

Learning Signals with Simple Fourier Transforms

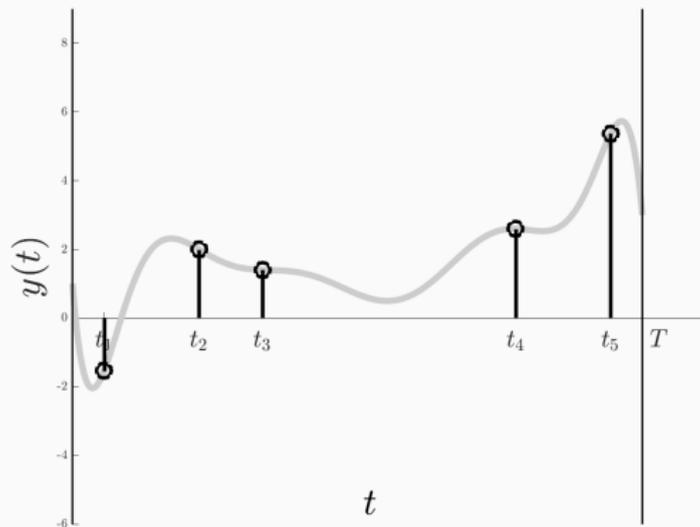
Christopher Musco, Princeton University

Solving an old problem in signal processing using tools from randomized algorithms.

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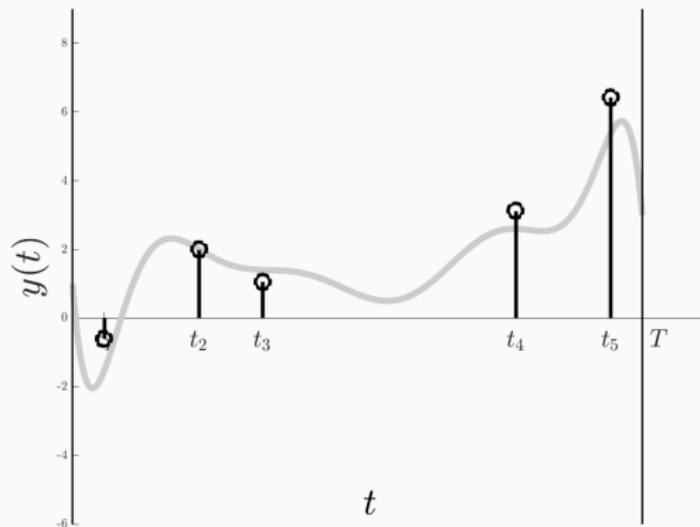
(matrix sketching, randomized SVD, Laplacian linear systems.)

BASIC PROBLEM



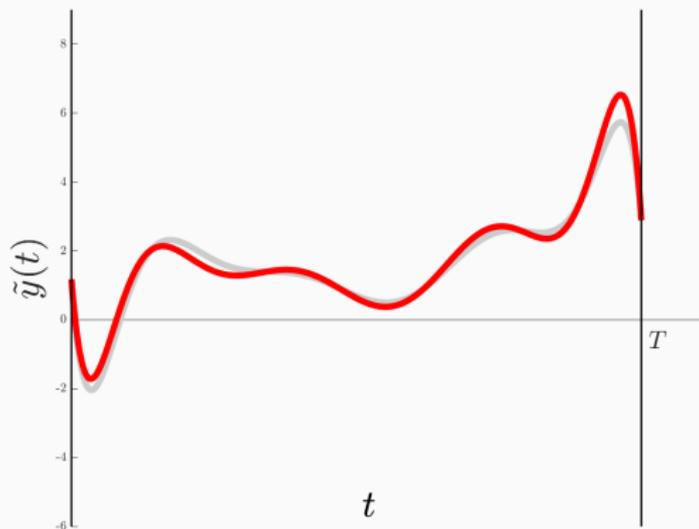
Observe signal y at sample locations $t_1, \dots, t_q \in [0, T]$.

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(possibly with noise)

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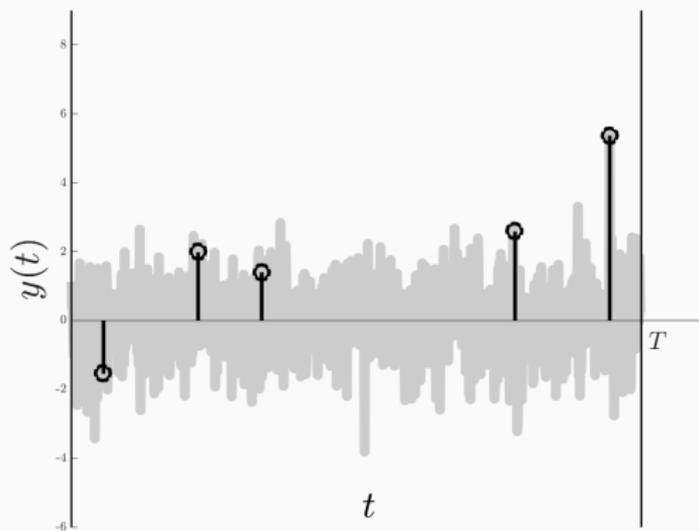


Goal: Recover signal \tilde{y} which is close to y .

Central questions:

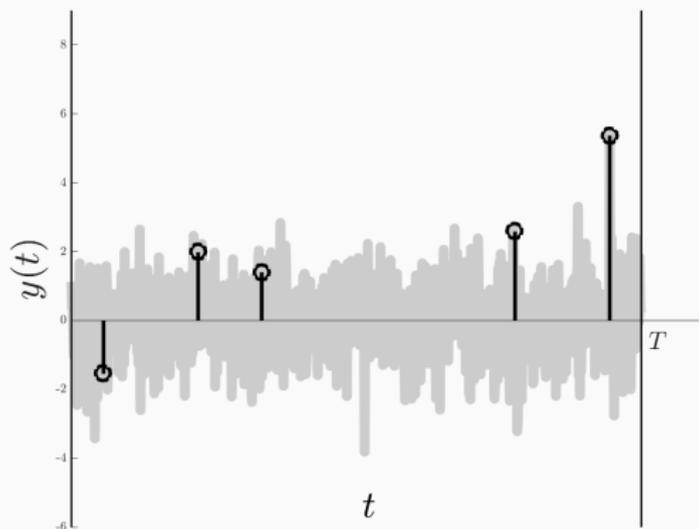
- How many samples do we need to approximately reconstruct y on $[0, T]$?
- How can we compute and represent \tilde{y} in a computationally efficient way?

CONTINUOUS SIGNAL RECONSTRUCTION



Naively, this problem is ill-posed.

CONTINUOUS SIGNAL RECONSTRUCTION



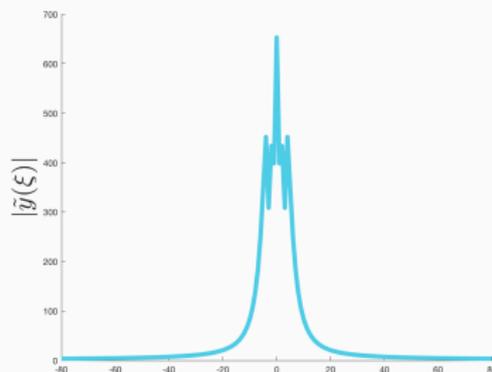
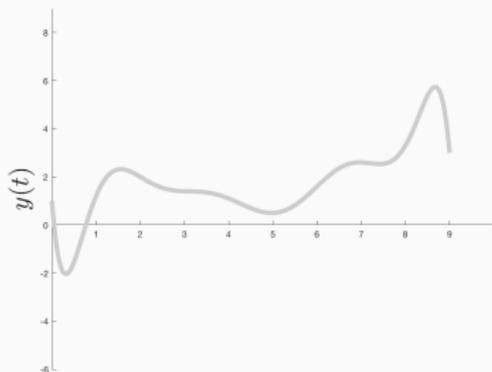
Naively, this problem is ill-posed.

We need to assume y is smooth or structured in some way.

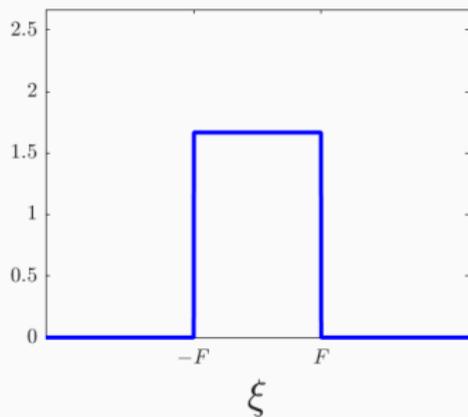
FOURIER TRANSFORM STRUCTURE

In science and engineering, we often impose structure by assuming y has a “simple” Fourier transform.

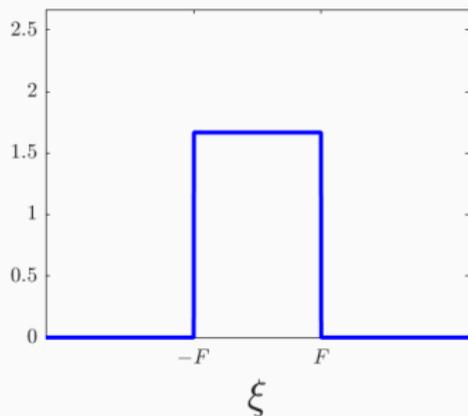
$$\hat{y}(\xi) = \int_{-\infty}^{\infty} y(t)e^{-2\pi i t \xi} dt.$$



Standard assumption: y is **bandlimited**, i.e. $\hat{y}(\xi) = 0$ for $|\xi| > F$.



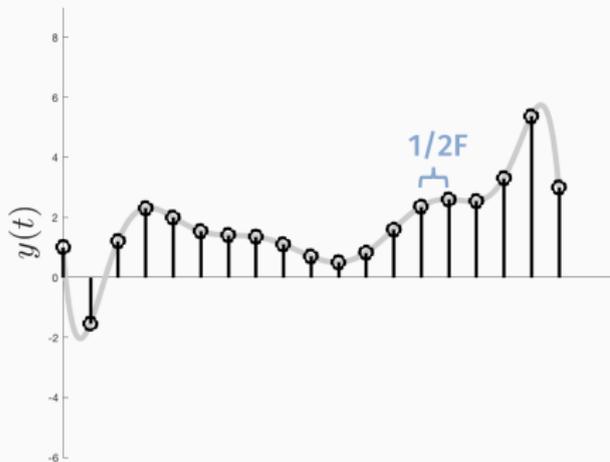
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Shannon, Whitaker, Nyquist, Kotelnikov – foundations of modern signal processing and information theory.

BANDLIMITED SIGNAL RECONSTRUCTION

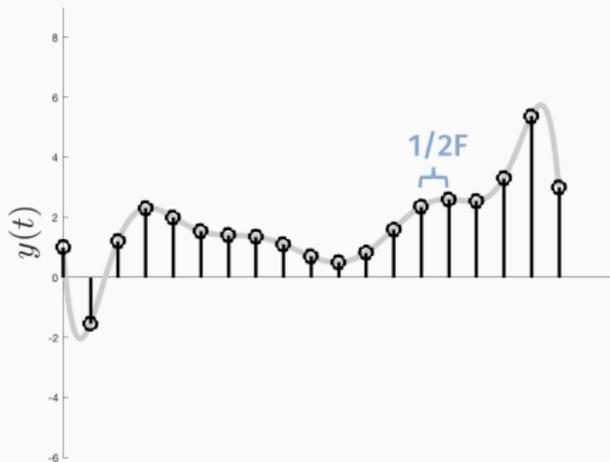
Shannon-Nyquist Theorem: $O(F)$ samples per second suffice.



Uniform Nyquist sampling.

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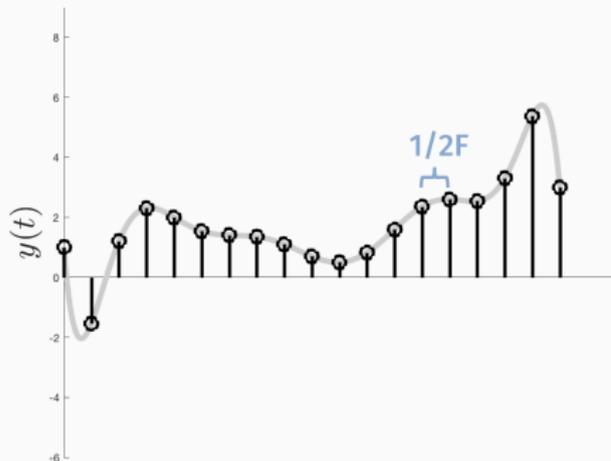
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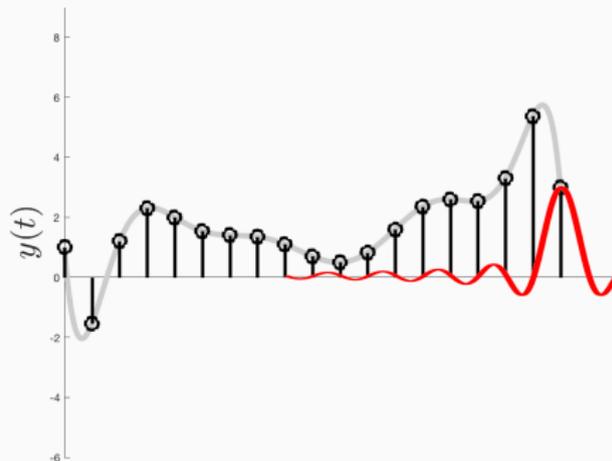
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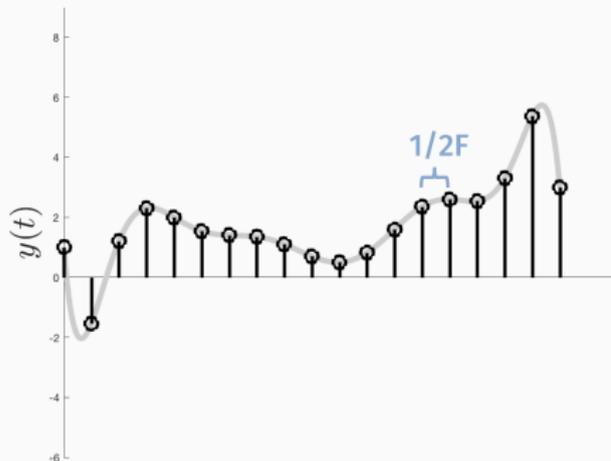


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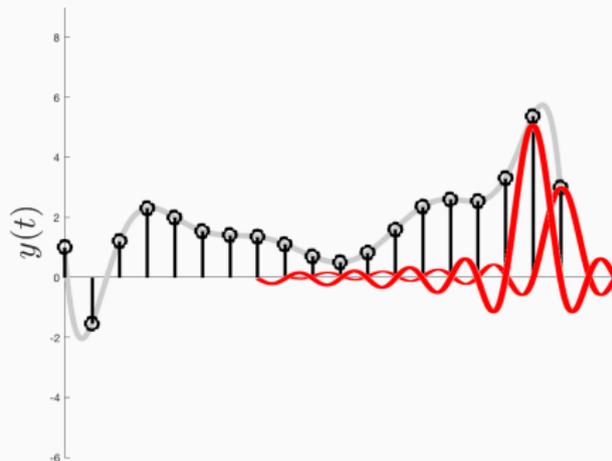
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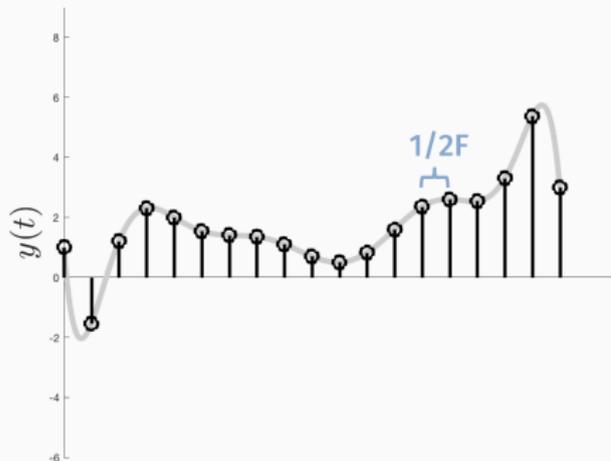


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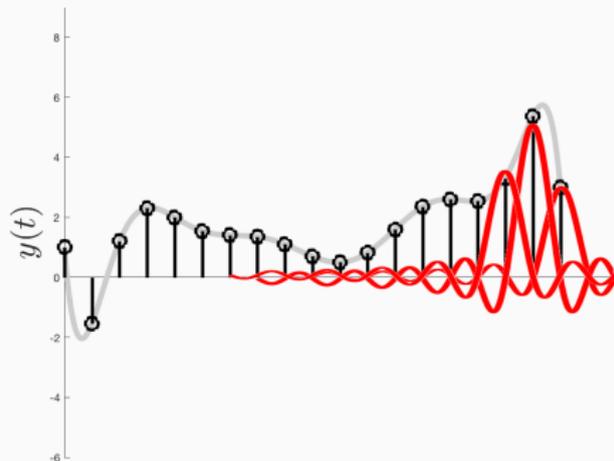
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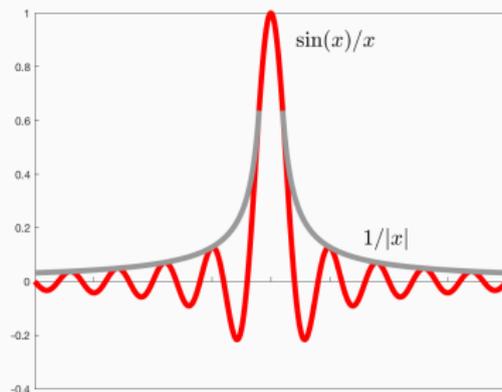
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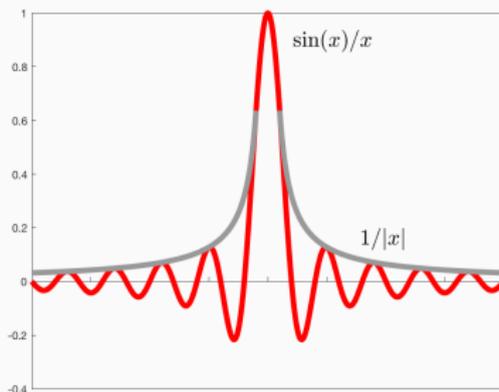
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$$\tilde{y}(t) = \sum_{s=0}^{FT} \text{sinc}(y(t+s) \cdot F)$$

$O(FT/\epsilon)$ samples for ϵ error at best.

This problem was resolved in the 1960s (+ some):

Theorem (Slepian, Landau, Pollak and Rokhlin et al.)

*Suppose $y = \mathcal{F}^*x$ for bandlimited x and we observe $y + n$.*

With $O(FT + \log(1/\epsilon))$ samples, it's possible to recover \tilde{y} with:

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PROLATE SPHEROIDAL WAVE FUNCTIONS

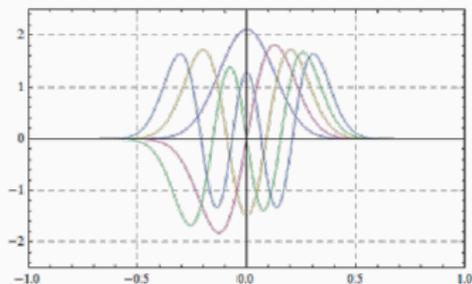
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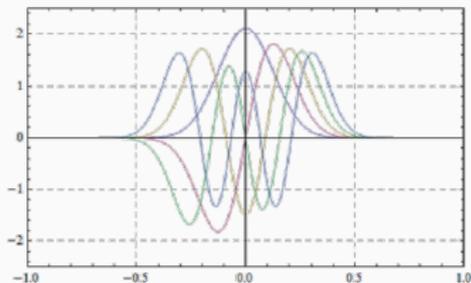
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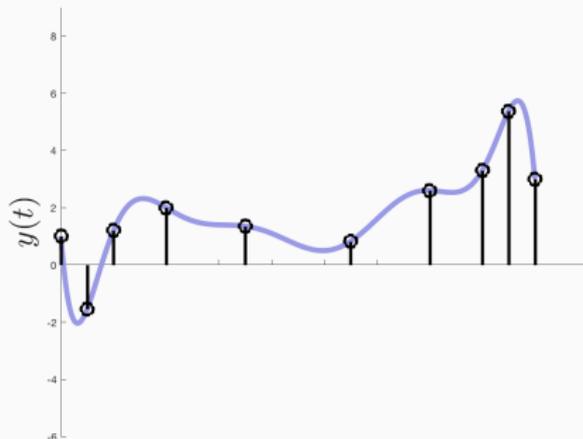
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Can project to this basis with numerical quadrature.

Important conclusion: For interpolation on a finite interval $[0, T]$, uniform sampling is suboptimal.

NON-UNIFORM SAMPLING

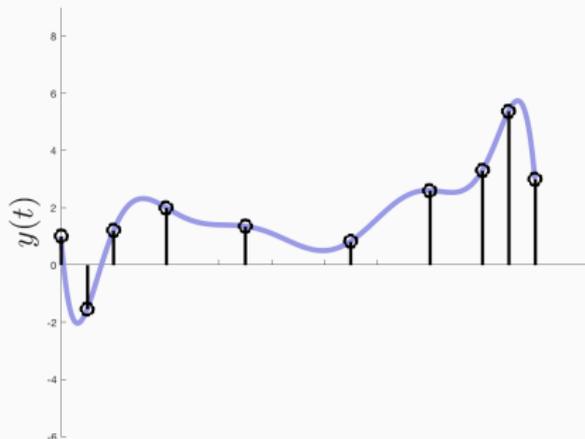
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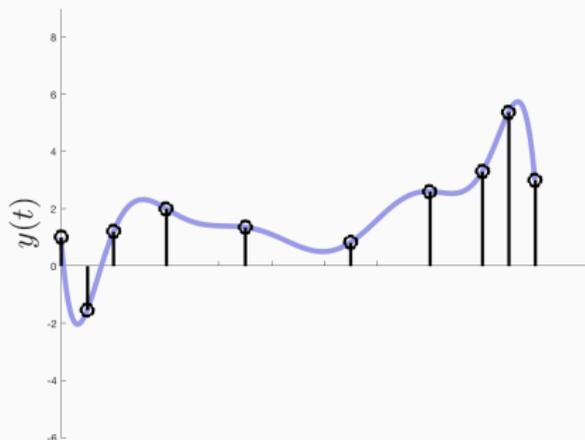


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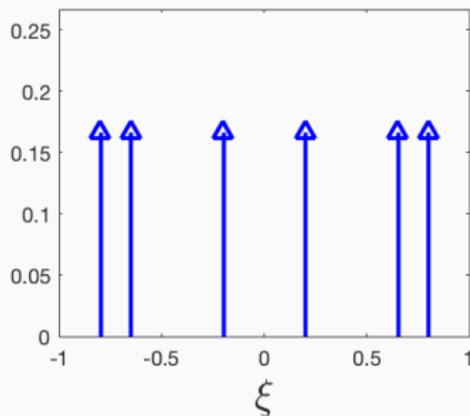
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Not surprising if you think about polynomial interpolation.

What about Fourier structure beyond a bandlimit?

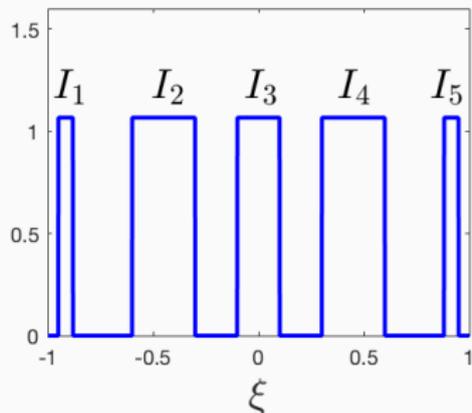
FOURIER TRANSFORM STRUCTURE

E.g. y is **Fourier sparse**. $\hat{y}(\xi)$ is supported on k frequencies.



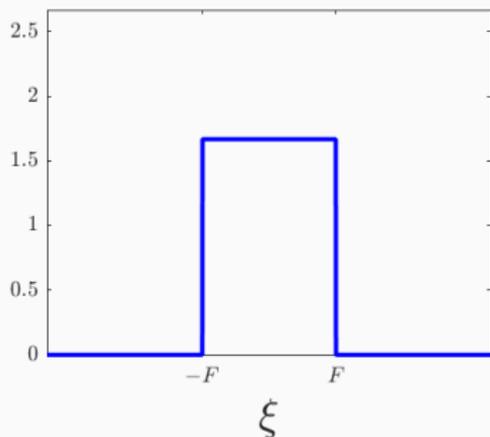
Compressed sensing, applications in medical imaging,
microscopy, RADAR, etc.

E.g. y is **multiband**, i.e. $\hat{y}(\xi)$ supported on k intervals.



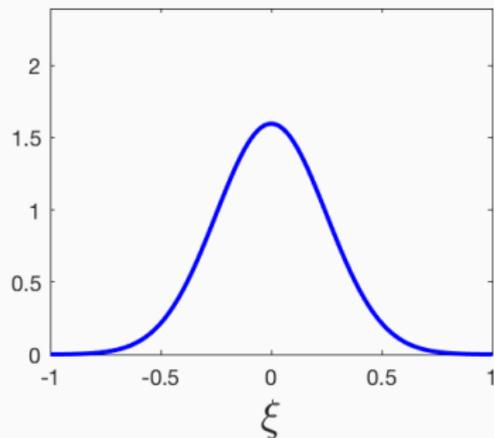
FOURIER TRANSFORM STRUCTURE

Bayesian perspective: instead of “allowing” or “disallowing” certain frequencies, we can consider any prior distribution on y 's power spectral density.



Bandlimited.

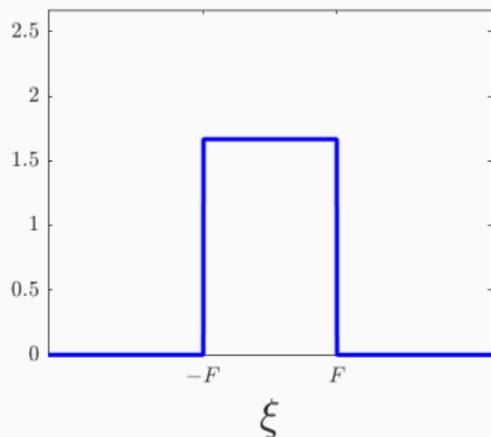
VS.



Gaussian prior.

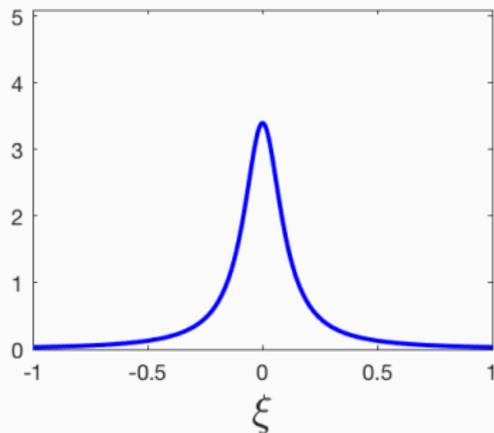
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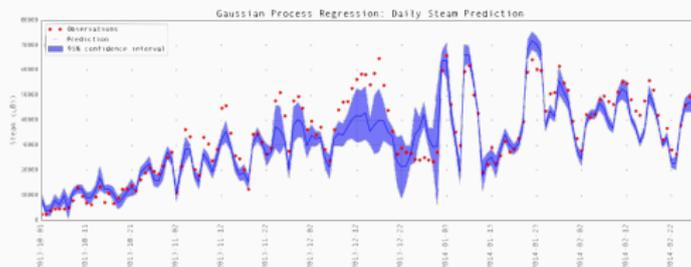
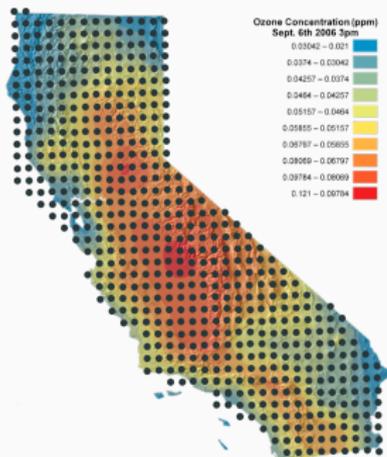
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Cauchy-Lorentz Prior.

Smooth penalties underly Gaussian process regression,
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Countless applications in environmental science, geostatistics, image processing, economics, time series analysis, etc.

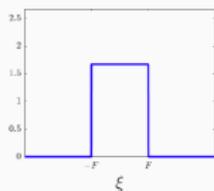
Knowledge gap: 50 years after PSWFs, no finite sample guarantees or efficient recovery algorithms for these popular problems on a finite interval $[0, T]$.

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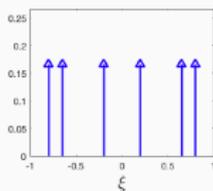
With the exception of Fourier sparse functions.
(Chen, Kane, Price, Song FOCs 2016, Chen, Kane 2018).

Our results:

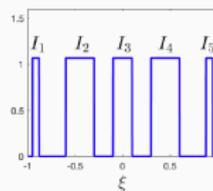
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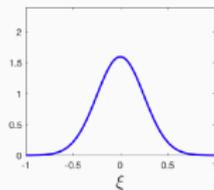
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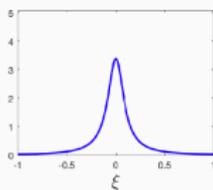
Sparse.



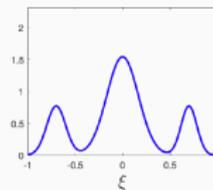
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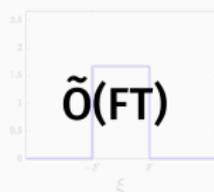
Cauchy-
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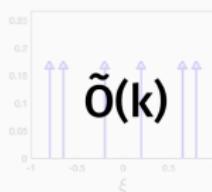
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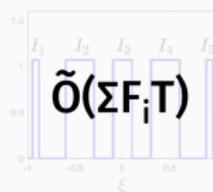
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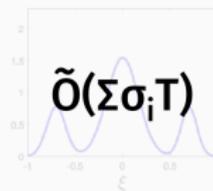
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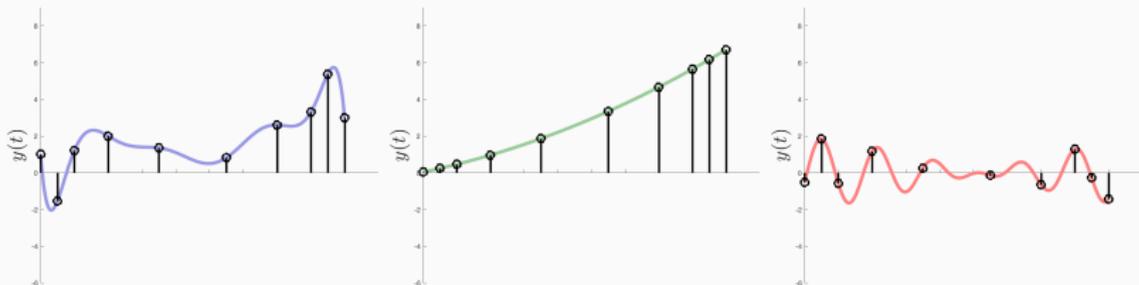
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Typically a quadratic improvement on uniform sampling.

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1. Characterize optimal sample complexity for any prior distribution μ .
2. **Universal non-uniform sampling scheme** that matches this complexity up to logarithmic factors.
3. Efficient algorithm to pair with this sampling scheme that works for essentially all priors used in practice.

All using tools from discrete randomized algorithms!

On arXiv soon:

“Universal Sampling Strategies for Learning Signals with Simple Fourier Transforms”

Joint work with:

Haim Avron (TAU)

Michael Kapralov (EPFL)

Cameron Musco (MSR)

Ameya Velingker (EPFL)

Amir Zandieh (EPFL)

Definition (Weighted Inverse Fourier Transform)

For any probability distribution μ over \mathbb{R} and frequency domain function g , let:

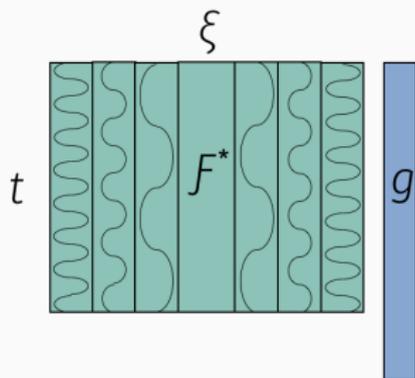
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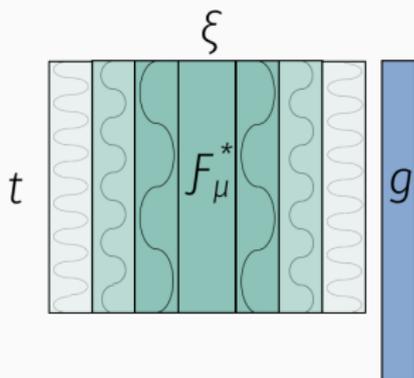
Standard inverse Fourier transform.

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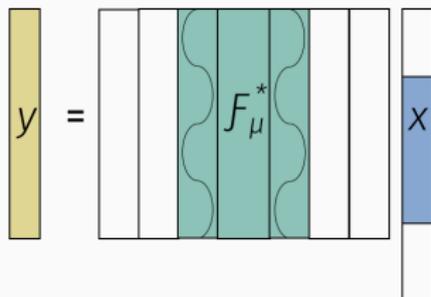
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Given: $y = \mathcal{F}_\mu^* x + n$.

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Easiest to understand for bandlimited, sparse, or multiband.



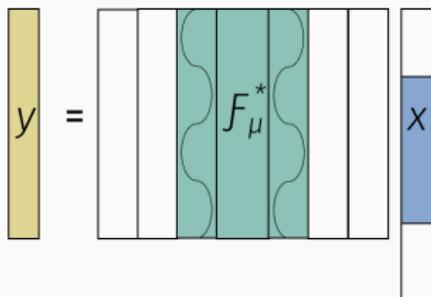
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There's a natural Bayesian formulation for non-uniform priors.

NATURAL APPROACH

Suffices to return $\tilde{y} = F_\mu^* \tilde{g}$ for any constant factor approximation \tilde{g} to the regression problem:

$$\min_g \left\| \begin{array}{c} t \in [0, T] \\ \xi \in [-\infty, \infty] \\ F_\mu^* \\ g \end{array} \right\|_T^2 - \left\| \begin{array}{c} y \\ n \end{array} \right\|_T^2 + \epsilon \left\| \begin{array}{c} g \end{array} \right\|_\mu^2$$

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If \tilde{g} satisfies:

$$\|y + n - \mathcal{F}_\mu^* \tilde{g}\|_T^2 + \epsilon \|\tilde{g}\|_\mu^2 \leq C \cdot [\min_g \|y + n - \mathcal{F}_\mu^* g\|_T^2 + \epsilon \|g\|_\mu^2],$$

then: $\|y - \mathcal{F}_\mu^* \tilde{g}\|_T^2 \leq O(C) \cdot [\|n\|_T^2 + \epsilon \|x\|_\mu^2].$

NATURAL APPROACH

Suffices to return $\tilde{y} = F_\mu^* \tilde{g}$ for any constant factor approximation \tilde{g} to the regression problem:

$$\min_g \left\| \begin{array}{c} t \in [0, T] \\ \xi \in [-\infty, \infty] \\ F_\mu^* \\ g \end{array} \right\|_T^2 - \left\| \begin{array}{c} y \\ n \end{array} \right\|_T^2 + \epsilon \left\| \begin{array}{c} g \end{array} \right\|_\mu^2$$

If \tilde{g} satisfies:

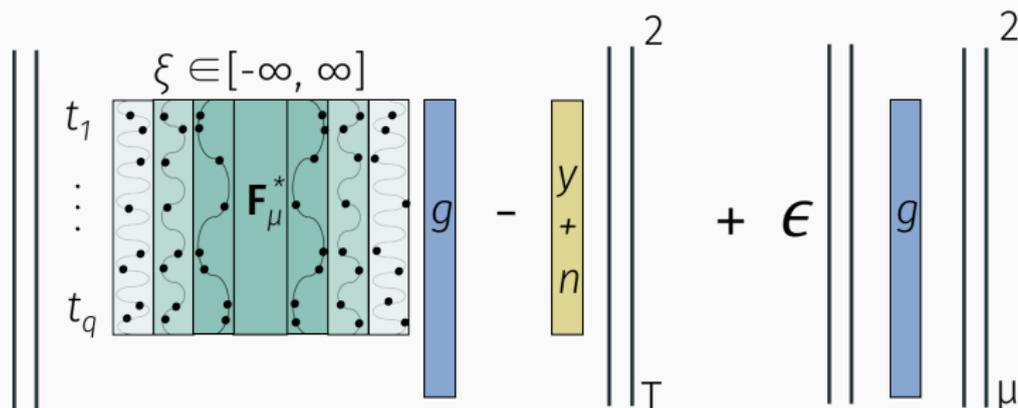
$$\|y + n - \mathcal{F}_\mu^* \tilde{g}\|_T^2 + \epsilon \|\tilde{g}\|_\mu^2 \leq C \cdot [\min_g \|y + n - \mathcal{F}_\mu^* g\|_T^2 + \epsilon \|g\|_\mu^2],$$

then: $\|y - \mathcal{F}_\mu^* \tilde{g}\|_T^2 \leq O(C) \cdot [\|n\|_T^2 + \epsilon \|x\|_\mu^2].$

Solution by discretization.

TIME DOMAIN DISCRETIZATION

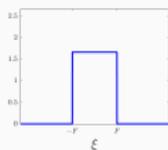
Selecting time samples discretizes time domain.



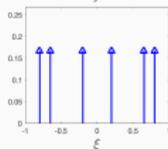
What about Fourier domain?

We can avoid discretization entirely as long as we have a closed form representation of $\hat{\mu}(t)$.

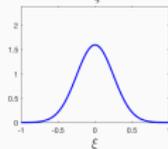
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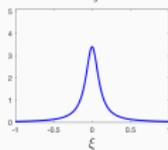
$$\hat{\mu}(t) = \text{sinc}(t)$$



$$\hat{\mu}(t) = \sum_{j=1}^k e^{-2\pi i(t)}$$



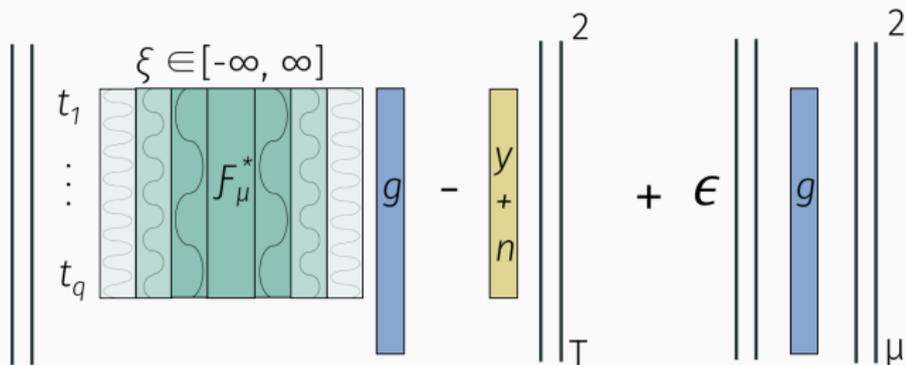
$$\hat{\mu}(t) = e^{-|t|^2}$$



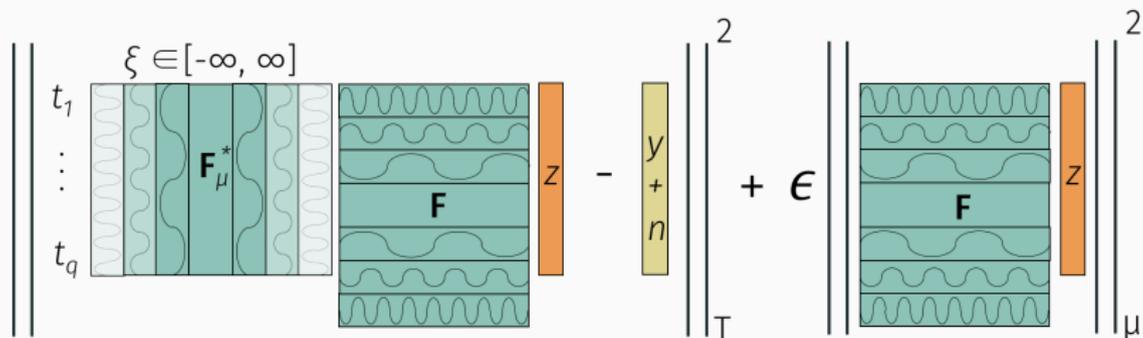
$$\hat{\mu}(t) = e^{-|t|}$$

$\hat{\mu}$ is referred to as the autocorrelation function, the semivariogram, the kernel function, etc.

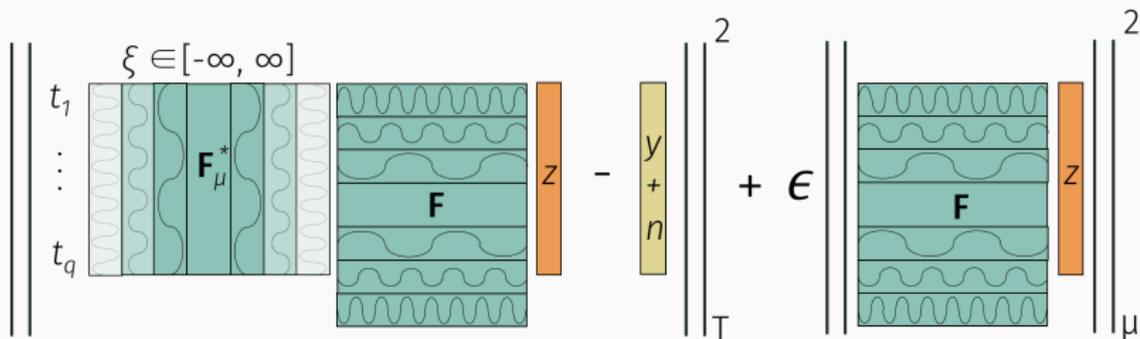
HANDLING FOURIER DOMAIN



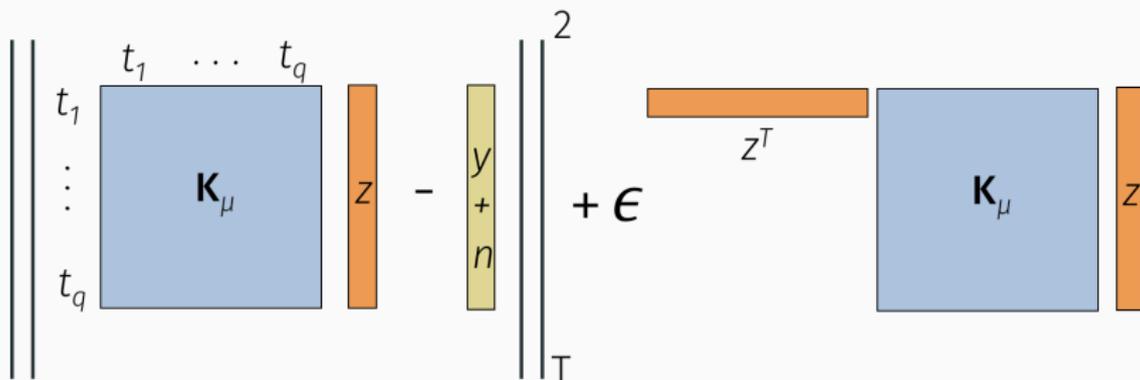
equivalent to



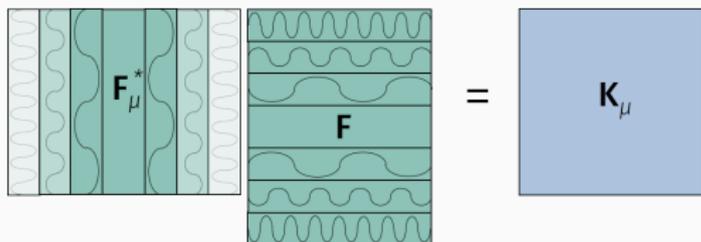
HANDLING FOURIER DOMAIN



equivalent to

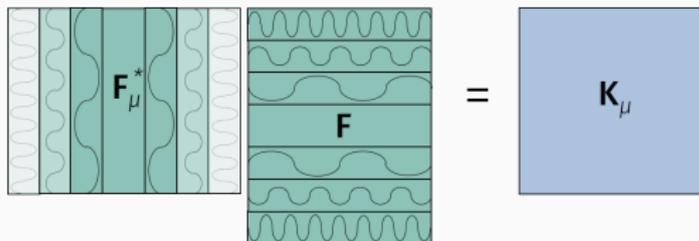


EQUIVALENCE TO KERNEL REGRESSION



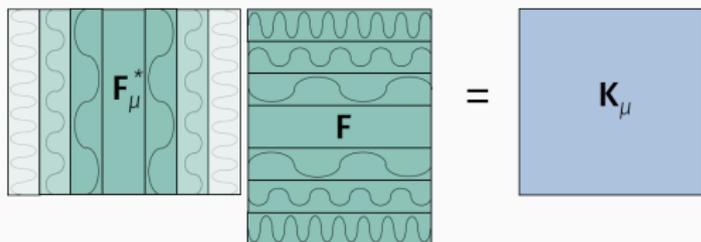
$$K_{\mu}(i, j) = \int_{-\infty}^{\infty} e^{2\pi i t_i \xi} e^{-2\pi i t_j \xi} \mu(\xi) d\xi$$

EQUIVALENCE TO KERNEL REGRESSION



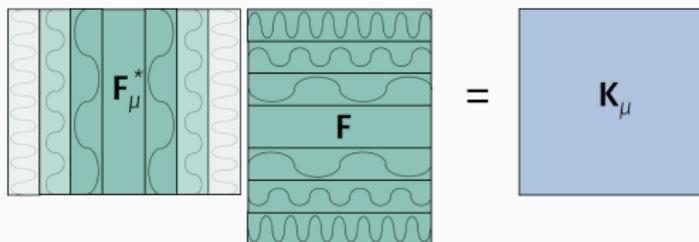
$$\begin{aligned} K_{\mu}(i, j) &= \int_{-\infty}^{\infty} e^{2\pi i t_i \xi} e^{-2\pi i t_j \xi} \mu(\xi) d\xi \\ &= \int_{-\infty}^{\infty} \mu(\xi) e^{-2\pi i (t_j - t_i) \xi} d\xi \end{aligned}$$

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We can construct \mathbf{K} in $O(q^2)$ time.

EQUIVALENCE TO KERNEL REGRESSION

- Sample t_1, \dots, t_q .
- Compute $\hat{\mu}(t_i - t_j)$ to build $q \times q$ kernel matrix \mathbf{K} .
- Solve $\mathbf{z} = (\mathbf{K} + \epsilon \mathbf{I})^{-1} [y_n(t_1), \dots, y_n(t_q)]$.
- Evaluate $\tilde{y}(t) = \sum_{i=1}^q \mathbf{z}_i \hat{\mu}(t_i - t) \cdot f(t_i)$.

$$\begin{pmatrix} t_1 \\ \vdots \\ t_q \end{pmatrix} \begin{matrix} t_1 & \dots & t_q \\ \mathbf{K}_\mu \end{matrix} \begin{matrix} \mathbf{z} \end{matrix} - \begin{matrix} y_n \\ + \\ n \end{matrix} + \epsilon \begin{matrix} \mathbf{z}^T \\ \mathbf{K}_\mu \end{matrix} \begin{matrix} \mathbf{z} \end{matrix}$$

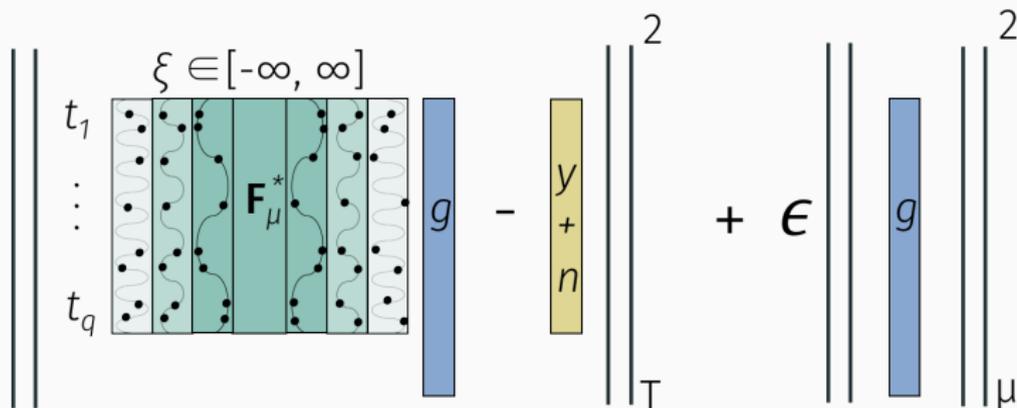
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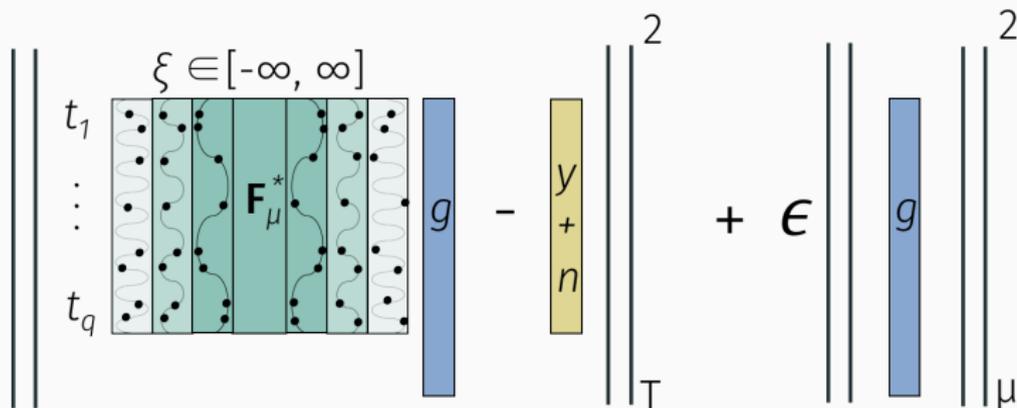
TIME DOMAIN DISCRETIZATION

Key Challenge: How to select samples in time domain.



TIME DOMAIN DISCRETIZATION

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Approach: Lean on well developed theory for randomly sampling discrete regression problems.

For an approximate solution, suffices to sample rows (i.e. time points) by their **statistical leverage score**:

$$\tau_{\mu,\epsilon}(t) = \max_g \frac{\frac{1}{T} |\mathcal{F}_\mu^* g(t)|^2}{\|\mathcal{F}_\mu^* g\|_T^2 + \epsilon \|g\|_\mu^2}$$

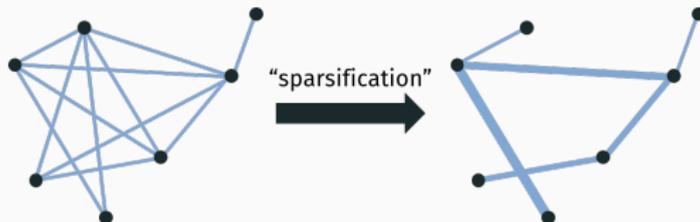
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$$0 \leq \tau_{\mu,\epsilon}(t) \leq 1$$

LEVERAGE SCORE SAMPLING

$\tau_{\mu, \epsilon}$ is a regularized version of **effective resistance**, a central quantity in recent work on randomized algorithms for graph problems and matrix sketching.



[Drineas, Mahoney, Muthukrishnan 2006]

[Spielman, Srivastava 2008]

How many samples are required?

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$$\int_0^T \tau_{\mu,\epsilon}(t) dt = S_{\mu,\epsilon} = \text{“statistical dimension”}.$$

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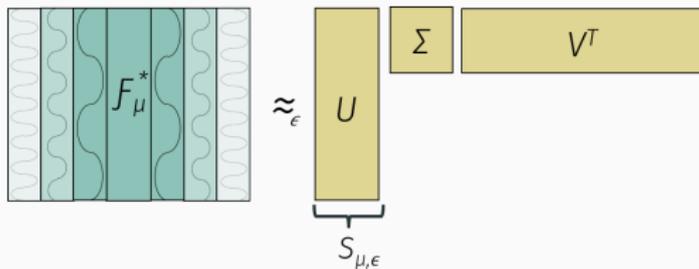
We need to take $S_{\mu,\epsilon}$ total samples to approximate the regression problem.

For finite dimension problems, $S_{\mu,\epsilon}$ is bounded by d .

$$\begin{aligned} S_{\mu,\epsilon} &= \text{tr} (\mathcal{K}_\mu + \epsilon \mathcal{I})^{-1} \mathcal{K}_\mu \\ &= \sum_{i=1}^{\infty} \frac{\lambda_i}{\lambda_i + \epsilon} \end{aligned}$$

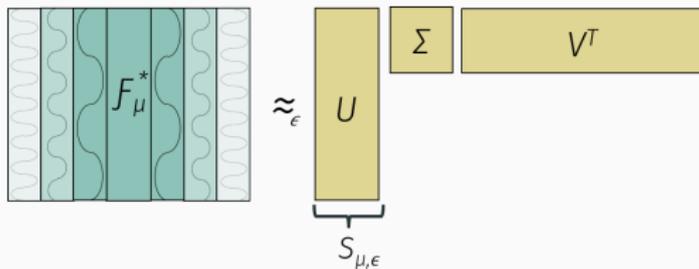
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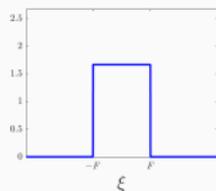


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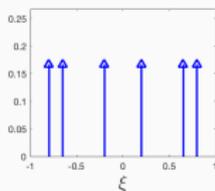
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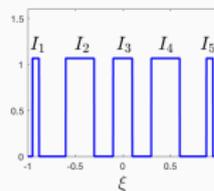
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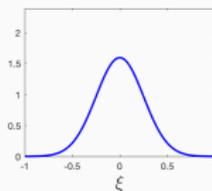
$$O(F T + \log \frac{1}{\epsilon})$$



$$k$$

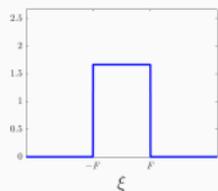


$$O(\sum F_i T + \log \frac{1}{\epsilon})$$

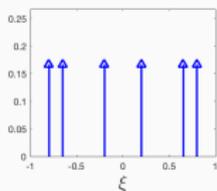


$$\tilde{O}(\sigma T + \log \frac{1}{\epsilon})$$

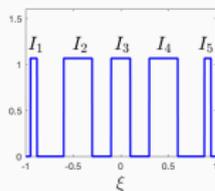
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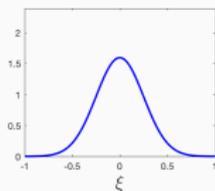
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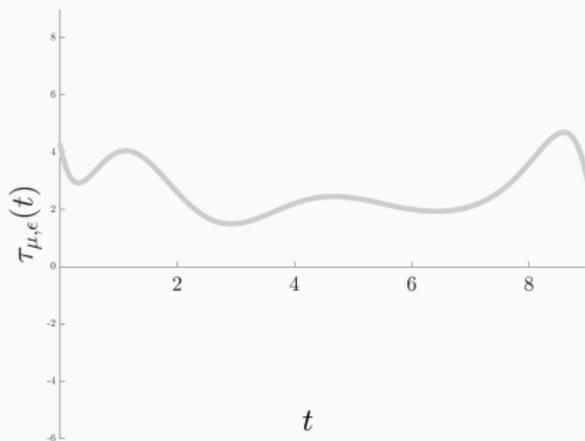


$$\tilde{O}(\sigma T + \log \frac{1}{\epsilon})$$

Bound of $S_{\mu, \epsilon}$ samples is tight.

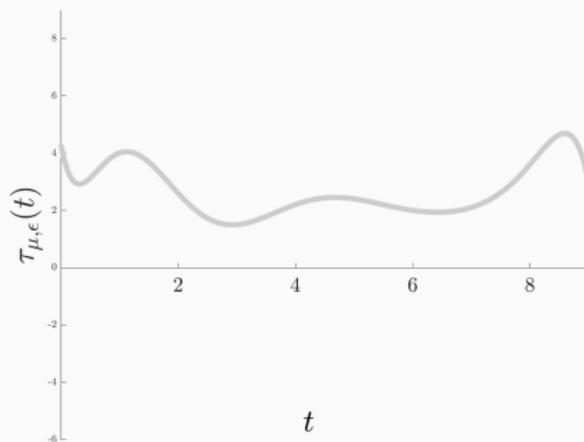
Leverage scores are hard to compute, even for discrete regression problems.

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LEVERAGE SCORE SAMPLING

Leverage scores are hard to compute, even for discrete regression problems. For our problem the challenge is even more daunting... we need scores for a continuum of values.



But... we have structure on our side.

What is the leverage score?

$$\tau_{\mu, \epsilon}(t) = \frac{1}{T} \max_g \frac{|\mathcal{F}_\mu^* g(t)|^2}{\|\mathcal{F}_\mu^* g\|_T^2 + \epsilon \|g\|_\mu^2}$$

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Squared value of a function at t over the average squared value. I.e. how much can the function “spike” at time t .

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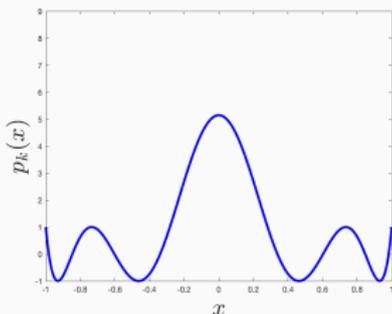
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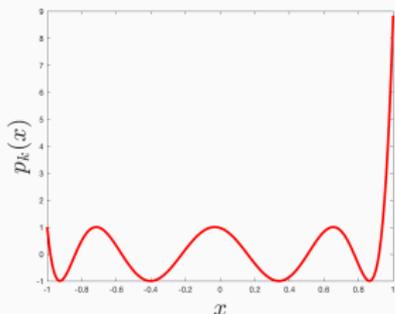
Worst case, but over a restricted class of functions – need to have small norm under μ .

POLYNOMIAL LEVERAGE

Leverage for degree k polynomials:



Bernstein Inequality.
 $\tau(t) \leq k / \sqrt{\min(t, T-t)}$

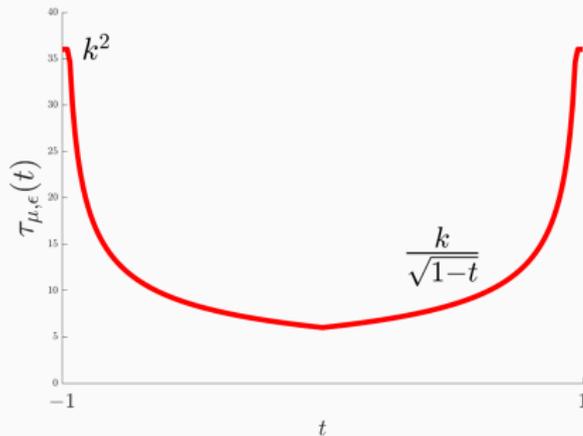


Markov Brother's Inequality.
 $\tau(t) \leq k^2$

In general, a polynomial can “spike” more near the edge of an interval.

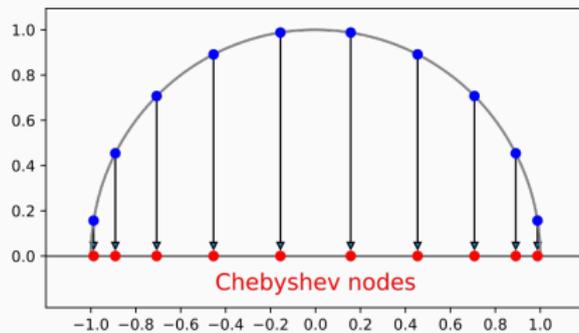
POLYNOMIAL LEVERAGE

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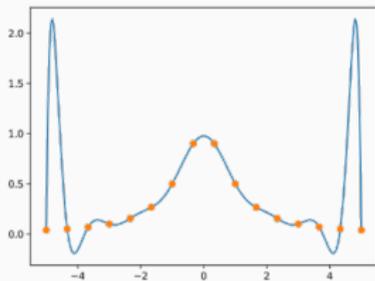
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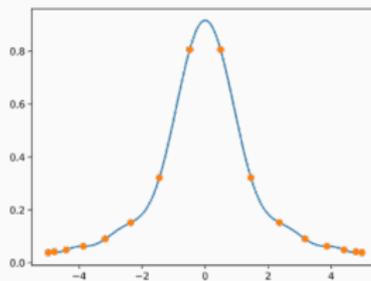


POLYNOMIAL LEVERAGE

Leverage for degree k polynomials:



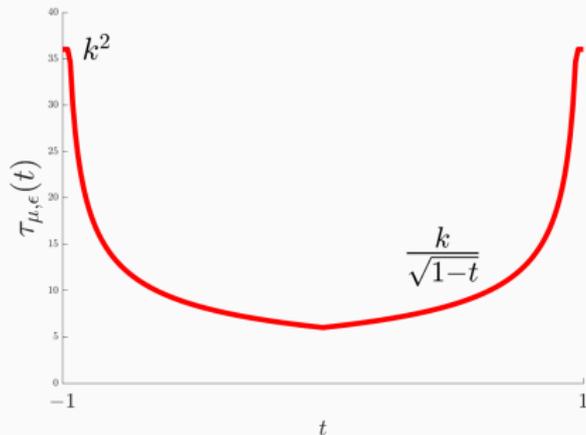
Uniform samples.



Chebyshev samples.

POLYNOMIAL LEVERAGE

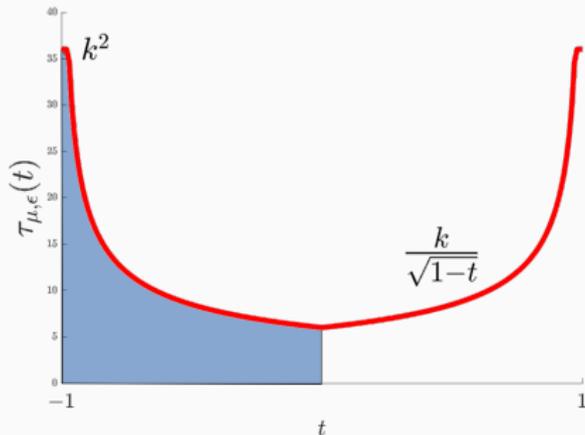
Leverage for degree k polynomials:



Total leverage:

POLYNOMIAL LEVERAGE

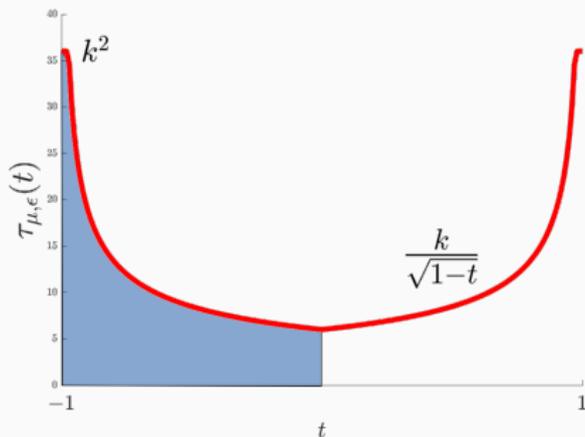
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POLYNOMIAL LEVERAGE

Leverage for degree k polynomials:



Total leverage: $O(k)$

Extends to bandlimited functions, which can be approximated by degree $k = O(FT + \log(1/\epsilon))$ degree polynomials.

[Chen, Kane, Price, Song, FOCS 2016], [Chen, Price 2018]

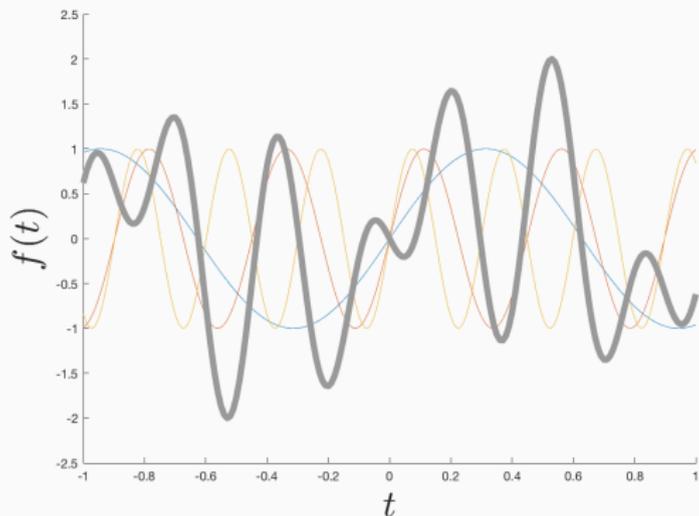
Nearly the same bounds holds for k -sparse Fourier functions.

[Chen, Kane, Price, Song, FOCS 2016], [Chen, Price 2018]

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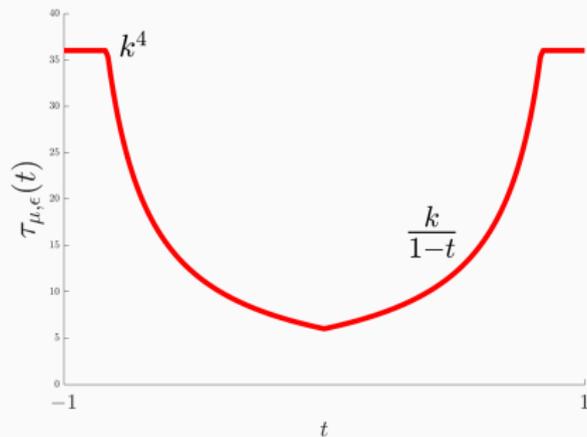
$$\frac{|f_k(t)|^2}{\|f_k\|_T^2} = \tilde{O}(\min[k^4, k/\min(t, T-t)])$$

Intuition: Sums of close frequencies look like modulated polynomials. Far frequencies are nearly orthogonal.



FOURIER LEVERAGE

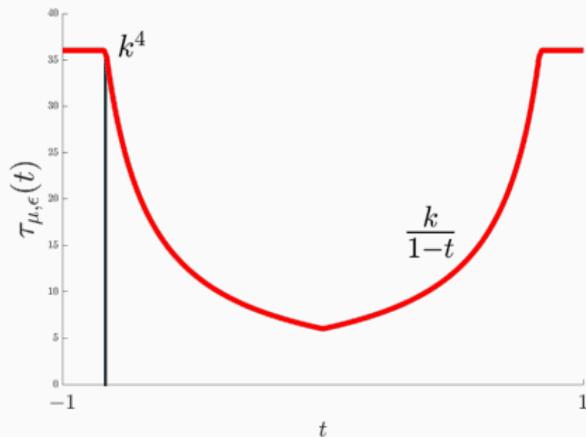
Leverage for k sparse Fourier functions:



Total leverage:

FOURIER LEVERAGE

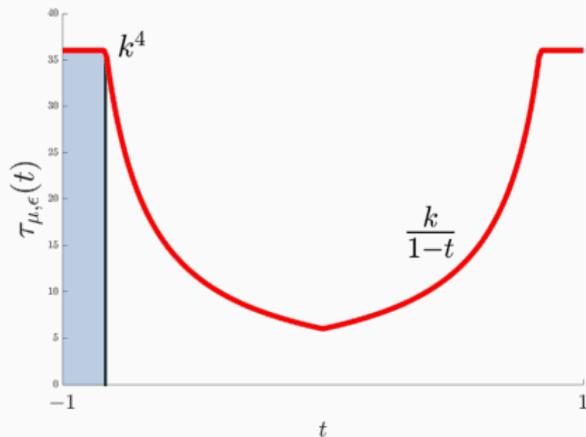
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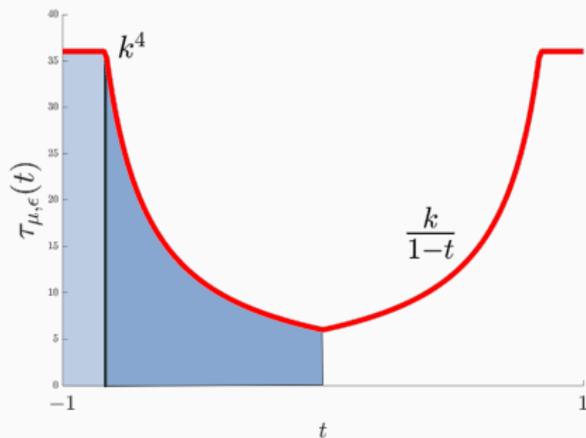
Leverage for k sparse Fourier functions:



Total leverage: k

FOURIER LEVERAGE

Leverage for k sparse Fourier functions:



Total leverage: $k + O(k \log k)$

How do we extend these bounds to more general constraint distributions μ ? Want $\tilde{O}(S_{\mu,\epsilon})$ samples.

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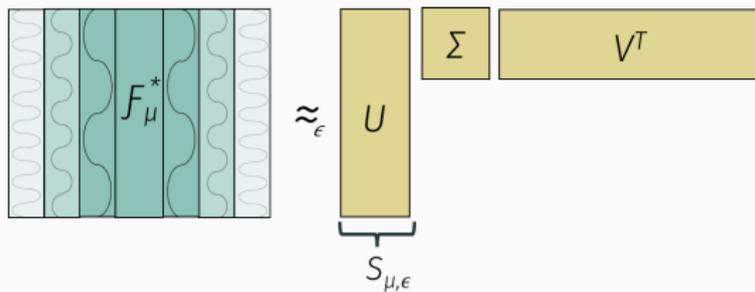
More tools from randomized matrix algorithms!

Lesson from past decade: Top q singular vectors of a matrix are approximately spanned by $O(q)$ columns from that matrix.

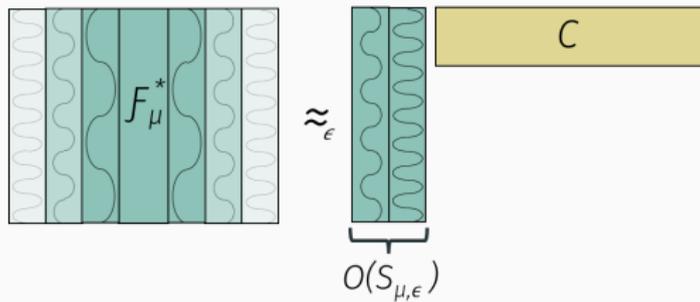
Lesson from past decade: Top q singular vectors of a matrix are approximately spanned by $O(q)$ columns from that matrix.

(rank-revealing QR, randomized SVD, columns subset selection, CUR decomposition, Nyström approximation, graph sparsification, random Fourier features, etc.)

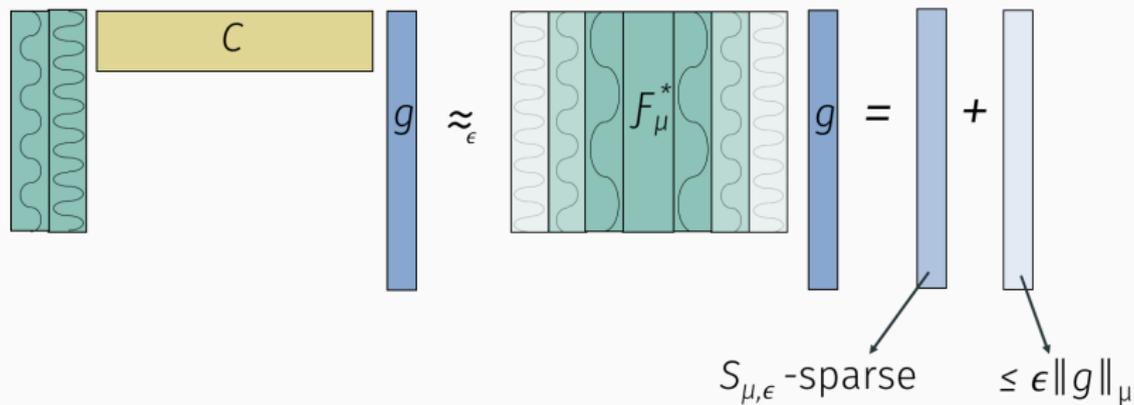
SPARSE APPROXIMATION OF WEIGHTED FOURIER TRANSFORM



SPARSE APPROXIMATION OF WEIGHTED FOURIER TRANSFORM

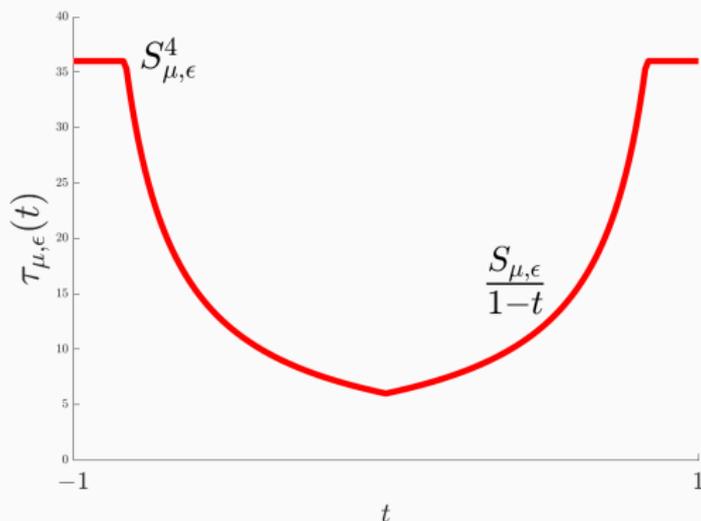


SPARSE APPROXIMATION OF WEIGHTED FOURIER TRANSFORM



$$\tau_{\mu,\epsilon}(t) = \max_g \frac{\frac{1}{T} |\mathcal{F}_\mu^* g(t)|^2}{\|\mathcal{F}_\mu^* g\|_T^2 + \epsilon \|g\|_\mu^2} \leq \tilde{O}(\min[S_{\mu,\epsilon}^4, S_{\mu,\epsilon} / \min(t, T-t)])$$

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Total number of samples: $\tilde{O}(S_{\mu,\epsilon})$.

Simple Fourier function fitting:

- Sample t_1, \dots, t_q according to \mathcal{D} .

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Simple Fourier function fitting:

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Matches known results for sparse and bandlimited function up to log factors, while achieving nearly optimal sample complexity for any other Fourier constraints.

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Other connections between random graph/matrix sampling
and classic function interpolation?

THANK YOU!