaginary part is due to the numerical error and very small.

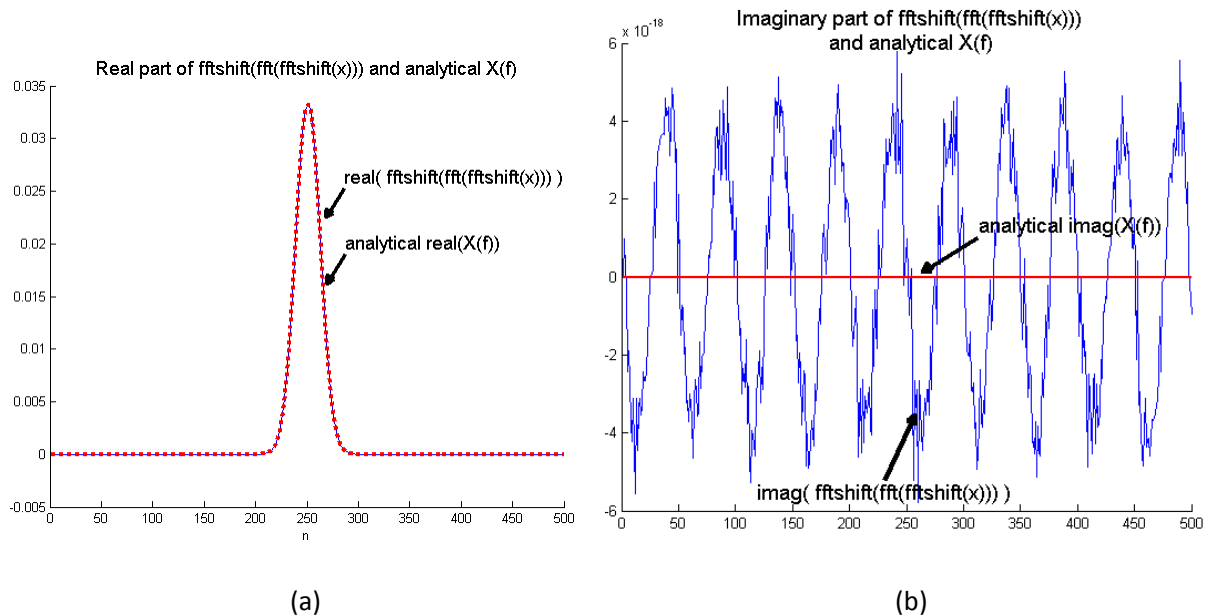


Fig. 3 Real and imaginary part of $\text{fftshift}(\text{fft}(\text{fftshift}(x)))$ and analytical $X(f)$

Finally, we have the formula for FFT and IFFT in Matlab:

FFT: `fftshift(fft(fftshift(x))) * dt`

IFFT: `fftshift(ifft(fftshift(X))) * fs %fs = 1/dt`

For other language, we need to multiply IFFT by 1/N based on the theory of discrete Fourier transform, which is done by Matlab automatically. E.g. type “help fft” in Matlab command window, we would find that:

$$X(k) = \sum_{n=1}^N x(n) \exp(-j \cdot 2 \cdot \pi \cdot (k-1) \cdot (n-1) / N), \quad 1 \leq k \leq N$$
$$x(n) = (1/N) \sum_{k=1}^N X(k) \exp(j \cdot 2 \cdot \pi \cdot (k-1) \cdot (n-1) / N), \quad 1 \leq n \leq N$$

Acknowledge

This little note is based on Daniele Disco 's work in *A guide to the Fast Fourier Transform, 2nd Edition*. More details of `fft()` and Matlab codes could be found in the following link:
<http://www.mathworks.com/matlabcentral/fileexchange/5654>