

Reconstruction and Cancellation of Sampled Multiband Signals Using Discrete Prolate Spheroidal Sequences

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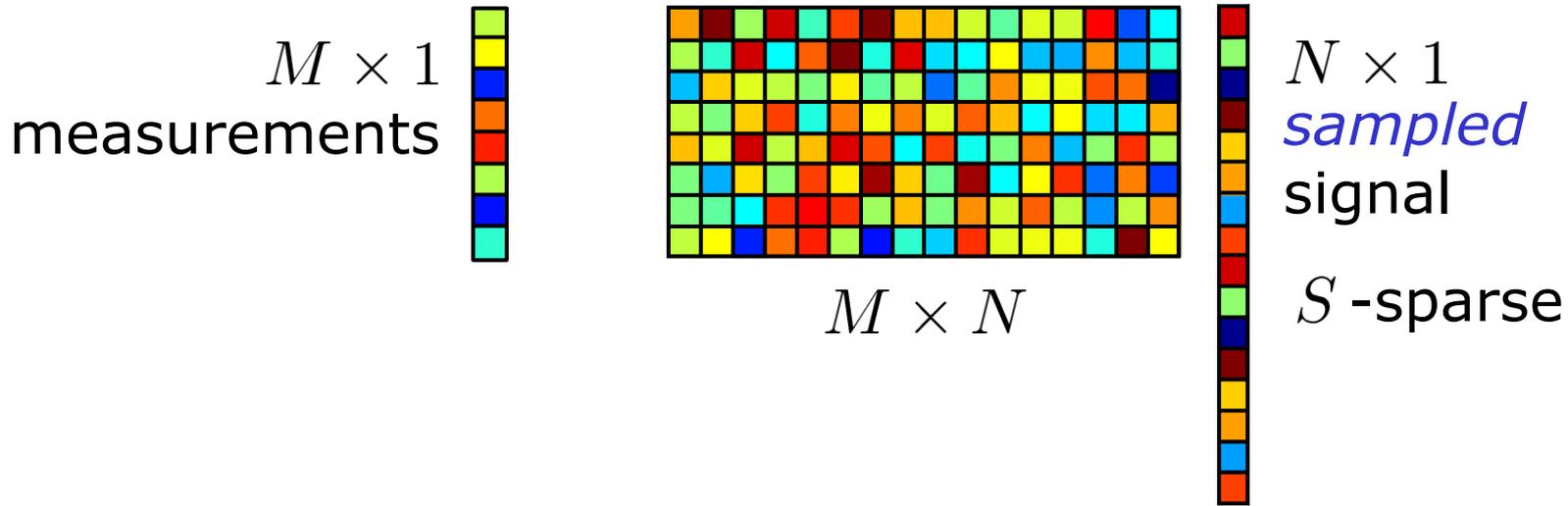
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Compressive Sensing (CS)

$$y = \Phi x$$



Can we really acquire analog signals with "CS"?

Potential Obstacles



Obstacle 1: CS is discrete, finite-dimensional

Obstacle 2: Analog sparse representations

Obstacle 1

Obstacle 1: CS is discrete, finite-dimensional

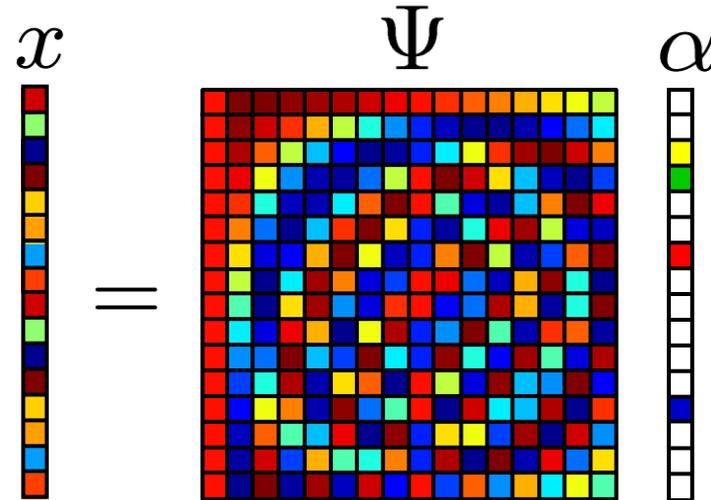
For any bandlimited signal $x(t)$,

$$\begin{aligned} y[m] &= \langle \phi_m(t), x(t) \rangle \\ &= \sum_{n=-\infty}^{\infty} x[n] \langle \phi_m(t), \text{sinc}(t/T_s - n) \rangle \end{aligned}$$

For many practical architectures, $y[m]$ will depend on only a finite window of $x[n]$.

Obstacle 2

Obstacle 2: Analog sparse representations

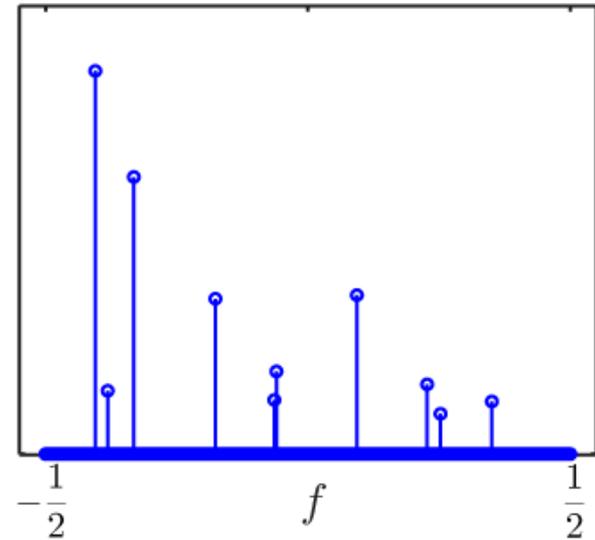


The structure of Ψ will derive from a continuous-time signal model.

Candidate Analog Signal Models

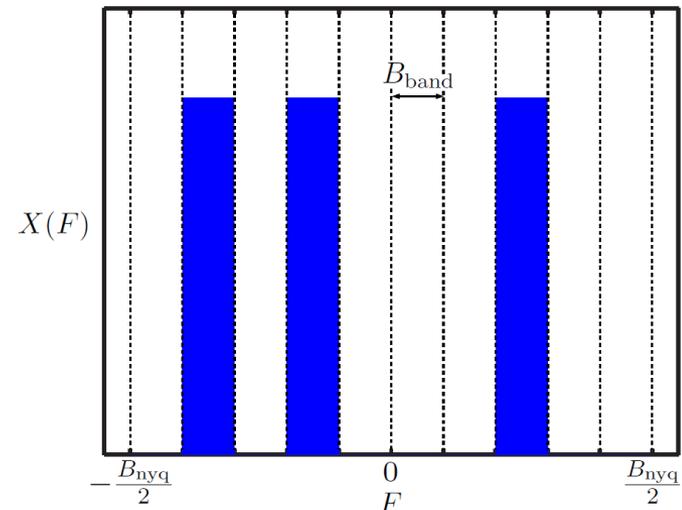
Multitone model:

- periodic signal
- DFT with S *tones*
- unknown *amplitude*



Multiband model:

- aperiodic signal
- DTFT with K *bands* of bandwidth B_{band}
- unknown *spectra*

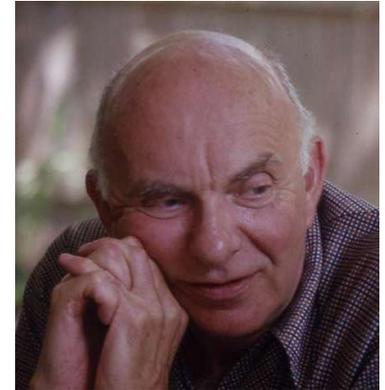


Discrete Prolate Spheroidal Sequences (DPSS's)

DPSS's (Slepian sequences)

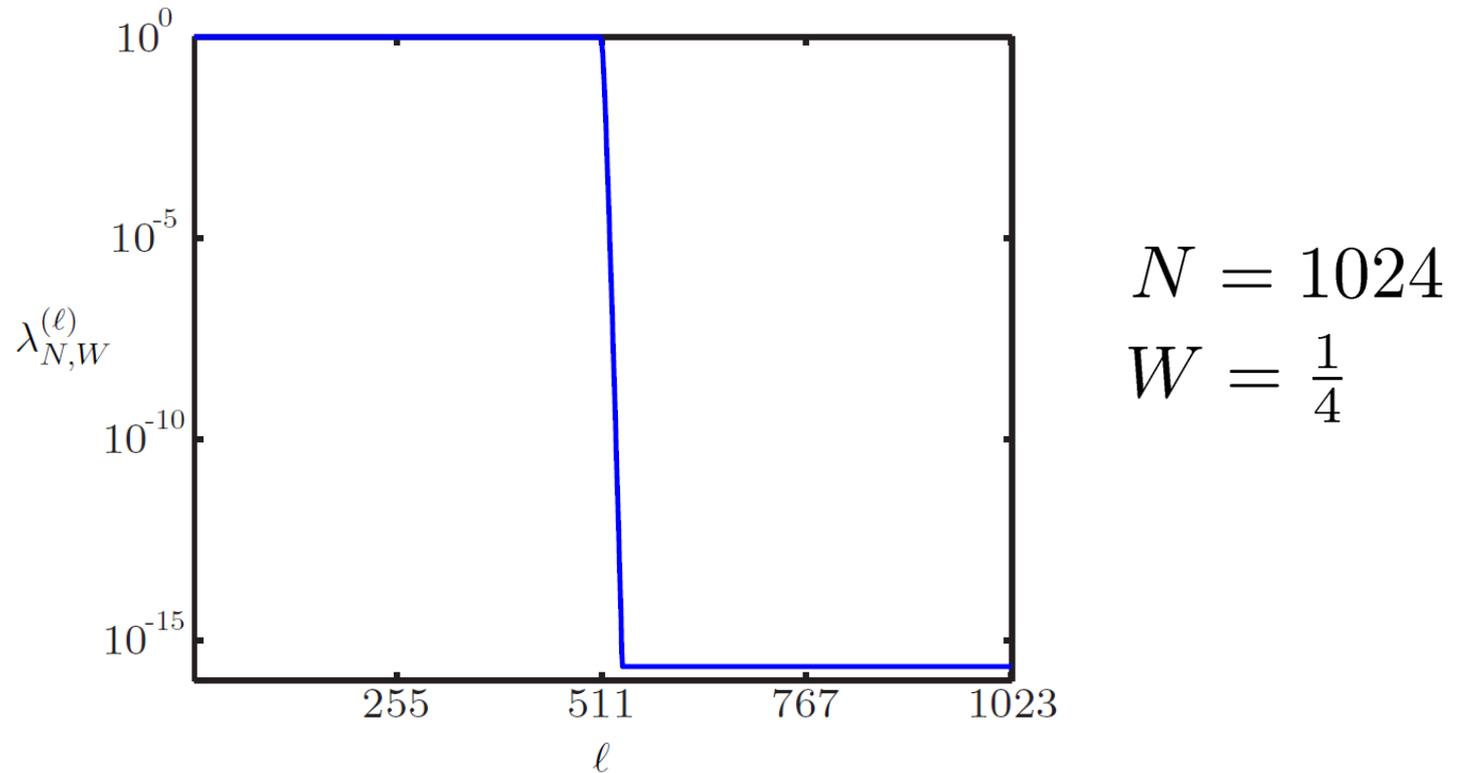
Given N and $W \leq \frac{1}{2}$, the DPSS's are a collection of N real-valued discrete-time sequences $s_{N,W}^{(0)}, s_{N,W}^{(1)}, \dots, s_{N,W}^{(N-1)}$ such that for all ℓ

$$\mathcal{B}_W(\mathcal{T}_N(s_{N,W}^{(\ell)})) = \lambda_{N,W}^{(\ell)} s_{N,W}^{(\ell)}.$$



The DPSS's are perfectly bandlimited, but when $\lambda_{N,W}^{(\ell)} \approx 1$ they are highly concentrated in time.

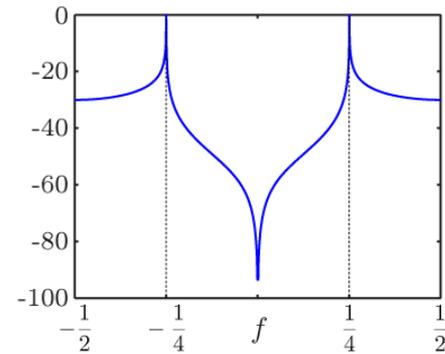
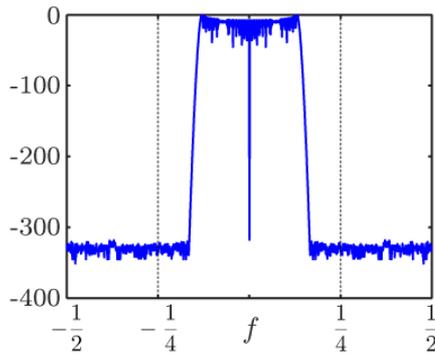
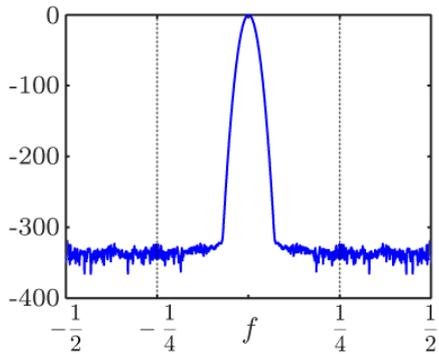
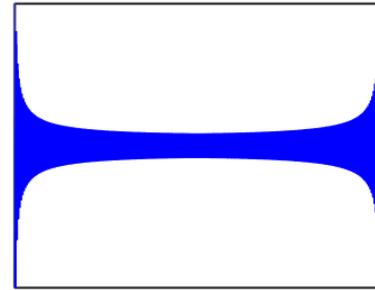
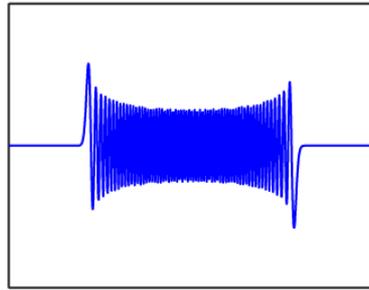
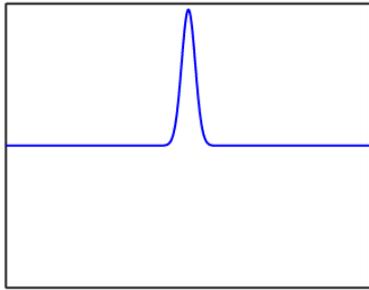
DPSS Eigenvalue Concentration



The first $\approx 2NW$ eigenvalues ≈ 1 .
The remaining eigenvalues ≈ 0 .

DPSS Examples

$$N = 1024 \quad W = \frac{1}{4}$$



$$l = 0$$

$$l = 127$$

$$l = 511$$

Why DPSS's?

Suppose that we wish to minimize

$$\frac{1}{2W} \cdot \int_{-W}^W \|e_f - P_Q e_f\|_2^2 df$$

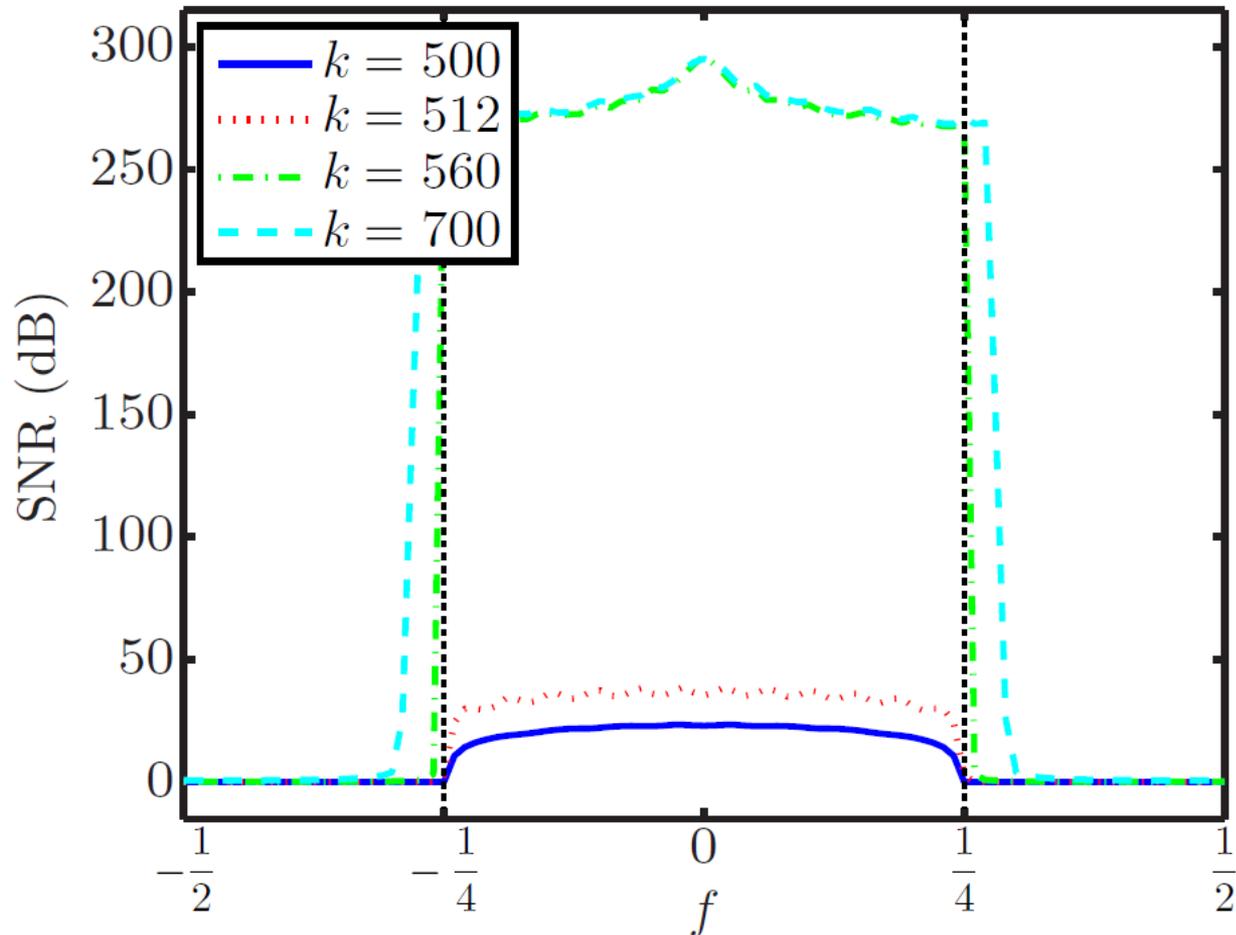
over Q where $e_f := \left[e^{j2\pi f0}, e^{j2\pi f}, \dots, e^{j2\pi f(N-1)} \right]^T$.

Optimal subspace of dimension k is the one spanned by the first k DPSS vectors.

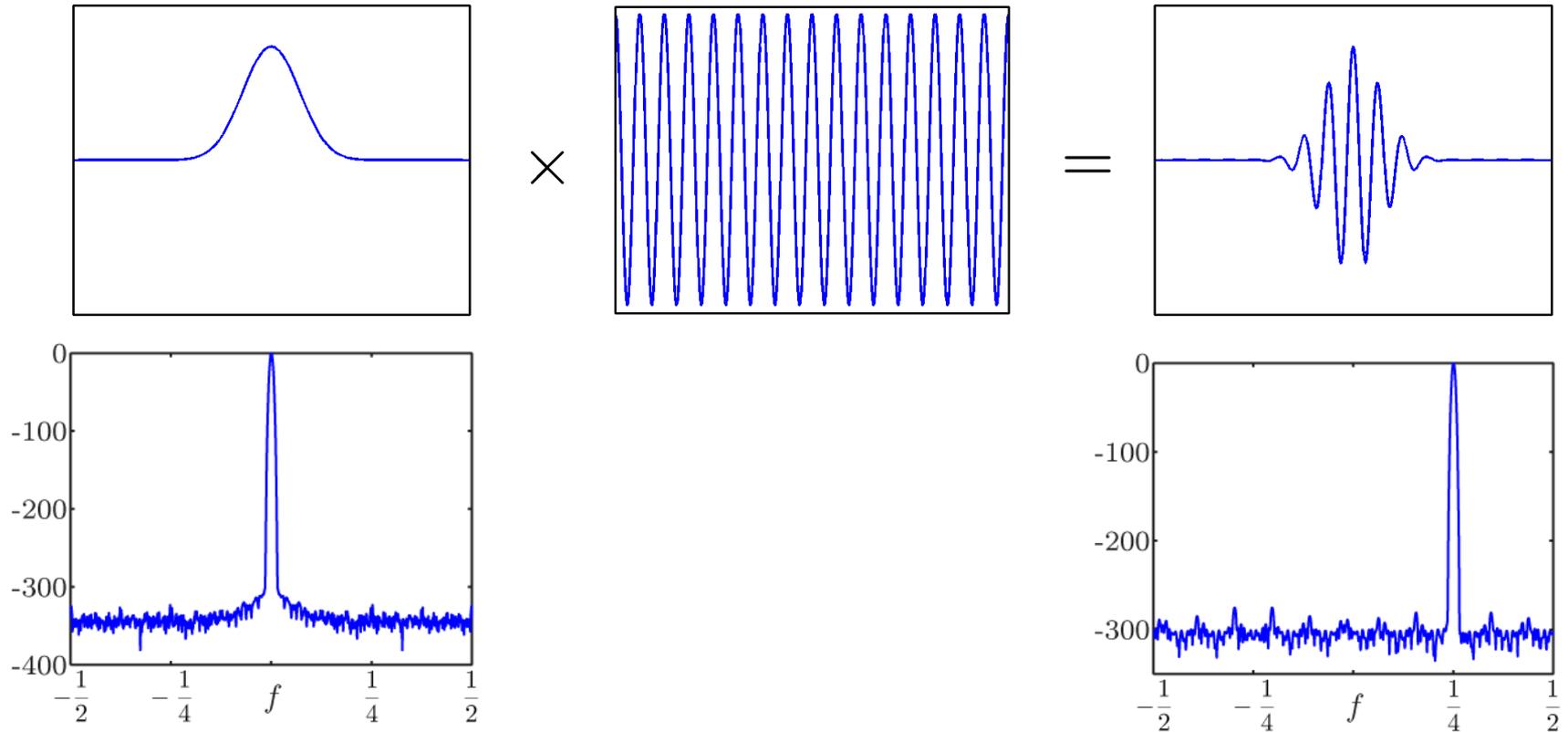
$$\frac{1}{2W} \cdot \int_{-W}^W \|e_f - P_Q e_f\|_2^2 df = \frac{1}{2W} \sum_{\ell=k}^{N-1} \lambda_{N,W}^{(\ell)}$$

Approximation Performance

$$\text{SNR} = 20 \log_{10} \left(\frac{\|e_f\|}{\|e_f - P_Q e_f\|} \right) \text{ dB}$$



DPSS's for Passband Signals



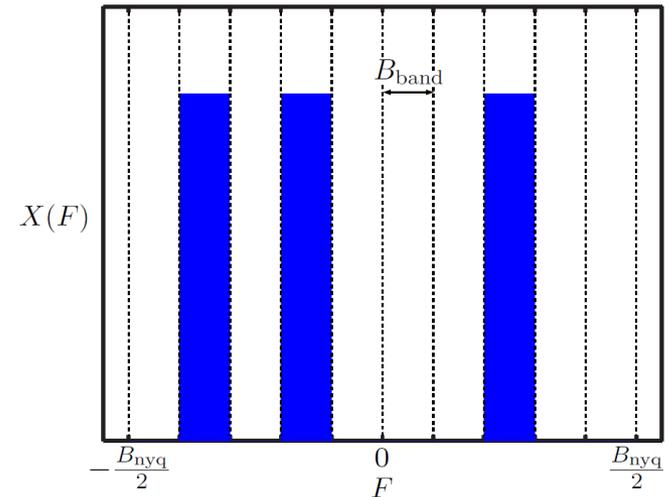
DPSS Dictionaries for CS

Construct dictionary Ψ as

$$\Psi = [\Psi_1, \Psi_2, \dots, \Psi_J]$$

where Ψ_i is the matrix of the first k DPSS's modulated to $f_i = -\frac{1}{2} + (i + \frac{1}{2}) (B_{\text{band}}/B_{\text{nyq}})$.

Ψ sparsely and accurately represents *most* sampled multiband signals.



DPSS Dictionaries and the RIP

Let $W = \frac{1}{2}(B_{\text{band}}/B_{\text{nyq}})$. Suppose that Φ is sub-Gaussian and that the Ψ_i are constructed with $k = (1 - \epsilon)2NW$. If

$$M \geq CS \log(N/S)$$

then with high probability $\Phi\Psi$ will satisfy

$$(1 - \delta)\|\alpha\|_2^2 \leq \|\Phi\Psi\alpha\|_2^2 \leq (1 + \delta)\|\alpha\|_2^2$$

for all S -sparse α .

K occupied bands $\longrightarrow S \approx KNB_{\text{band}}/B_{\text{nyq}}$

$$\frac{M}{N} \geq C' \frac{KB_{\text{band}}}{B_{\text{nyq}}} \log \left(\frac{B_{\text{nyq}}}{KB_{\text{band}}} \right)$$

Block-Sparse Recovery

Nonzero coefficients of α should be clustered in blocks according to the occupied frequency bands

$$x = [\Psi_1, \Psi_2, \dots, \Psi_J] \begin{bmatrix} \alpha_1 \\ \alpha_2 \\ \vdots \\ \alpha_J \end{bmatrix}$$

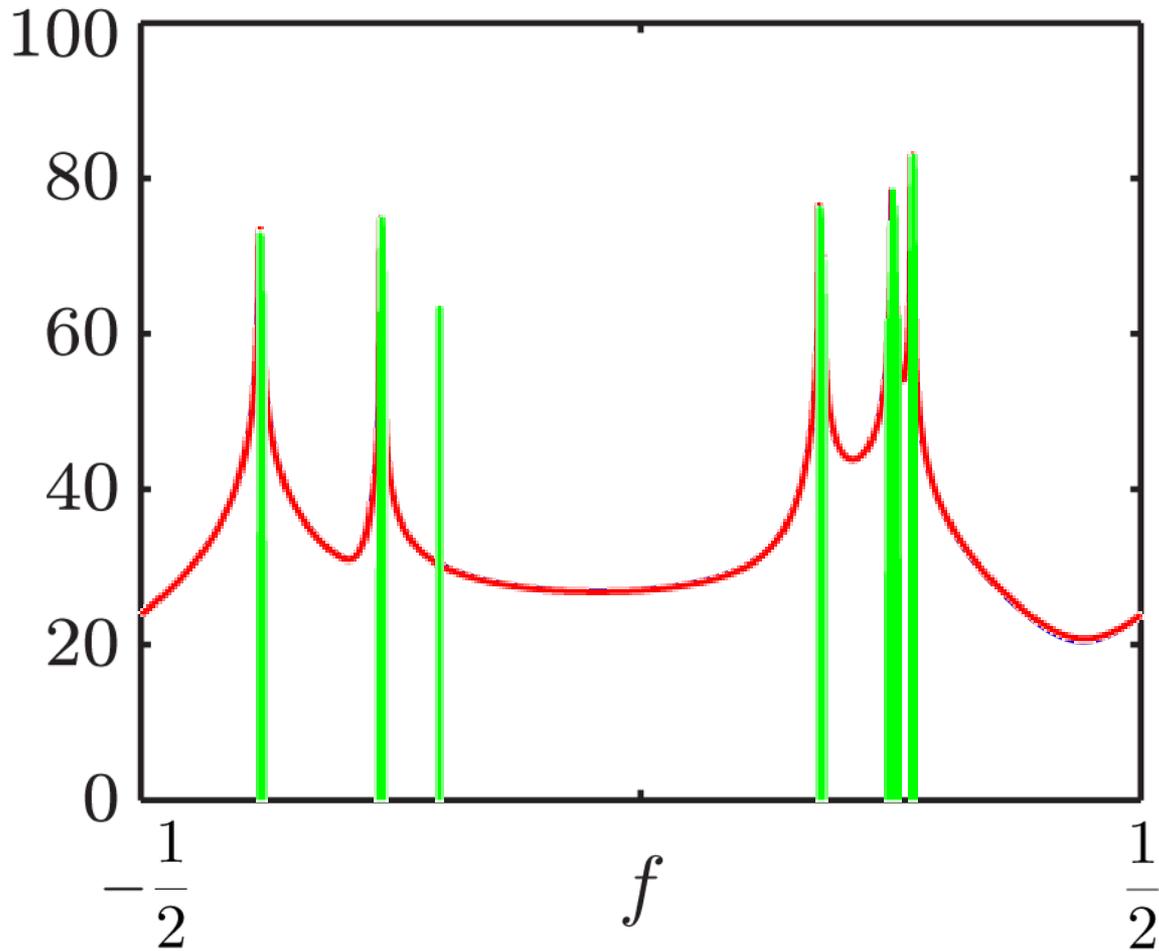
This can be leveraged to reduce the required number of measurements and improve performance through “model-based CS”

- Baraniuk et al. [2008, 2009, 2010]
- Blumensath and Davies [2009, 2011]

Recovery: DPSS vs DFT

$$N = 1024 \quad M = 128 \quad K = 5 \quad \frac{B_{\text{band}}}{B_{\text{nyq}}} = \frac{1}{512}$$

$$S \approx 45$$

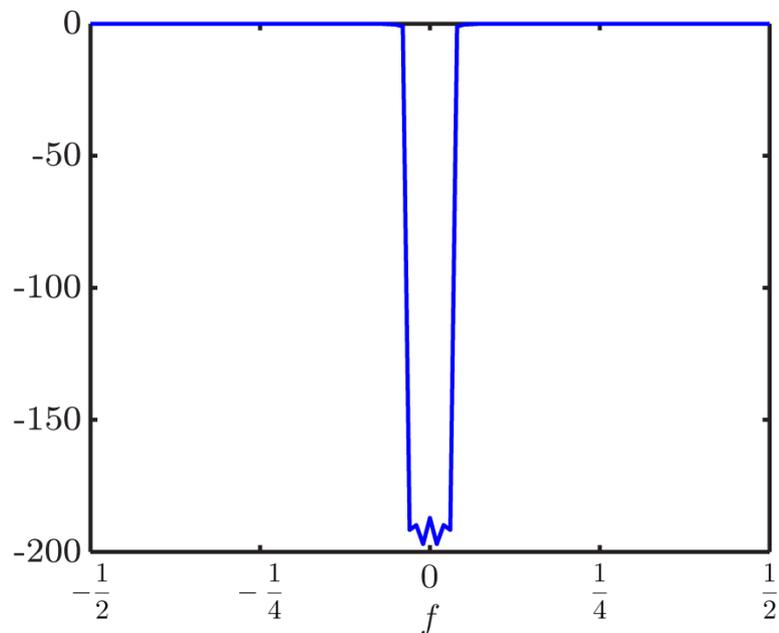


DPSS : SNR = 54dB

DFT : SNR = 12dB

Interference Cancellation

DPSS's can be used to cancel bandlimited interferers *without reconstruction*.



$$P = I - \Phi \Psi_i (\Phi \Psi_i)^\dagger$$

Extremely useful in *compressive signal processing* applications.

Summary

- DPSS's can be used to efficiently represent *most* sampled multiband signals
 - knowledge of occupied bands not necessary a priori
 - far superior to DFT
- Two types of error: *approximation* + *reconstruction*
 - approximation: small for most signals
 - reconstruction: zero for DPSS-sparse vectors
 - delicate balance in practice, but there is a sweet spot
- Applications
 - signal reconstruction
 - interference cancellation
 - compressive signal processing