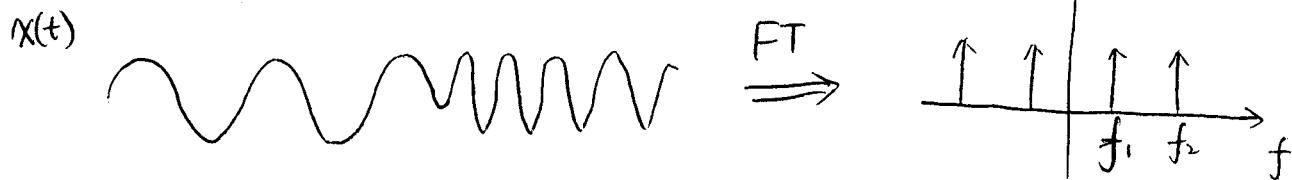
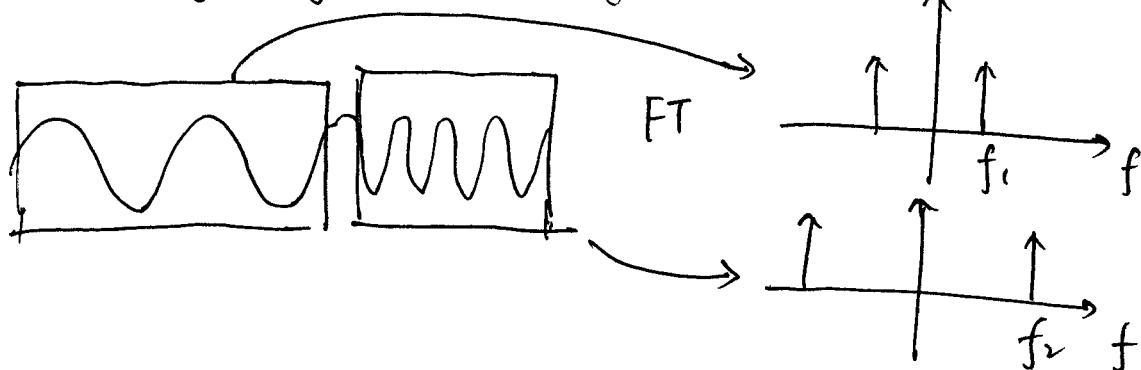


(6)

Frequency spectrum shows the aggregated frequency components of the entire signal of analysis. We don't when does each freq. component happen in the signal. But this information is important if we are dealing with non-stationary signal. What can we do?



Idea: Analyze signal window by window.



Time-frequency resolution tradeoff.

Δt : time resolution : Length of window, because freq. components ^{of signal} within the window cannot be resolved.

Δf : freq. resolution: (without zero-padding) freq. difference between adjacent bins.

Suppose $X[n]$ was sampled from $X(t)$ with f_s .

Take a window of M samples, then $\Delta t = \frac{M}{f_s}$ (second)

After DFT, we have M samples to represent the spectrum from 0 to f_s Hz

$$\therefore \Delta f = \frac{f_s}{M} \text{ (Hz)}$$

$$\therefore \Delta t \cdot \Delta f = 1$$

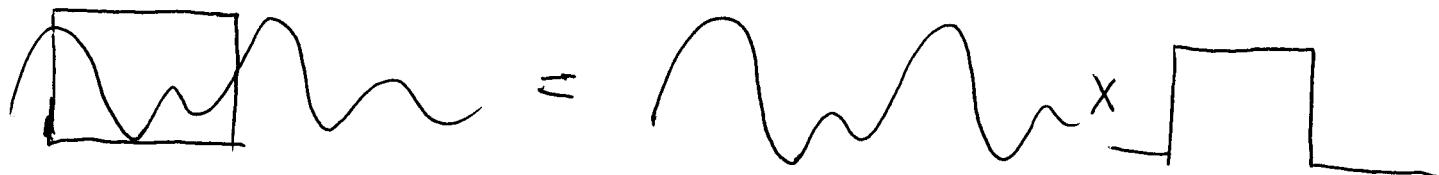
Therefore, to estimate Δf , the window length in second Δt is the only critical variable. f_s and M don't matter!

e.g. if $\Delta t = 100$ ms. $\Delta f = 10$ Hz.

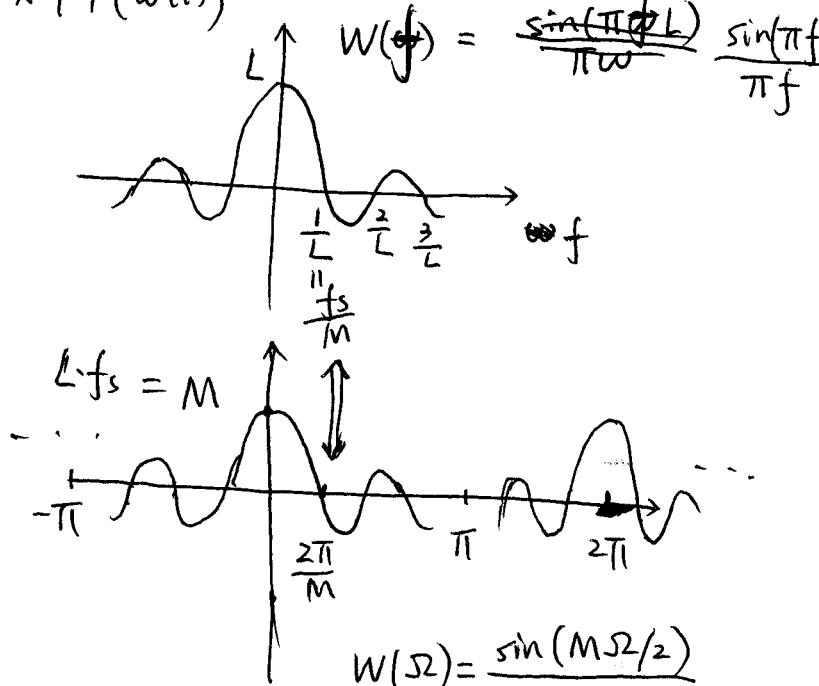
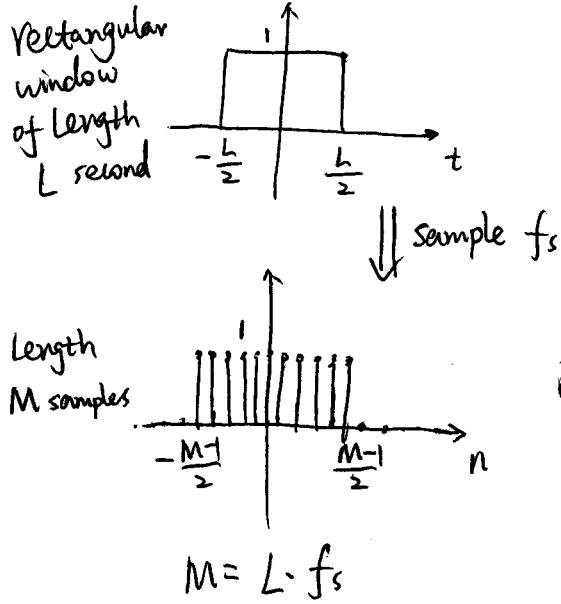
Effect of Windowing

(7)

Taking a chunk of $x(t)$ for analysis is equivalent to multiplying $x(t)$ with a rectangular window function.



$$\text{FT}(\mathbf{x}(t) \times w(t)) = \text{FT}(\mathbf{x}(t)) * \text{FT}(w(t))$$



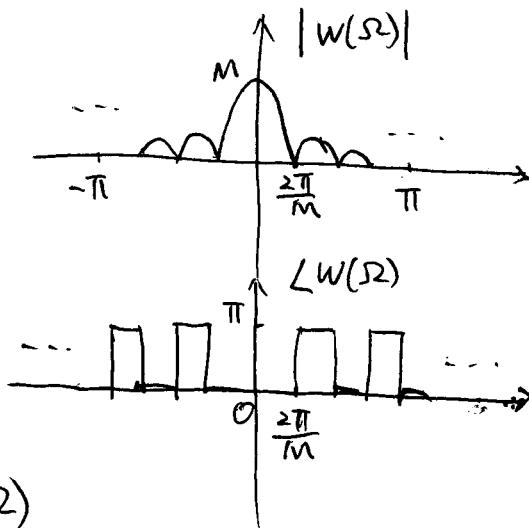
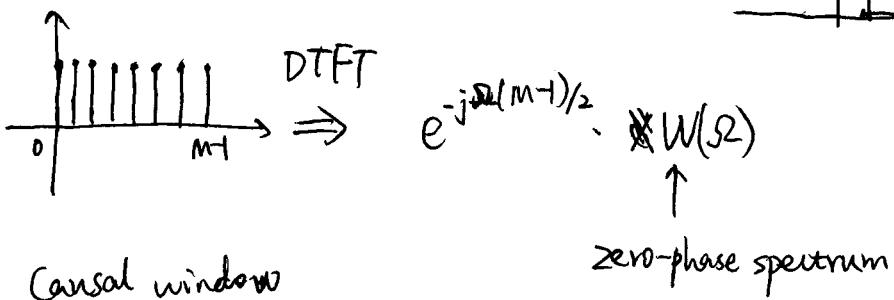
symmetric windows (zero-phase windows)

\Leftrightarrow spectrum is real

\Leftrightarrow phase spectrum only take 0 or π values

Shift window in time

\Leftrightarrow multiply a phase linear in spectrum



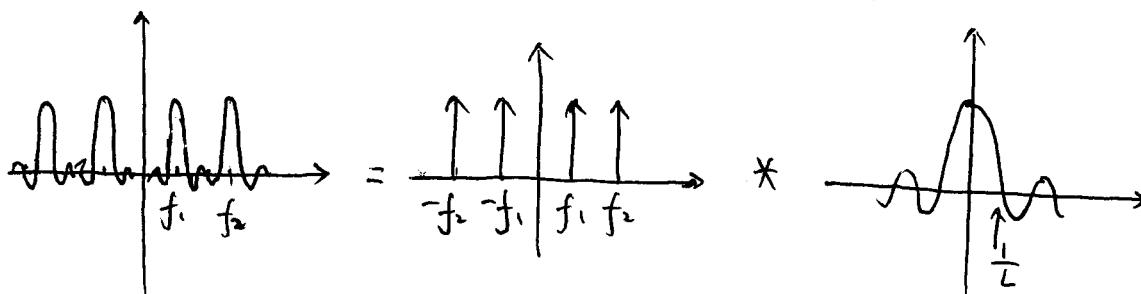
Windowing smears (smoothes) spectrum, i.e., reduces frequency resolution.

(8)

Ideal frequency resolution $\Delta f = \frac{1}{\Delta t}$

Considering window effect: Let $x(t) = \sin(\frac{2\pi}{L}f_1 t) + \sin(\frac{2\pi}{L}f_2 t)$ (in phase sinusoids).

$$FT(x(t) \cdot w(t)) = FT(x(t)) * FT(w(t))$$

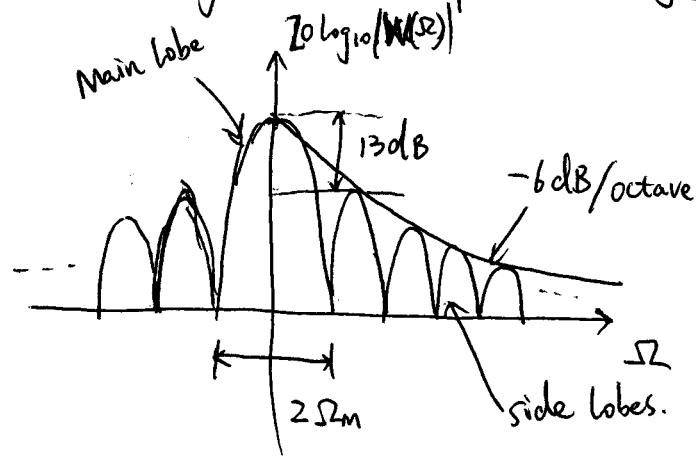


To make sure the two sinusoids are resolved, we need $|f_1 - f_2| > \frac{2}{L}$, i.e.

$$\Delta f > \frac{2}{\Delta t}$$

Clearly, the ^{real} frequency resolution is affected by the shape of window spectrum.

A log-amplitude rectangular window spectrum magnitude



$$\Omega_m = \frac{2\pi}{L}$$

translating to frequency in Hz

$$f = \frac{f_s}{m} = \frac{1}{L}$$

length in second

- We want 1) narrow main lobe
2) low side lobes.

Rect window is good at ①, but not good at ②.

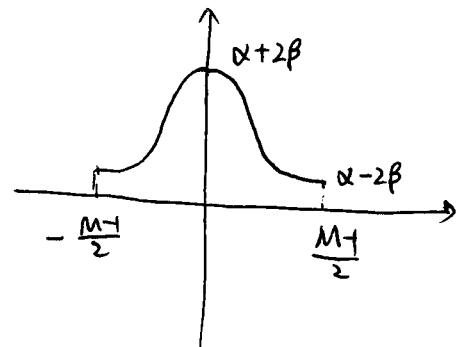
Can we design other windows to achieve these?

Reason why Rect window has high side-lobe is because of the abrupt change at the window boundary. (9)

Ideas: Smooth out the boundary.

Generalized Hamming Window Family.

$$W_H[n] = W_R[n] \left[\alpha + 2\beta \cos\left(\frac{2\pi n}{M}\right) \right]$$



- Hann window : $\alpha = \frac{1}{2}$, $\beta = \frac{1}{4}$.

$$W_H[n] = W_R[n] \left[\frac{1}{2} + \frac{1}{2} \cos\left(\frac{2\pi n}{M}\right) \right] = W_R[n] - \cos^2\left(\frac{\pi}{M}n\right)$$

Main lobe width : 452m

First side lobe height: -31 dB

Side lobe roll off : -18 dB/octave

- Hamming widow : $\alpha = 0.54$, $\beta = 0.23$

Main lobe width = 4 Ω_m

First side lobe height : -41 dB

Sidelobe roll off : - 6 dB/octave.

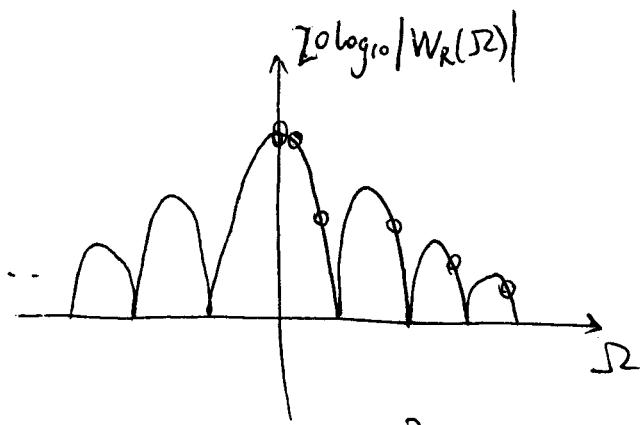
Remember $\Omega_m = \frac{2\pi}{M}$

$\Delta f > \frac{4}{\Delta t}$ for Hann and Hamming windows, which is worse than rectangular window. But because sidelobes are significantly reduced, they are often used in practice.

Spectral Interpolation, Zero padding

In DFT, suppose we take an M-point windowed signal and perform DFT to get its spectrum, its spectrum is represented by M points as well. This is often not enough to see the detailed content.

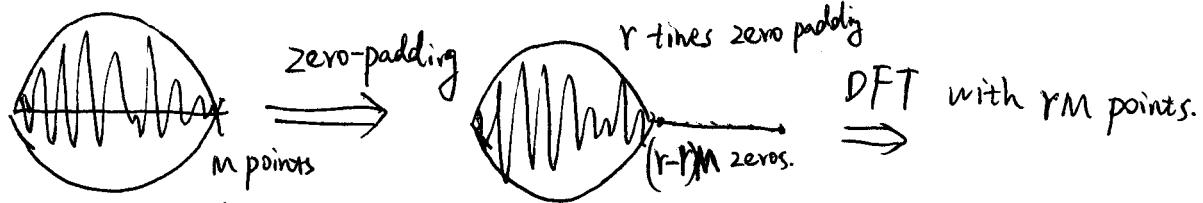
Analysis: M points to represent 2π . \Leftrightarrow each point for $2\pi/M = \Delta\theta$



This is clearly not enough to show the shape of the spectrum. (10)

Can we use more points?

Zero padding: append zeros to windowed signal before perform DFT.



Zero padding gives ideal interpolation in frequency domain

Proof: Consider ~~N-point~~ ^{N-point} $X[n]$ with DFT spectrum $X[k]$.

$$X[n] = \frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j \frac{2\pi k n}{N}} \quad (\text{IDFT})$$

Let's zero pad $X[n]$ with infinitely many zeros on both sides, then take DTFT to look at its spectrum.

$$\begin{aligned} X(S2) &= \sum_{n=-N/2}^{N/2} X[n] e^{-j S2 n} = \sum_{n=-\frac{N}{2}}^{\frac{N}{2}} \left[\frac{1}{N} \sum_{k=0}^{N-1} X[k] e^{j \frac{2\pi k n}{N}} \right] e^{-j S2 n} \\ &= \sum_{k=0}^{N-1} X[k] \left[\frac{1}{N} \sum_{n=-\frac{N}{2}}^{\frac{N}{2}} e^{j \left(\frac{2\pi k n}{N} - S2 \right) n} \right] \\ &= \sum_{k=0}^{N-1} X[k] \text{asinc}_N(S2 - \frac{2\pi k}{N}) \quad \text{asinc}_N(S2) = \frac{\sin(N S2 / 2)}{N \sin(S2 / 2)} \end{aligned}$$

This means $X(S2)$ can be interpolated from $X[k]$ using infinite zero padding, and the interpolation is ideal (as it uses asinc function!).

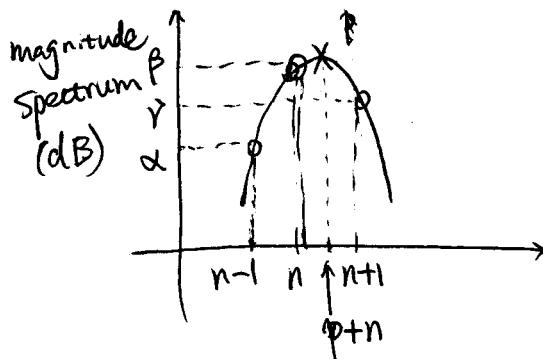
Note: After zero-padding, we are using more points to represent the spectrum. i.e., Δf between adjacent bins is smaller.

But, frequency resolution is Not Changed!

Sinusoids that were not resolved are still not resolved.

△ Zero padding does not increase frequency resolution! (11)

Quadratic Peak Interpolation



Real Peak may lie in between two frequency bins.

We could solve this by large number of zero padding,
but it's computationally expensive.

Another sub-optimal solution: fit neighboring bins with a parabola.

$$y(x) = \cancel{a(x-p)^2} + b \quad a[x-(p+n)]^2 + b$$

$$y(n-1) = a(p+1)^2 + b = \alpha$$

$$y(n) = ap^2 + b = \beta$$

$$y(n+1) = a(p-1)^2 + b = \gamma$$

$$\Rightarrow p = \frac{1}{2} \frac{\alpha - \gamma}{\alpha - 2\beta + \gamma} \in [-\frac{1}{2}, \frac{1}{2}]$$

$$y(\cancel{p}) = \beta - \frac{1}{4}(\alpha - \gamma)p$$

Quadratically Interpolated FFT procedure. for \checkmark peak detection.

1. Make sure window is long enough to resolve all sinusoids.
2. Use enough zero padding (e.g., 4 times) to oversample spectrum.
3. Use quadratic interpolation of three samples around a peak.
4. solve parabola equations to get interpolated amplitude and freq.
- (optional) 5. Use quadratic or linear interpolation on the unwrapped phase of the peak