

ESE 531: Digital Signal Processing

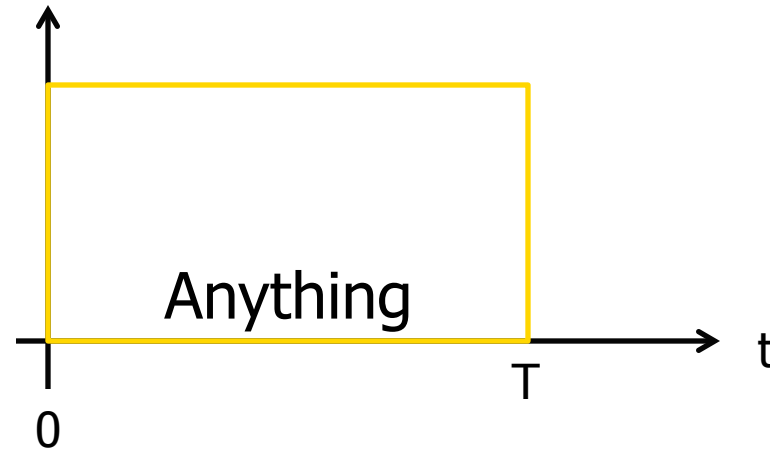
Lec 24: April 23, 2020
Compressive Sensing



Today

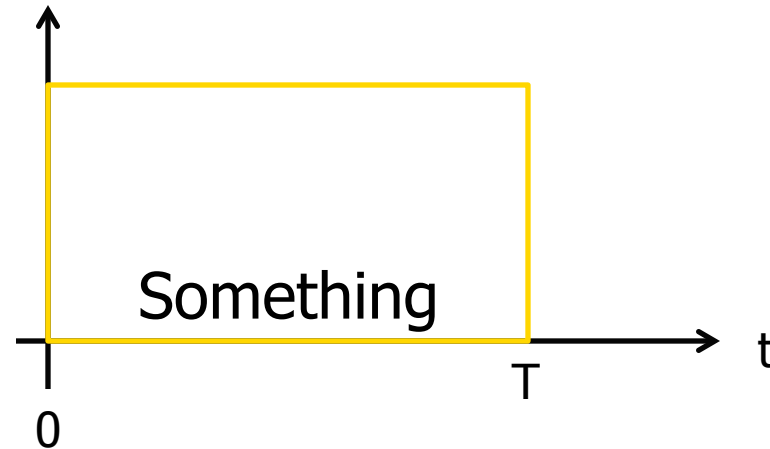
- ❑ Compressive Sampling/Sensing

Compressive Sampling



- ❑ What is the rate you need to sample at?
 - At least Nyquist

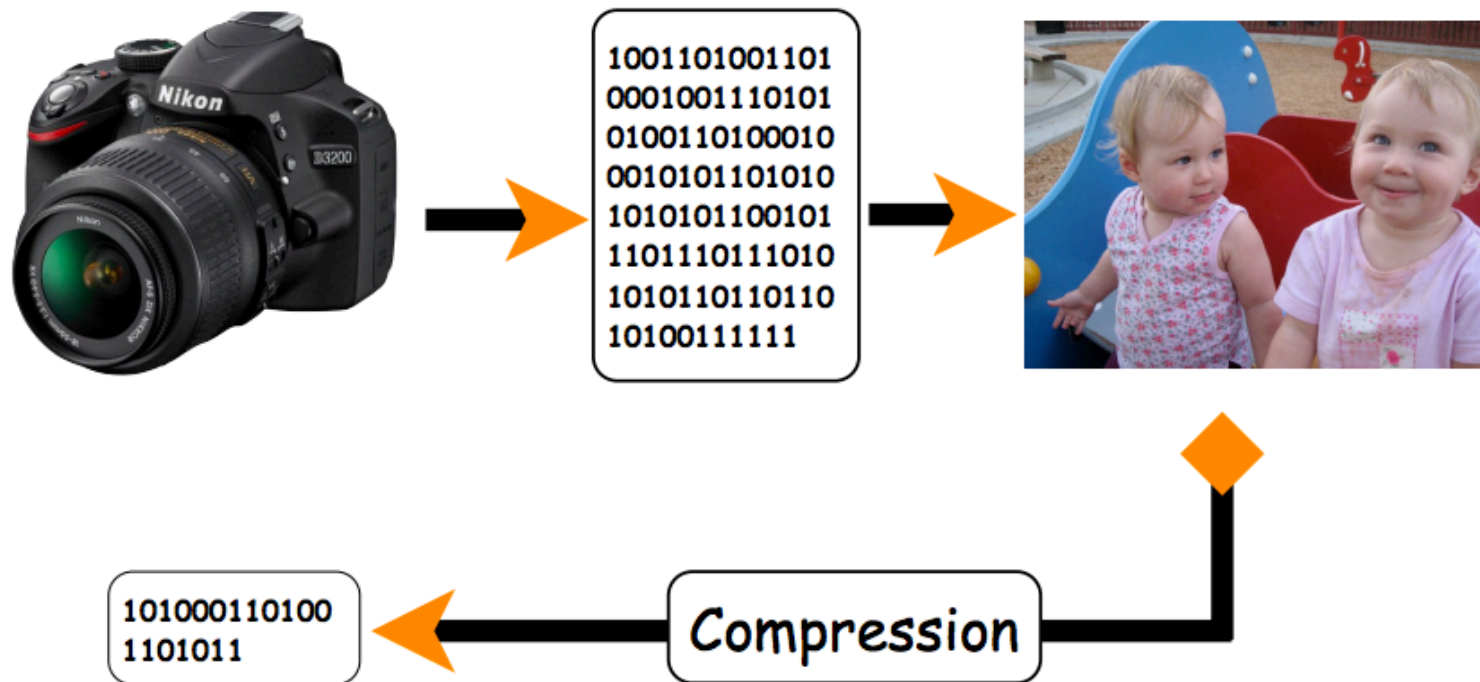
Compressive Sampling



- ❑ What is the rate you need to sample at?
 - Maybe less than Nyquist...

First: Compression

- ❑ Standard approach
 - First collect, then compress
 - Throw away unnecessary data





First: Compression

□ Examples

■ Audio – 10x

- Raw audio: 44.1kHz, 16bit, stereo = 1378 Kbit/sec
- MP3: 44.1kHz, 16 bit, stereo = 128 Kbit/sec

■ Images – 22x

- Raw image (RGB): 24bit/pixel
- JPEG: 1280x960, normal = 1.09bit/pixel

■ Videos – 75x

- Raw Video: $(480 \times 360) \text{p/frame} \times 24 \text{b/p} \times 24 \text{frames/s} + 44.1 \text{kHz} \times 16 \text{b} \times 2 = 98,578 \text{ Kbit/s}$
- MPEG4: 1300 Kbit/s

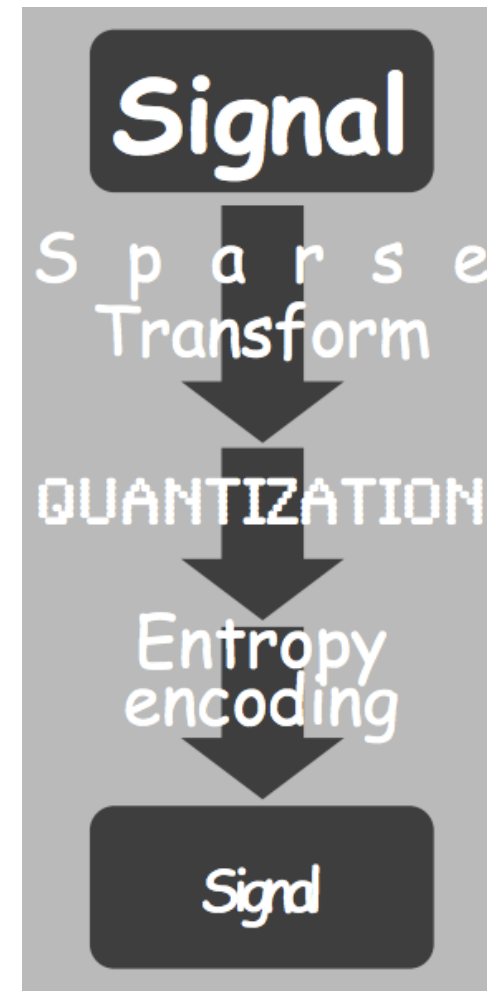


First: Compression

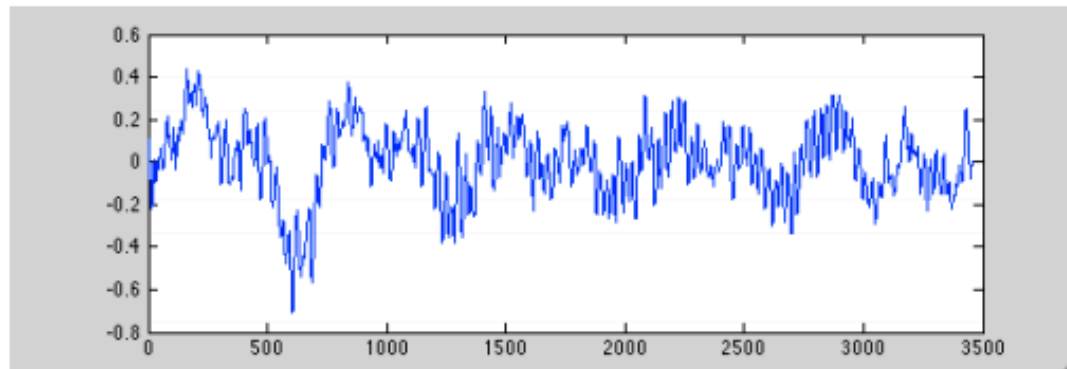
- ❑ Almost all compression algorithm use transform coding
 - mp3: DCT
 - JPEG: DCT
 - JPEG2000: Wavelet
 - MPEG: DCT & time-difference

First: Compression

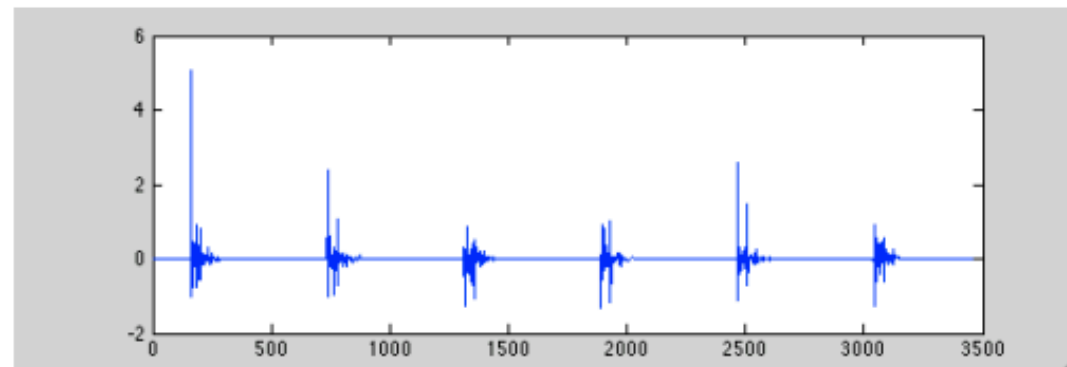
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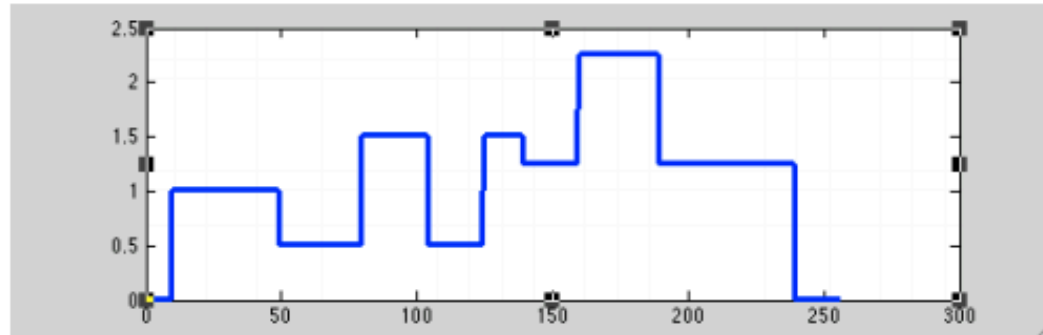
Sparse Transform



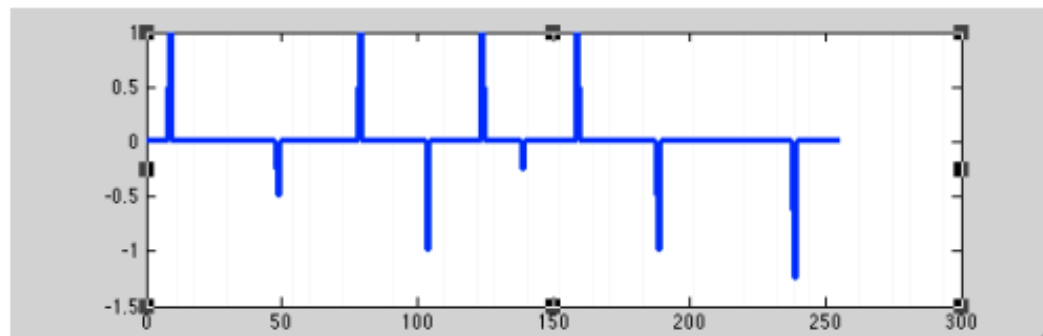
DCT



Sparse Transform

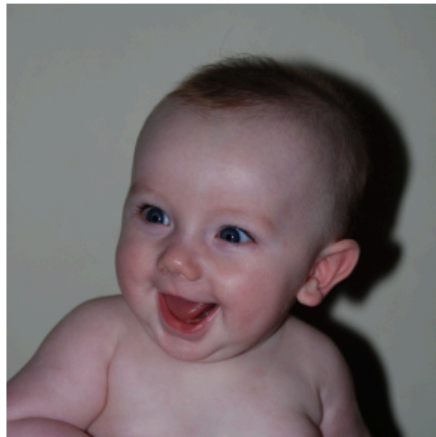


Difference



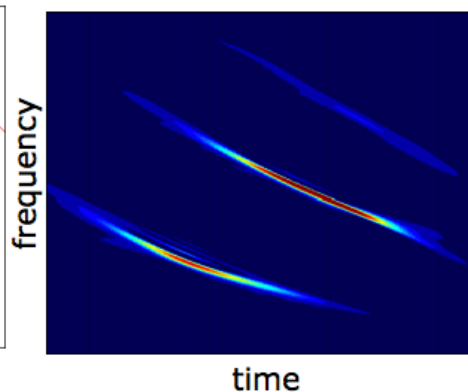
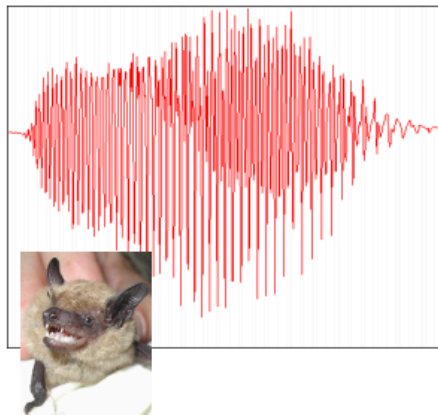
Sparsity

N
pixels



$K \ll N$
large
wavelet
coefficients
(blue = 0)

N
wideband
signal
samples



$K \ll N$
large
Gabor (TF)
coefficients



Signal Processing Trends

- ❑ Traditional DSP → sample first, ask questions later



Signal Processing Trends

- ❑ Traditional DSP → sample first, ask questions later
- ❑ Explosion in sensor technology/ubiquity has caused two trends:
 - Physical capabilities of hardware are being stressed, increasing speed/resolution becoming expensive
 - gigahertz+ analog-to-digital conversion
 - accelerated MRI
 - industrial imaging
 - Deluge of data
 - camera arrays and networks, multi-view target databases, streaming video...

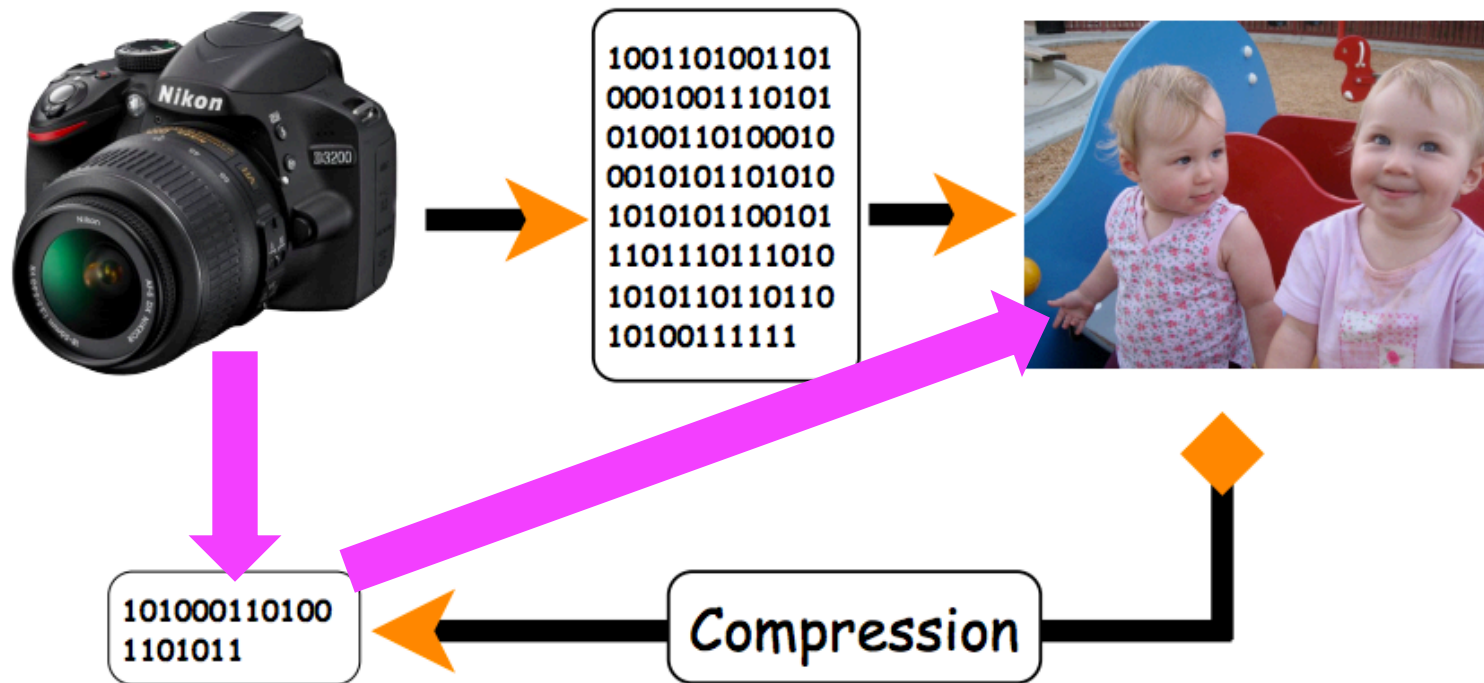


Signal Processing Trends

- ❑ Traditional DSP → sample first, ask questions later
- ❑ Explosion in sensor technology/ubiquity has caused two trends:
 - Physical capabilities of hardware are being stressed, increasing speed/resolution becoming expensive
 - gigahertz+ analog-to-digital conversion
 - accelerated MRI
 - industrial imaging
 - Deluge of data
 - camera arrays and networks, multi-view target databases, streaming video...
- ❑ Compressive Sensing → sample smarter, not faster

Compressive Sensing/Sampling

- ❑ Standard approach
 - First collect, then compress
 - Throw away unnecessary data

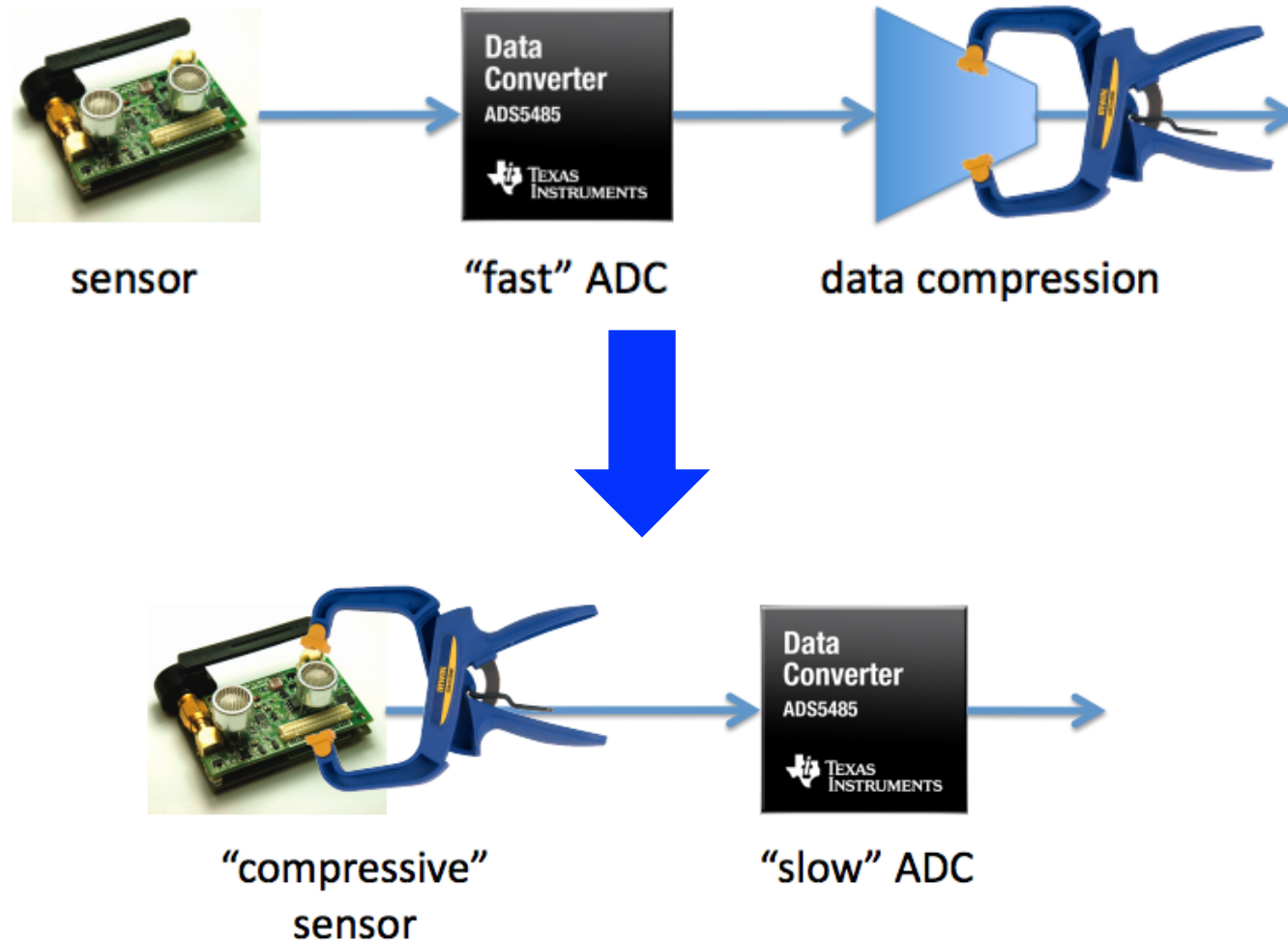




Compressive Sensing

- ❑ Shannon/Nyquist theorem is pessimistic
 - $2 \times$ bandwidth is the worst-case sampling rate — holds uniformly for any bandlimited data
 - sparsity/compressibility is irrelevant
 - Shannon sampling based on a linear model, compression based on a nonlinear model
- ❑ Compressive sensing
 - new sampling theory that leverages compressibility
 - key roles played by new uncertainty principles and randomness

Sensing to Data





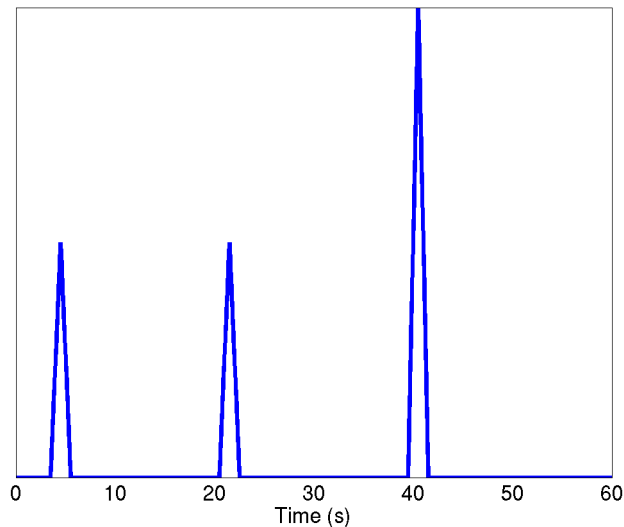
Compressive Sampling

- ❑ Sample at lower than the Nyquist rate and still accurately recover the signal, and in most cases *exactly* recover

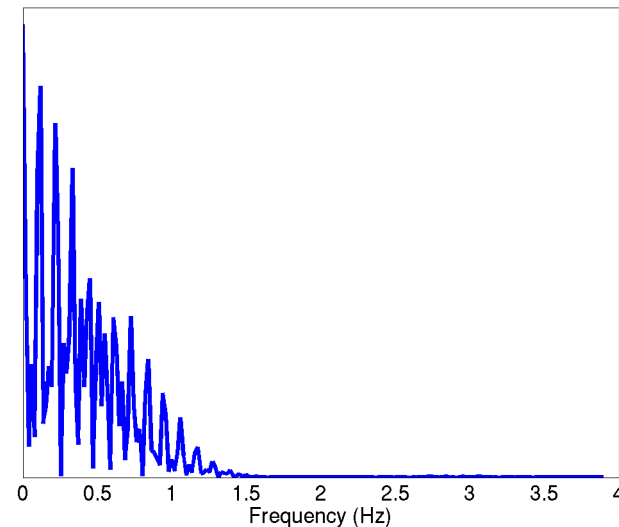
Compressive Sampling

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Sparse signal in time



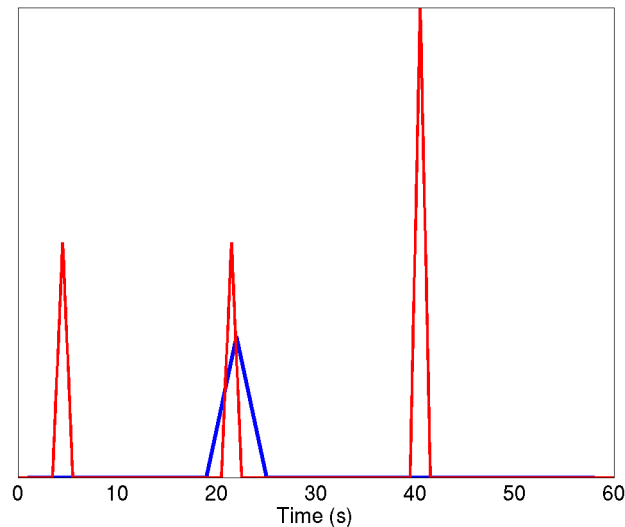
Frequency spectrum



Compressive Sampling

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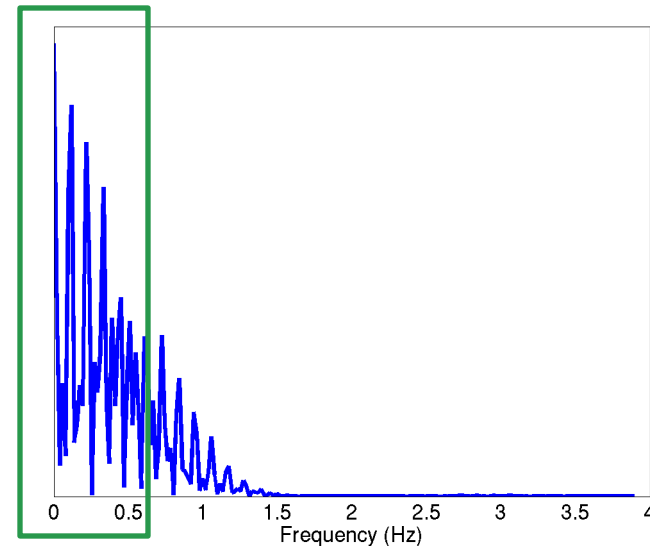
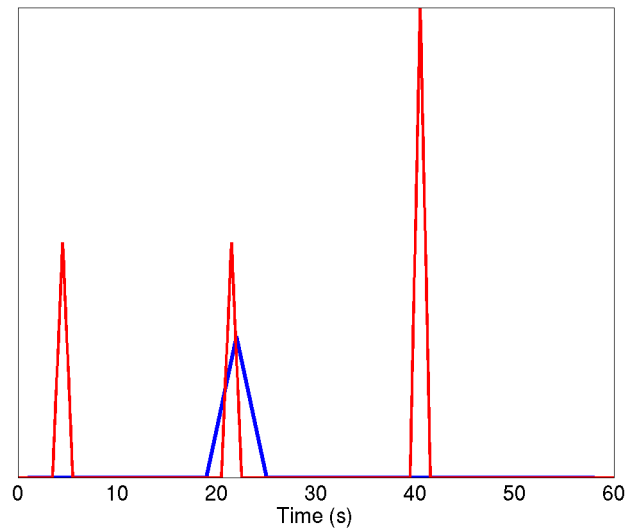
Undersampled in time



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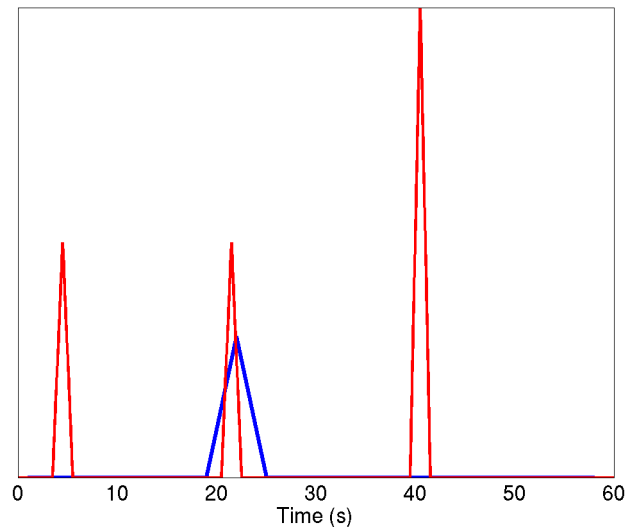
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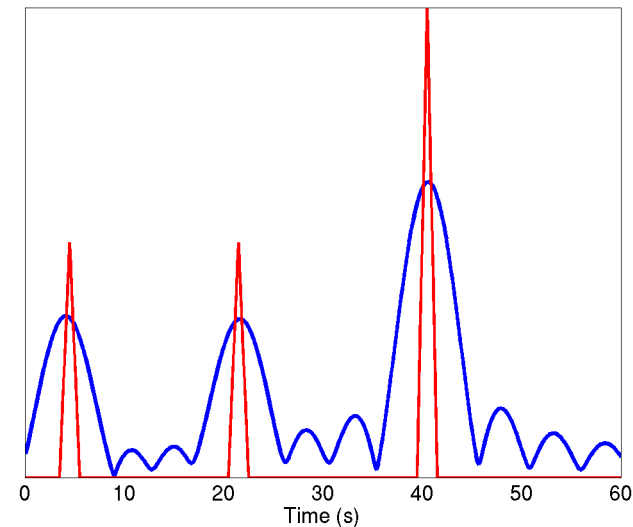
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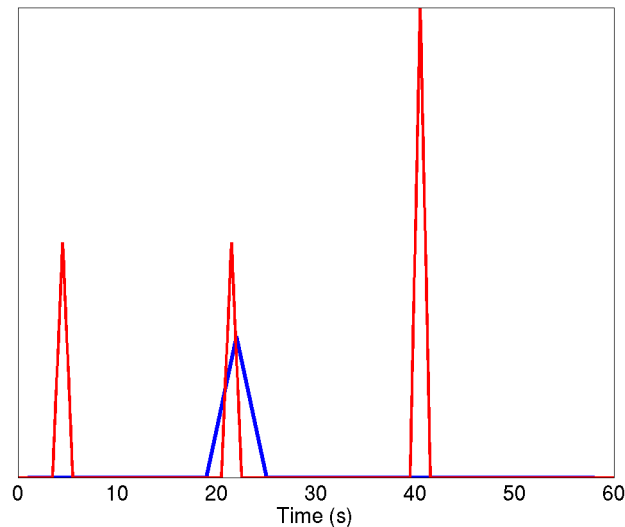
Undersampled in frequency
(reconstructed in time with IFFT)



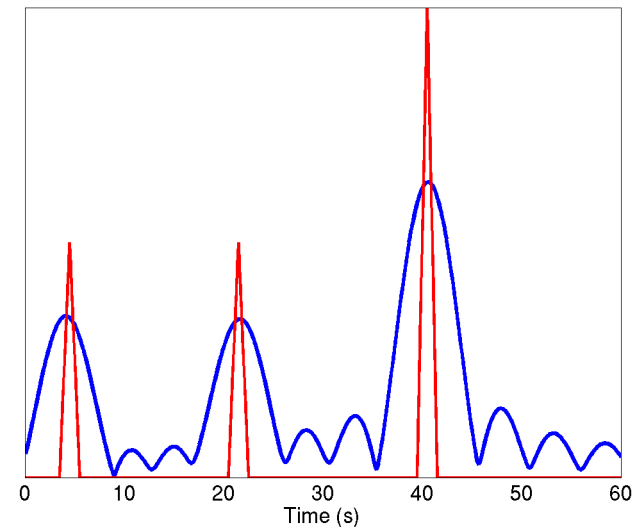
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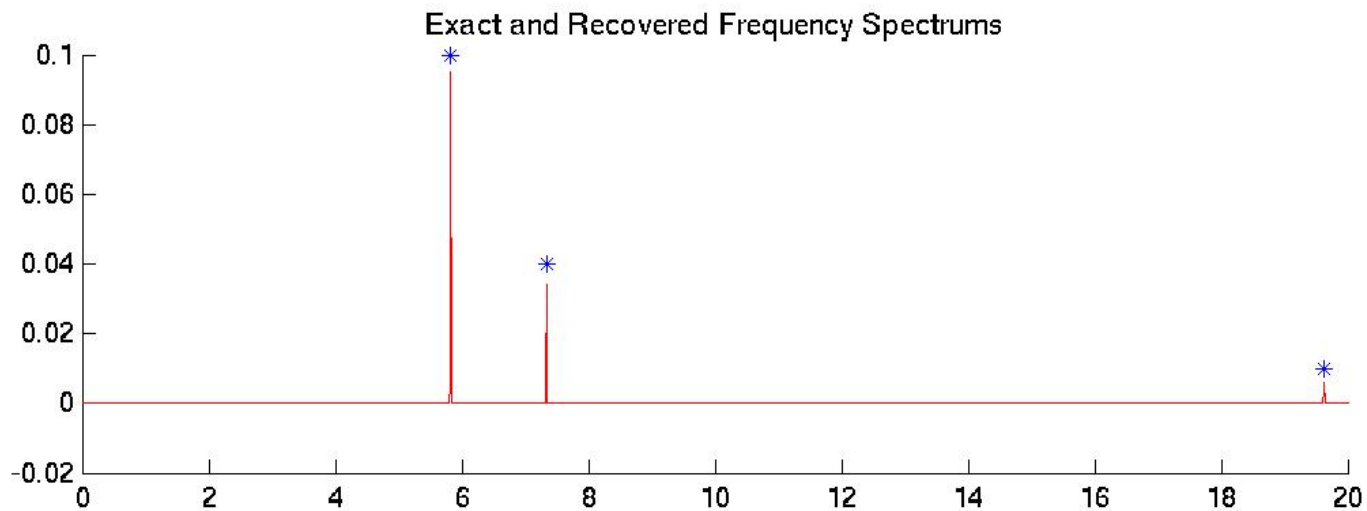
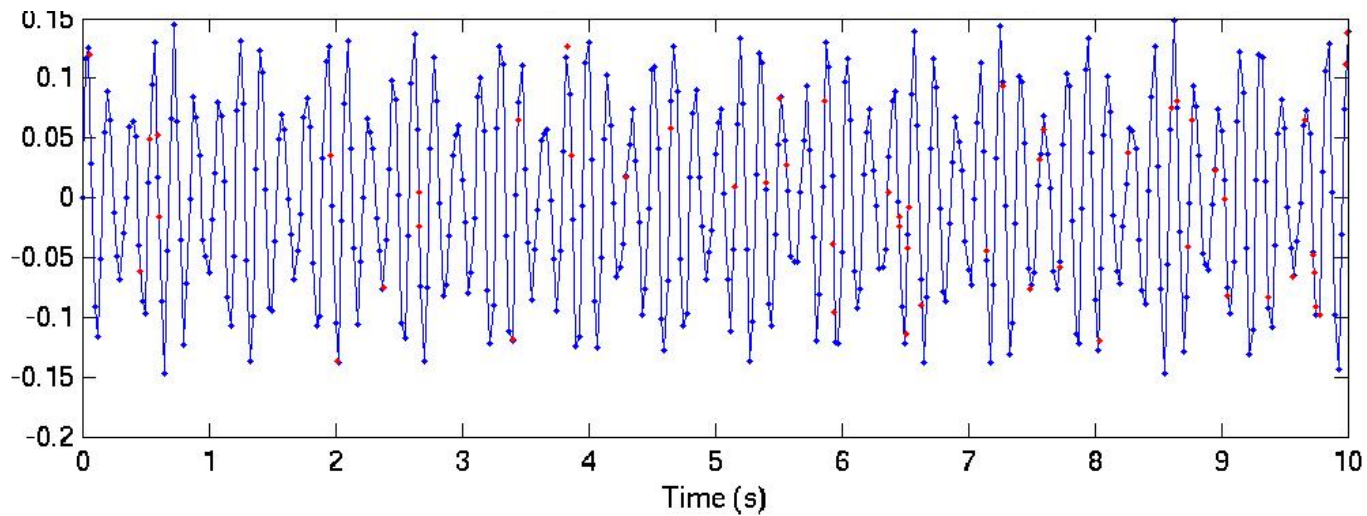


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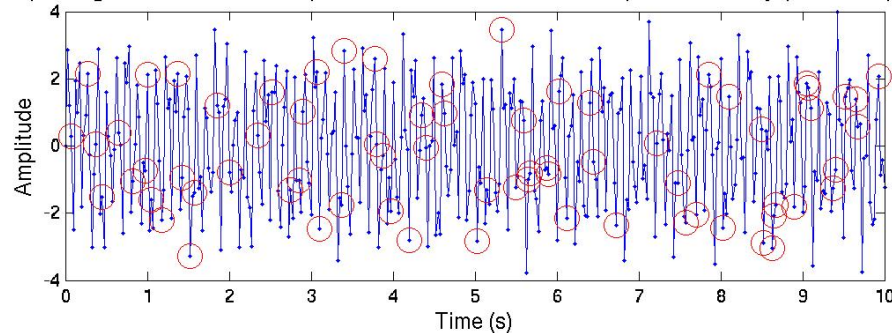
Requires sparsity and incoherent sampling

Compressive Sampling: Simple Example

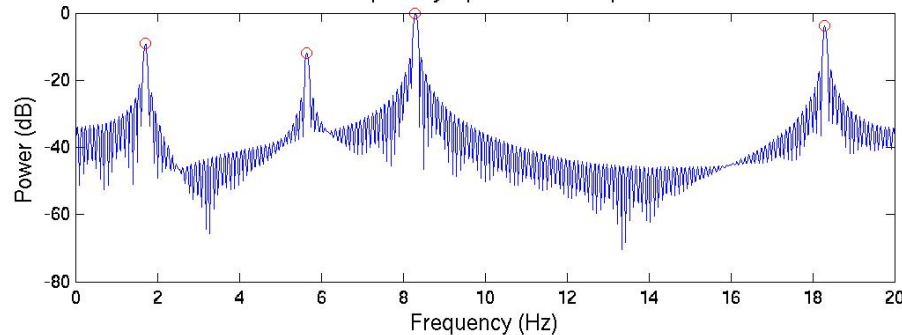


Compressive Sampling

Input signal with undersampled measurements circled ($\sim 17.5\%$ of Nyquist samples)



Frequency spectrum of input

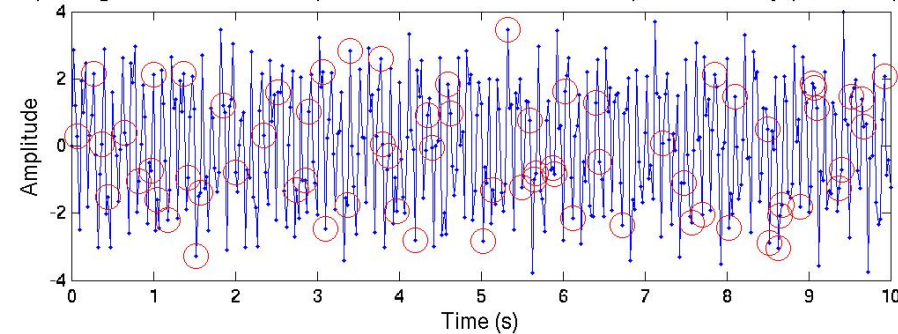


- Sense signal M times
- Recover with linear program

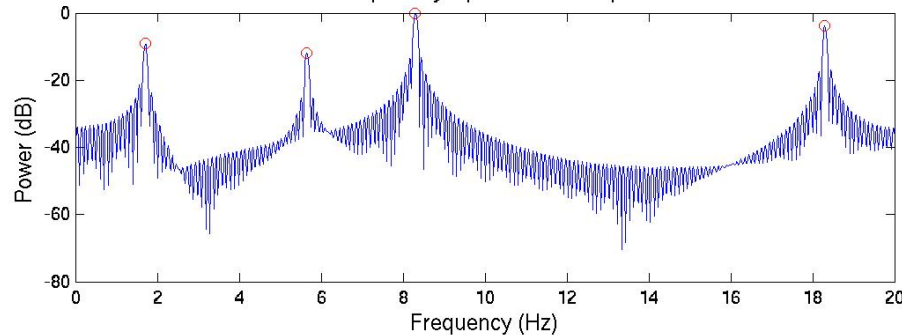
$$\min_{\omega} \sum |\hat{g}(\omega)| \quad \text{subject to} \quad g(t_m) = f(t_m), \quad m = 1, \dots, M$$

Compressive Sampling

Input signal with undersampled measurements circled (~17.5% of Nyquist samples)



Frequency spectrum of input

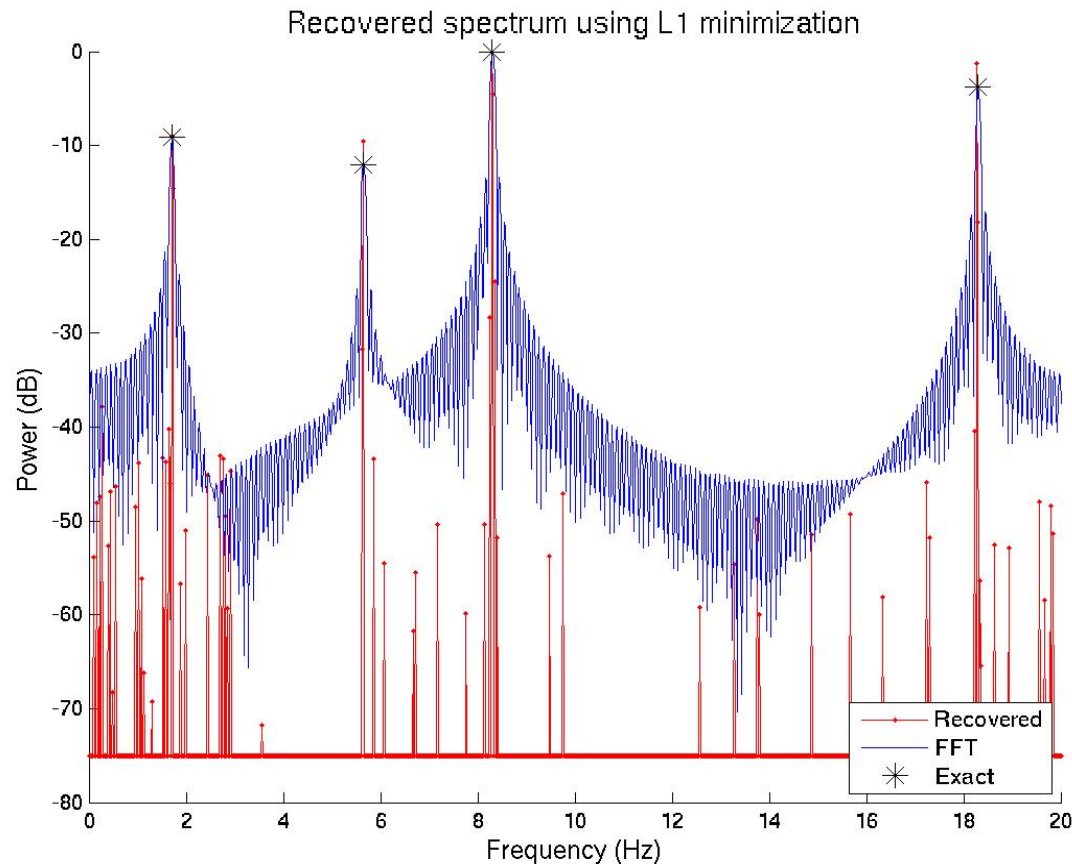


$$\hat{f}(\omega) = \sum_{i=1}^K \alpha_i \delta(\omega_i - \omega) \xleftrightarrow{\mathcal{F}} f(t) = \sum_{i=1}^K \alpha_i e^{i\omega_i t}$$

- Sense signal M times
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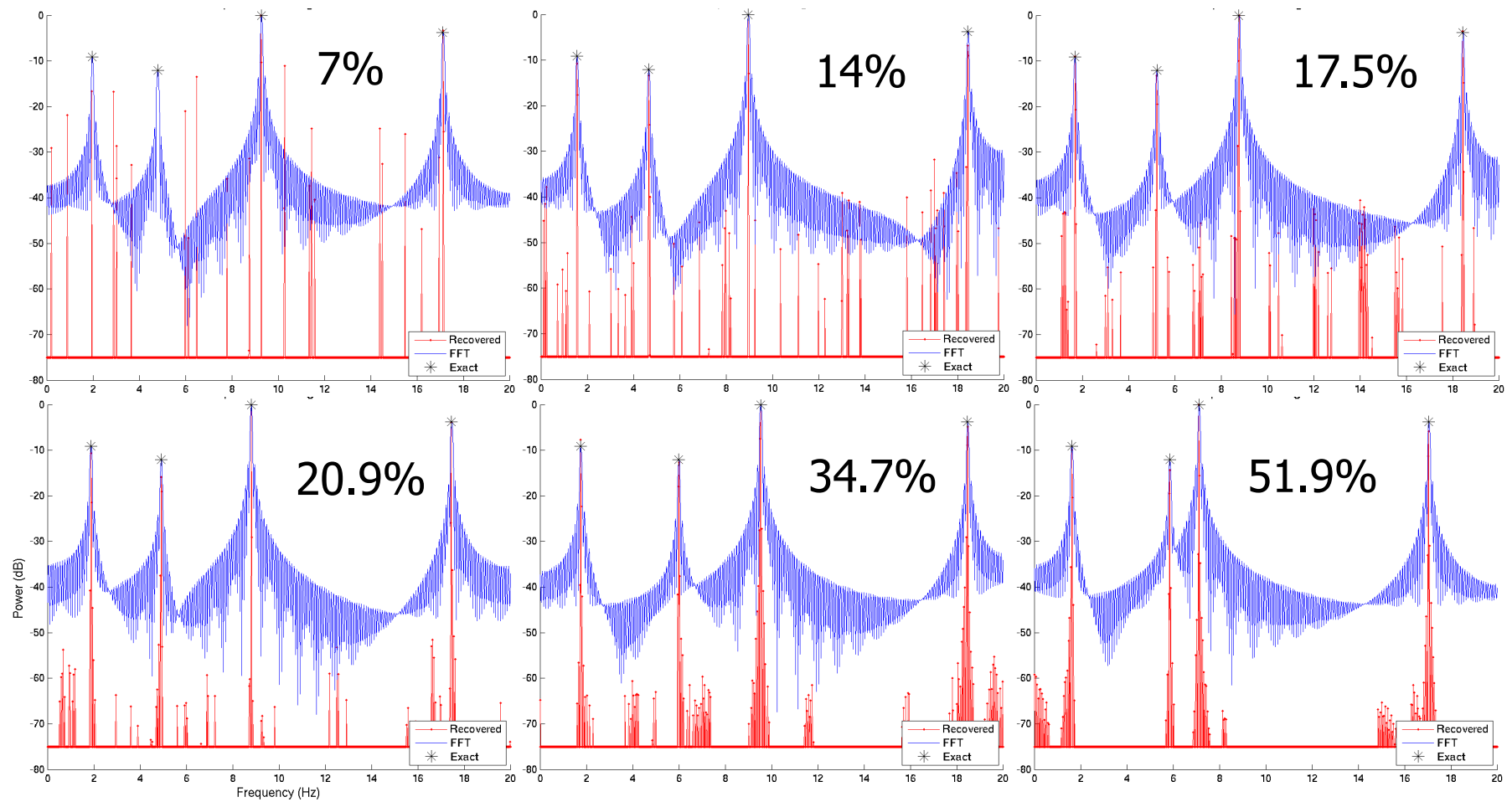
$$\min_{\omega} \sum |\hat{g}(\omega)| \quad \text{subject to} \quad g(t_m) = f(t_m), \quad m = 1, \dots, M$$

Example: Sum of Sinusoids

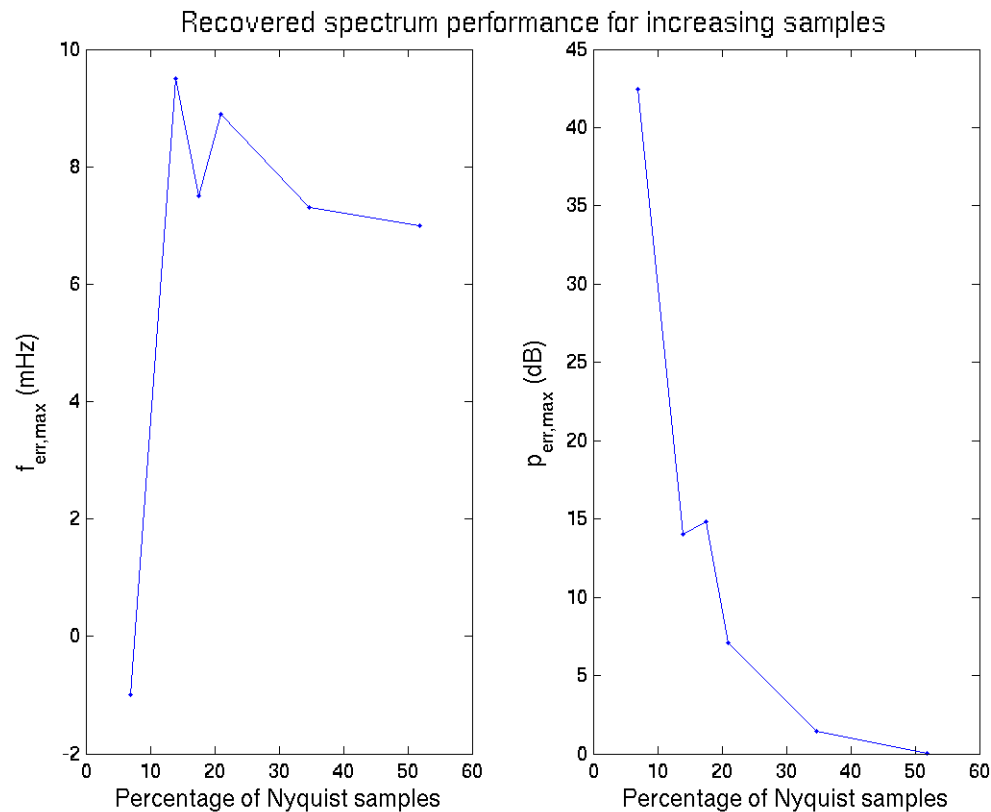


- Two relevant “knobs”
 - percentage of Nyquist samples as altered by adjusting number of samples, M
 - input signal duration, T
 - Data block size

Example: Increasing M

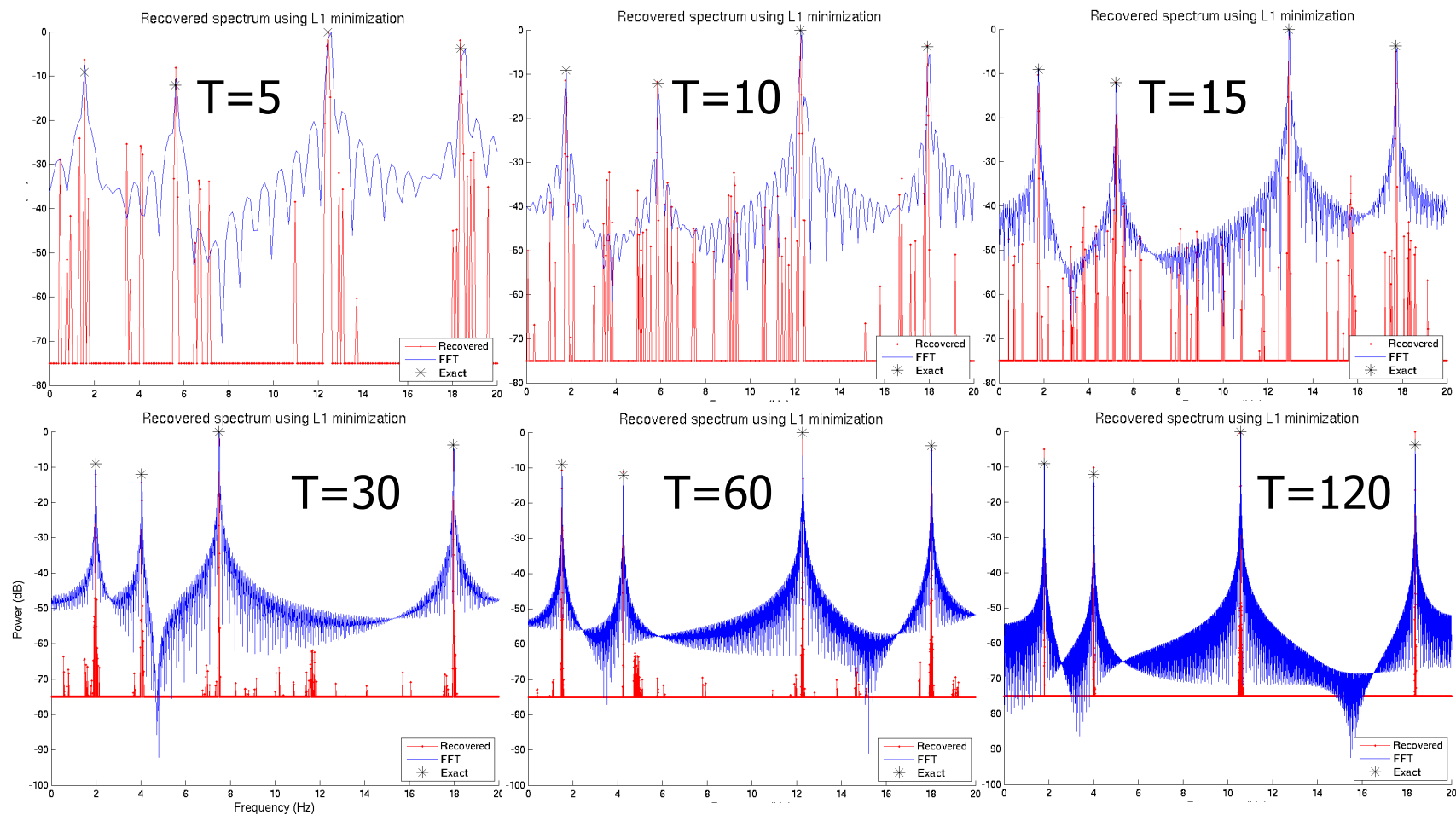


Example: Increasing M

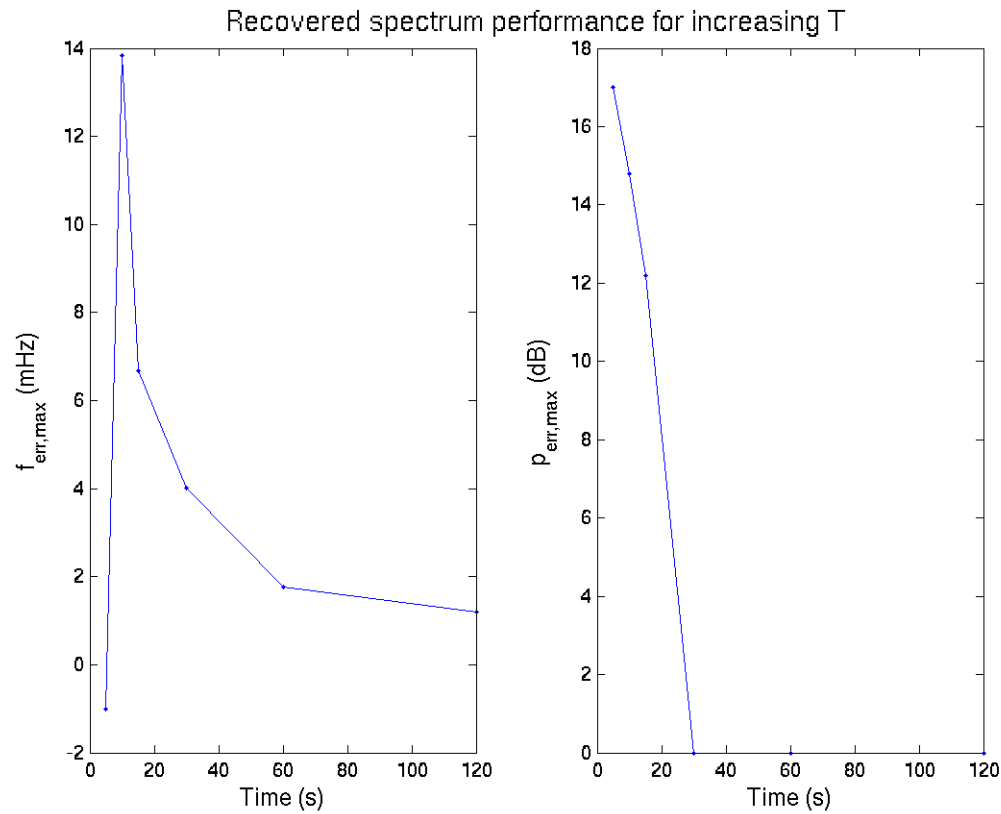


- $f_{err,max}$ within 10 mHz
- $p_{err,max}$ decreasing

Example: Increasing T



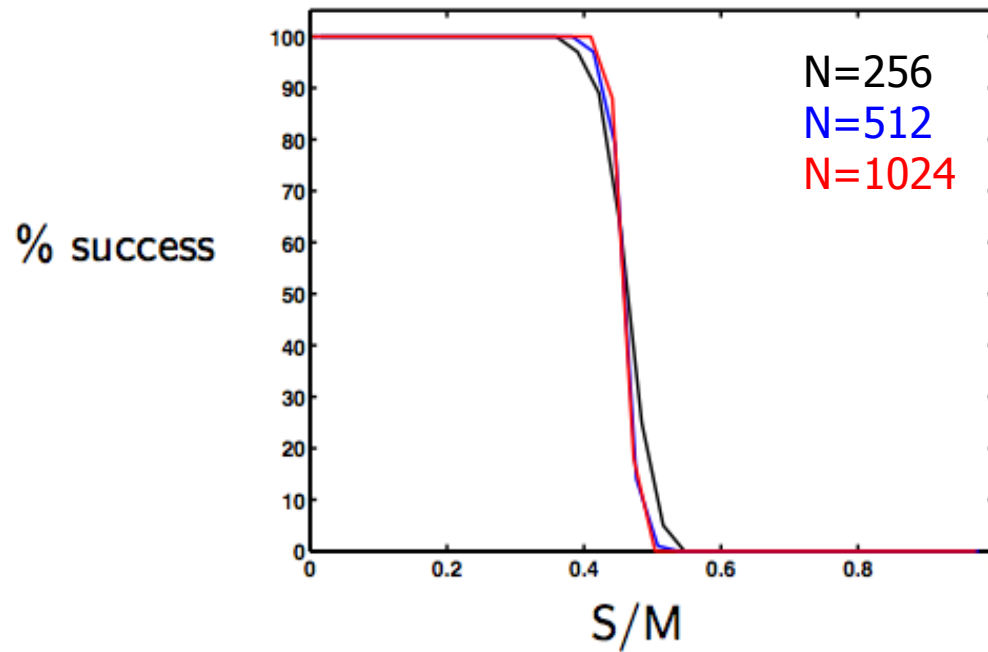
Example: Increasing T



- $f_{err,max}$ decreasing
- $p_{err,max}$ decreasing

Numerical Recovery Curves

- Sense S -sparse signal of length N randomly M times



- In practice, perfect recovery occurs when $M \approx 2S$ for $N \approx 1000$

A Non-Linear Sampling Theorem

- Exact Recovery Theorem (Candès, R, Tao, 2004):

- Select M sample locations $\{t_m\}$ “at random” with

$$M \geq \text{Const} \cdot S \log N$$

- Take time-domain samples (measurements)

$$y_m = x_0(t_m)$$

- Solve

$$\min_x \|\hat{x}\|_{\ell_1} \quad \text{subject to} \quad x(t_m) = y_m, \quad m = 1, \dots, M$$

- Solution is **exactly** recovered signal with extremely high probability

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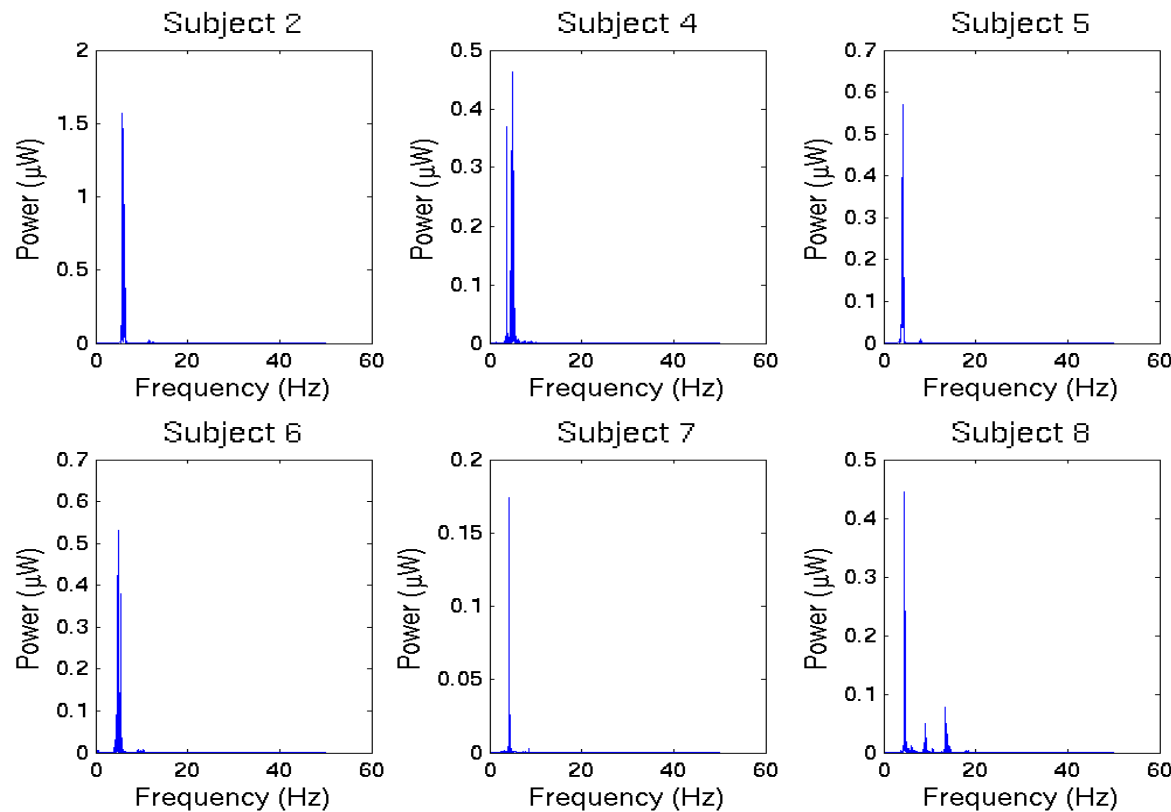
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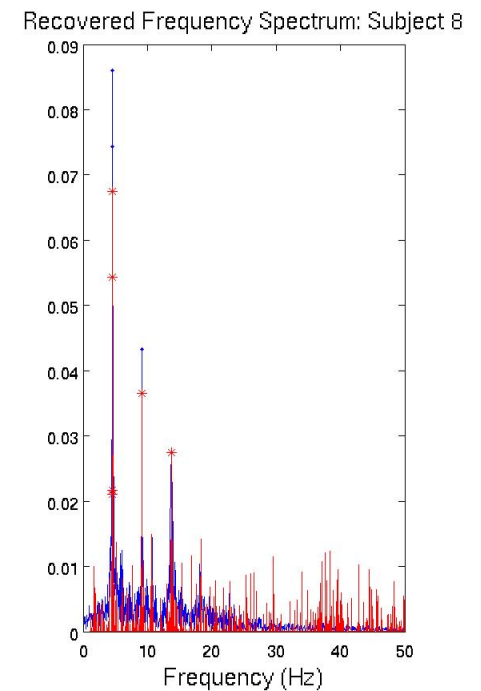
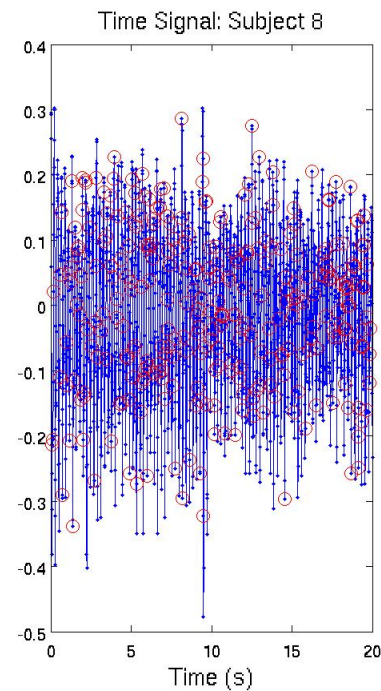
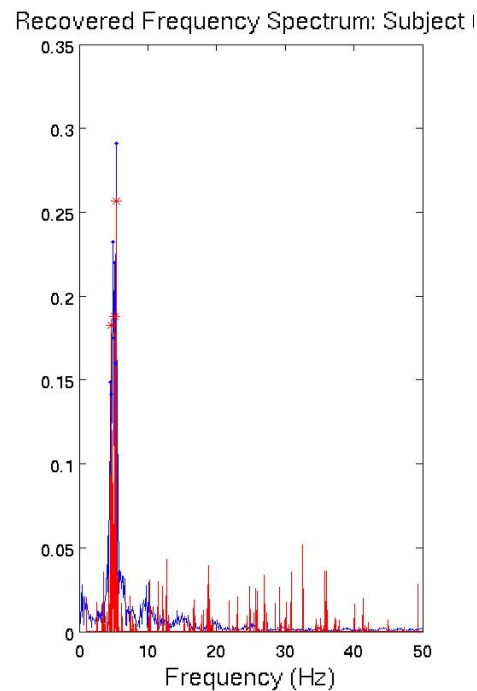
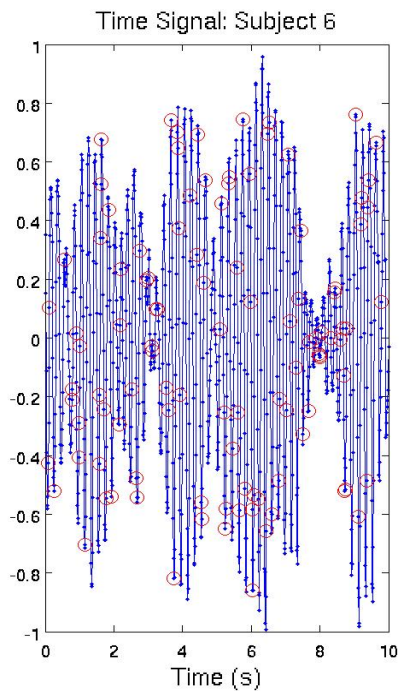
$$M > C \cdot \mu^2(\Phi, \Psi) \cdot S \cdot \log N$$

Biometric Example: Parkinson's Tremors

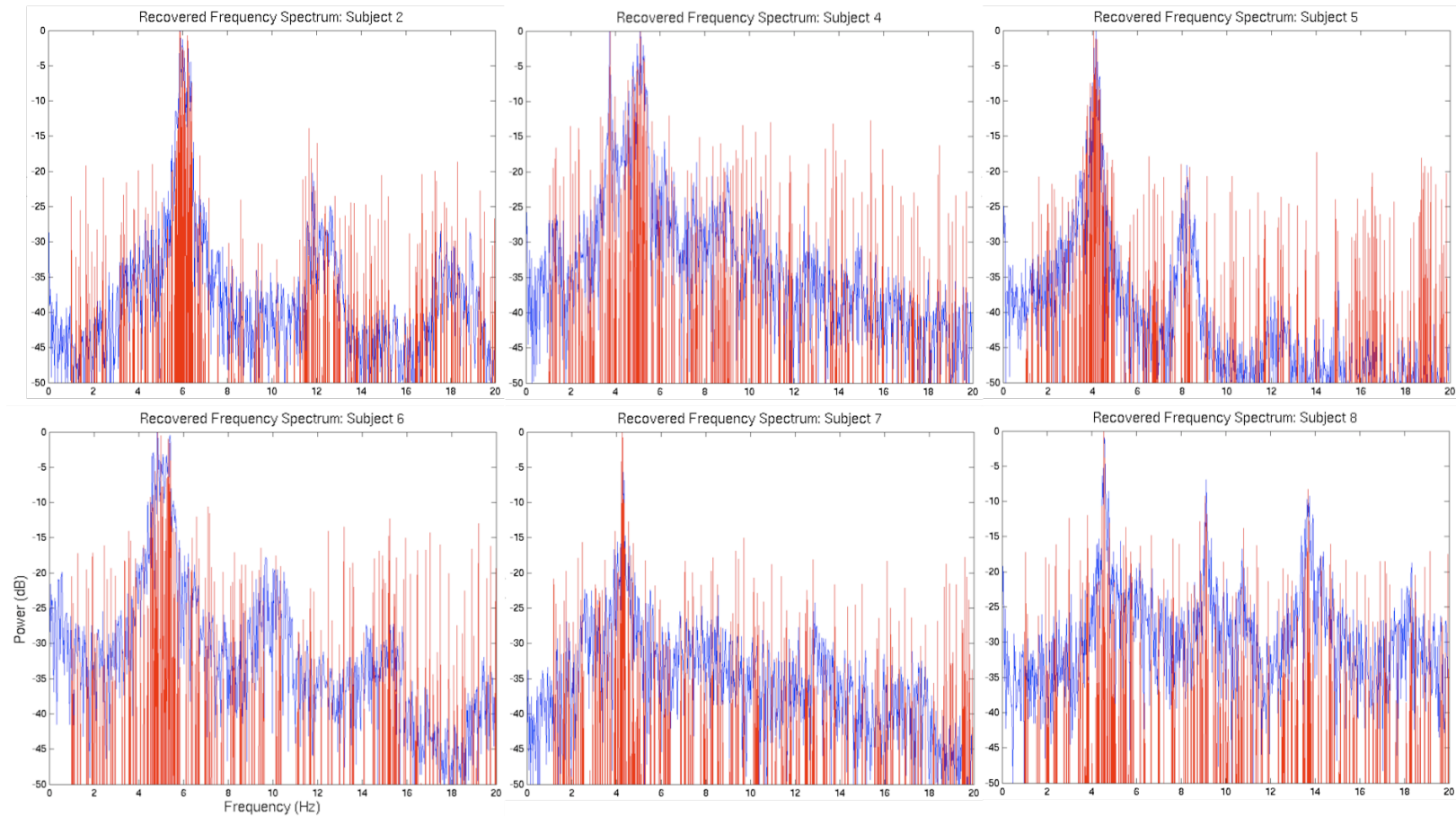


- 6 Subjects of real tremor data
 - collected using low intensity velocity-transducing laser recording aimed at reflective tape attached to the subjects' finger recording the finger velocity
 - All show Parkinson's tremor in the 4-6 Hz range.
 - Subject 8 shows activity at two higher frequencies
 - Subject 4 appears to have two tremors very close to each other in frequency

Compressive Sampling: Real Data



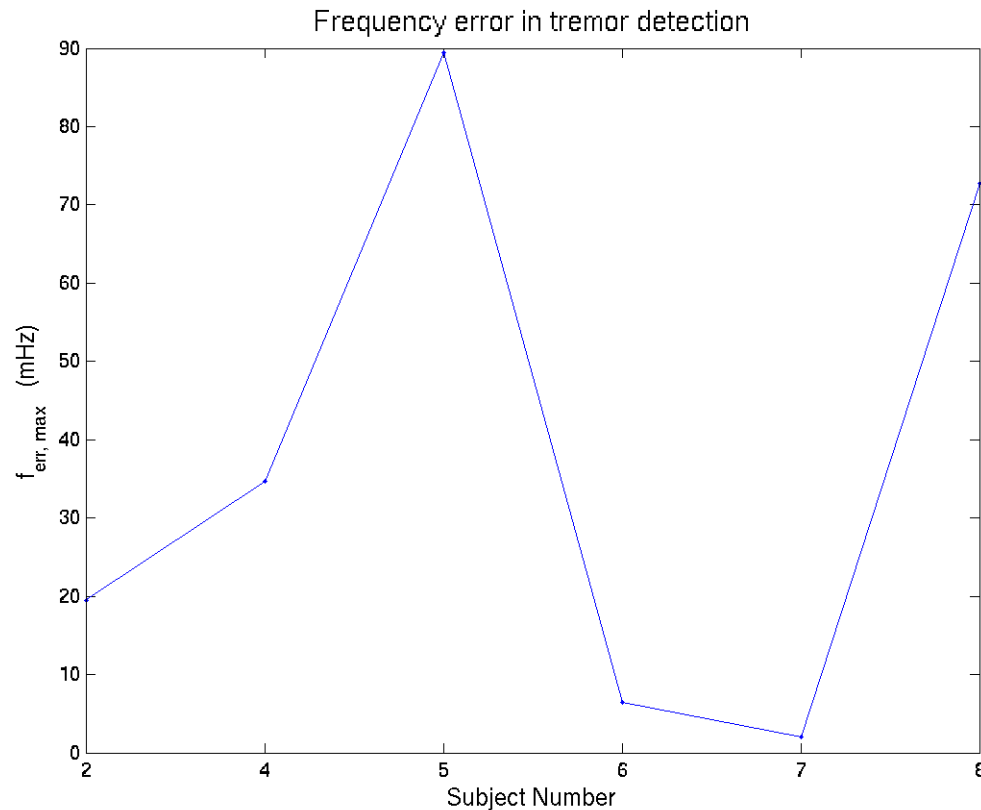
Biometric Example: Parkinson's Tremors



■ **C=10.5, T=30**

■ 20% Nyquist required samples

Biometric Example: Parkinson's Tremors



- Tremors detected within 100 mHz
- randomly sample 20% of the Nyquist required samples

Requires post processing to randomly sample!



Implementing Compressive Sampling

- ❑ Devised a way to randomly sample 20% of the Nyquist required samples and still detect the tremor frequencies within 100mHz
 - Requires post processing to randomly sample!

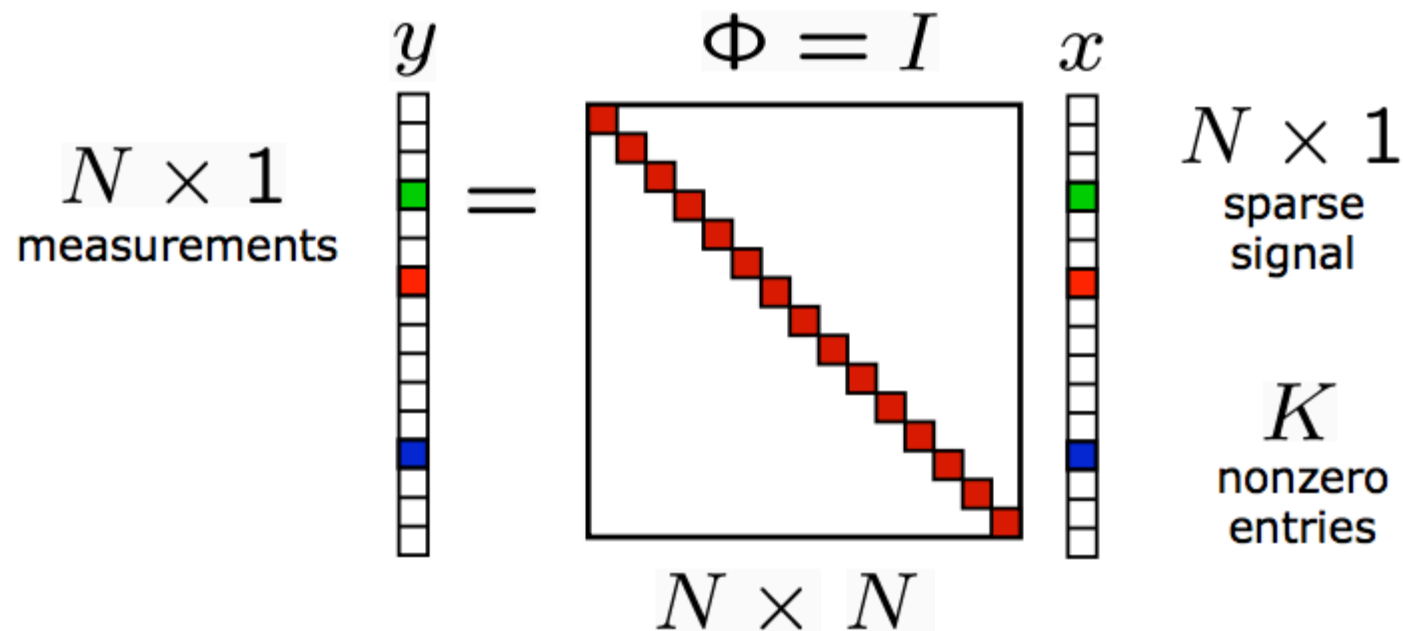
- ❑ Implement hardware on chip to “choose” samples in real time
 - Only write to memory the “chosen” samples
 - Design random-like sequence generator
 - Only convert the “chosen” samples
 - Design low energy ADC

CS Theory

Why does it work?

Sampling

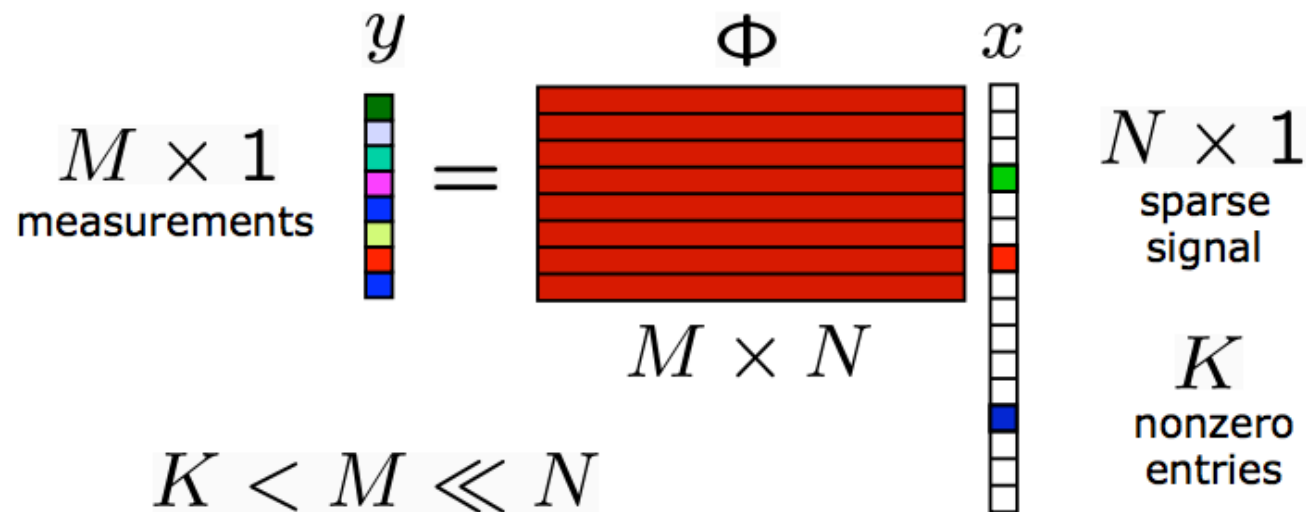
- Signal x is K -sparse in basis/dictionary Ψ
 - WLOG assume sparse in space domain $\Psi = I$
- Sampling**



Compressive Sampling

- When data is sparse/compressible, can directly acquire a **condensed representation** with no/little information loss through linear **dimensionality reduction**

$$y = \Phi x$$

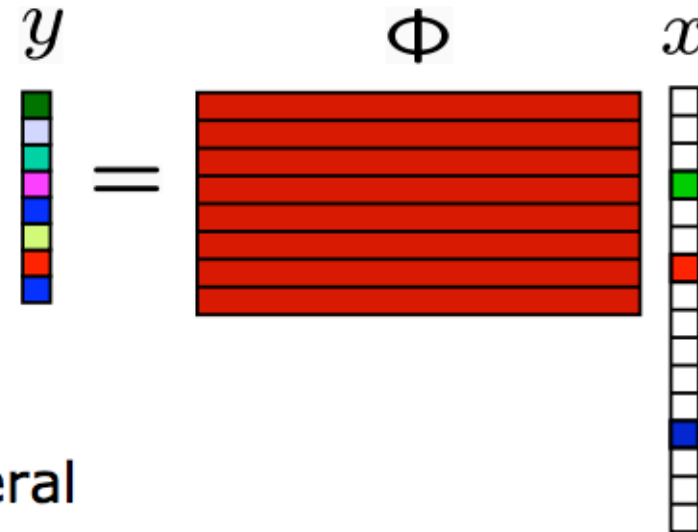


How Can It Work?

- Projection Φ
not full rank...

$$M < N$$

... and so
loses information in general



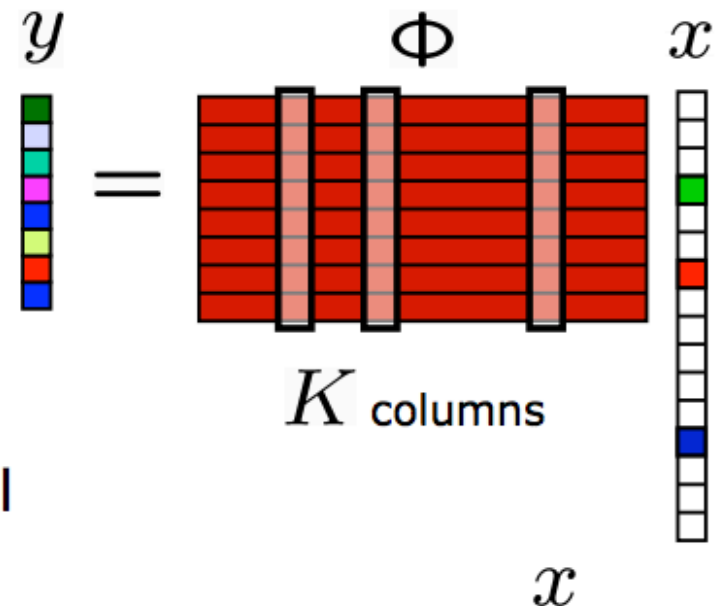
- Ex: Infinitely many x 's map to the same y
(null space)

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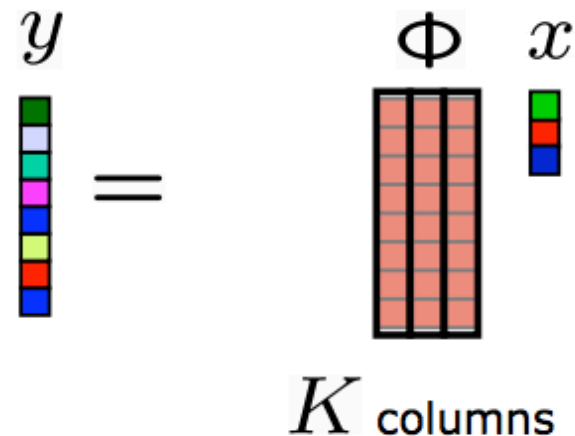
- But we are only interested in **sparse** vectors

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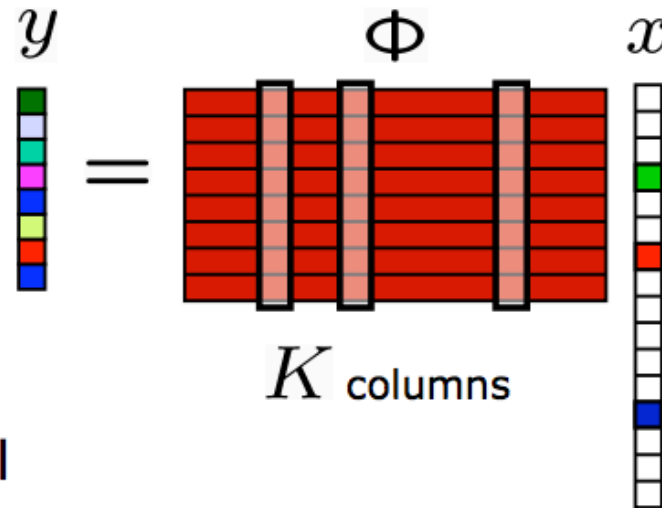
- But we are only interested in **sparse** vectors
- Φ is effectively $M \times K$

How Can It Work?

- Projection Φ not full rank...

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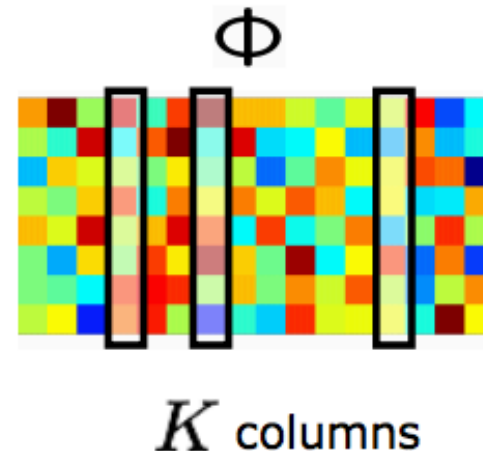
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- But we are only interested in **sparse** vectors
- **Design** Φ so that each of its $M \times K$ submatrices are full rank (ideally close to orthonormal)
 - **Restricted Isometry Property (RIP)**

RIP

- Draw Φ at **random**
 - iid Gaussian
 - iid Bernoulli ± 1
 - ...



- Then Φ has the RIP with high probability provided

$$M = O(K \log(N/K)) \ll N$$

CS Signal Recovery

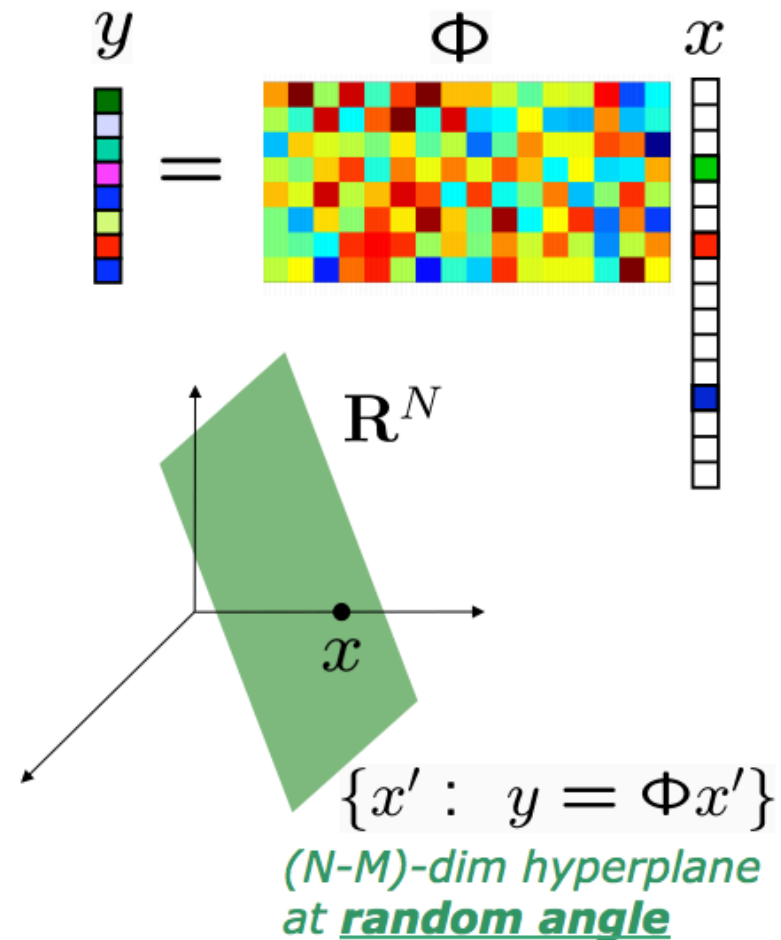
- **Goal:** Recover signal x from measurements y

$$y = \Phi x$$

- **Problem:** Random projection Φ not full rank (ill-posed inverse problem)
- **Solution:** Exploit the sparse/compressible **geometry** of acquired signal x

CS Signal Recovery

- Random projection Φ not full rank
- Recovery problem:
given $y = \Phi x$
find x
- **Null space**
- Search in null space for the “best” x according to some criterion
 - ex: least squares



L₂ Signal Recovery

- Recovery:
(ill-posed inverse problem)

given $y = \Phi x$
find x (sparse)

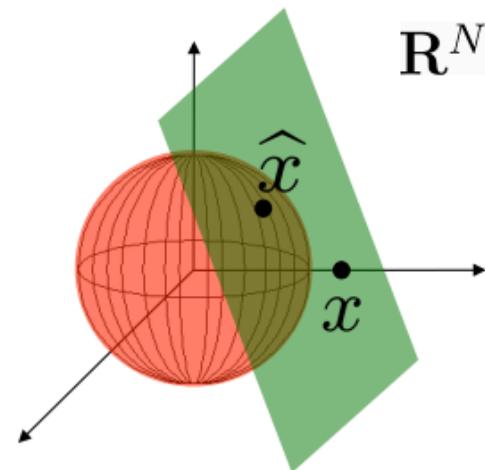
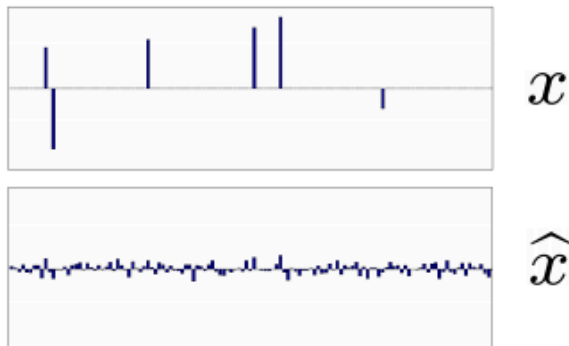
- Optimization:

$$\hat{x} = \arg \min_{y=\Phi x} \|x\|_2$$

- Closed-form solution:

$$\hat{x} = (\Phi^T \Phi)^{-1} \Phi^T y$$

- **Wrong answer!**



L_0 Signal Recovery

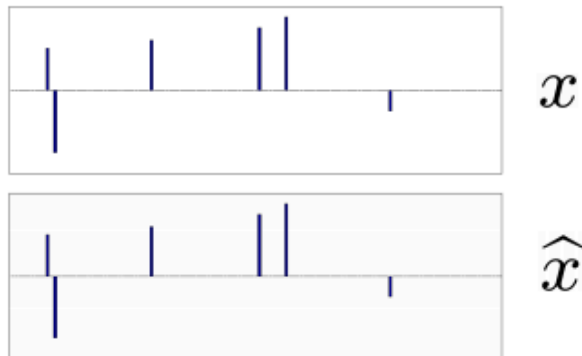
- Recovery:
(ill-posed inverse problem)

given $y = \Phi x$
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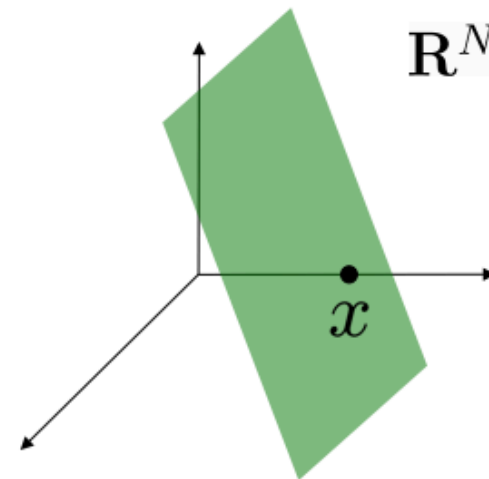
- Optimization:

$$\hat{x} = \arg \min_{y=\Phi x} \|x\|_0$$

- **Correct!**



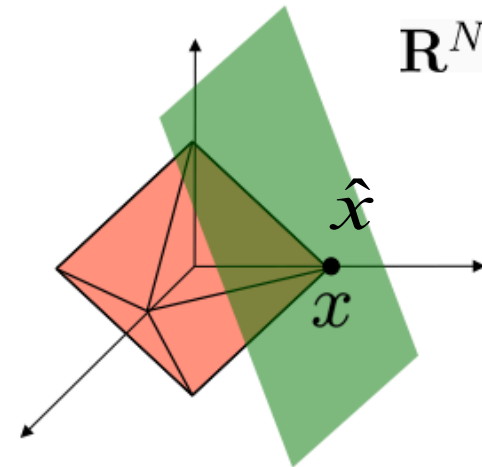
*“find **sparsest** vector
in translated nullspace”*



- But **NP-Complete** alg

L_1 Signal Recovery

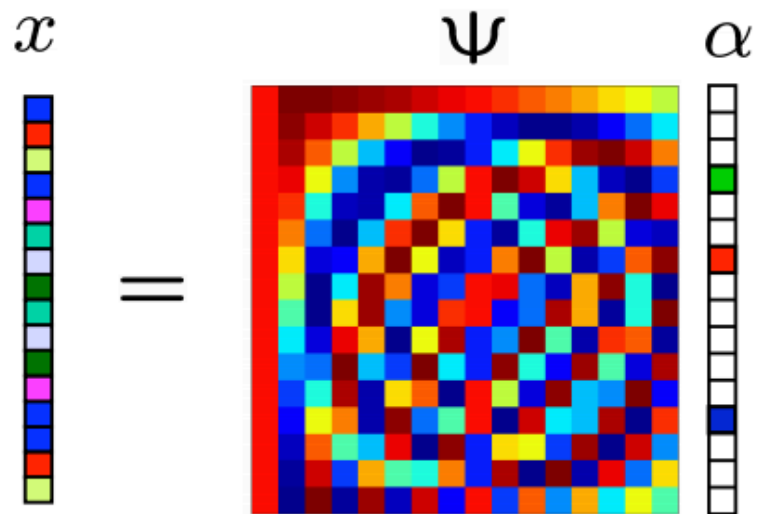
- Recovery:
(ill-posed inverse problem) given $y = \Phi x$
find x (sparse)
- Optimization: $\hat{x} = \arg \min_{y=\Phi x} \|x\|_1$
- **Convexify** the ℓ_0 optimization
- **Correct!**
- **Polynomial time** alg
(linear programming)
- Much recent alg progress
– greedy, Bayesian approaches, ...



Universality

- Random measurements can be used for signals sparse in *any* basis

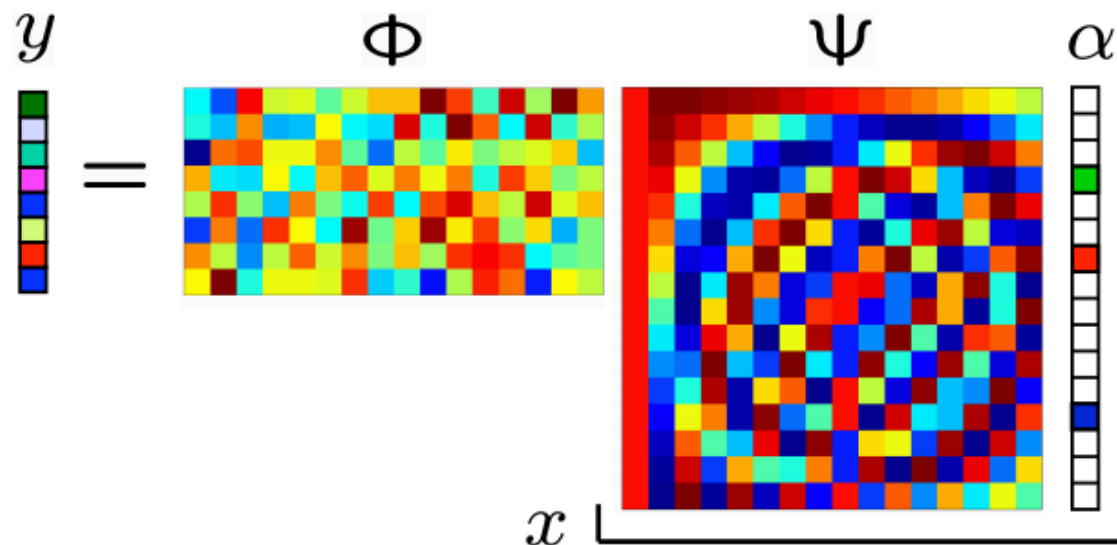
$$x = \Psi \alpha$$



Universality

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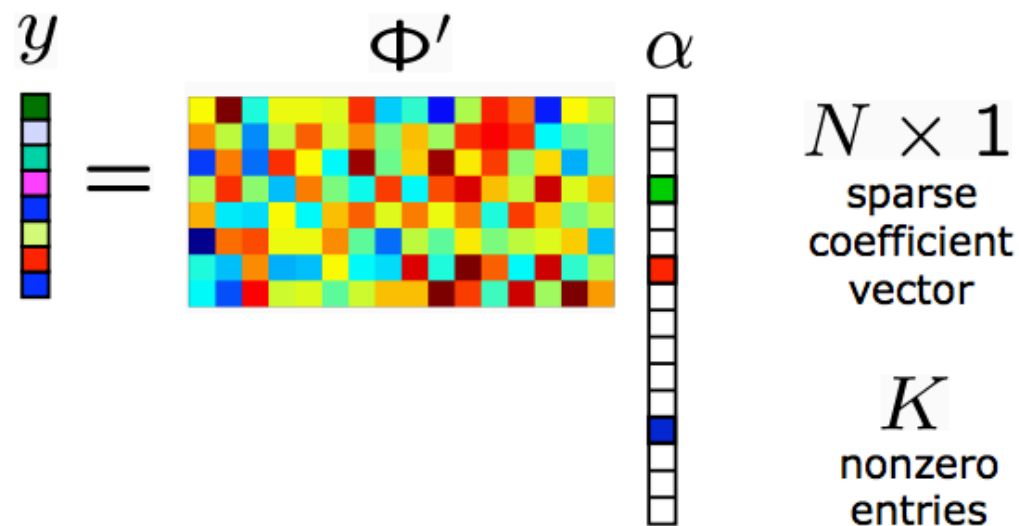
$$y = \Phi x = \Phi \Psi \alpha$$



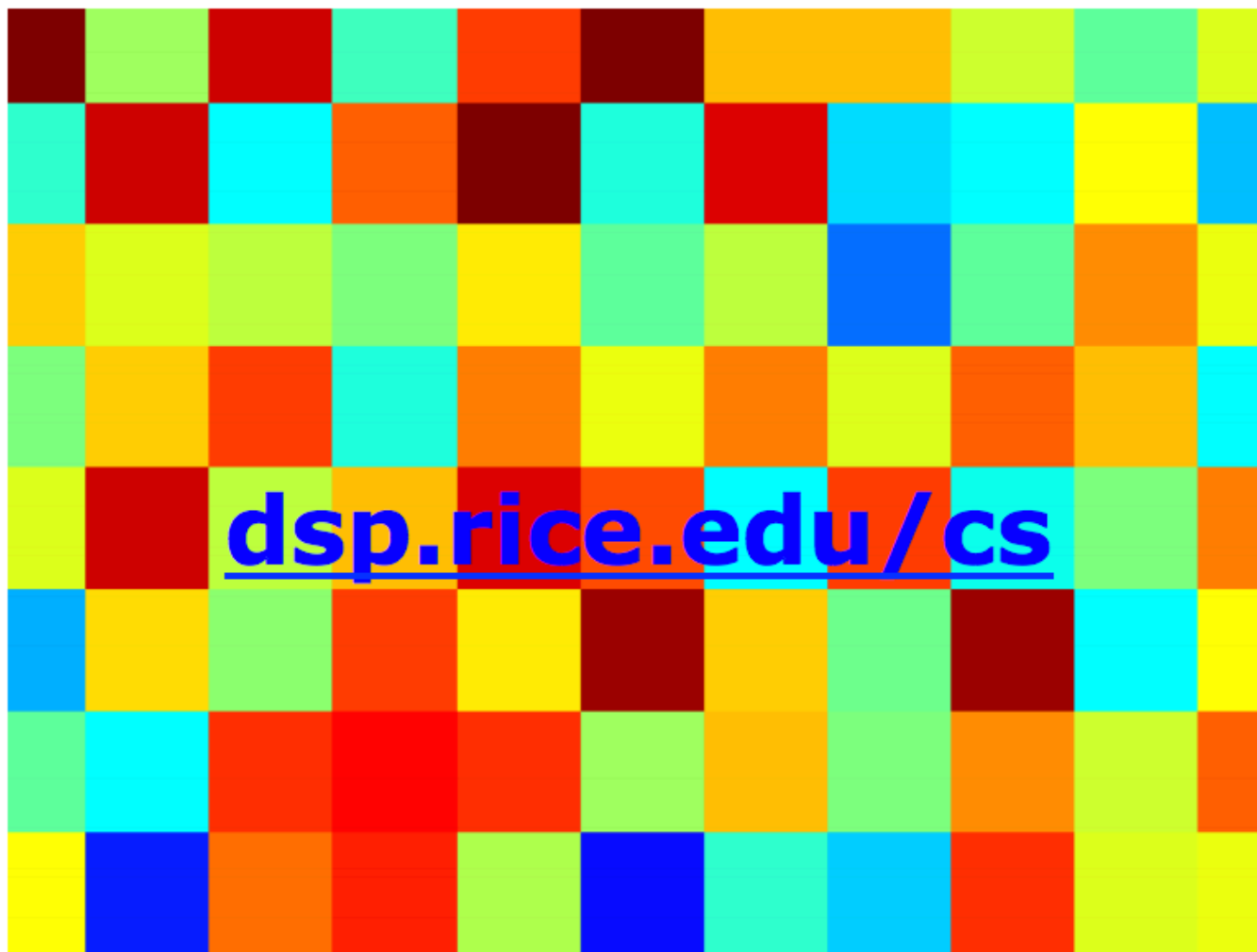
Universality

- Random measurements can be used for signals sparse in *any* basis

$$y = \Phi x = \Phi \Psi \alpha = \Phi' \alpha$$



Reference Slide





Big Ideas

- ❑ Compressive Sampling
 - Integrated sensing/sampling, compression and processing
 - Based on sparsity and incoherency



Admin

- ❑ Final Project due – Apr 28th
 - TA advice – “The report takes time. Leave time for it.”
- ❑ Last day of TA office hours – Apr 28th
 - Piazza still available
 - Review session for exam TBD
- ❑ Last day of Tania office hours – May 1st
- ❑ Final Exam – May 7th



Final Exam Admin

- ❑ Final Exam – 5/7 (3pm-5pm)
 - In Canvas
 - Will have a 2 hr window to complete within a 12 hr time block
 - Open course notes and textbook, but cannot communicate with each other about the exam
 - Students will have randomized and different questions
 - Reminder, it is not in your best interest to share the exam
 - Old exams posted on old course websites
 - Covers Lec 1- 20
 - Does not include lec 12 (data converters and noise shaping) or IIR Filters