

# ESE 531: Digital Signal Processing

---

Lec 26: April 25, 2019  
Compressive Sensing

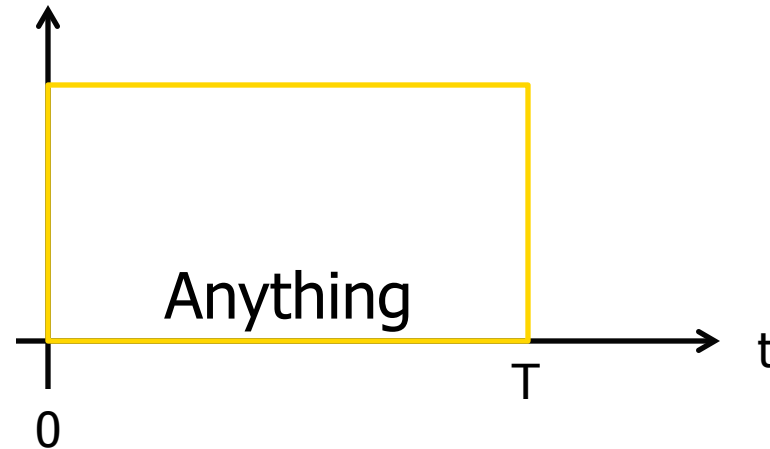


# Today

---

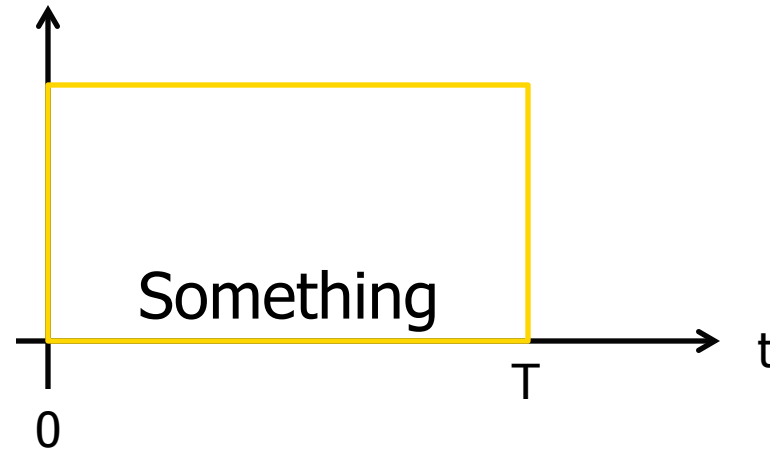
- ❑ Compressive Sampling/Sensing

# Compressive Sampling



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

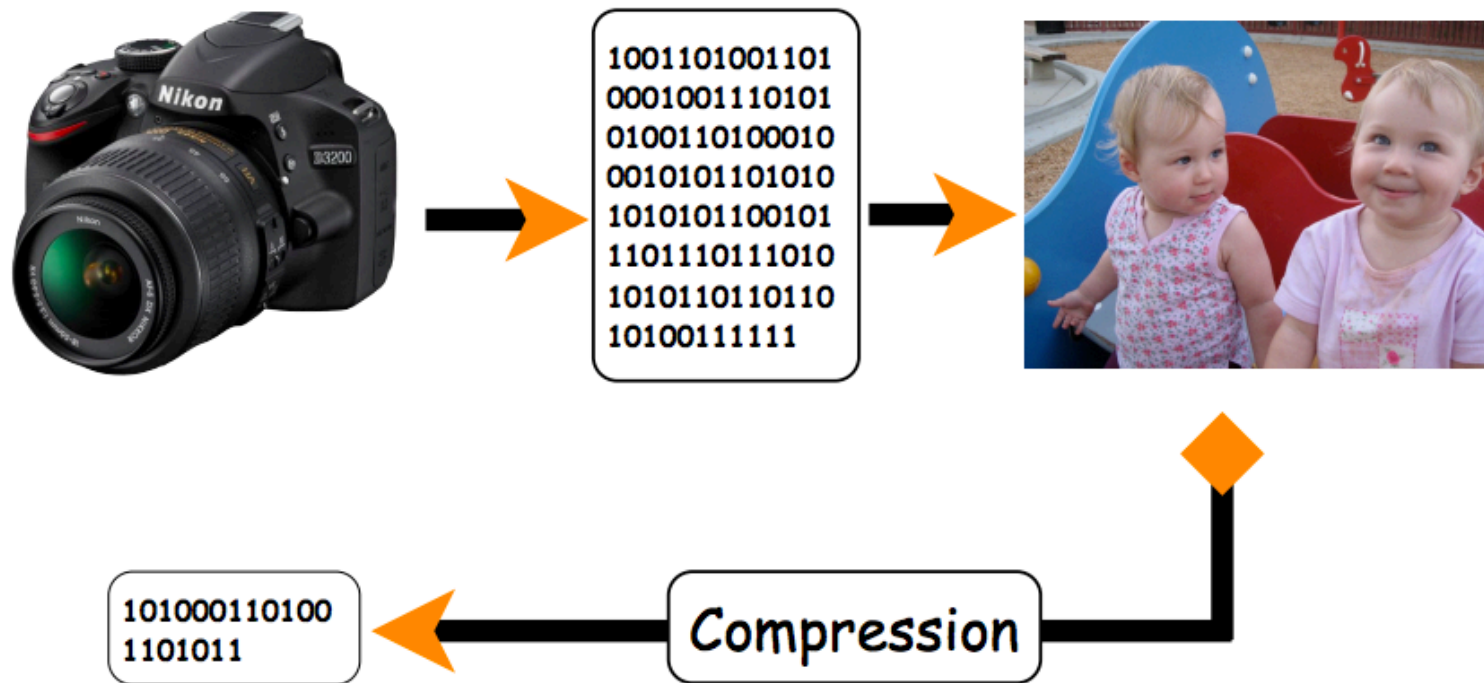
# Compressive Sampling



- ❑ 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



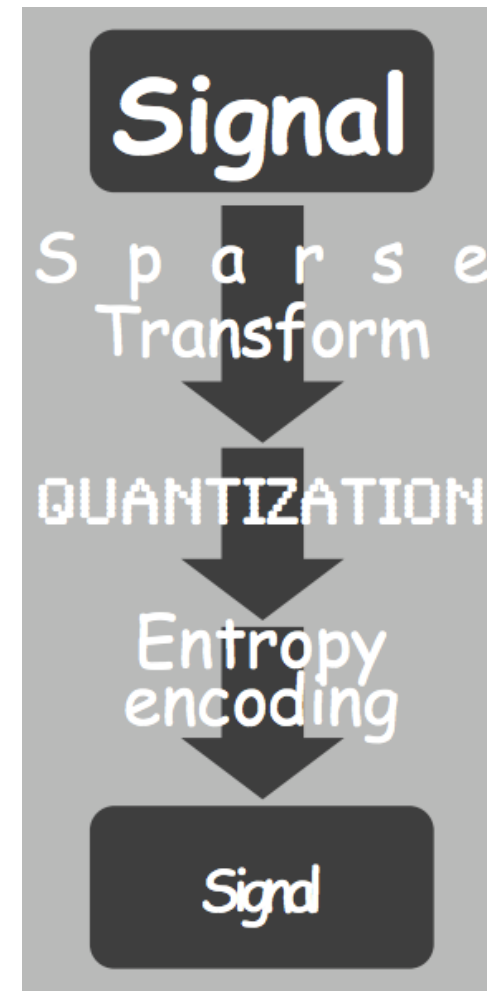
# First: Compression

---

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

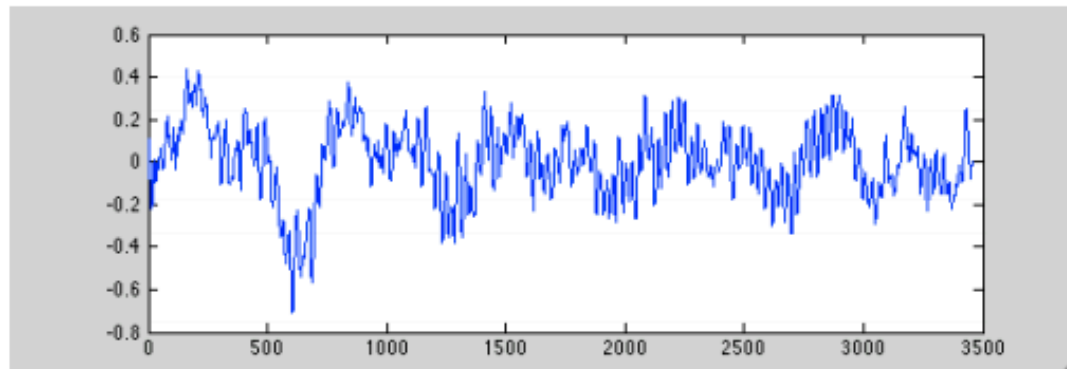
# First: Compression

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

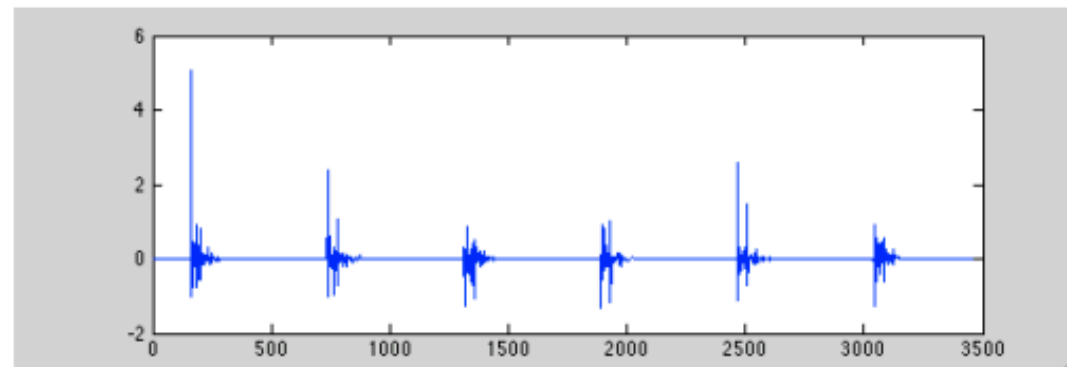




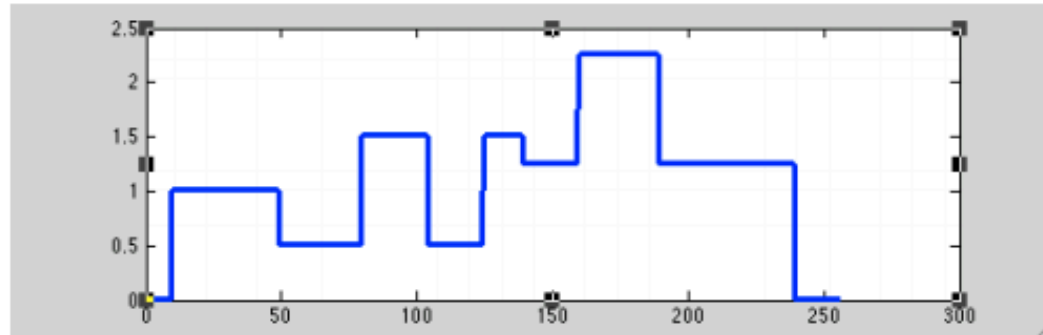
# Sparse Transform



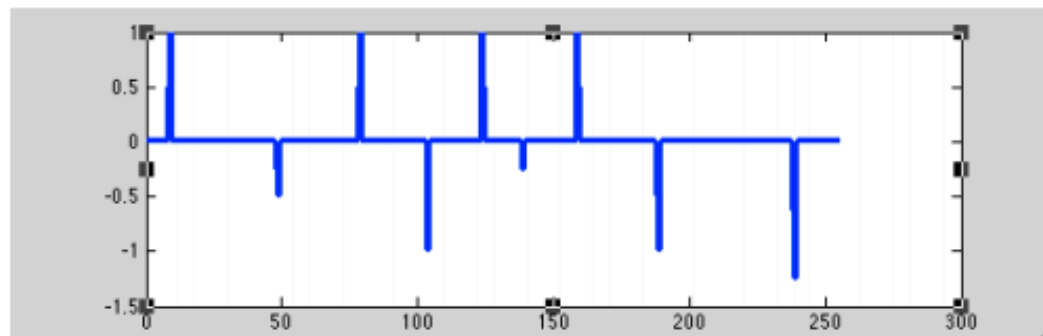
DCT



# Sparse Transform

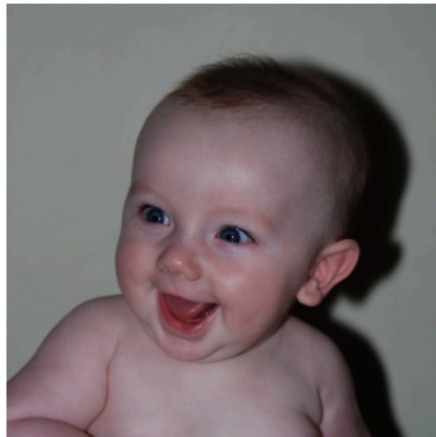


Difference



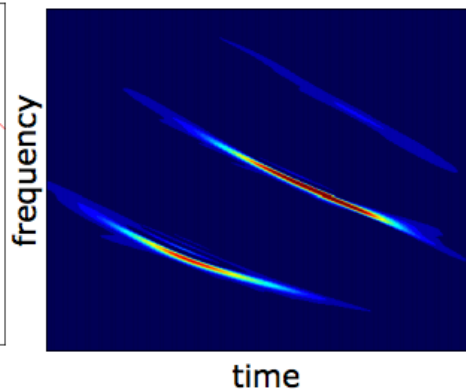
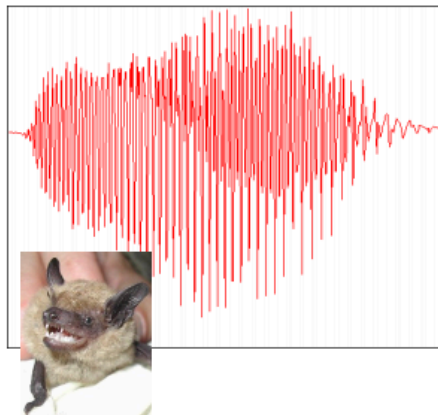
# Sparsity

$N$   
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...



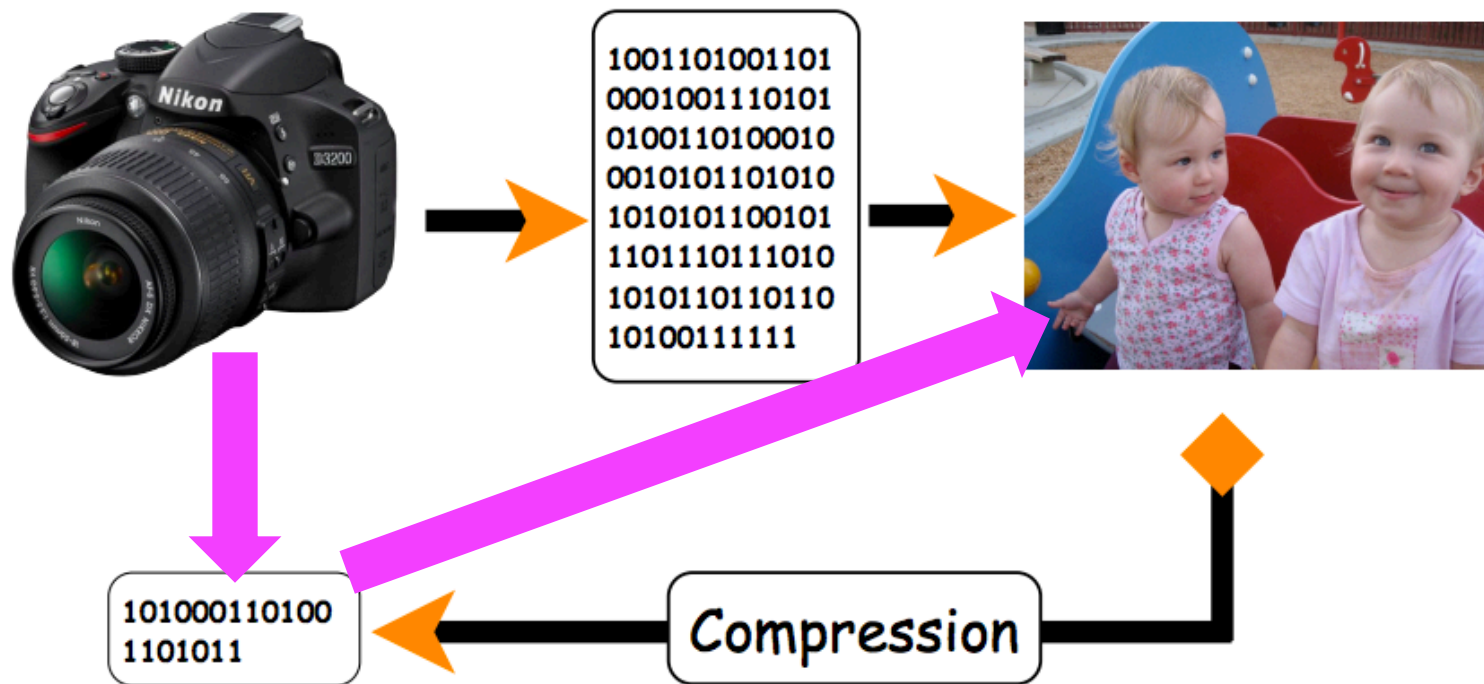
# Signal Processing Trends

---

- ❑ 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





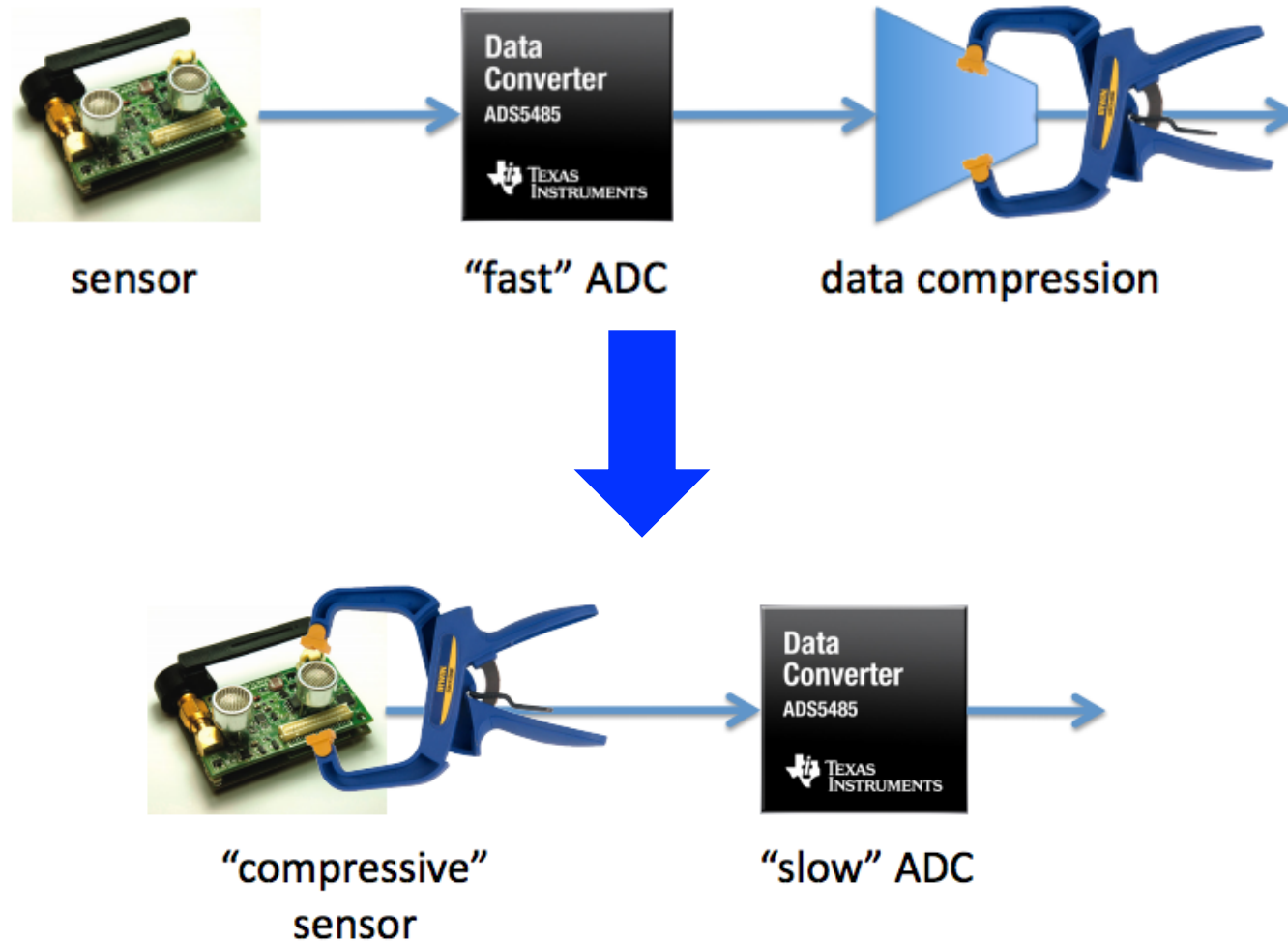
# Compressive Sensing

---

- ❑ 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

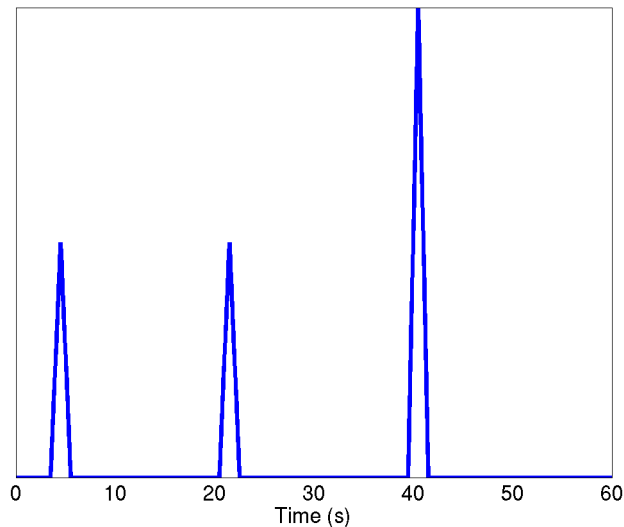
---

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

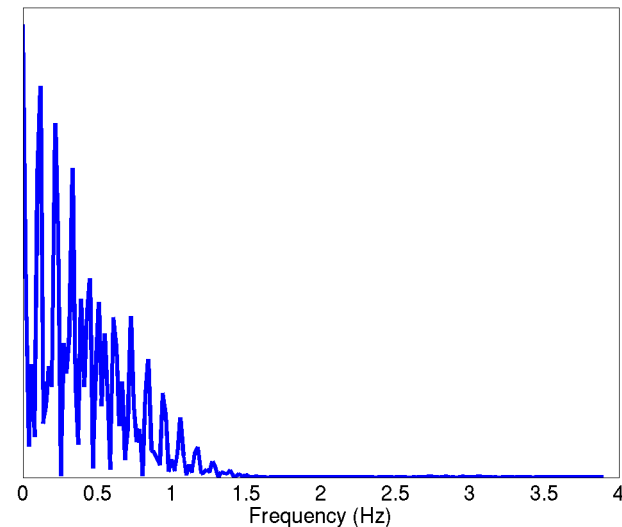
# Compressive Sampling

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

Sparse signal in time



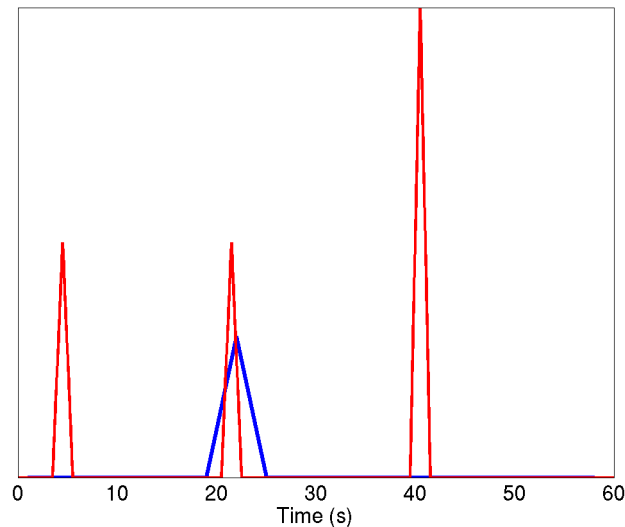
Frequency spectrum



# Compressive Sampling

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

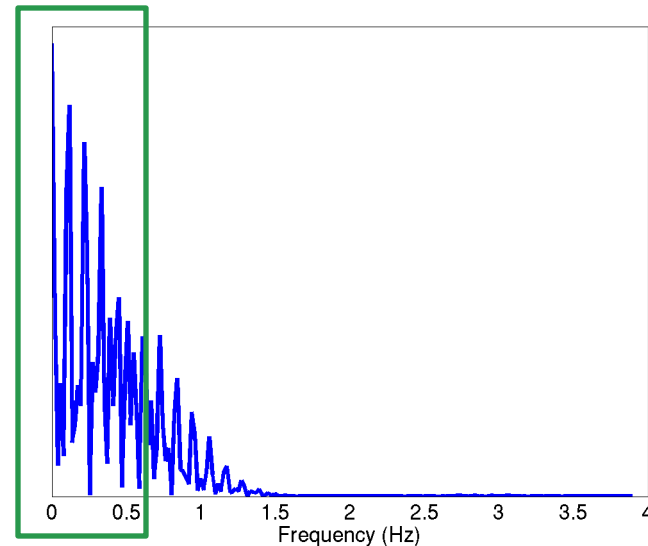
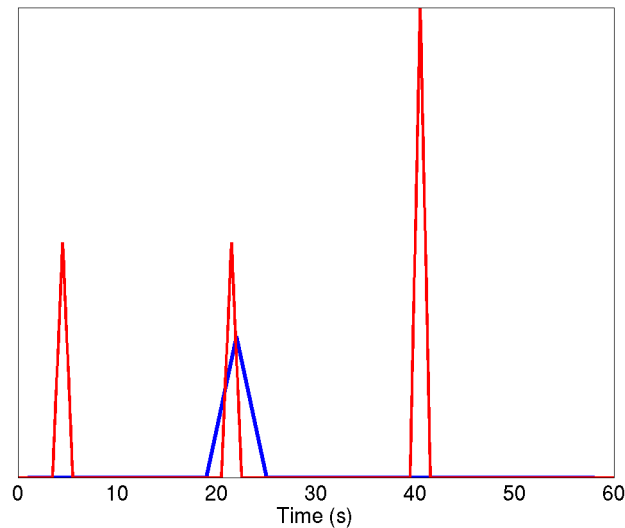
Undersampled in time



# Compressive Sampling

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

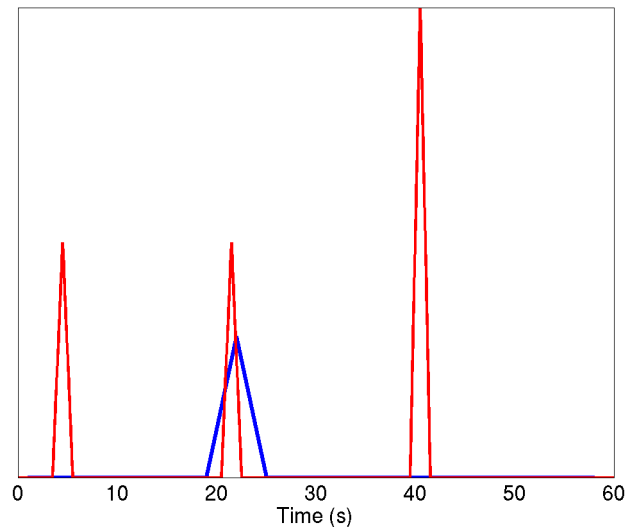
Undersampled in time



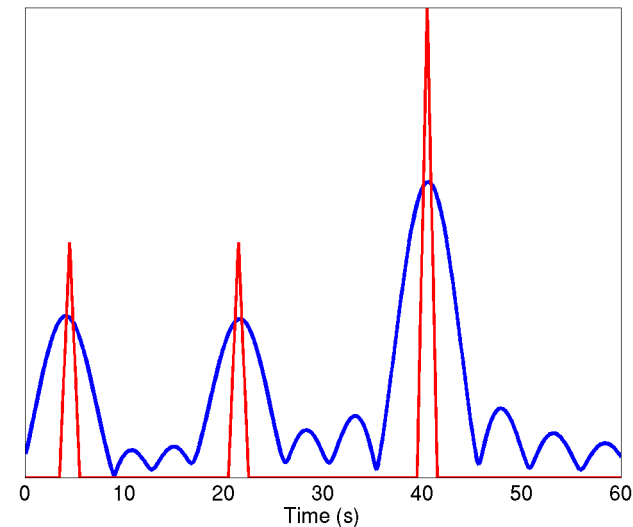
# Compressive Sampling

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

Undersampled in time



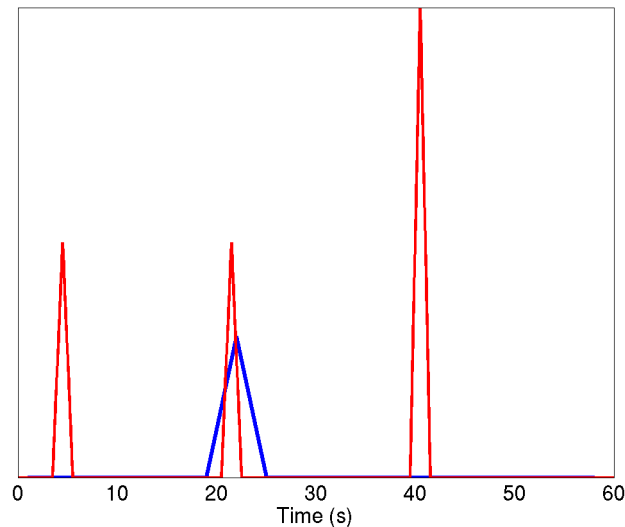
Undersampled in frequency  
(reconstructed in time with IFFT)



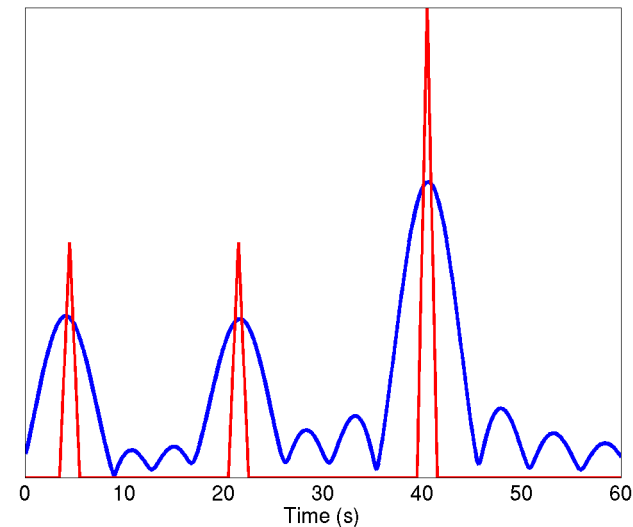
# Compressive Sampling

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

Undersampled in time

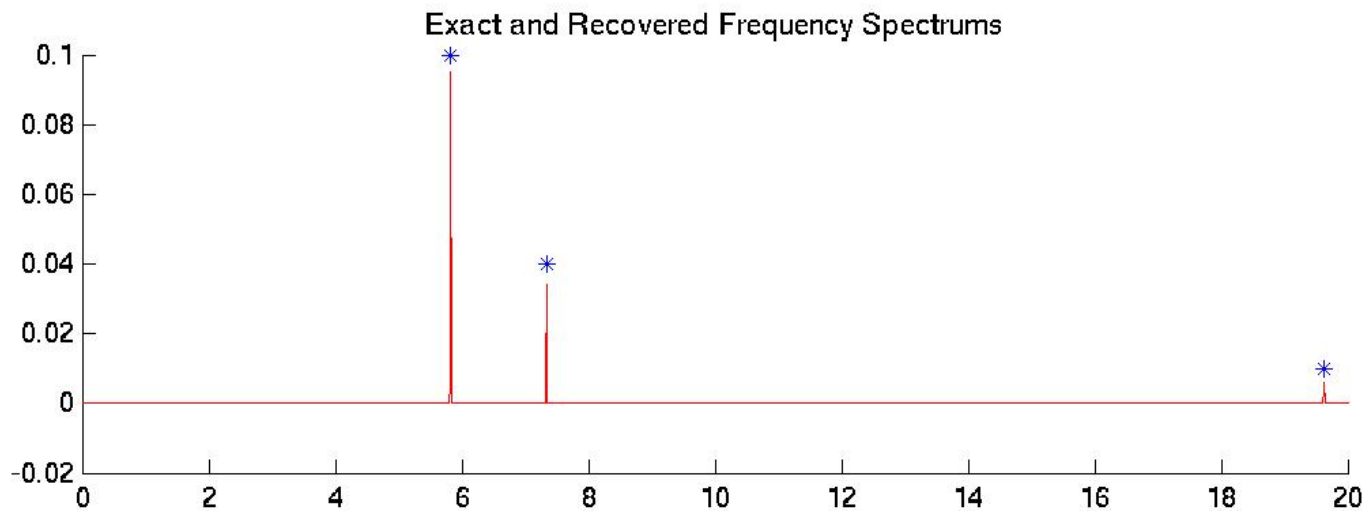
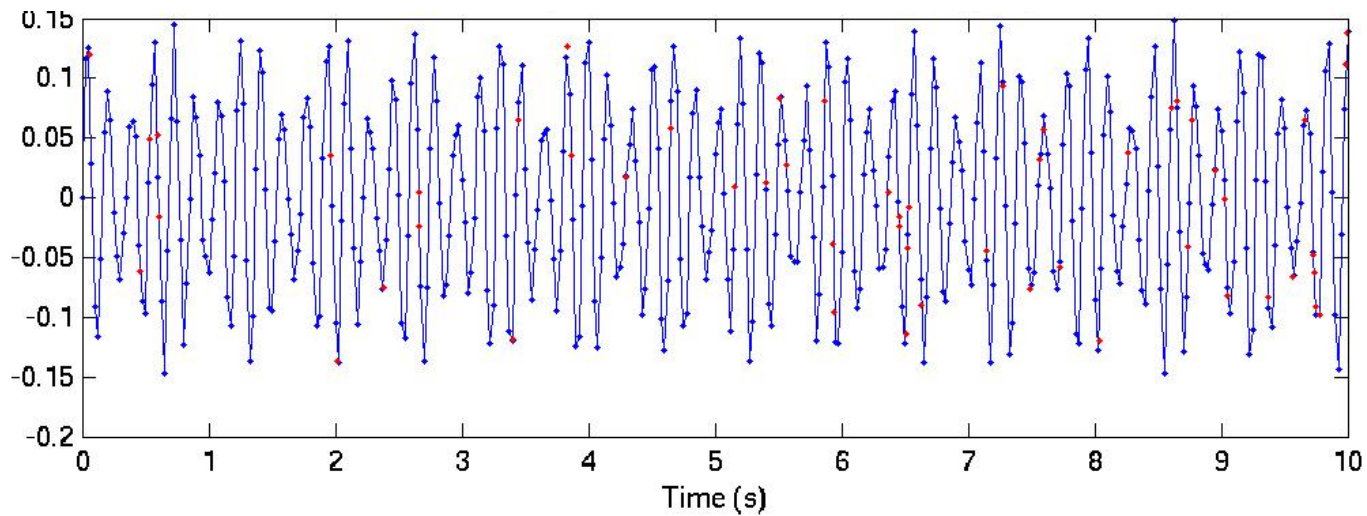


Undersampled in frequency  
(reconstructed in time with IFFT)



Requires sparsity and incoherent sampling

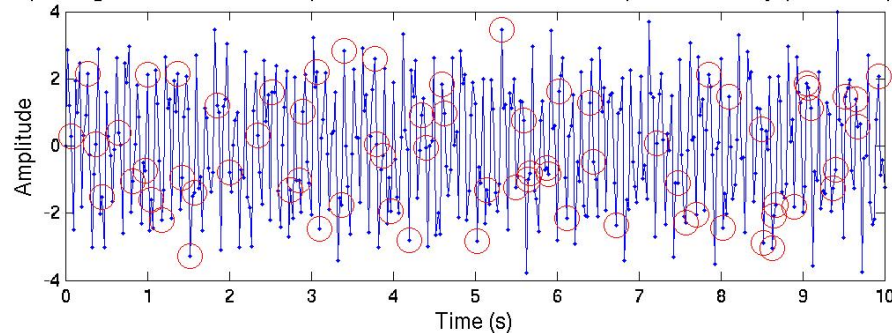
# Compressive Sampling: Simple Example



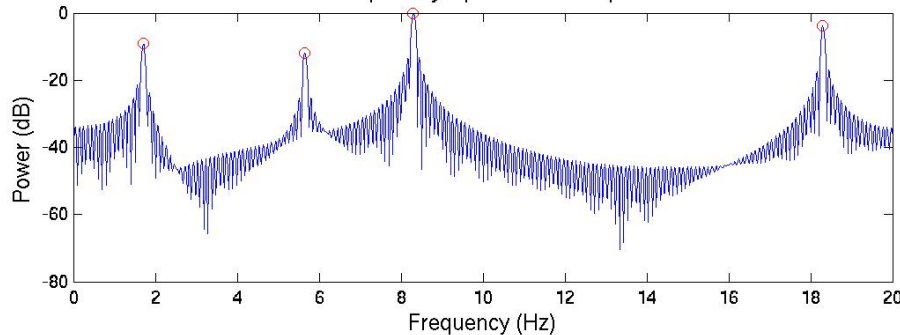


# Compressive Sampling

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



Frequency spectrum of input

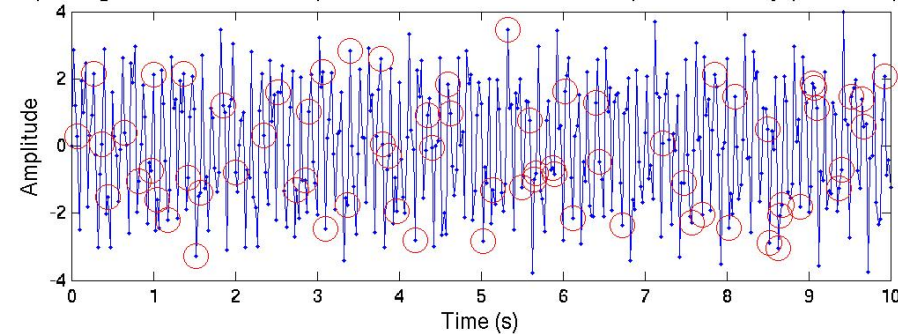


- Sense signal  $M$  times
- Recover with linear program

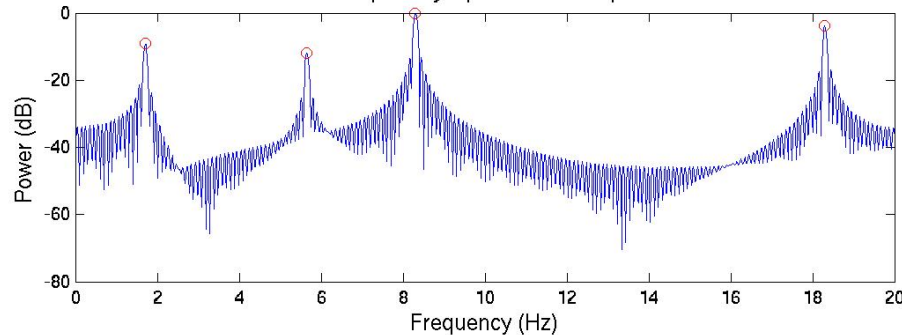
$$\min_{\omega} \sum |\hat{g}(\omega)| \quad \text{subject to} \quad g(t_m) = f(t_m), \quad m = 1, \dots, M$$

# Compressive Sampling

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



Frequency spectrum of input

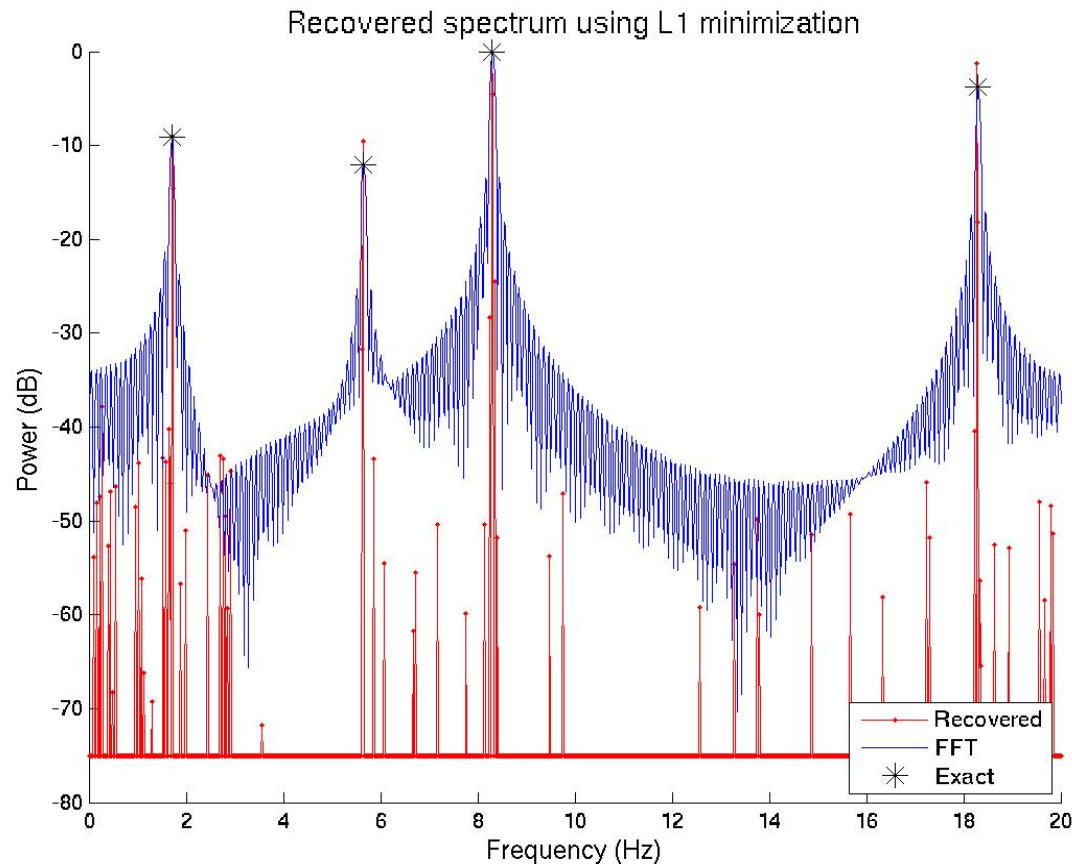


$$\hat{f}(\omega) = \sum_{i=1}^K \alpha_i \delta(\omega_i - \omega) \xleftrightarrow{\mathcal{F}} f(t) = \sum_{i=1}^K \alpha_i e^{i\omega_i t}$$

- Sense signal  $M$  times
- Recover with linear program

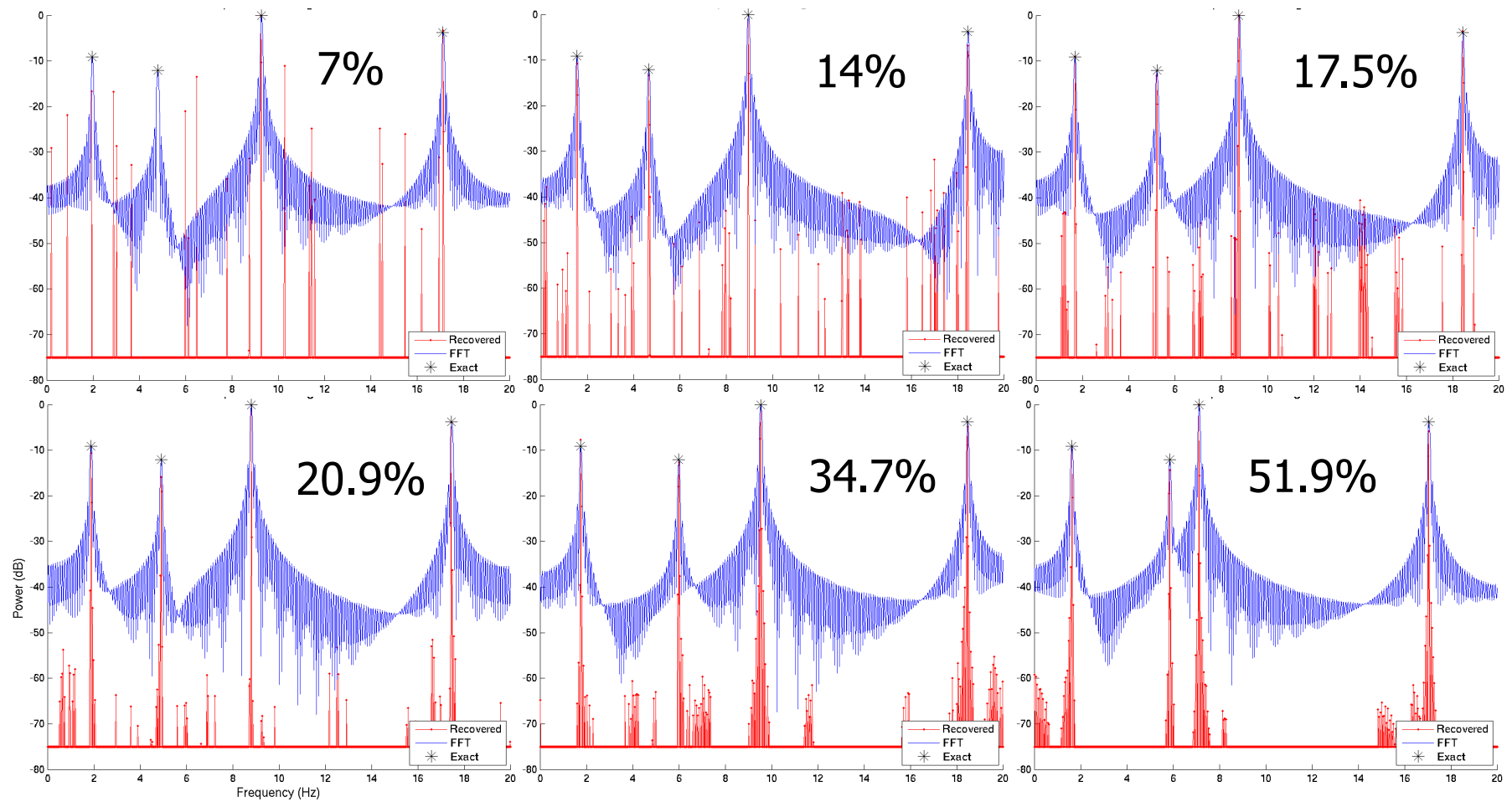
$$\min_{\omega} \sum |\hat{g}(\omega)| \quad \text{subject to} \quad g(t_m) = f(t_m), \quad m = 1, \dots, M$$

# Example: Sum of Sinusoids

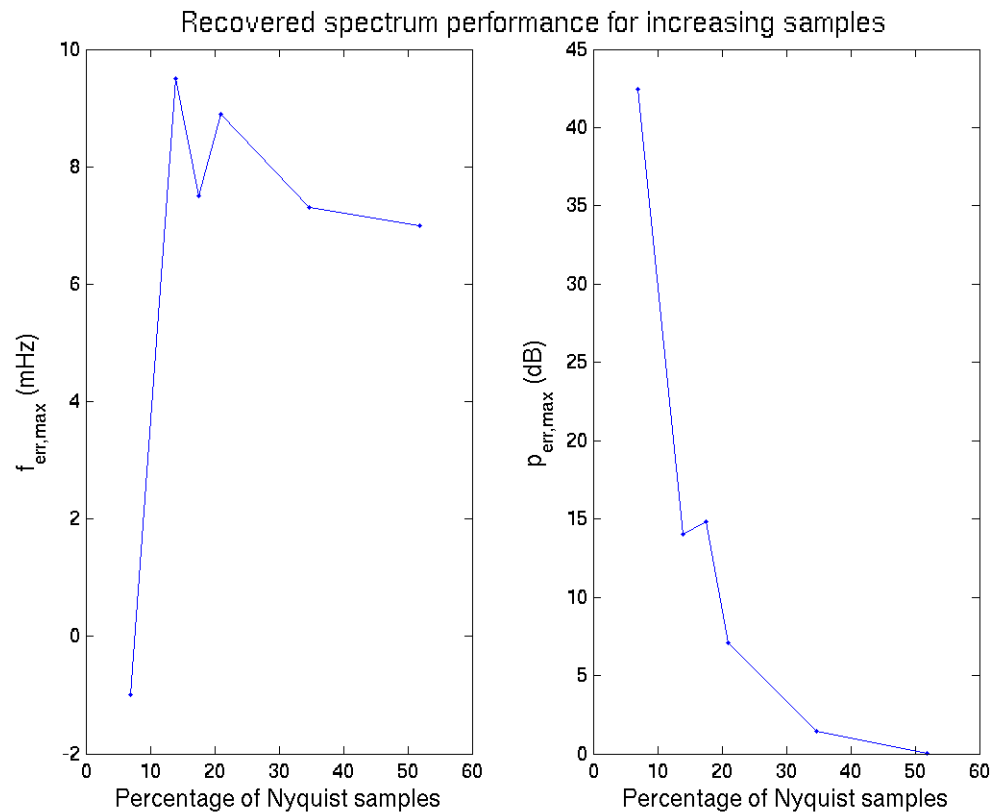


- Two relevant “knobs”
  - percentage of Nyquist samples as altered by adjusting number of samples,  $M$
  - input signal duration,  $T$ 
    - Data block size

# Example: Increasing M

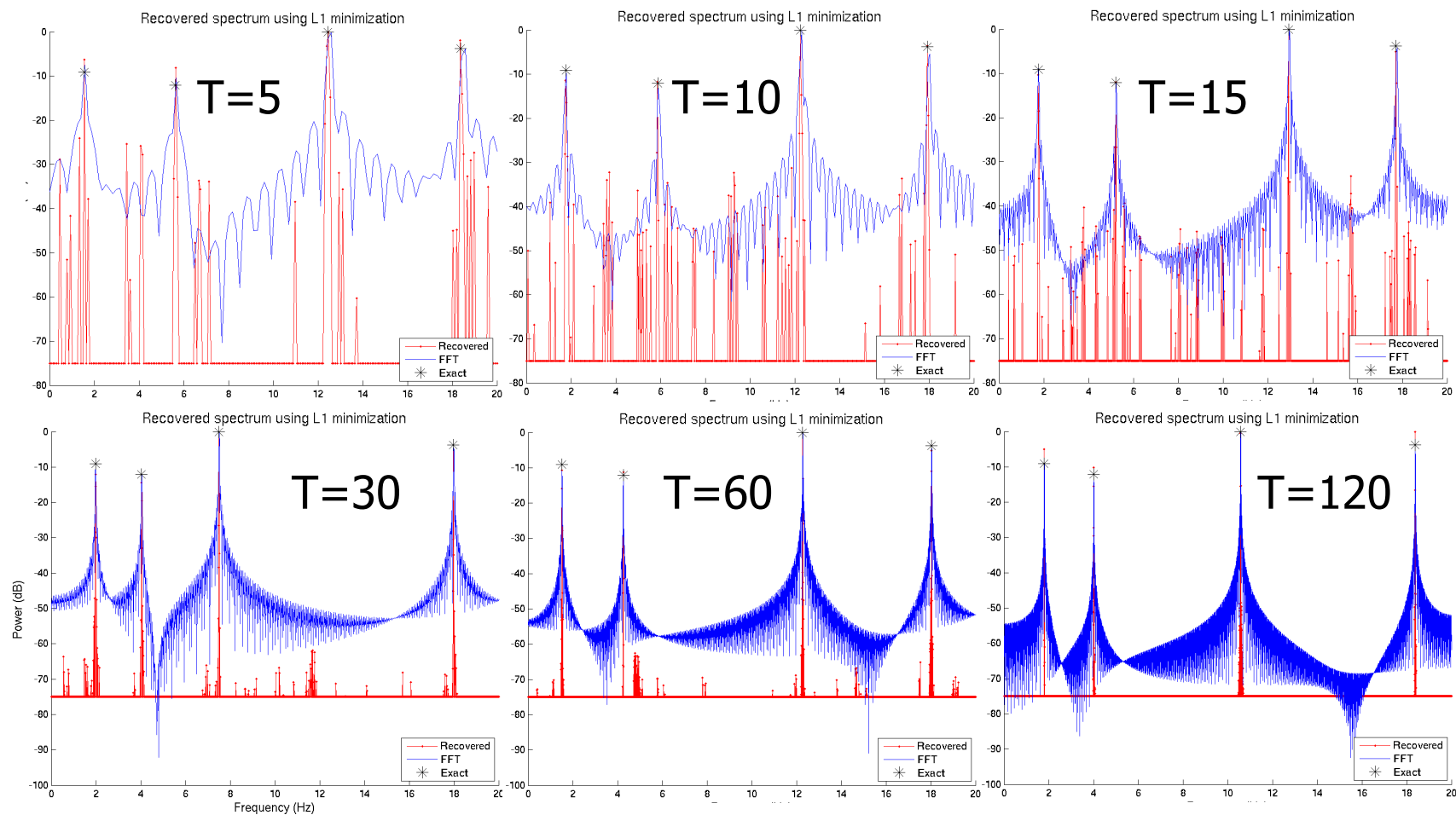


# Example: Increasing M

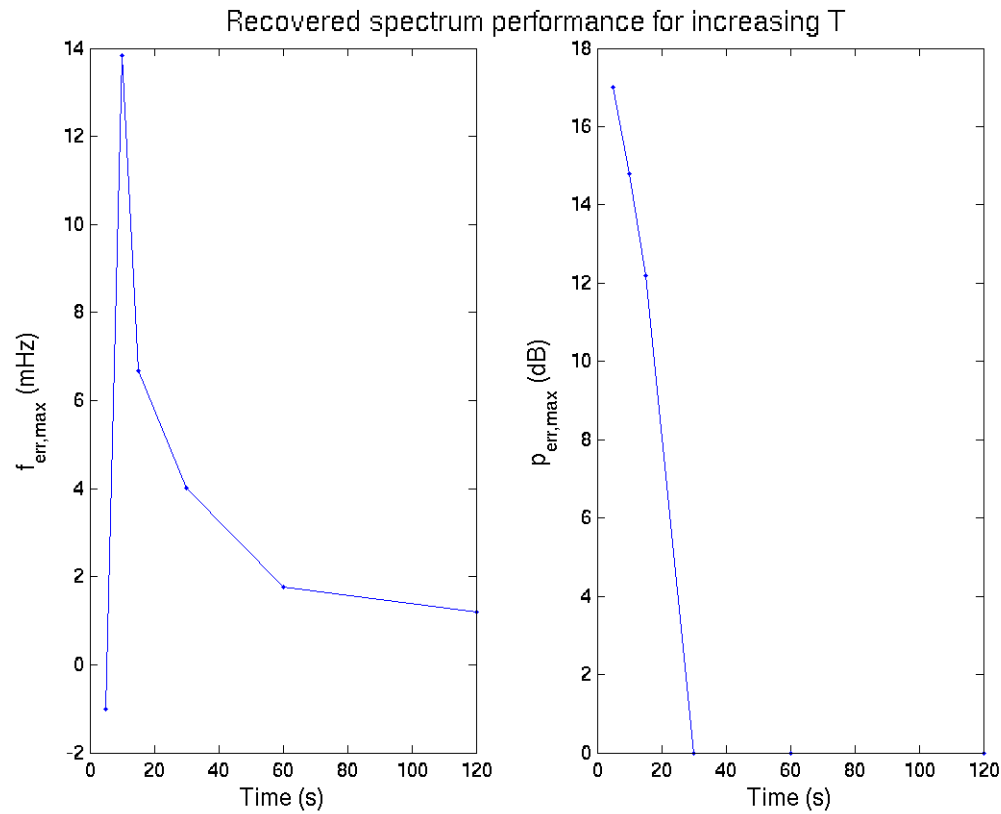


- $f_{err,max}$  within 10 mHz
- $p_{err,max}$  decreasing

# Example: Increasing $T$



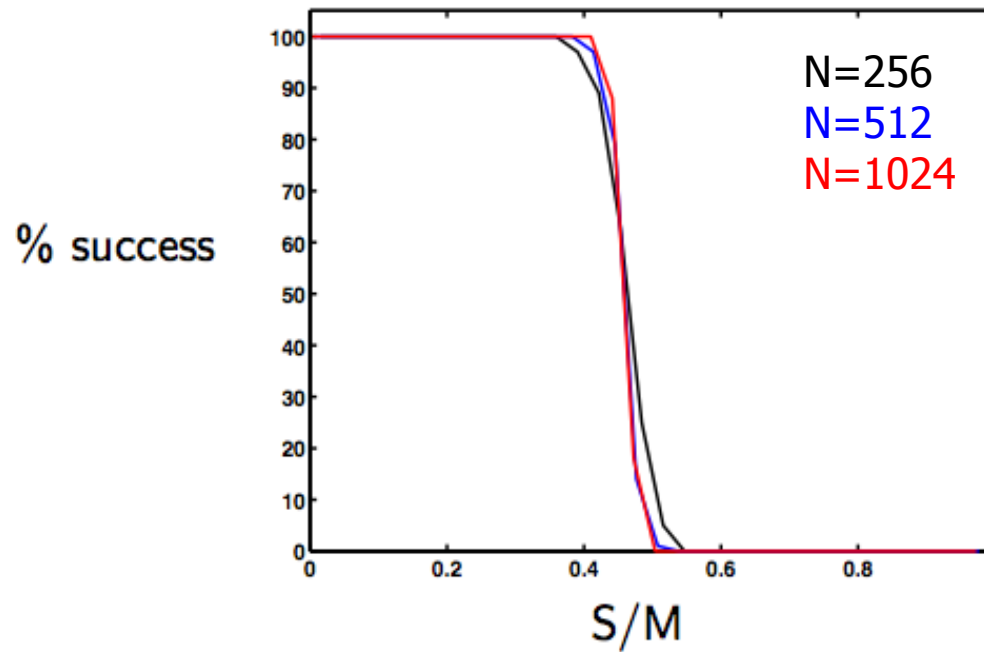
# Example: Increasing T



- **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

# 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)

$$y_m = x_0(t_m)$$

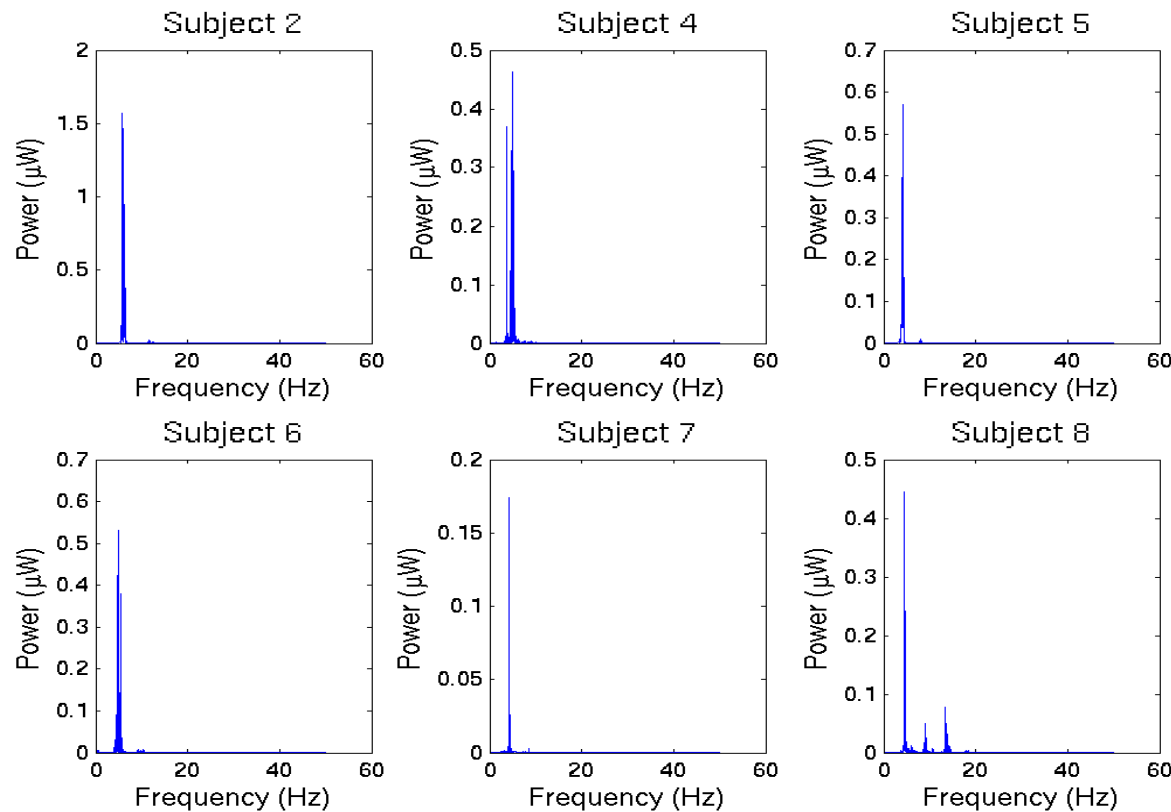
- Solve

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

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

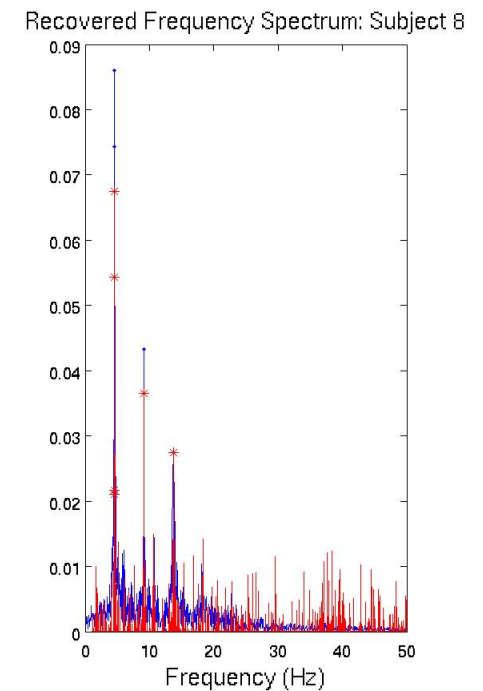
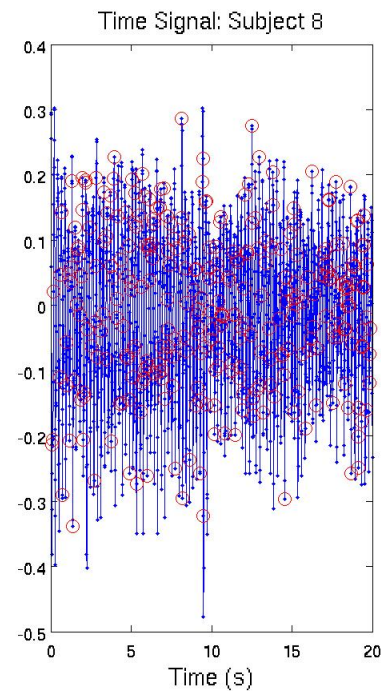
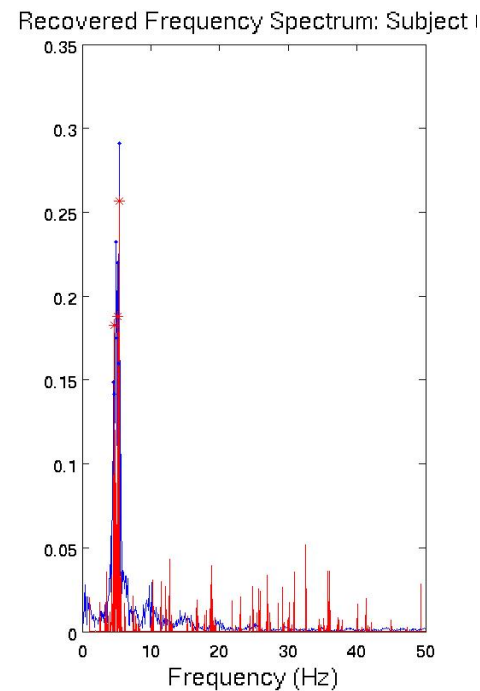
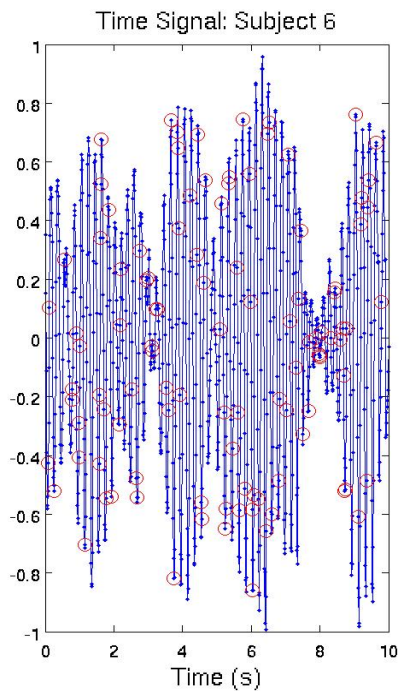
$$M > C \cdot \mu^2(\Phi, \Psi) \cdot S \cdot \log N$$

# Biometric Example: Parkinson's Tremors

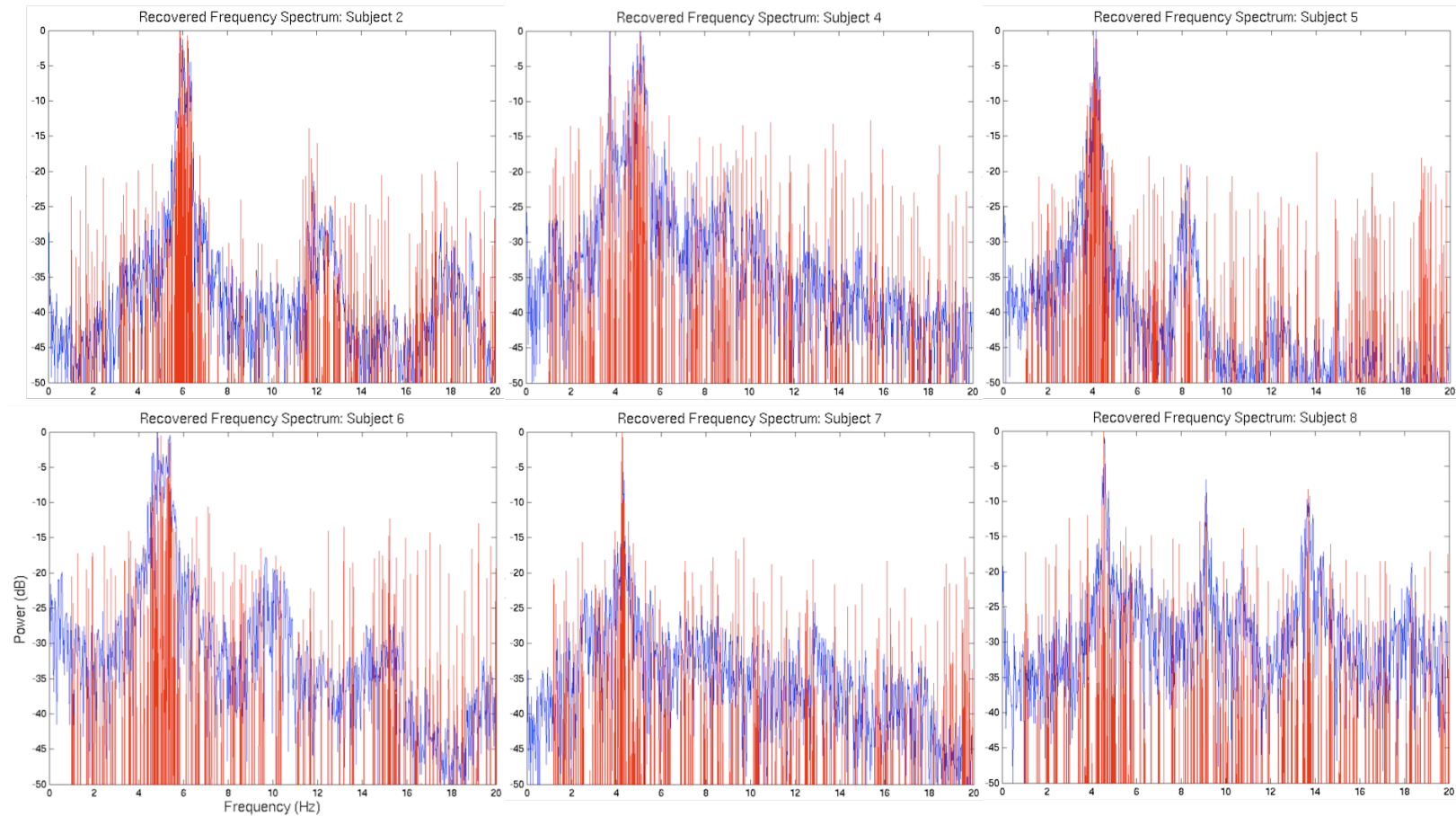


- 6 Subjects of real tremor data
  - collected using low intensity velocity-transducing laser recording aimed at reflective tape attached to the subjects' finger recording the finger velocity
  - All show Parkinson's tremor in the 4-6 Hz range.
  - Subject 8 shows activity at two higher frequencies
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

# Compressive Sampling: Real Data



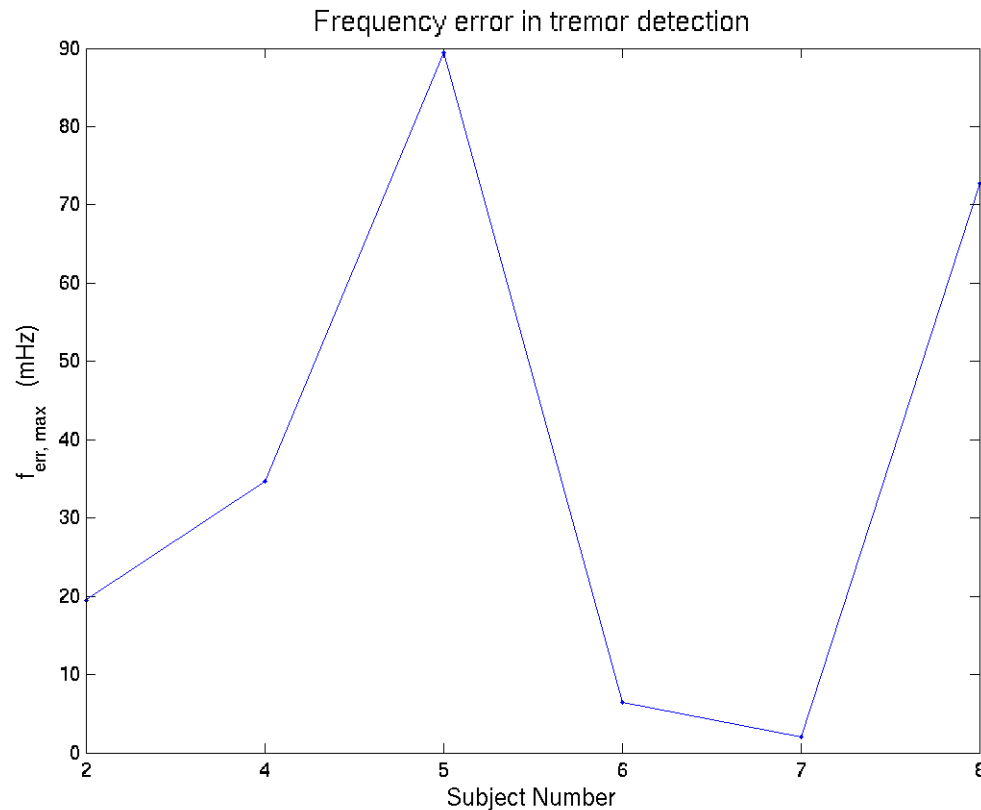
# Biometric Example: Parkinson's Tremors



■ **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

---

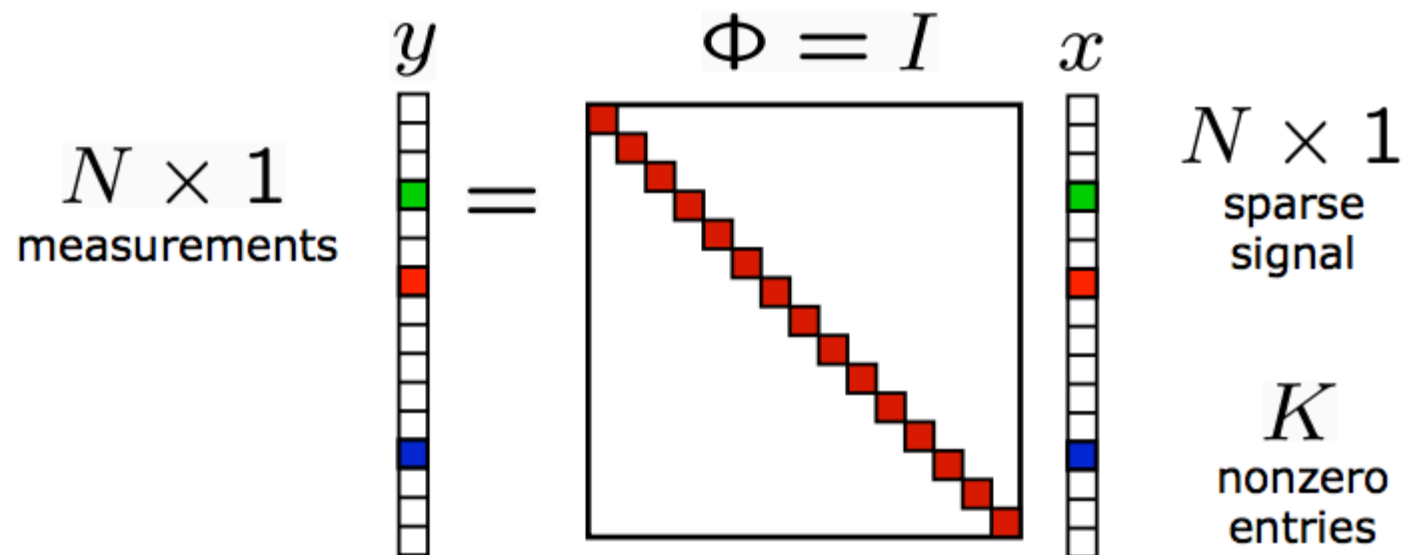
Why does it work?



# Sampling

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

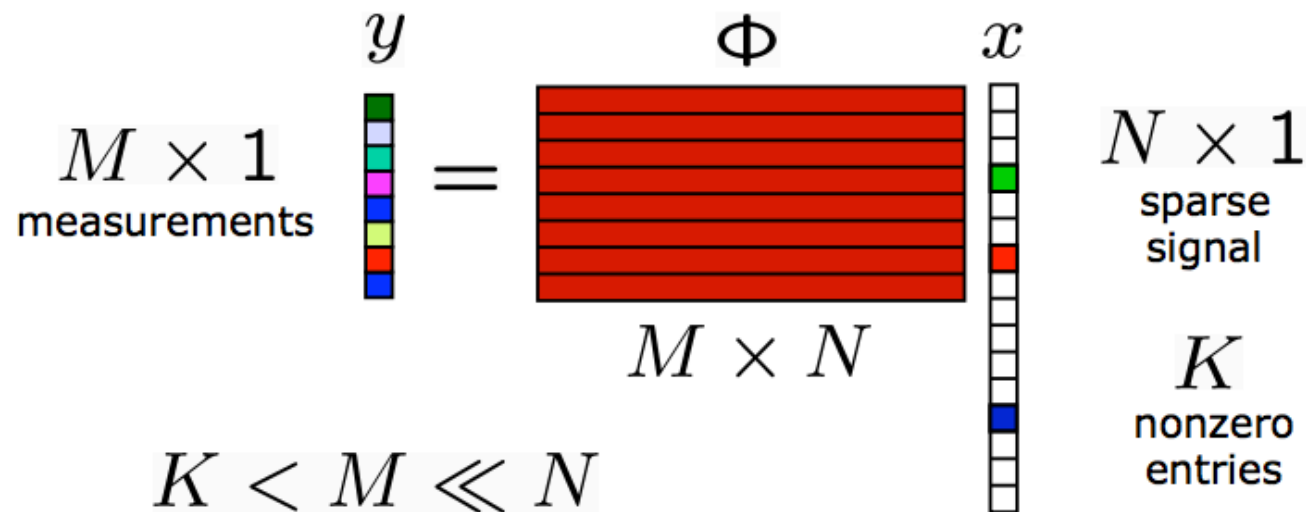
- Sampling**



# Compressive Sampling

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

$$y = \Phi x$$

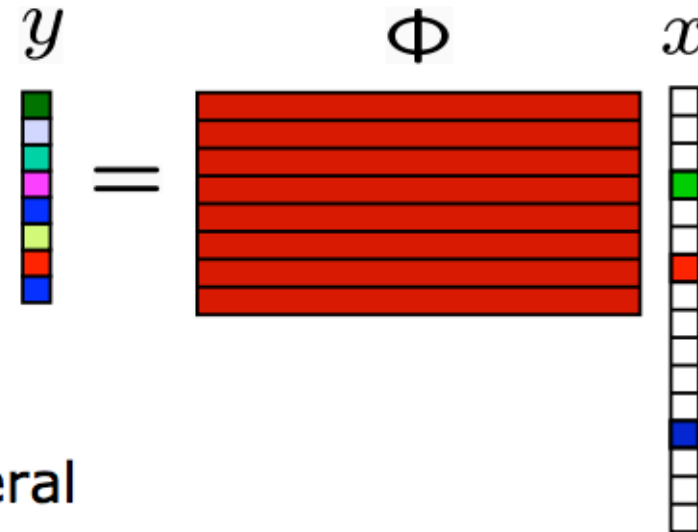


# How Can It Work?

- Projection  $\Phi$   
**not full rank...**

$$M < N$$

... and so  
**loses information** in general



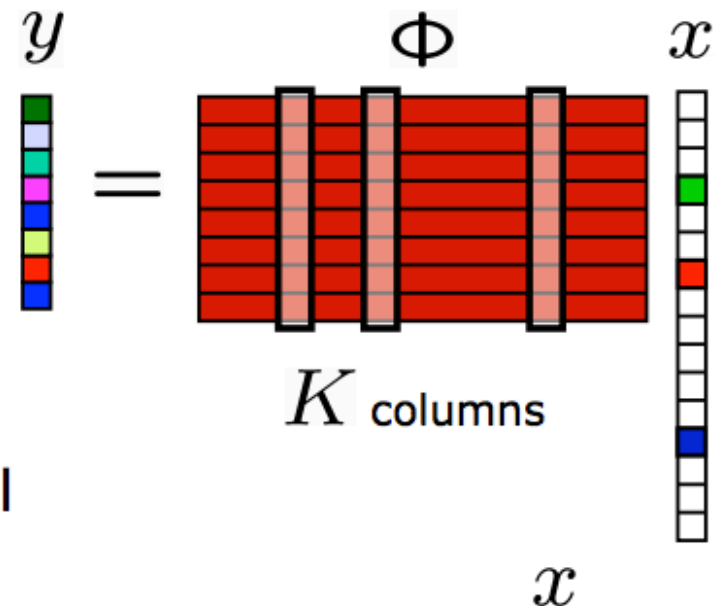
- Ex: Infinitely many  $x$ 's map to the same  $y$   
(null space)

# How Can It Work?

- Projection  $\Phi$   
not full rank...

$$M < N$$

... and so  
loses information in general



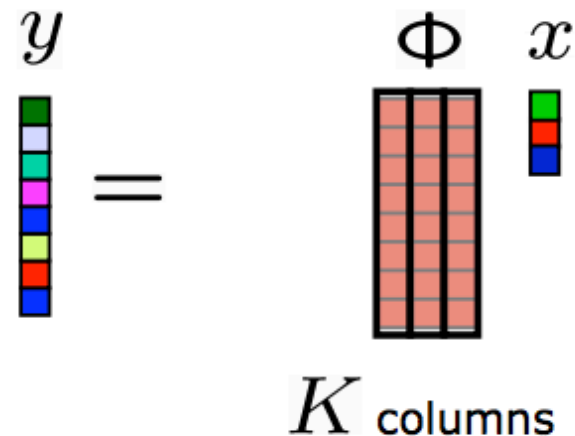
- But we are only interested in **sparse** vectors

# How Can It Work?

- Projection  $\Phi$  not full rank...

$$M < N$$

... and so  
loses information in general



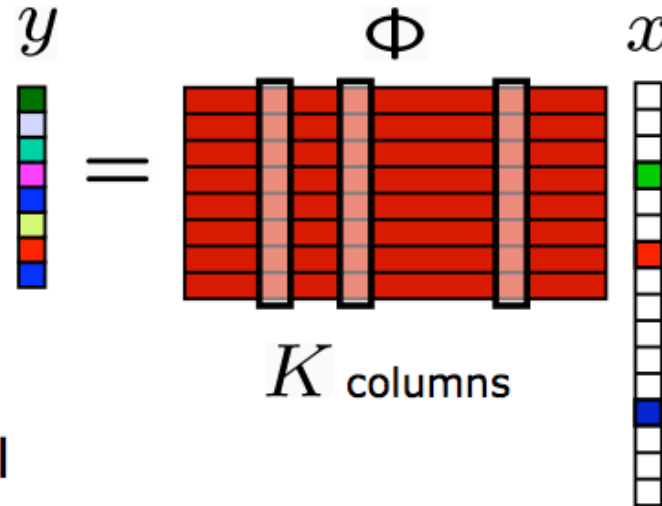
- But we are only interested in **sparse** vectors
- $\Phi$  is effectively  $M \times K$

# How Can It Work?

- Projection  $\Phi$  not full rank...

$$M < N$$

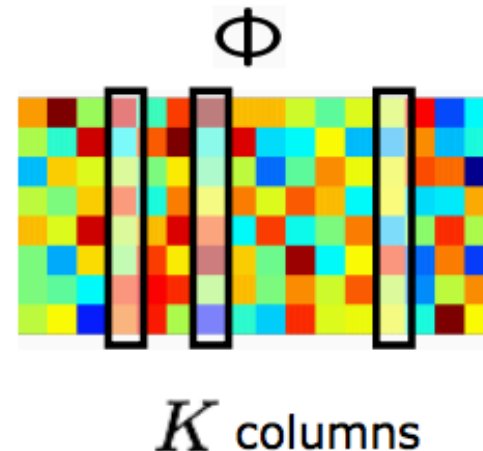
... and so  
loses information in general



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

# RIP

- Draw  $\Phi$  at **random**
  - iid Gaussian
  - iid Bernoulli  $\pm 1$
  - ...

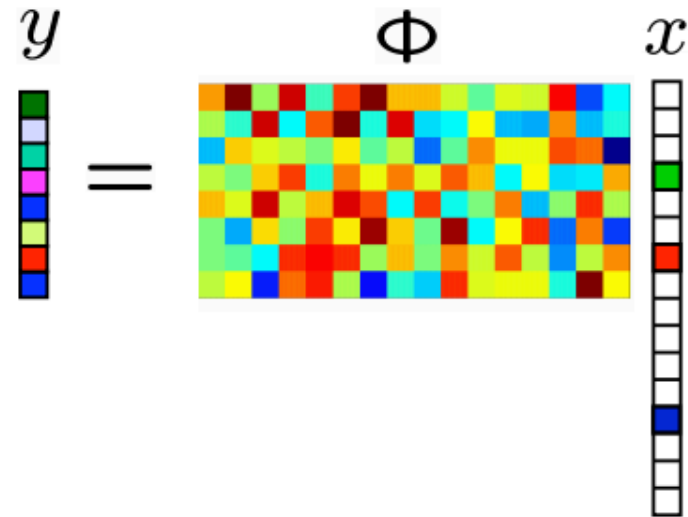


- Then  $\Phi$  has the RIP with high probability provided

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

# CS Signal Recovery

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

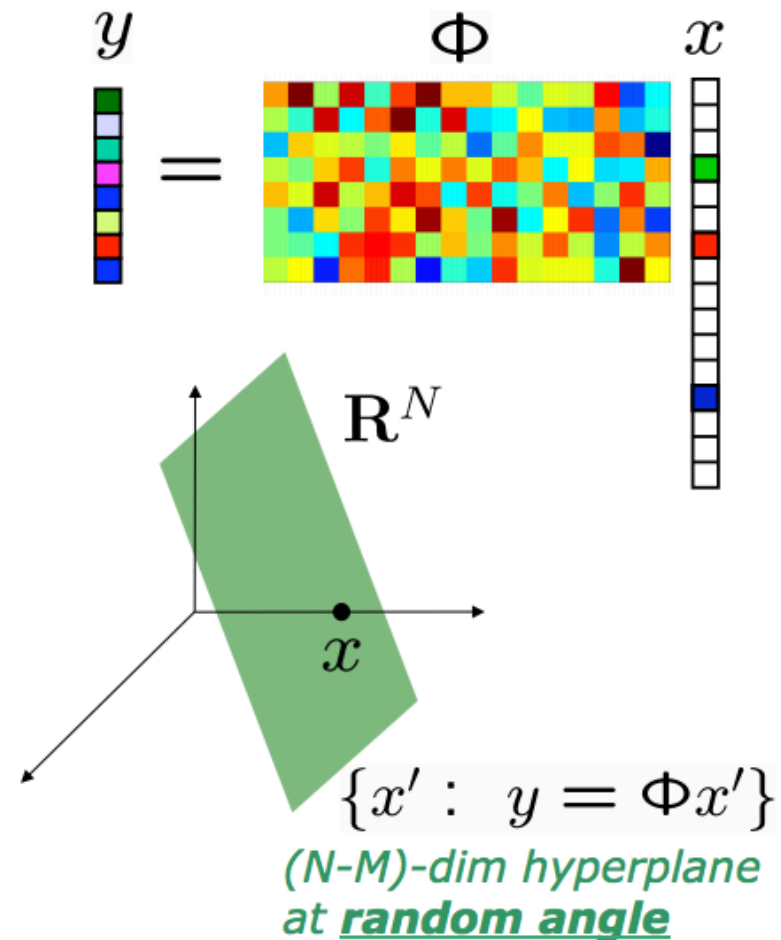
$$y = \Phi x$$


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



# CS Signal Recovery

- Random projection  $\Phi$  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



# L<sub>2</sub> Signal Recovery

- Recovery:  
(ill-posed inverse problem)

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

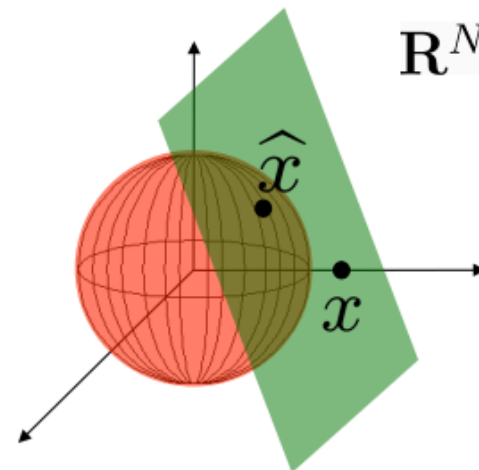
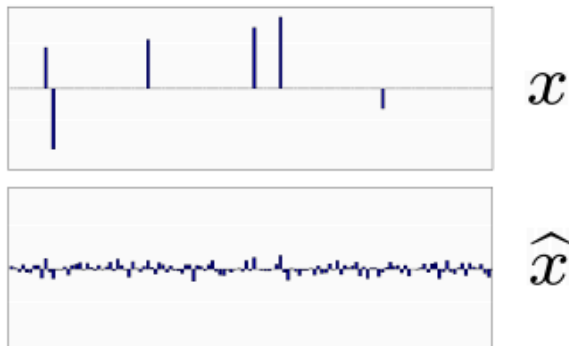
- Optimization:

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

- Closed-form solution:

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

- **Wrong answer!**



# $L_0$ Signal Recovery

- Recovery:  
(ill-posed inverse problem)

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

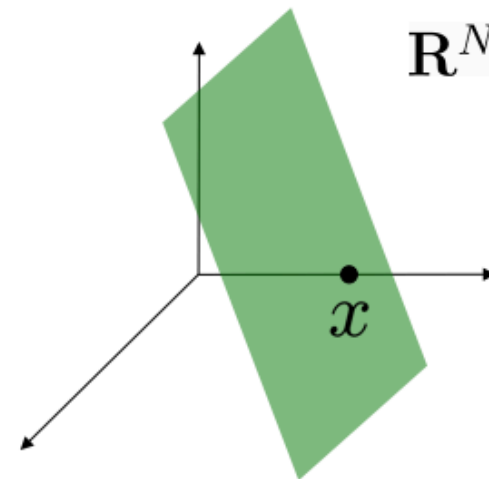
- Optimization:

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

- **Correct!**



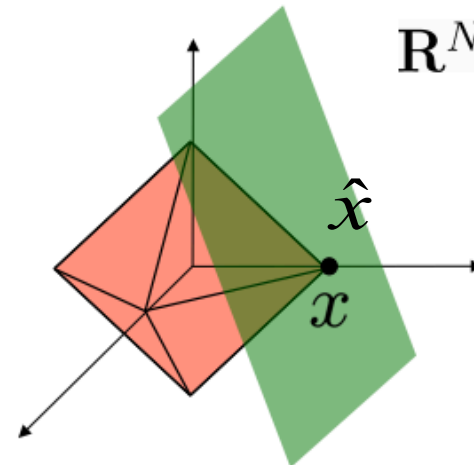
*“find **sparsest** vector  
in translated nullspace”*



- But **NP-Complete** alg

# $L_1$ Signal Recovery

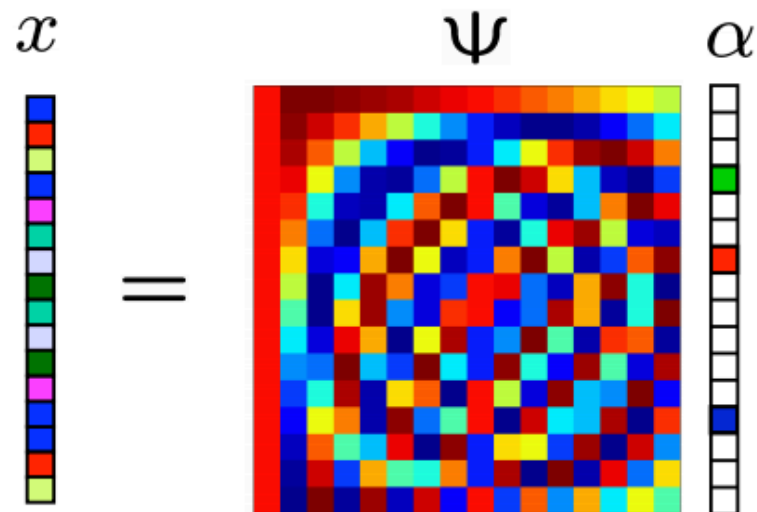
- Recovery:  
(ill-posed inverse problem)      given  $y = \Phi x$   
find  $x$  (sparse)
- Optimization:  $\hat{x} = \arg \min_{y=\Phi x} \|x\|_1$
- **Convexify** the  $\ell_0$  optimization
- **Correct!**
- **Polynomial time** alg  
(linear programming)
- Much recent alg progress
  - greedy, Bayesian approaches, ...



# Universality

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

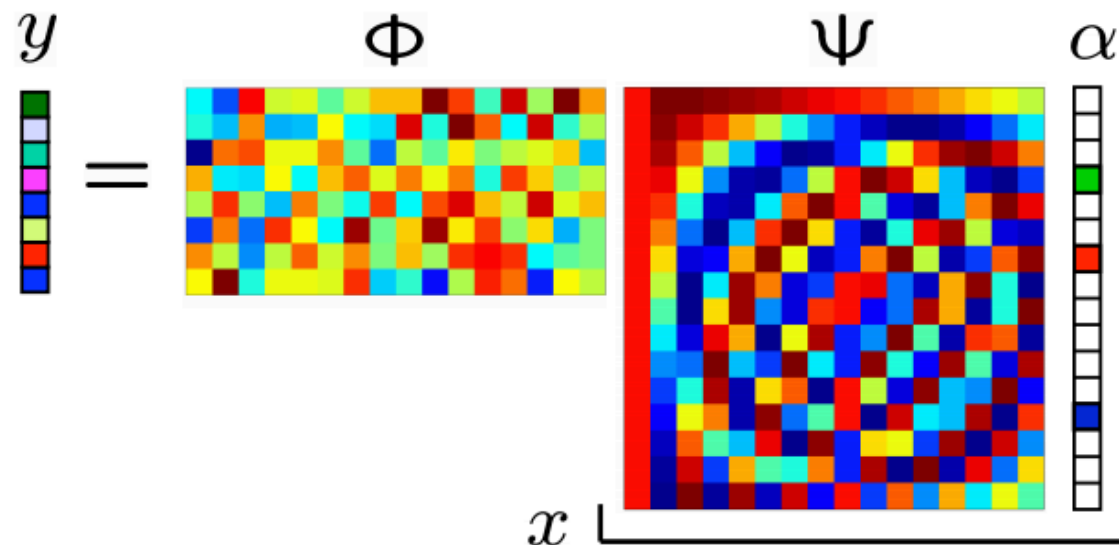
$$x = \Psi \alpha$$



# Universality

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

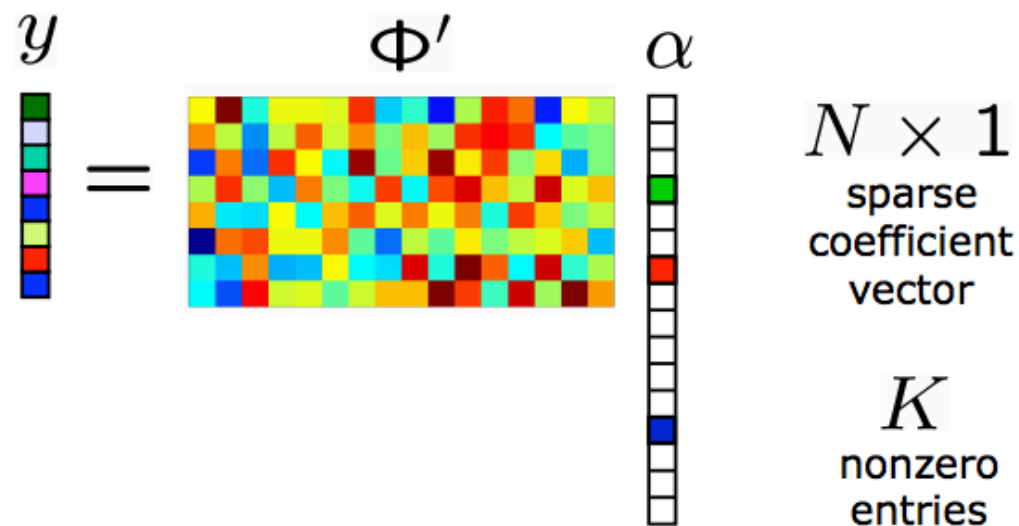
$$y = \Phi x = \Phi \Psi \alpha$$



# Universality

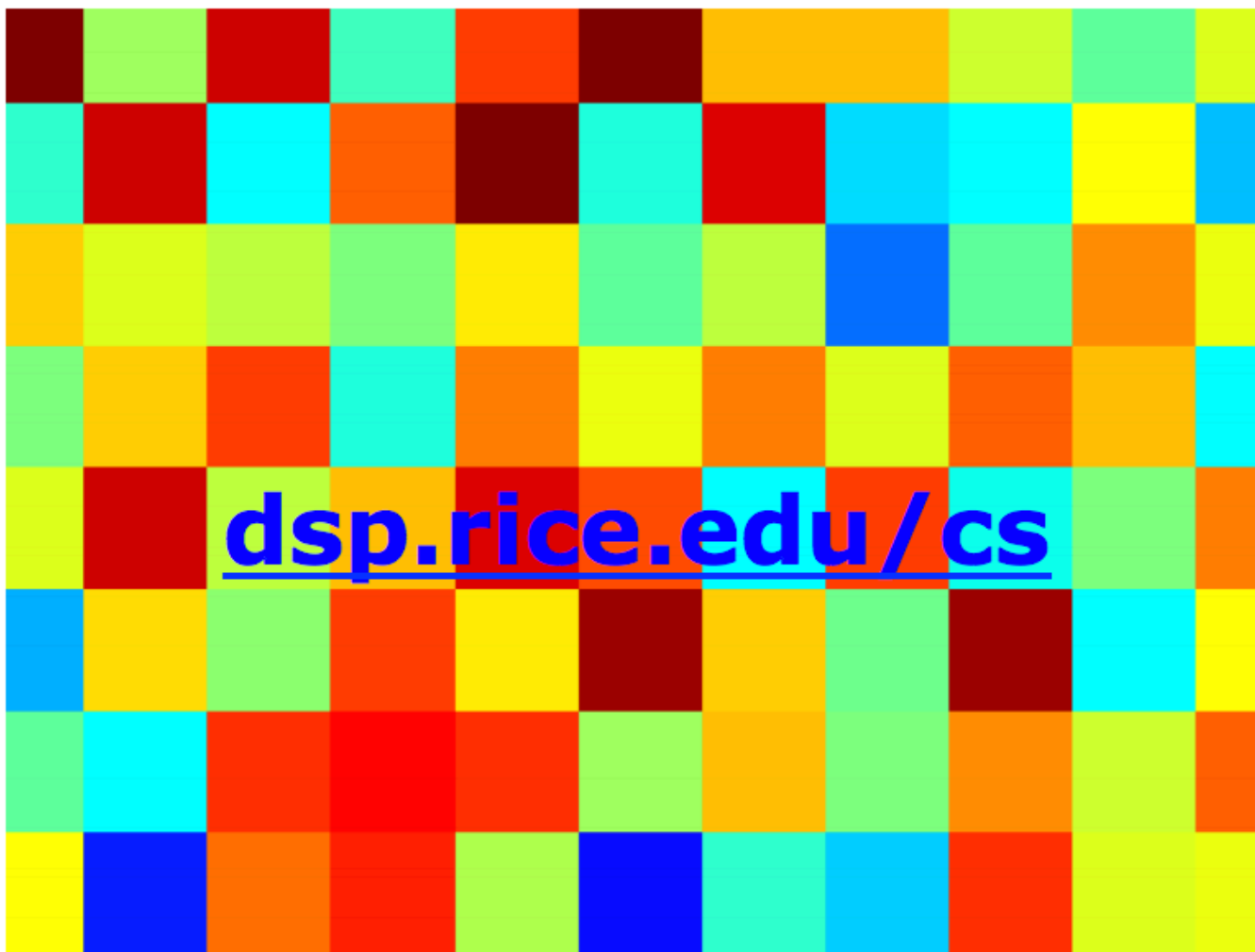
- Random measurements can be used for signals sparse in *any* basis

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



# Reference Slide

---







# Big Ideas

---

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



# Admin

---

- ❑ Final Project due – Apr 30<sup>th</sup>
  - TA advice – “The report takes time. Leave time for it.”
  - No late accepted. Turn into Canvas on time.
- ❑ Last day of TA office hours – Apr 30<sup>th</sup>
  - Piazza still available
- ❑ Last day of Tania office hours – May 8<sup>th</sup>
- ❑ Final Exam Review Session – May 10<sup>th</sup> (time TBD)
  - Watch Piazza for details
- ❑ Final Exam – May 13<sup>th</sup>



# Final Exam Admin

---

- ❑ Final – 5/13
  - Location Levine 101
  - Starts at exactly 3:00pm, ends at exactly 5:00pm (120 minutes)
  - Cumulative – covers entire course
    - Except data converters, noise shaping (lec 12), adaptive filters (lec 23), wavelet transform (lec 25), and compressive sampling (lec 26)
  - Closed book
    - Data/Equation sheet provided by me
    - 2 8.5x11 two-sided cheat sheets allowed
    - Calculators allowed, no smart phones
  - Old exams posted
  - TA Review session on 5/10, Time and Place TBD
    - Watch Piazza for details