Fast Fourier Transform (FFT) and Short-Time Fourier Transform (STFT)
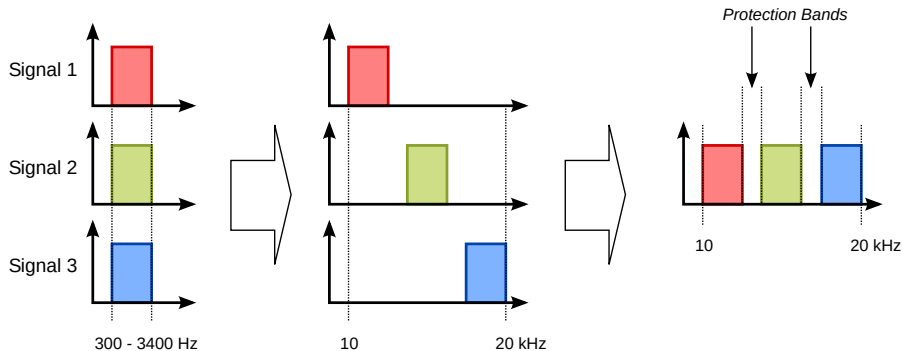
Digital Signal Processing

February 15, 2024

UNIVERSITY *of* VIRGINIA

Application of FFT: Communications

Frequency Division Multiplexing (FDM)

Protection Bands

Signal 1

Signal 2

Signal 3

300 - 3400 Hz

10   20 kHz

10   20 kHz

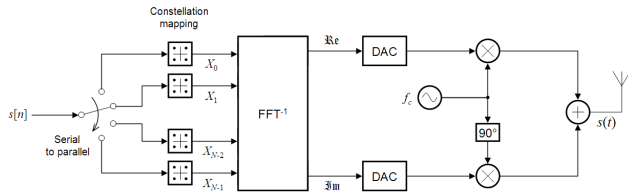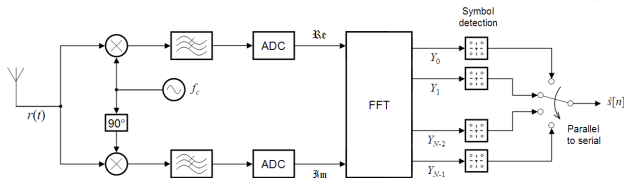https://commons.wikimedia.org/w/index.php?curid=18327256

Application of FFT: Communications

Orthogonal Frequency Division Multiplexing (OFDM)

Transmitter:

Receiver:

Technology behind most digital wireless communication!
(WiFi, 4G, 5G, HD Radio, HDTV)

The Fast Fourier Transform (FFT)

Naïve DFT Algorithm

Recall the DFT equation:

$$X[k] = \frac{1}{\sqrt{L}} \sum_{n=0}^{L-1} e^{-i\omega_0 nk} x[n]$$
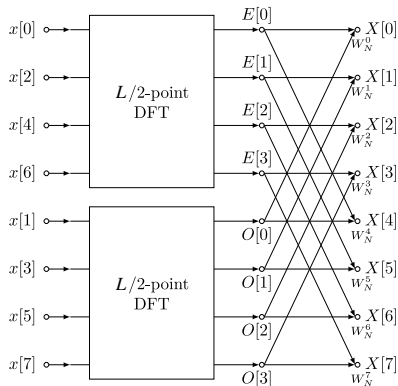
Naïve DFT Algorithm

Loop over $k = 0, \ldots, L-1$
     Compute $X[k]$ by sum (loop) over $n = 0, \ldots, L-1$

Complexity is $O(L^2)$

Fast Fourier Transform

Cooley-Tukey Algorithm

Divide-and-Conquer

Radix-2 version:

1. compute "even" DFT
2. compute "odd" DFT
3. combine and reuse results

Recursively apply to each $L/2$ block

Complexity is $O(L \log L)$

Fast Fourier Transform

- More general radix-p FFT breaks the DFT into p blocks, where p is a prime factor of the signal length L
- Recursively applied to each L/p block
- Recursion stops when the remaining block lengths are prime numbers (can't be factored any further)
- **Bottom line:** The FFT is most efficient when the input signal length has small prime factors, preferrably L is a power of 2.
- Sometimes it is more efficient to pad a signal with zeros to get a good prime factorization.

First Half of FFT

Compute $X[k]$ for $k = 0, 1, \ldots, \frac{L}{2} - 1$,

$$
\begin{aligned}
X[k] &= \frac{1}{\sqrt{L}} \sum_{n=0}^{L-1} e^{-i\omega_0 nk} x[n] \\
&= \frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2mk} x[2m] + \frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 (2m+1)k} x[2m+1] \\
&= \underbrace{\frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2mk} x[2m]}_{E[k] = \text{sum of even terms}} + e^{-i\omega_0 k} \underbrace{\frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2mk} x[2m+1]}_{O[k] = \text{sum of odd terms}} \\
&= E[k] + e^{-i\omega_0 k} O[k]
\end{aligned}
$$

Second Half of FFT

Compute $X[k + \frac{L}{2}]$ for $k = 0, 1, \ldots, \frac{L}{2} - 1$,

$$X\left[k + \frac{L}{2}\right] = \frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2m(k+\frac{L}{2})} x[2m] +$$

$$+ e^{-i\omega_0(k+\frac{L}{2})} \frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2m(k+\frac{L}{2})} x[2m+1]$$

$$= E[k] - e^{-i\omega_0 k} O[k]$$

Using $e^{-i\omega_0 2m(k+\frac{L}{2})} = e^{-i\omega_0 2mk}$ and $e^{-i\omega_0(k+\frac{L}{2})} = -e^{-i\omega_0 k}$

Second Half of FFT

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$$+ e^{-i\omega_0(k+\frac{L}{2})} \frac{1}{\sqrt{L}} \sum_{m=0}^{L/2-1} e^{-i\omega_0 2m(k+\frac{L}{2})} x[2m+1]$$

$$= E[k] - e^{-i\omega_0 k} O[k] \qquad \text{Reused!}$$

Using $e^{-i\omega_0 2m(k+\frac{L}{2})} = e^{-i\omega_0 2mk}$ and $e^{-i\omega_0(k+\frac{L}{2})} = -e^{-i\omega_0 k}$

FFT Historical Trivia

- FFT actually invented by Gauss in 1805! (but lost)
- Re-invented by Cooley and Tukey in 1965
- Tukey coined the term "bit" (for "binary digit") and was first to use the term "software" in writing
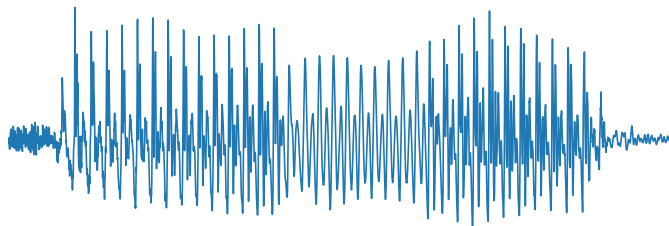
Carl Friedrich Gauss   John Tukey
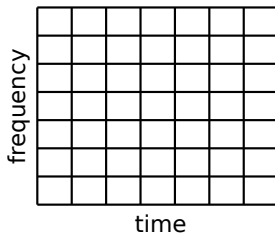
The Short-Time Fourier Transform (STFT)

STFT Procedure

Multiply signal $x[n]$ by a sliding window $w[n]$ and take FFT.

$x[n]$

$X[k, m]$

time

STFT Procedure

Multiply signal $x[n]$ by a sliding window $w[n]$ and take FFT.
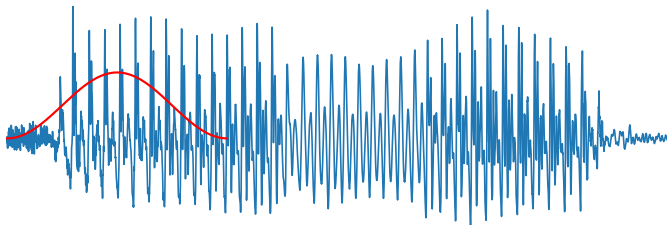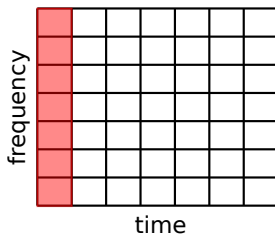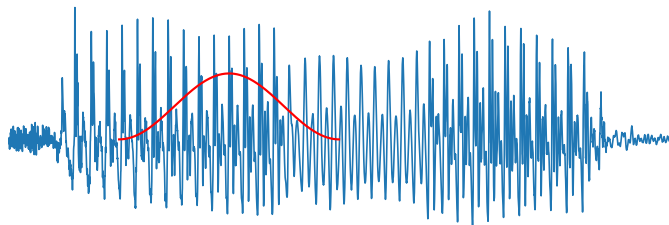

$x[n]$

$X[k, m]$

STFT Procedure

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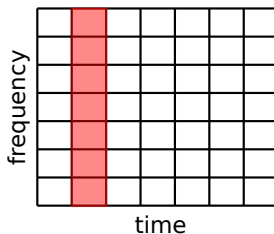
STFT Procedure

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STFT Procedure

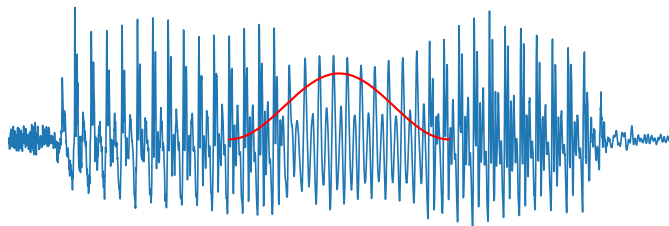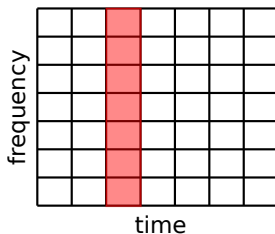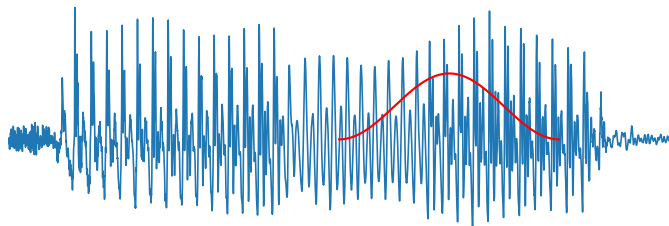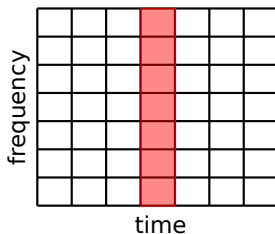Multiply signal $x[n]$ by a sliding window $w[n]$ and take FFT.

$x[n]$

$X[k, m]$

STFT Definition

The STFT of a signal, $x[n]$, is a function of frequency, k, and time, m, given by:

$$X[k,m] = \frac{1}{\sqrt{W}} \sum_{n=0}^{W-1} x[n+mh]w[n]e^{-\frac{i2\pi kn}{W}},$$

where W is the window length, h is the hop length.

STFT Pseudocode

```
x = input signal
w = window function
h = hop length
H = number of hops
X = output STFT

for m = 0 .. H
    x_clip = x[m*h : m*h + len(w)]
    X[:, m] = FFT(x_clip * w)
```

Overlap-Add (OLA)

We can reconstruct a signal $x[n]$ from its STFT, $X[k, m]$, using a method called **Overlap-Add (OLA)**:

1 Compute the inverse fast Fourier transform on each column of $X[k, m]$ to get

$$s[n, m] = \mathcal{DFT}^{-1}(X[k, m])$$

2 Scale s by a window, $w[n]$, and sum over m:

$$\tilde{x}[n] = \sum_m s[n - mh, m]w[n - mh],$$

where h is the hop length used to compute $X[k, m]$, and the range of the summation is $\left\lceil \frac{1 - W + n}{h} \right\rceil \leq m \leq \left\lfloor \frac{n}{h} \right\rfloor$.

Pseudocode for OLA

```
X = STFT
w = window function
x = output signal

initialize x[n] = 0 for all n
for m = 0 .. H
    s = IFFT(X[:, m]
    x[m*h : m*h + len(w)] += s * w
```

Perfect Reconstruction Conditions

The $\tilde{x}[n]$ resulting from OLA is a reconstruction of $x[n]$, but are they equal?

Yes, if window satisfies the constant overlap-add (COLA) condition:

$$\sum_m w[n - mh]^2 = 1$$

Note: We can also apply two different windows $w_f[n]$ and $w_b[n]$ during forward STFT and backward OLA, respectively. Then the condition is that $\sum_m w_f[n - mh]w_b[n - mh] = 1$.

Perfect Reconstruction Conditions

First, we have

$$s[n, m] = \mathcal{DFT}^{-1}\{X[k, m]\}$$
$$= \mathcal{DFT}^{-1}\left\{\mathcal{DFT}\left\{x[n + mh]w[n]\right\}\right\},$$
$$= x[n + mh]w[n]$$

Perfect Reconstruction Conditions

So, assuming $w[n]$ is COLA:

$$
\begin{aligned}
\tilde{x}[n] &= \sum_m s[n - mh, m]w[n - mh] \\
&= \sum_m (x[n]w[n - mh])w[n - mh] \\
&= x[n] \underbrace{\sum_m w[n - mh]^2}_{=1 \text{ if COLA}} \\
&= x[n]
\end{aligned}
$$