

Faster Algorithms for Sparse Fourier Transform

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Material from:

- Hassanieh, Indyk, Katabi, Price, “Simple and Practical Algorithms for Sparse Fourier Transform, SODA’12.
- Hassanieh, Indyk, Katabi, Price, “Nearly Optimal Sparse Fourier Transform”, STOC’12.

The Discrete Fourier Transform

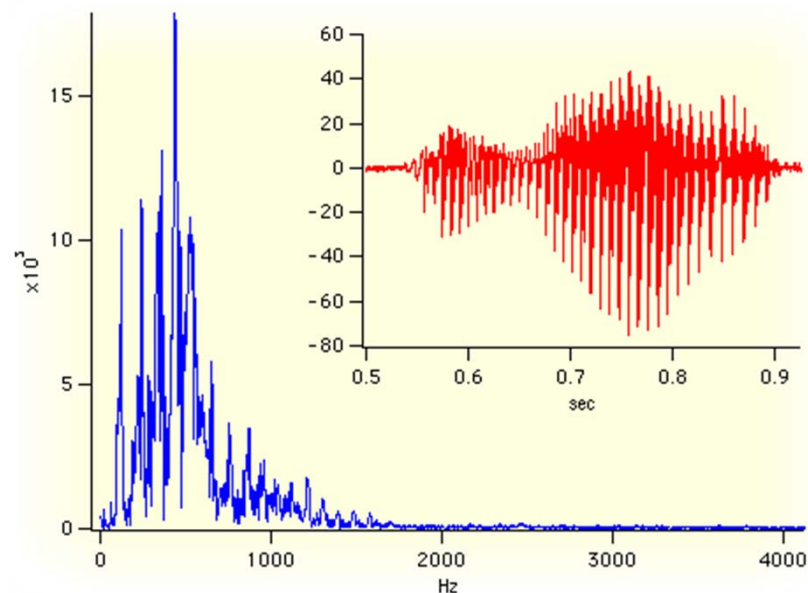
- Discrete Fourier Transform:
 - Given: a signal $\mathbf{x} \in \mathbb{C}^n$
 - Goal: compute the frequency vector $\hat{\mathbf{x}}$ such that for $f \in [1 \dots n]$:

$$\hat{\mathbf{x}}_f = \sum \mathbf{x}_t e^{-i 2\pi t f/n}$$

- Fundamental tool:
 - Compression (audio, image, video)
 - Signal processing
 - Data analysis
 - Wireless Communication
 - ...

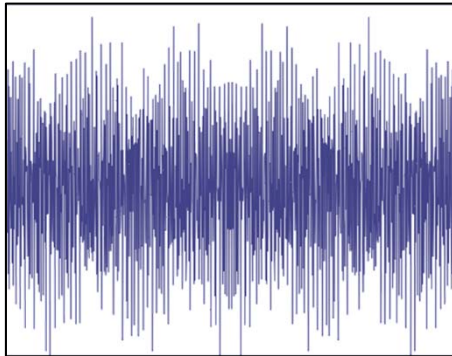
- Fastest algorithm since 1960s

FFT : $O(n \log n)$ time

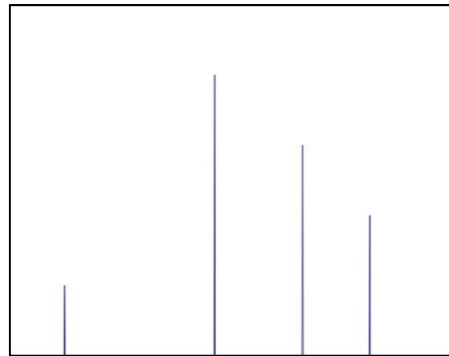


- **Sampled Audio Data (Time)**
- **DFT of Audio Samples (Frequency)**

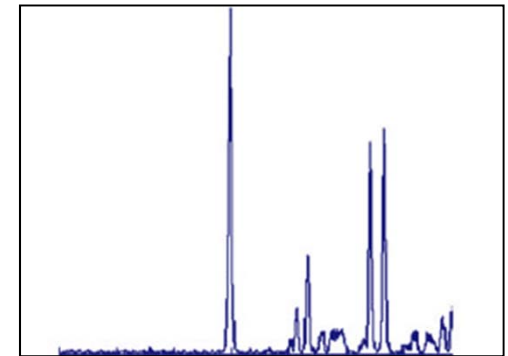
Sparse Fourier Transform



Time Domain Signal



Sparse Frequency
Spectrum



Approximately Sparse
Frequency Spectrum

- Often the Fourier transform is dominated by a small number of “peaks”
 - Only few of the frequency coefficients are nonzero.
 - An exactly k -sparse signal has only k nonzero frequency coefficients.
 - In practice : approximate a sparse signal using the k largest peaks.
- Problem : Can we recover the k -sparse frequency spectrum faster than FFT?

Previous Work

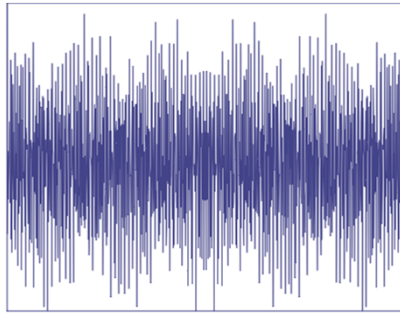
- Algorithms:
 - [KM92, Mansour92, GGIMS02, AGS03, GMS05, Iwen10, Aka10]
- Best running time: [GMS05] $O(k \log^4 n)$
 - In theory : Improves over FFT for $n/k \gg \log^3 n$
 - In practice : Large constants; need $n/k > 40,000$ to beat FFT
- Goal:
 - Theory: improve over FFT for **all** values of $k = o(n)$
 - Practice: faster runtime than FFT.

Our results

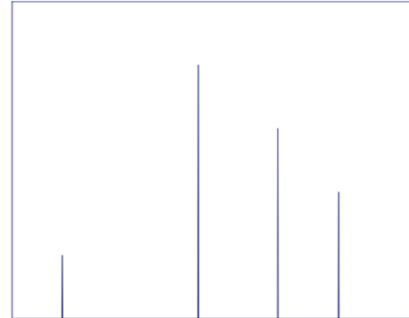
- Randomized algorithms, with large constant probability of success
- Exactly k -sparse case, recover $\hat{\mathbf{x}}$: $O(k \log n)$
 - Optimal if FFT optimal
- Approximately k -sparse case, recover $\hat{\mathbf{x}}'$:
 - Let $\text{Err}_2^k(\hat{\mathbf{x}}) = \min_{k \text{ sparse } \hat{\mathbf{x}}_k} \|\hat{\mathbf{x}} - \hat{\mathbf{x}}_k\|_2$
 - l_2/l_2 guarantee $\|\hat{\mathbf{x}}' - \hat{\mathbf{x}}\|_2 \leq c \times \text{Err}_2^k(\hat{\mathbf{x}})$: $O(k \log(n) \log(n/k))$
 - Improves over FFT for any $k \ll n$
 - l_∞/l_2 guarantee $\|\hat{\mathbf{x}}' - \hat{\mathbf{x}}\|_\infty \leq \frac{c}{\sqrt{k}} \text{Err}_2^k(\hat{\mathbf{x}})$: $O(\sqrt{nk \log n} \log n)$
 - Improves over FFT for $k \ll n/\log n$

Sparse FFT - Algorithm

Intuition



Time Domain Signal



Frequency Domain

n-point DFT : $n \log(n)$

$$\mathbf{x} \longrightarrow \hat{\mathbf{x}}$$



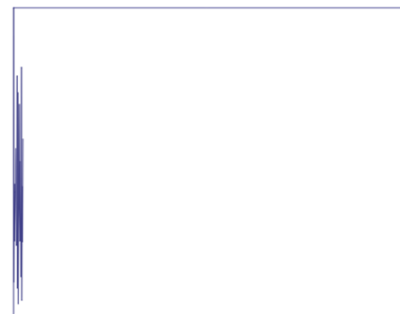
Cut off Time signal



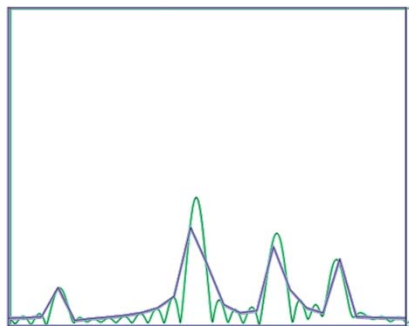
Frequency Domain

n-point DFT of first B terms : $n \log(n)$

$$\mathbf{x} \times \text{Boxcar} \longrightarrow \hat{\mathbf{x}} * \text{sinc}$$



First B samples



Frequency Domain

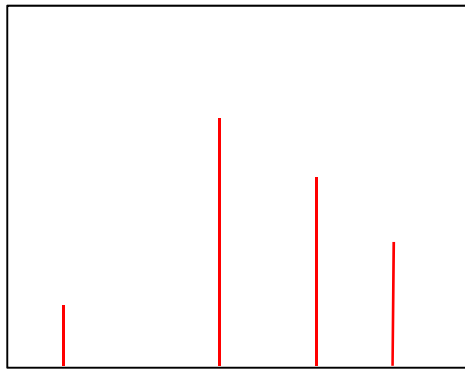
B-point DFT of first B terms : $B \log(B)$

Alias ($\mathbf{x} \times \text{Boxcar}$)

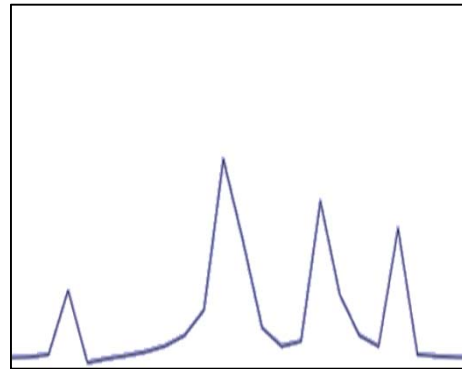


Subsample ($\hat{\mathbf{x}} * \text{sinc}$)

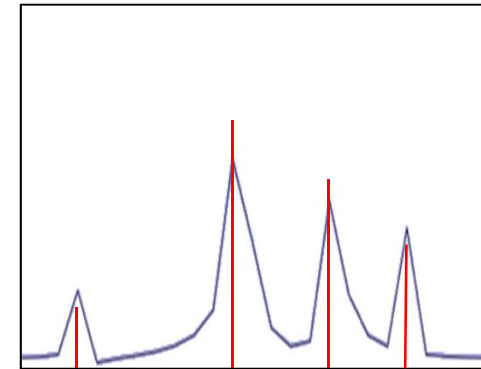
Framework



n-point DFT
of all n samples



B-point DFT
of first B sample



n frequencies hash
into B buckets

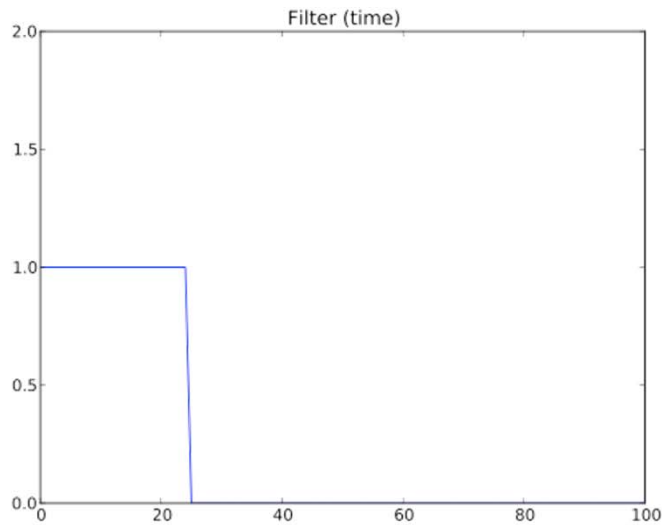
- “Hashes” the n frequency coefficients into B buckets in $O(B \log B)$ time
- n/B frequencies coefficients hash into each bucket.
- Ideally we want:
 - Value of each bucket = sum over n/B frequencies that hash to it.
 - If one large frequency in the bucket \rightarrow Estimate its value from value of the bucket.

Issues

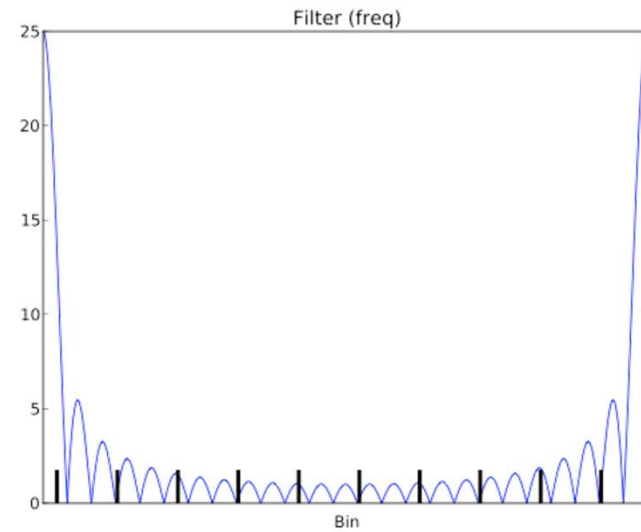
- Leakage
 - value of bucket = **Subsample** ($\hat{x} * \text{sinc}$)
 - sum over all frequencies weighted by sinc
 - frequencies outside the bucket leak power into the bucket.
 - Replace **sinc** with a better **Filter**
 - **GOAL** : **Subsample** ($\hat{x} * \text{Filter}$) = sum only over n/B frequencies that hash to the bucket
- Given these B buckets, how can we estimate the locations and values the k large frequencies?



Filter: Sinc



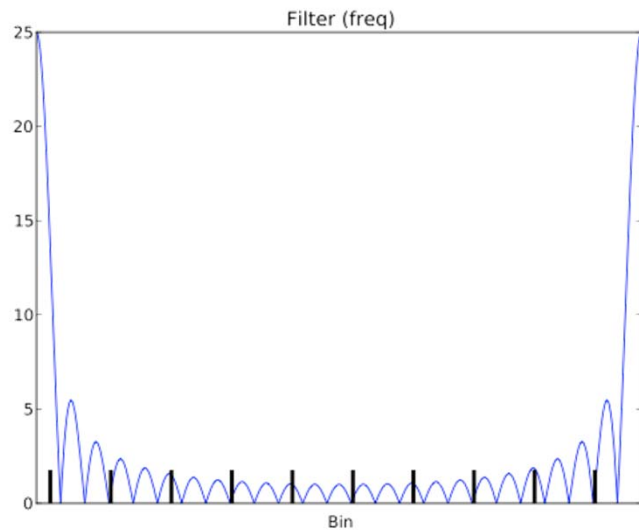
$$F = \text{Boxcar}$$



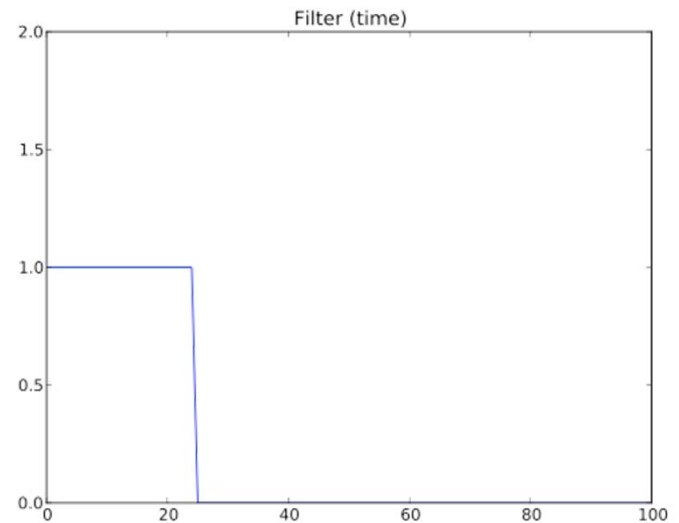
$$\hat{F} = \text{Sinc}$$

- Polynomial decay
- Leaking many buckets

Filter: Boxcar



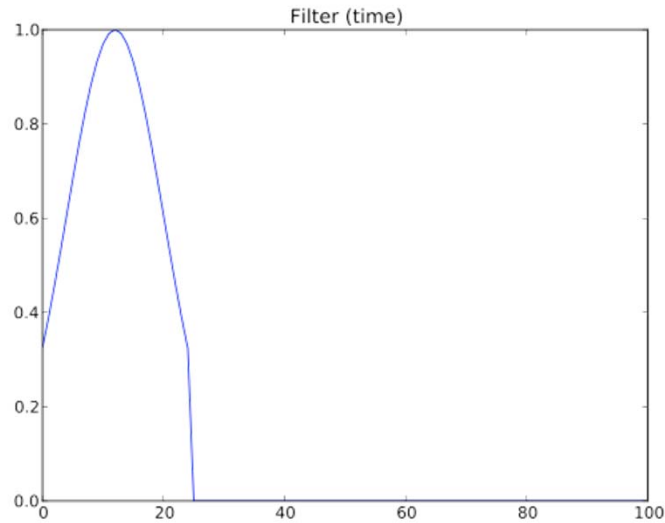
$$F = \text{Sinc}$$



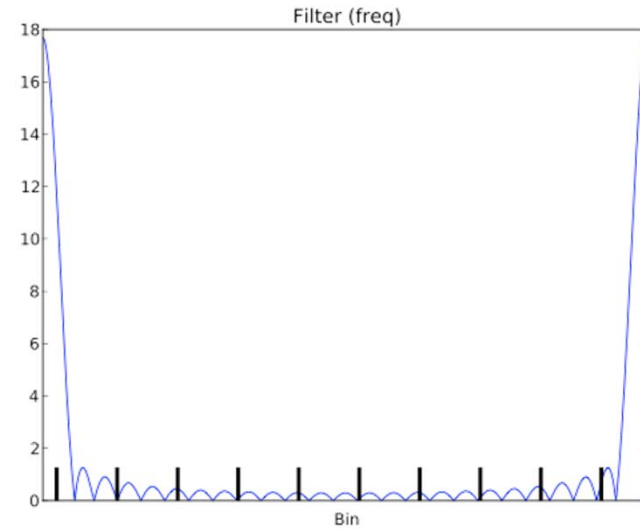
$$\hat{F} = \text{Boxcar}$$

- Large support in time domain → Cannot truncate
- Need all n time domain samples

Filter: Gaussian



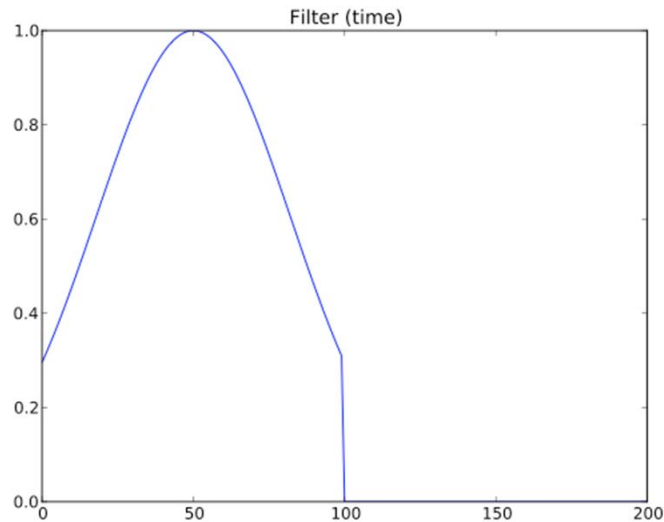
$F = \text{Gaussian}$



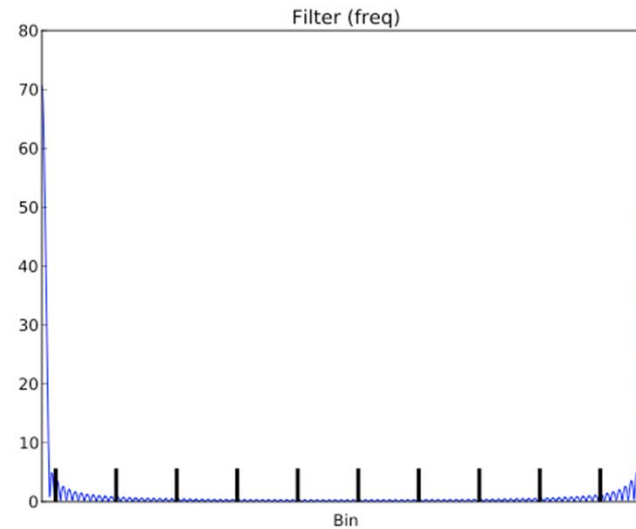
$\hat{F} = \text{Gaussian}$

- Exponential decay
- Leaking to $\sqrt{\log n}$ buckets

Filters: Wider Gaussian



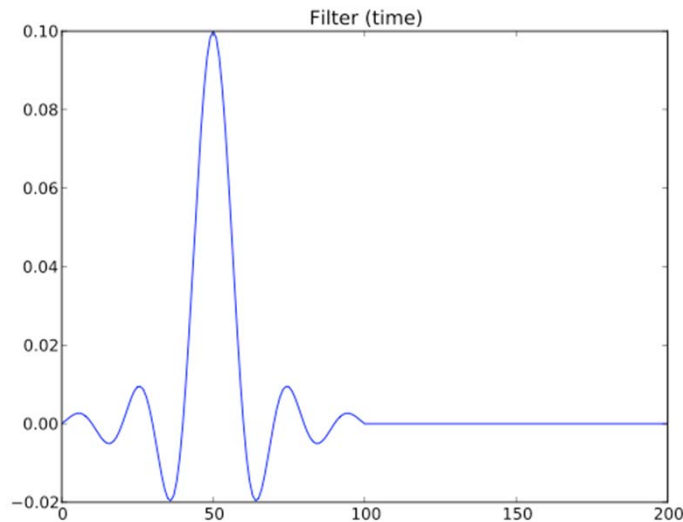
F = Wider Gaussian



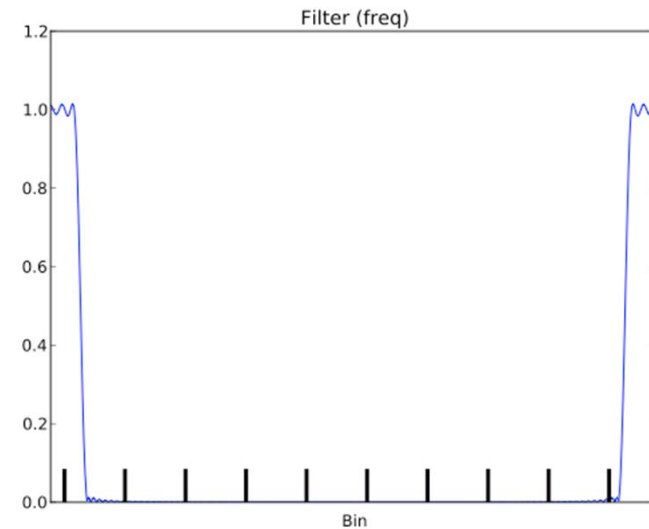
\hat{F} = Narrow Gaussian

- Exponential decay
- Leaking to 0 buckets
- But trivial contribution to the correct bucket

Filters: Sinc \times Gaussian



$$F = \text{Sinc} \times \text{Gaussian}$$

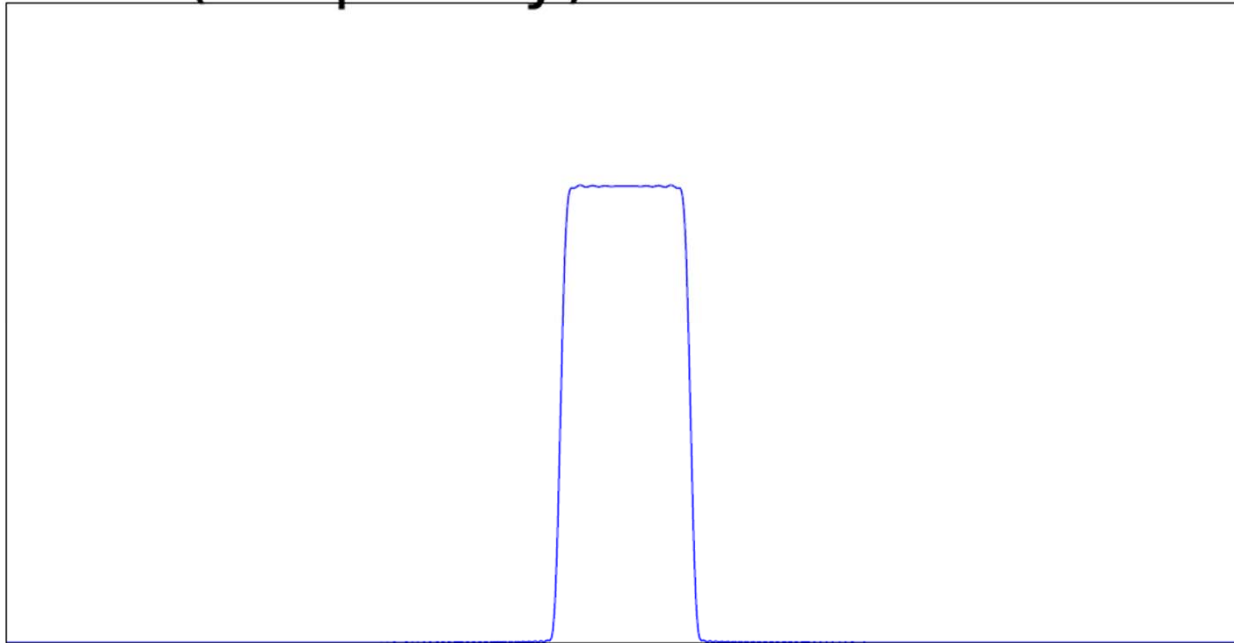


$$\hat{F} = \text{Boxcar} * \text{Gaussian}$$

- Still exponential decay
- Leaking to at most 1 bucket
- Sufficient contribution to the correct bucket
- Small support in time domain

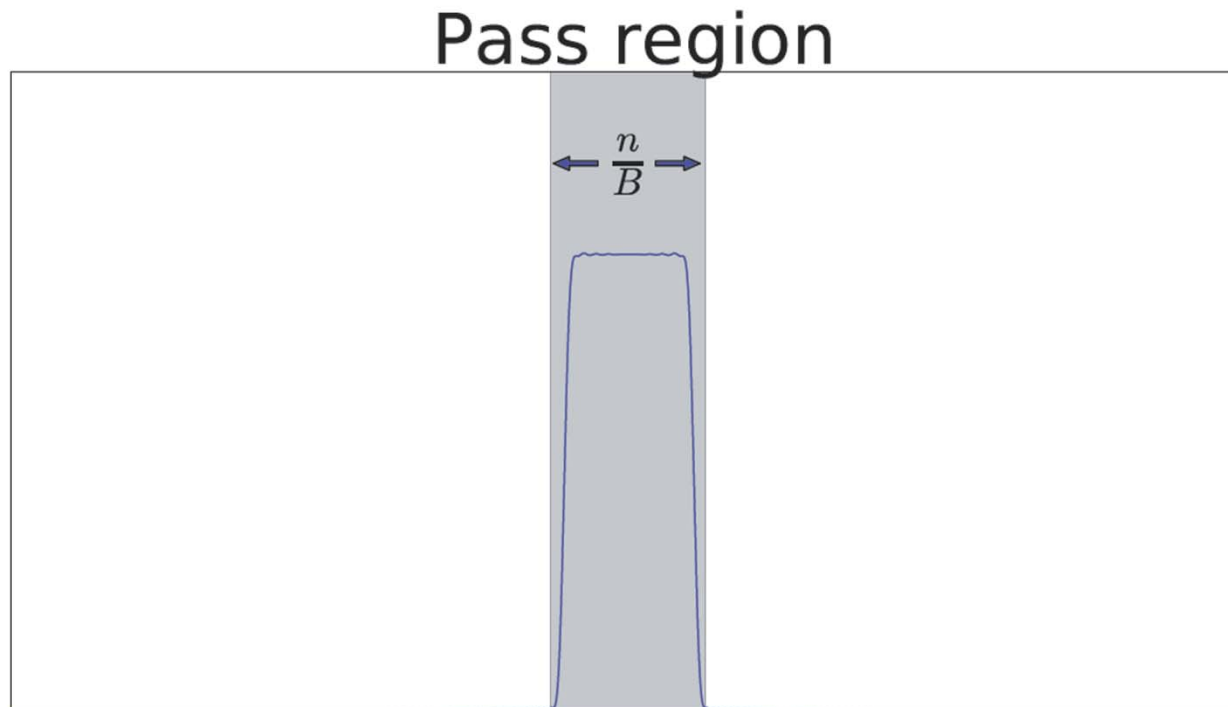
Filters: Sinc \times Gaussian

Filter (frequency): Gaussian * boxcar



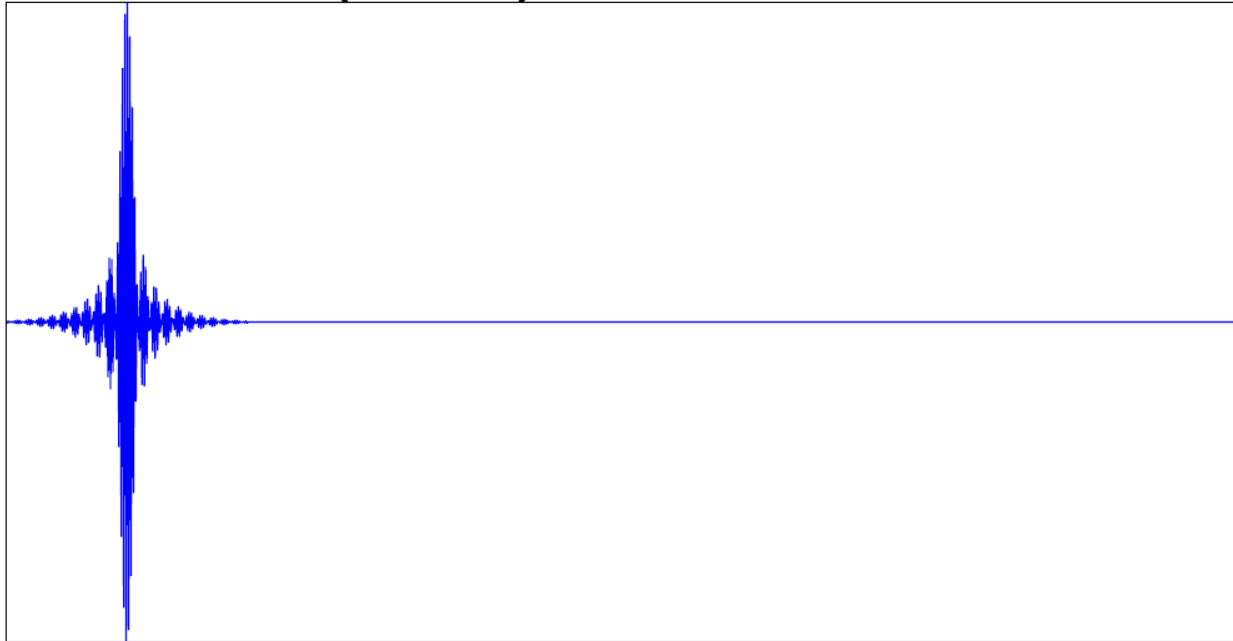
– Gaussian with standard deviation : $B\sqrt{\log n}$

Filters: Sinc \times Gaussian



- Gaussian with standard deviation : $B\sqrt{\log n}$

Filters: Sinc \times Gaussian



- Gaussian with standard deviation : $B\sqrt{\log n}$
- Filter F has a support of $B \log n$ in the time domain
- **Alias** ($\mathbf{x} \times F$) into B samples

Finding the support

- $\hat{\mathbf{y}} = \text{B-point DFT}(\mathbf{x} \times F) = \text{Subsample}(\hat{\mathbf{x}} * \hat{F})$
- Assume no collisions:
 - At most one large frequency hashes into each bucket.
 - Large frequency f_1 hashes to bucket b_1 :

$$\hat{\mathbf{y}}_{b_1} = \hat{\mathbf{x}}_{f_1} \times \hat{F}_\Delta + \text{leakage}$$

- Recall: $\text{DFT}(\mathbf{x}^\tau) = \hat{\mathbf{x}} \times e^{-i 2\pi \tau f/n}$

- $\hat{\mathbf{y}}^\tau = \text{B-point DFT}(\mathbf{x}^\tau \times F) :$

$$\hat{\mathbf{y}}^\tau_{b_1} = \hat{\mathbf{x}}_{f_1} \times e^{-i 2\pi \tau f_1/n} \times \hat{F}_\Delta + \text{leakage}$$

Finding the support

- $\hat{\mathbf{y}} = \text{B-point DFT}(\mathbf{x} \times F) = \text{Subsample}(\hat{\mathbf{x}} * \hat{F})$
- Assume no collisions:
 - At most one large frequency hashes into each bucket.
 - Large frequency f_1 hashes to bucket b_1 :

$$\hat{\mathbf{y}}_{b_1} = \hat{\mathbf{x}}_{f_1} \times \hat{F}_\Delta$$

$$\hat{\mathbf{y}}_{b_1}^1 = \hat{\mathbf{x}}_{f_1} \times e^{-i 2\pi f_1/n} \times \hat{F}_\Delta$$

$$\frac{\hat{\mathbf{y}}_{b_1}}{\hat{\mathbf{y}}_{b_1}^1} = e^{-i 2\pi f_1/n} \rightarrow \text{angle}\left(\frac{\hat{\mathbf{y}}_{b_1}}{\hat{\mathbf{y}}_{b_1}^1}\right) = -2\pi f_1/n$$

$$f_1 = -\frac{n}{2\pi} \cdot \text{angle}\left(\frac{\hat{\mathbf{y}}_{b_1}}{\hat{\mathbf{y}}_{b_1}^1}\right) \bmod n \quad \hat{\mathbf{x}}_{f_1} = \frac{\hat{\mathbf{y}}_{b_1}}{\hat{F}_\Delta}$$

Random Hashing

- Some Large frequencies collide:
 - Subtract and recurse
 - Small number of collisions \rightarrow converges in few iterations
- Every iteration needs new random hashing:
 - Permute time domain signal \rightarrow permute frequency domain
 - σ is invertible mod n :

$$\mathbf{x}'_t = \mathbf{x}_{\sigma t} \times e^{-i 2\pi t\beta/n} \qquad \hat{\mathbf{x}}'_f = \hat{\mathbf{x}}_{\sigma^{-1}f + \beta}$$

Algorithm

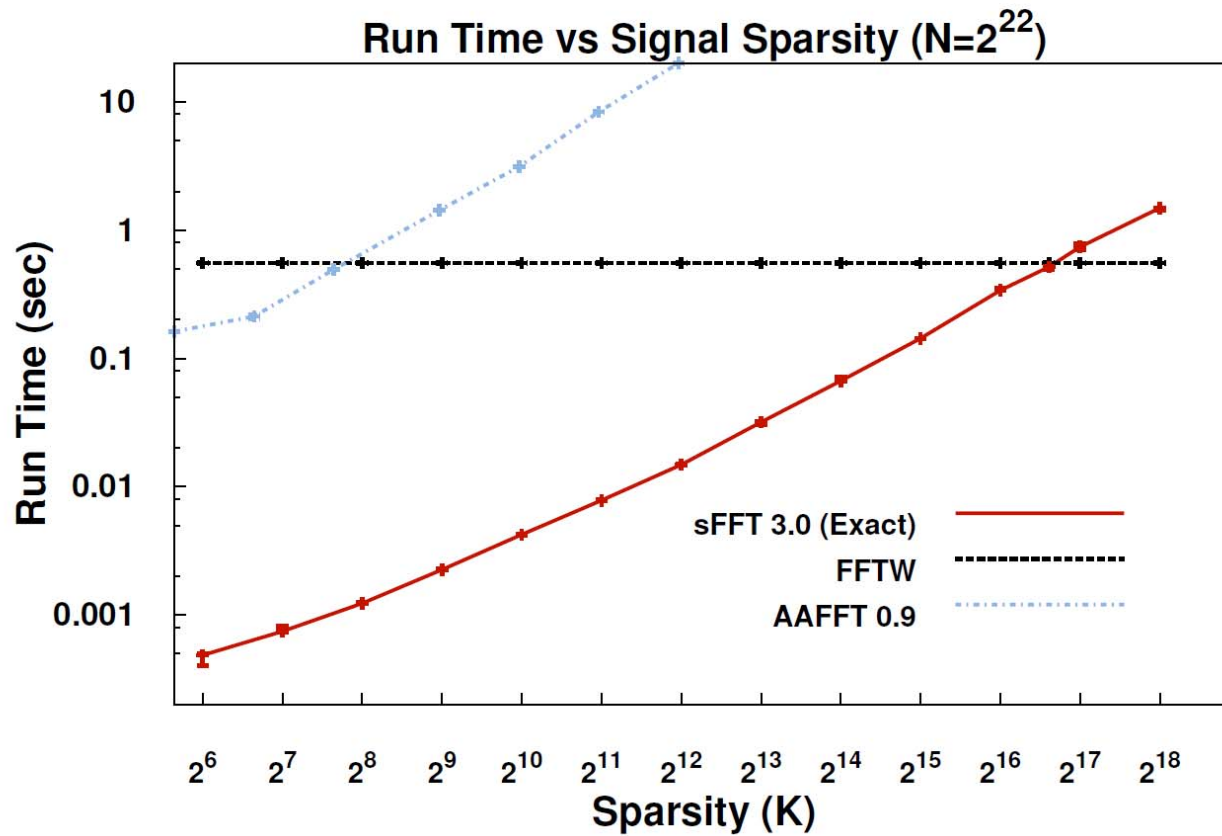
- Iteration i :
 - $B_i \propto k / 2^{i-1}$
 - Permute spectrum : $\mathbf{x}'_t = \mathbf{x}_{\sigma t} \times e^{-i 2\pi t\beta/n}$
 - $\hat{\mathbf{y}} = B_i$ -point DFT ($\mathbf{x}' \times F$) = Subsample ($\hat{\mathbf{x}}' * \hat{F}$)
 - Repeat with time shift to get $\hat{\mathbf{y}}^\tau$
 - Subtract large frequencies recovered in previous iterations
 - Recover locations and values of remaining large frequencies
- Iteration i recovers $k / 2^i$ of the large frequencies with probability $3/4$ in $O(B_i \log n)$ time

Theorem: Recover $\hat{\mathbf{x}}$ in $O(k \log n)$ with probability $3/4$

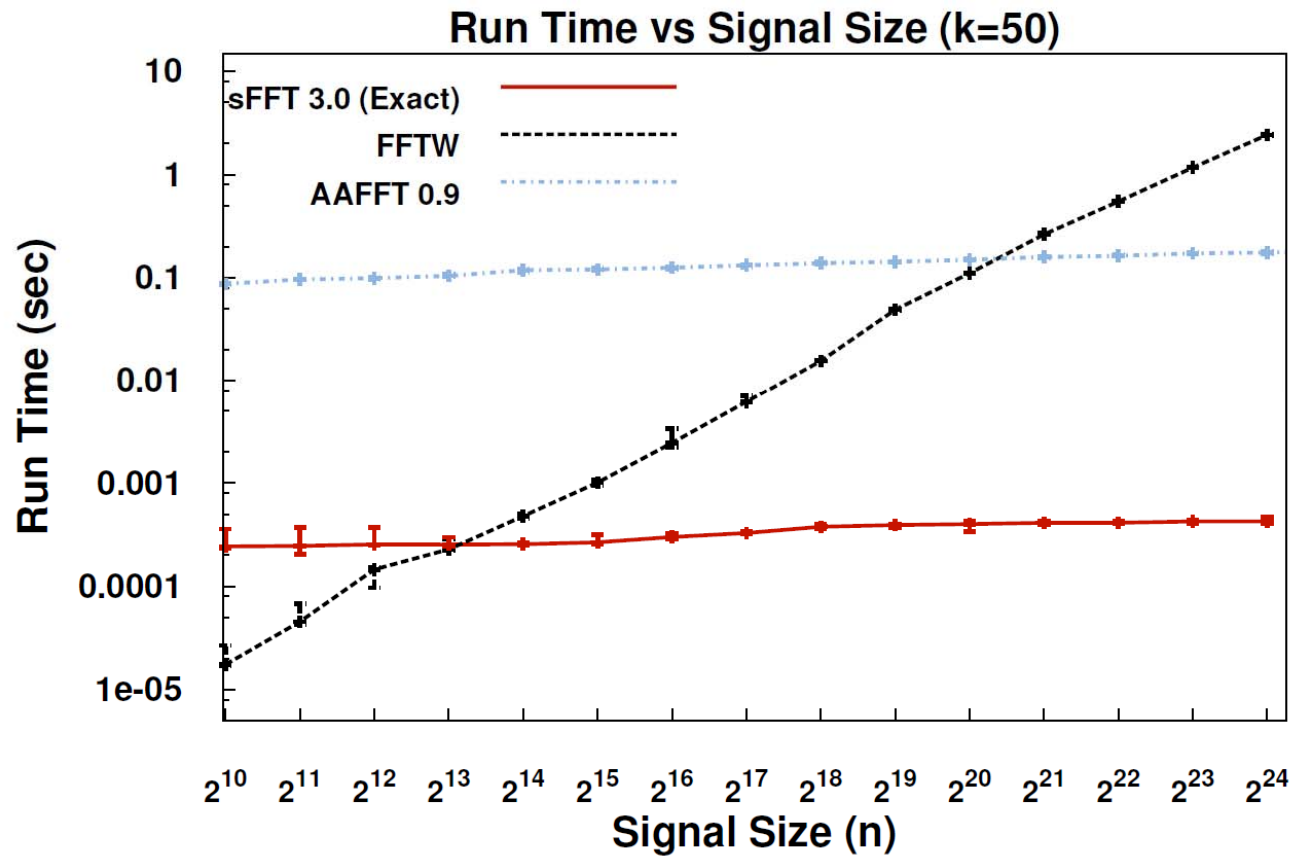
Experiments

(variant exactly k-sparse algorithm)

Experiments



Experiments, ctd



Conclusions

- Sparse Fourier Transform with running times :
 - $O(k \log n)$ for exactly sparse case
 - $O(k \log(n) \log(n/k))$ for approximately sparse case
 - Improves over FFT for $k \ll n$
- Significant improvement in practice : $n/k > 32$
- $k \log n$ time for **approximately** sparse signals?
 - Not clear: $k \log(n/k)$ samples needed, extra $\log n$ for processing.