

DSP : Discrete Fourier
Transform (DFT)

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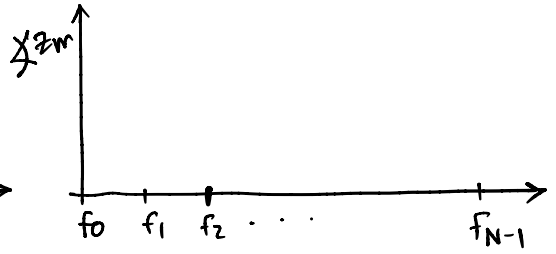
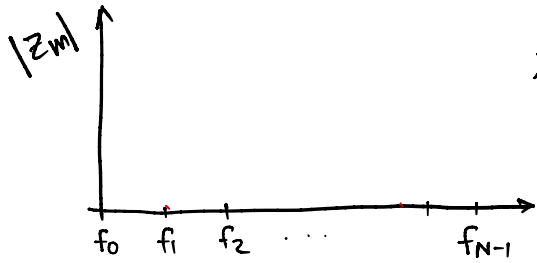
let's do 3 simple examples.

✓✓ ① DFT ($\cos 2\pi f_m nts$)

② DTT ($e^{j2\pi f_m nts}$)

③ DFT ($\sin 2\pi f_m nts$)

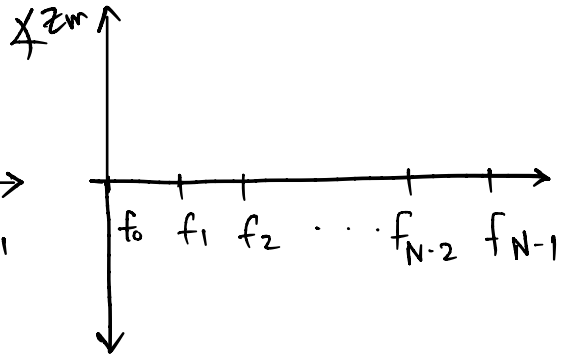
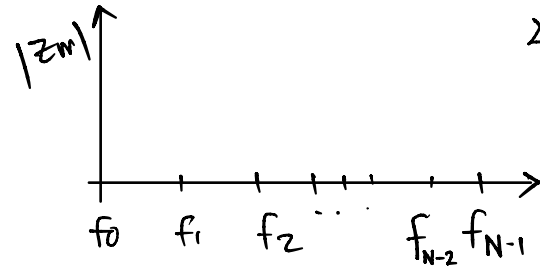
② Example 2: DFT ($x[n] = e^{j2\pi f_1 n t_s}$)



Just one stick rotating at

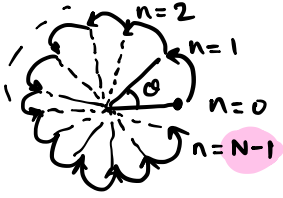
since it gives $e^{j0 \cdot n}$

③ Example 3: DFT ($x[n] = \sin 2\pi f_1 n t_s$)



like $\cos 2\pi f_1 n t_s$, you need 2 sticks to cancel out the complex values, but

② Translating to real-world frequencies



Now, slowest freq = N samples/cycle

Say sampling freq. = f_s

N samples take

1 cycle takes

∴ slowest frequency =

$\frac{f_s}{N} \Rightarrow$ Called Fundamental freq.

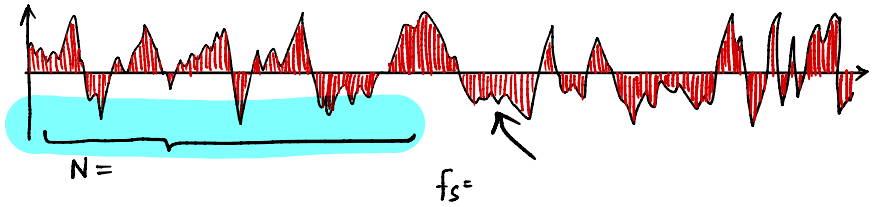
Faster freq. \Rightarrow

For large N , $(N-1) \approx N$,

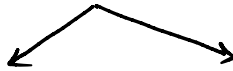
∴

② let's put it all together

Consider sampling your voice signal at 8000 Hz and taking a 1000 point FFT (i.e.,)



③ Thus, when analyzing a given signal, we have we can control

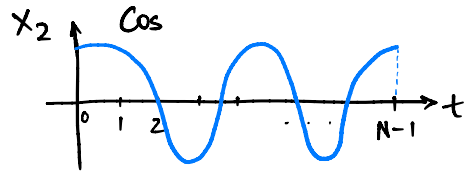
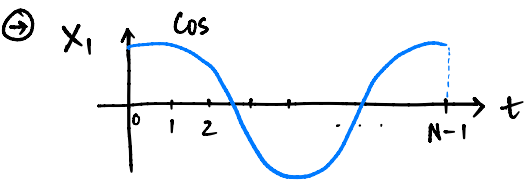


Large f_s means

Large N means we have finer

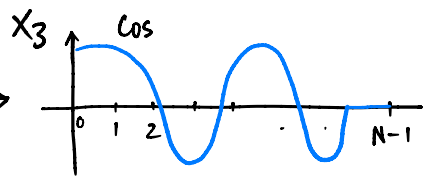
$$\text{Width of bin} = f_s/N$$

⊙ BEWARE OF INCOMPLETE CYCLES.



What is the FFT mag. spectrum for these signals?

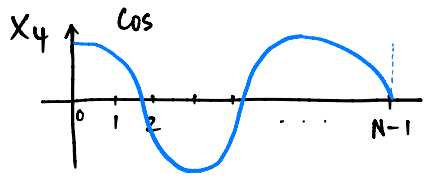
⊙ Now, what is the FFT mag. spectrum of this →



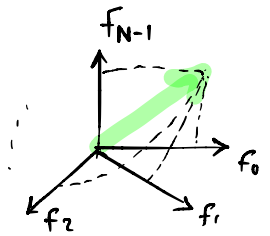
Think of this X_3 signal as →

Thus, $DFT(X_3)$ $DFT(X_2)$ since the DFT (|||||ooo) influences results

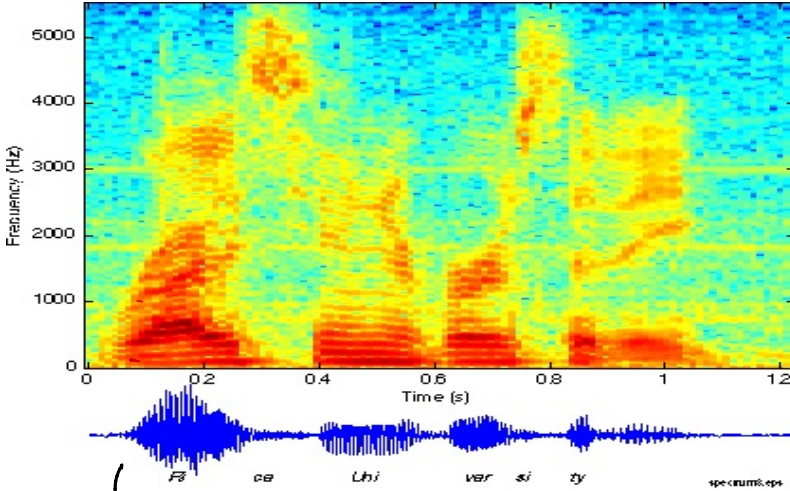
⊙ In fact:
is $DFT(X_4) = DFT(X_2)$



Many different rotations are needed ⇒
Think of this as



REAL SPECTROGRAM : Voice Signal



Shows various
freq. components
at this time
slice.

Shows how a specific freq. comes and goes over time.

→ 3 important properties of DFT

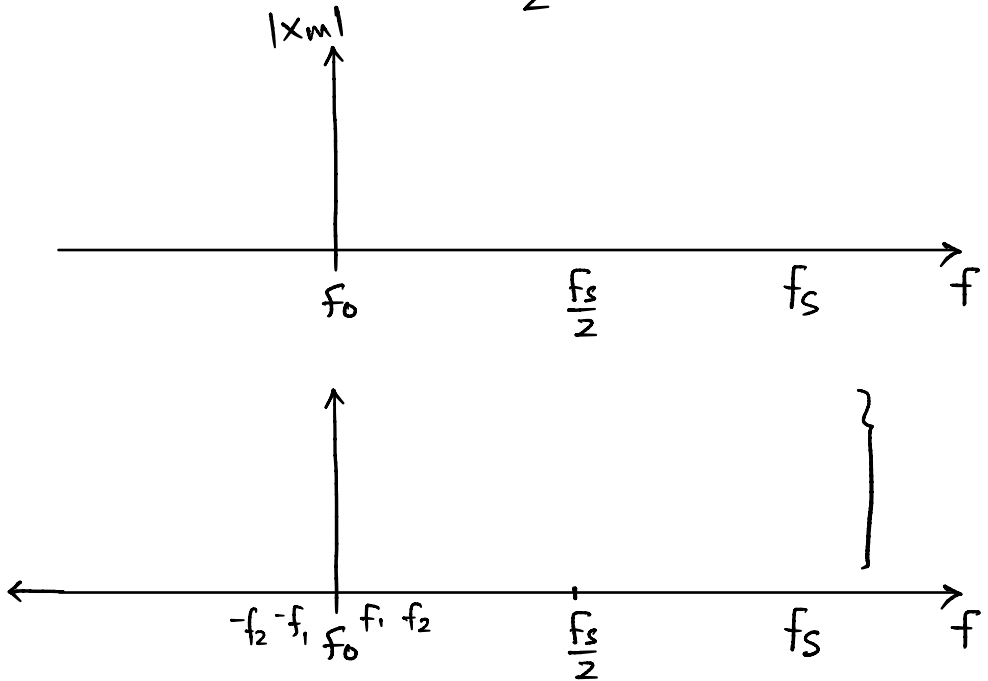
① DFT is linear

$$\text{DFT}(x[n] + y[n]) =$$
$$=$$

So what is DFT of $\cos 2\pi f_1 n t_s + \sin 2\pi f_2 n t_s$?

② For real signals, DFT is symmetric

↳ Not only around $\frac{f_s}{2}$



→ So take frequencies in $[\frac{f_s}{2}, f_s]$ and move to the

→ Its like the sticks are going clockwise (i.e., (-)ve θ) to cancel the complex part.

③ DFT of a shifted signal is original DFT with a phase shift.

$$X'_m = X_m \cdot e^{j \frac{2\pi}{N} \cdot m \cdot k}$$

let's shift x by k samples

$$x'[n] = x[n+k]$$

$$X'_m = \text{DFT} \{ x[n+k] \}$$

$$\therefore X'_m = \sum_{n=0}^{N-1}$$

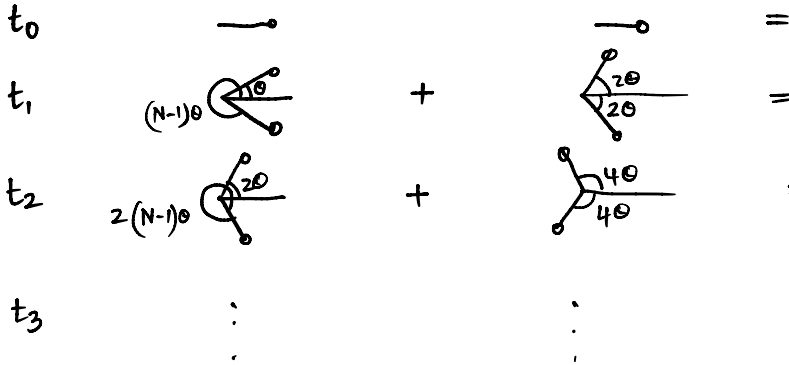
$$= \sum_{h=0}^{N-1}$$

What is this?

$$X'_m = X_m e^{j \frac{2\pi}{N} \cdot m \cdot k}$$

This phase shift is proportional to k , and

$$x_n = \cos 2\pi f_1 n t_s + \cos 2\pi(2f_1) n t_s$$



Now	$z_1 e^{j \frac{2\pi}{N} \cdot (m=1)(k=1)}$		$z_2 e^{j \frac{2\pi}{N} \cdot (m=2)(k=1)}$
	$z_{N-1} e^{j \frac{2\pi}{N} \cdot (m=N-1)(k=1)}$		$z_{N-2} e^{j \frac{2\pi}{N} \cdot (m=N-2)(k=1)}$

