

# Intelligent Signal Processing

## Streaming and Compressive Sensing

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# Introduction

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- Compressive Sensing (o Compressed Sensing) technique
- client-server architecture
- Compressive Sensing for
  - compression
  - packet loss reconstruction



# Compressive Sensing

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- Compressive Sensing (CS)
  - is a new sensing modality, which compresses the signal being acquired at the time of sensing
  - Signals can have sparse or compressible representation either in original domain or in some transform domain
  - Relying on the sparsity of the signals, CS allows us to sample the signal at a rate much below the Nyquist sampling rate
  - the varied reconstruction algorithms of CS can faithfully reconstruct the original signal back from fewer compressive measurements



# Compressive Sensing

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- CS was introduced by Donoho, Candès, Romberg, and Tao in 2004



# Compressive Sensing

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- Emerging technique for **signal processing**
  - acquisition/reconstruction that violates the Nyquist-Shannon limit
    - less samples
- A signal can have **sparse/compressible representation** either in **original domain** or in **some transform domains**
  - Fourier transform, cosine transform, wavelet transform, etc. A few examples of signals having sparse
- **Domains**
  - **natural images** which have sparse representation in wavelet domain
  - **speech signal** can be represented by fewer components using Fourier transform
  - better model for **medical images** can be obtained using Radon transform
  - etc.



# Linear inverse problems

- Many classic problems in computer can be posed as **linear inverse problems**

- **Notation**

- Signal of interest

$$x \in \mathbb{R}^N$$

- Observations

$$y \in \mathbb{R}^M$$

- Measurement model

$$y = \Phi x + e$$

measurement  
matrix

measurement  
noise

- Problem definition: given  $y$ , recover  $x$



# Linear inverse problems

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## ■ Scenario 1

$$M \geq N$$

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

## ■ Scenario 2

$$M < N$$

- Measurement matrix has a  $(N-M)$  dimensional null-space
- Solution is no longer unique
- Under-sampling ratio  $M/N$



# Image super-resolution

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**Low resolution  
input/observation**

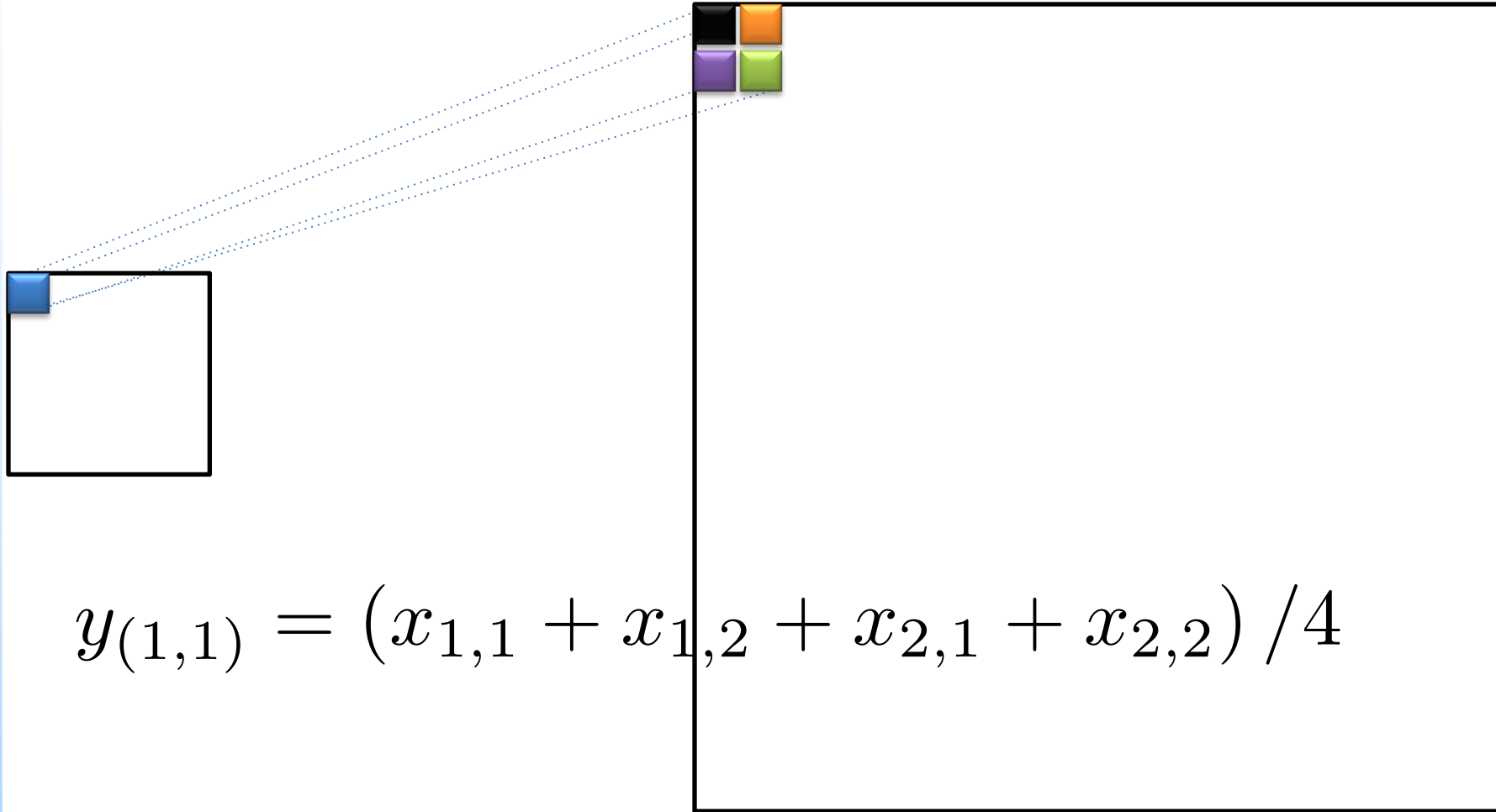
128x128 pixels





# Image super-resolution

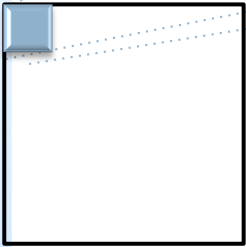
2x super-resolution



$$y_{(1,1)} = (x_{1,1} + x_{1,2} + x_{2,1} + x_{2,2}) / 4$$



# Image super-resolution



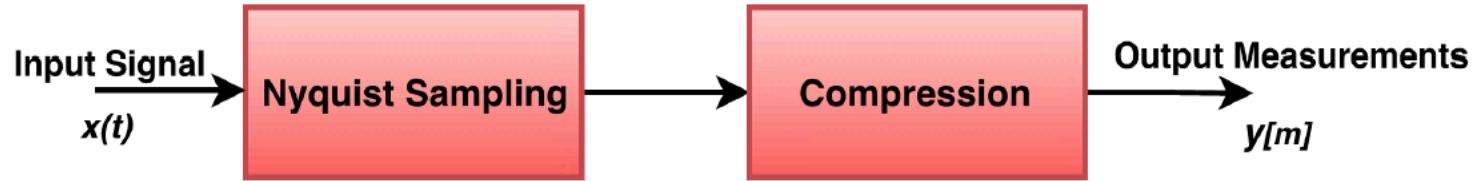
$$y_{(1,1)} = (x_{1,1} + x_{1,2} + \dots + x_{4,3} + x_{4,4}) / 16$$

Super-resolution factor  $D$

**Under-sampling factor**  $M/N = 1/D^2$



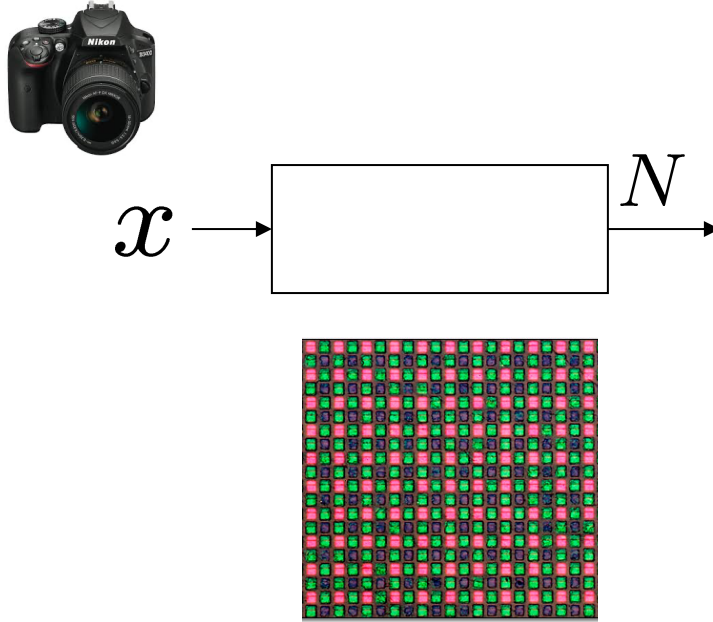
# Compressive Sensing



A comparison of sampling techniques: (a) traditional sampling, (b) compressive sensing.



# Sampling



**too  
much  
data!**



# Compression



$x$



$N$



$K$

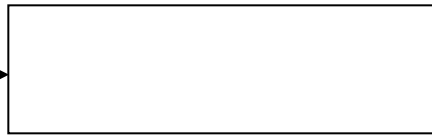
$N \gg K$



**JPEG**  
**JPEG2000**

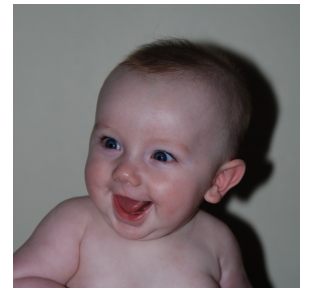
...

$K$



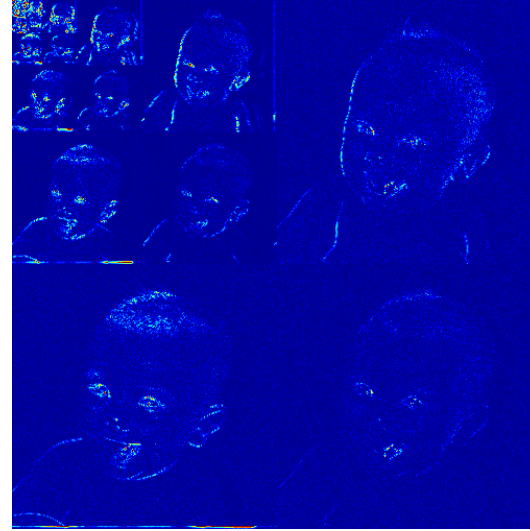
$N$

$\hat{x}$



# Sparsity

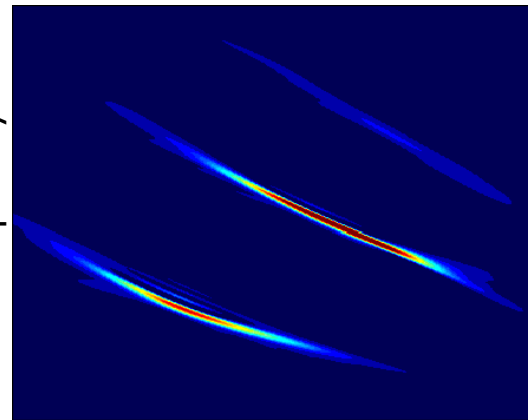
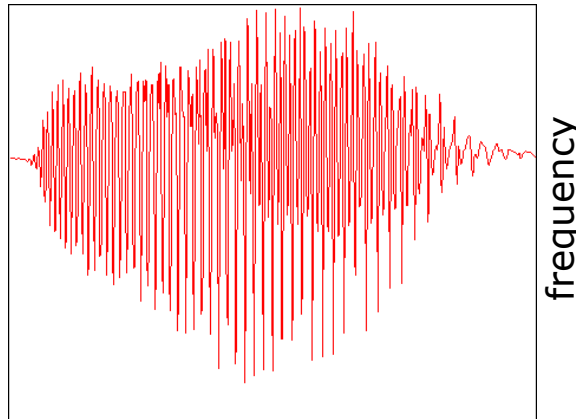
$N$   
pixels



$K \ll N$   
large  
wavelet  
coefficients

(blue = 0)

$N$   
wideband  
signal  
samples



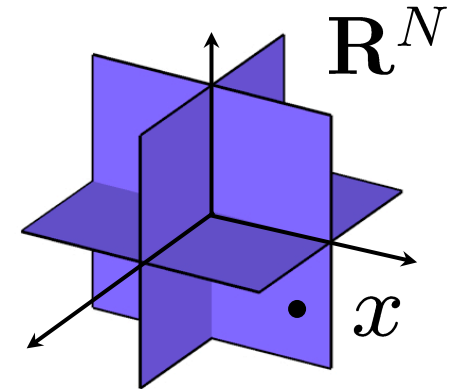
$K \ll N$   
large  
Gabor (TF)  
coefficients

time



# Sparsity

- Sparse signal
  - only  $K$  out of  $N$  coordinates nonzero
  - Model – union of  $k$ -dimensional subspaces

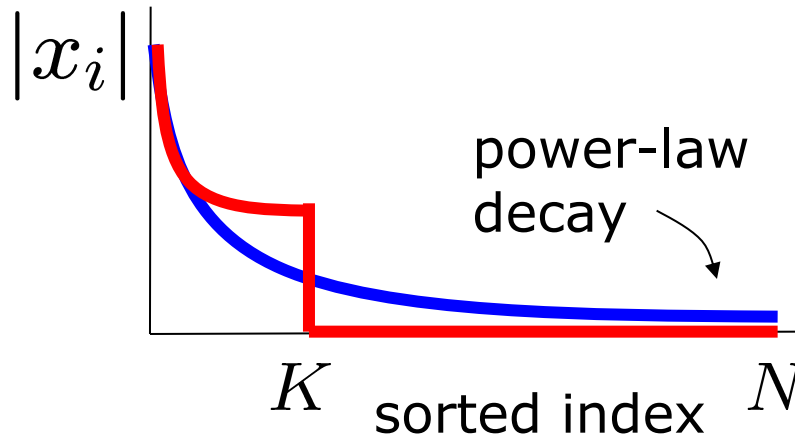
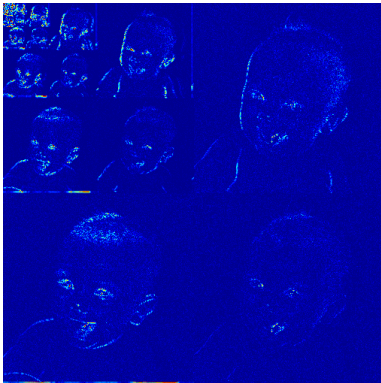
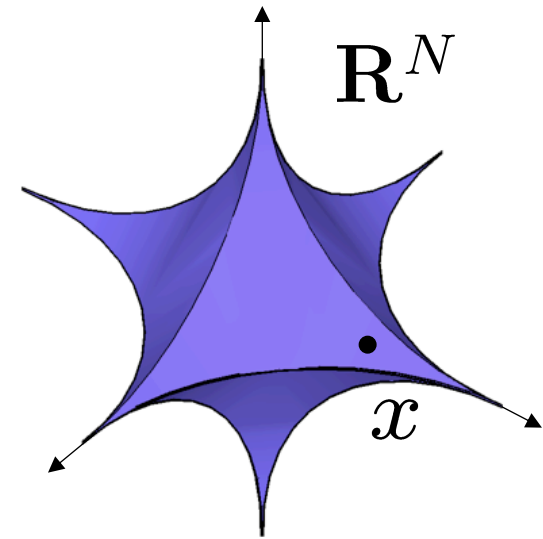


# Sparsity

## ■ Compressible signal

- sorted coordinates decay rapidly with power-law
- Model based on  $\ell_p$

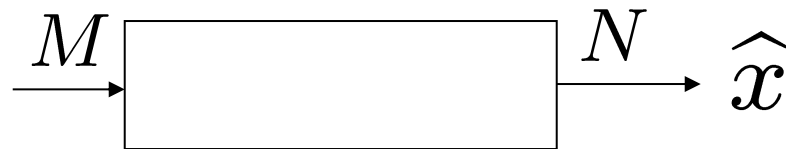
$$\|x\|_p^p = \sum_i |x_i|^p \leq 1, \quad p \leq 1$$



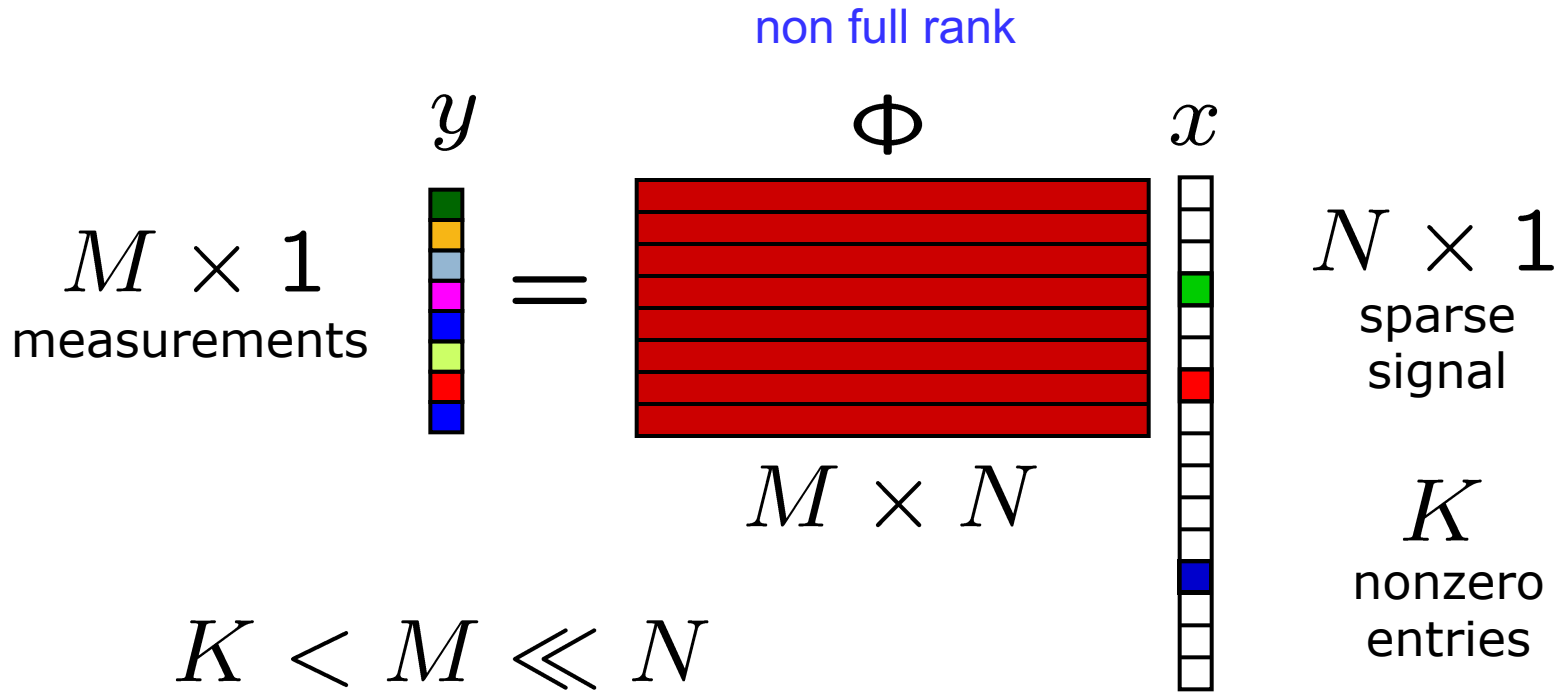


# Compressive sensing

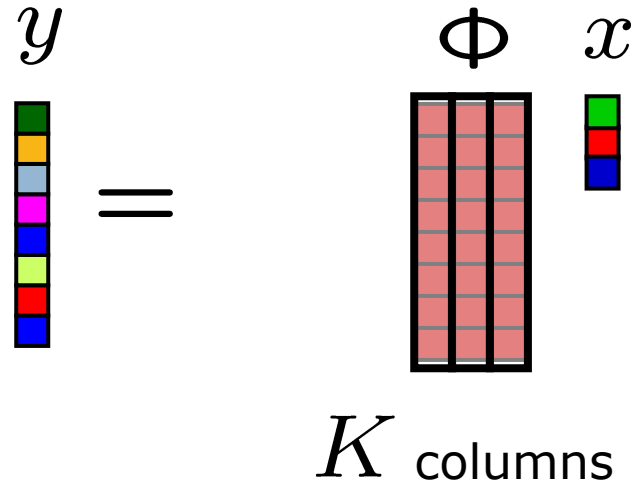
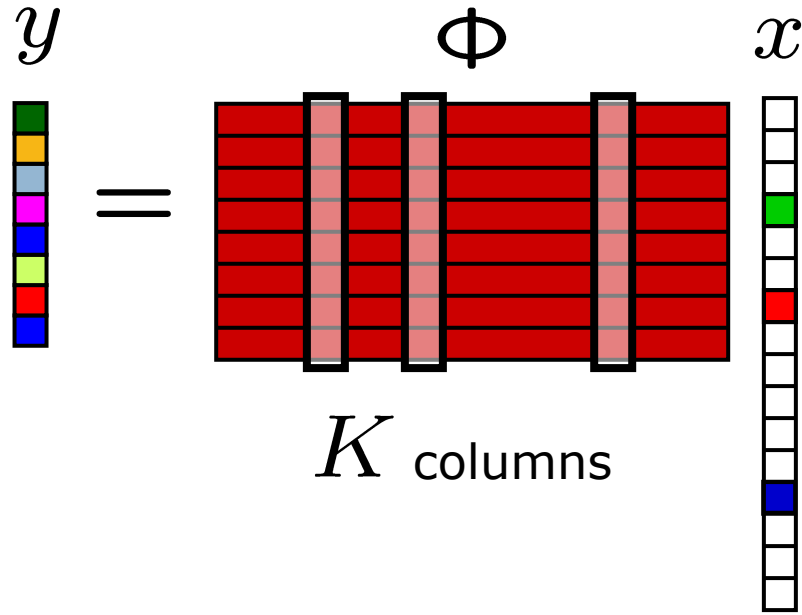
$$K \approx \underline{M} \ll N$$



# Compressive Sampling



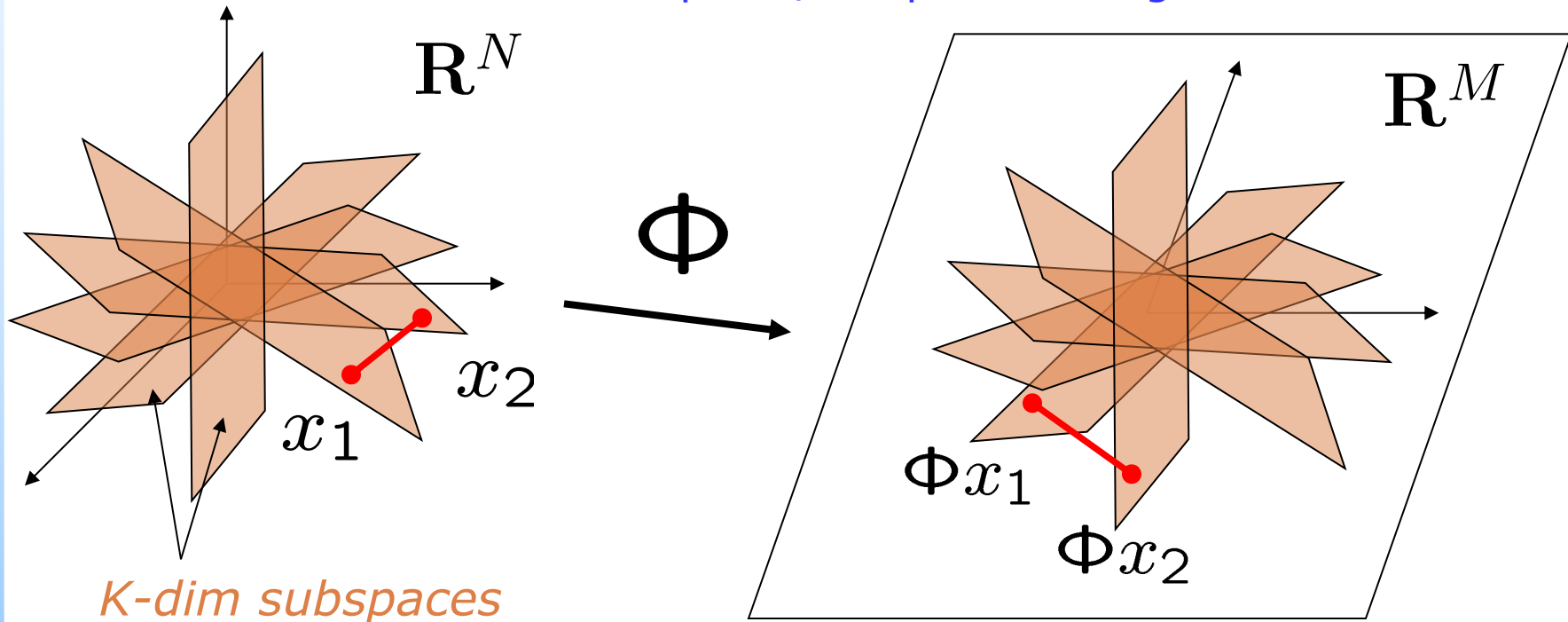
# How can it work?



# Restricted Isometry Property

- Design  $\Phi$  so that each of its  $M \times K$  submatrices are full rank (ideally close to orthobasis)
  - Restricted Isometry Property (RIP)

Preserve the structure of sparse/compressible signals



*K-dim subspaces*



# Restricted Isometry Property

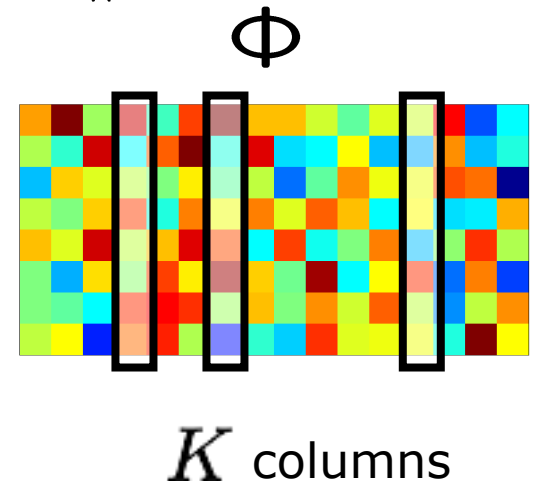
- RIP of order  $2K$  implies
  - for all  $K$ -sparse  $x_1$  and  $x_2$

$$(1 - \delta_{2K}) \leq \frac{\|\Phi x_1 - \Phi x_2\|_2^2}{\|x_1 - x_2\|_2^2} \leq (1 + \delta_{2K})$$

- Ensure that

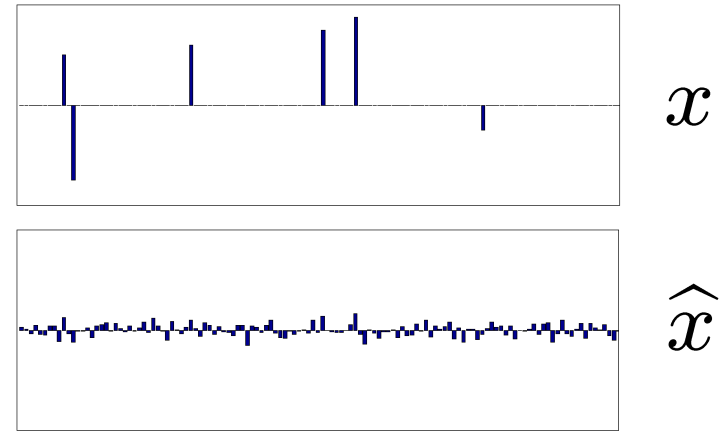
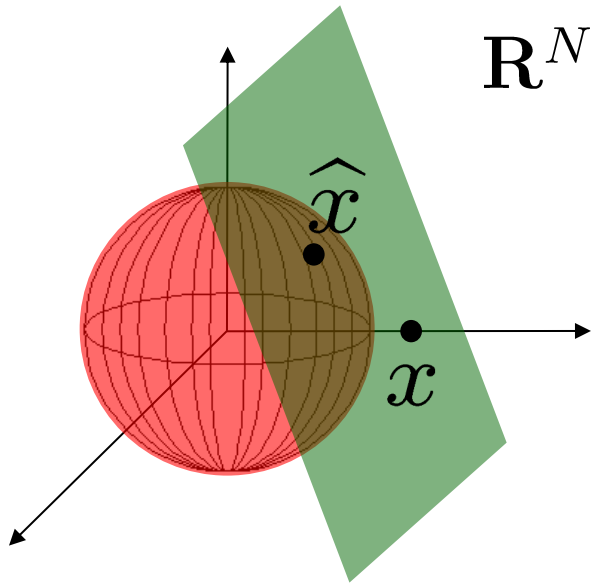
$$\|x_1 - x_2\|_2 \approx \|\Phi x_1 - \Phi x_2\|_2$$

- Draw  $\Phi$  at random
  - iid Gaussian
  - iid Bernoulli



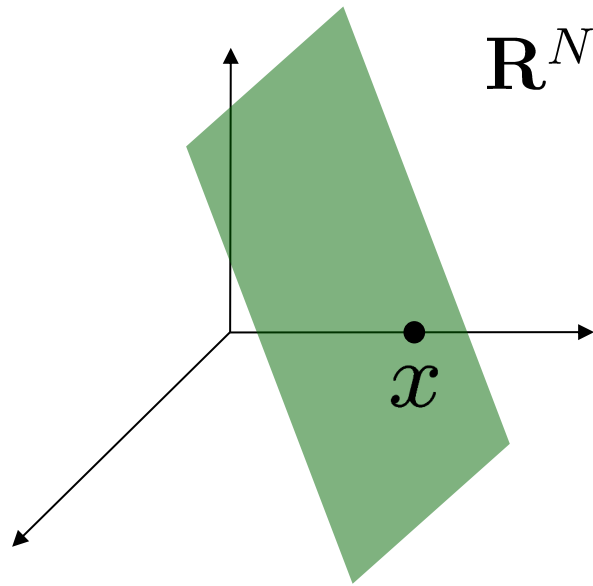
# $L_2$ signal recovery

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

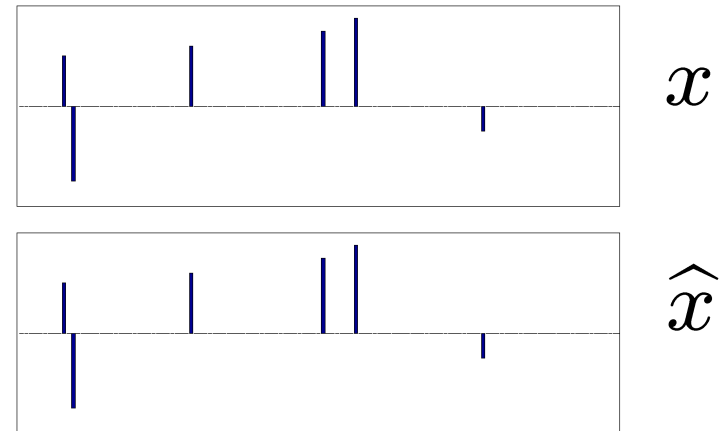


# $L_0$ signal recovery

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

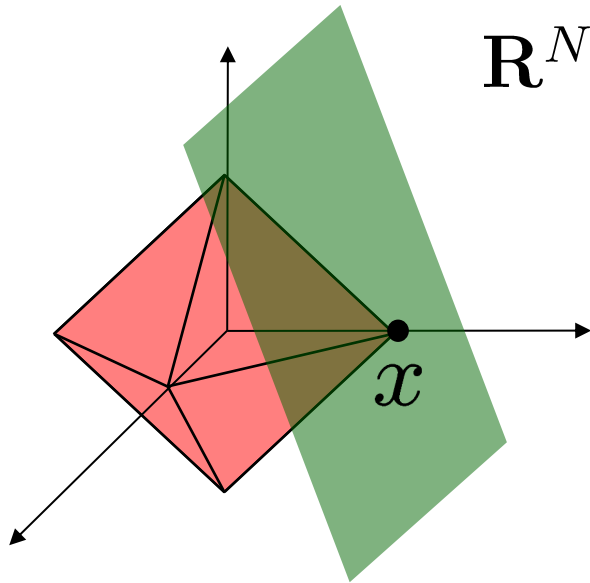


**NP-Complete**

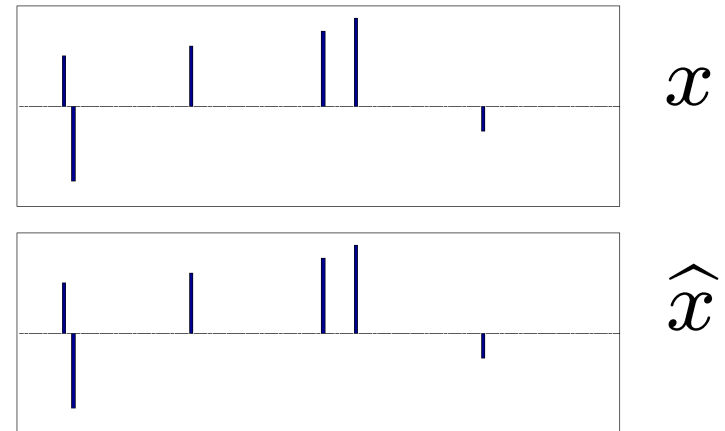


# $L_0$ signal recovery

$$\hat{x} = \arg \min_{y=\Phi x} \|x\|_1$$



**Polynomial time** alg  
(linear programming)





# Compressive Sensing

observed signal  $\longrightarrow$   $y = \Phi_s \mathbf{f} \in \mathbf{R}^m$

sensing matrix      source signal

$$\mathbf{f} = \sum_{i=1}^n x_i \psi_i = \Psi \mathbf{x}$$

orthonormal basis matrix

sparse representation

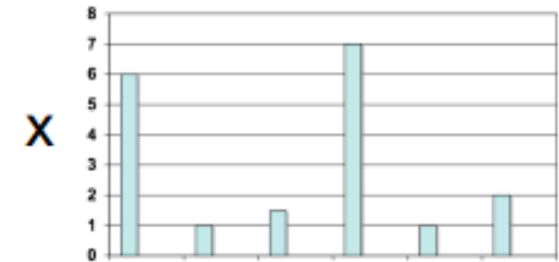
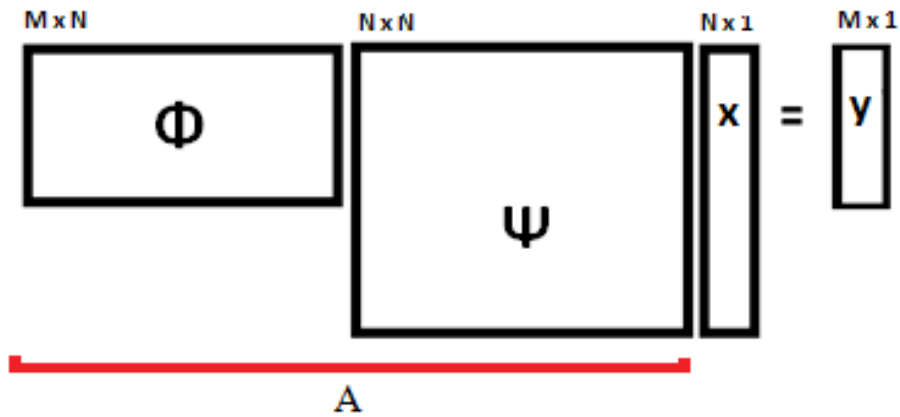
$$\mu(\Phi, \bar{\Psi})$$

coherence measure

In our case  $\phi$  is the identity matrix and  $\psi$  is a dictionary (learned or obtained by DCT)



# Compressive Sensing



$k=2$



# Optimization algorithm

$l_0$  -minimization  
problem is NP-hard

$$\mathbf{f}^* = \Psi \mathbf{x}^* \quad \text{reconstruction}$$

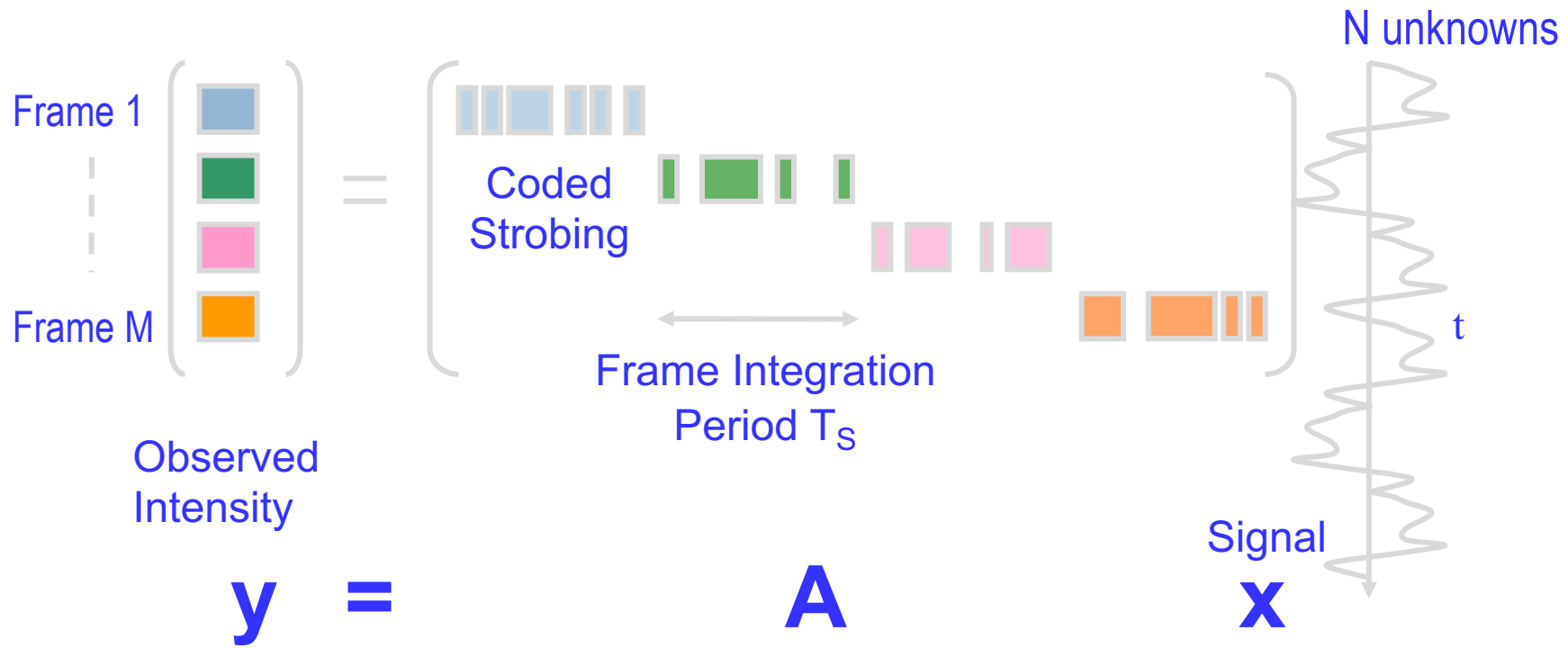


$$\begin{aligned} \min \|\mathbf{x}\|_{L_1} \quad \text{subject to} \quad & y_k = \langle \phi_k, \Psi \mathbf{x} \rangle \quad \forall k \in M \\ \mathbf{x} \in R^n \end{aligned}$$

Convex optimization algorithm

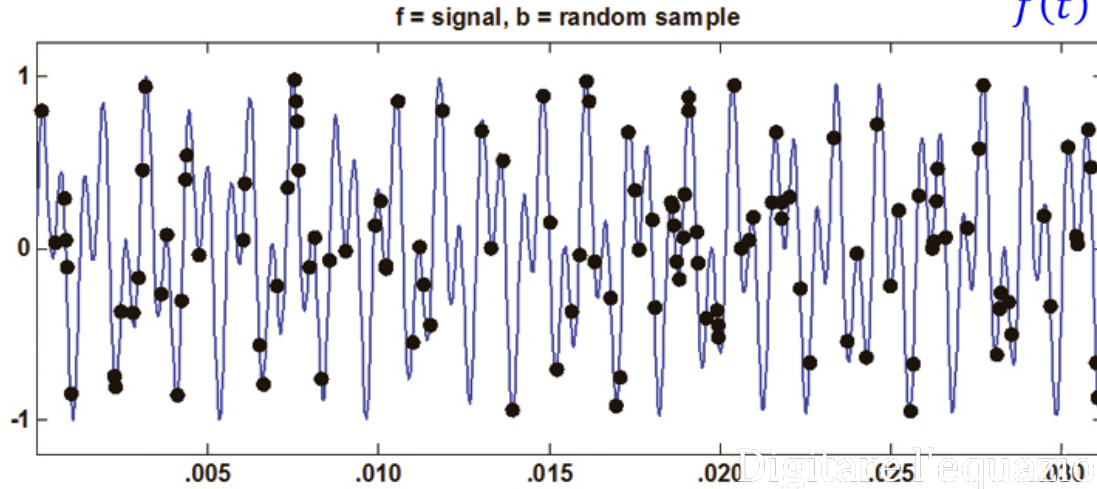
<https://statweb.stanford.edu/~candes/l1magic/#code>





# Introduction

$$f(t) = \sin(1394 \pi t) + \sin(3266 \pi t)$$

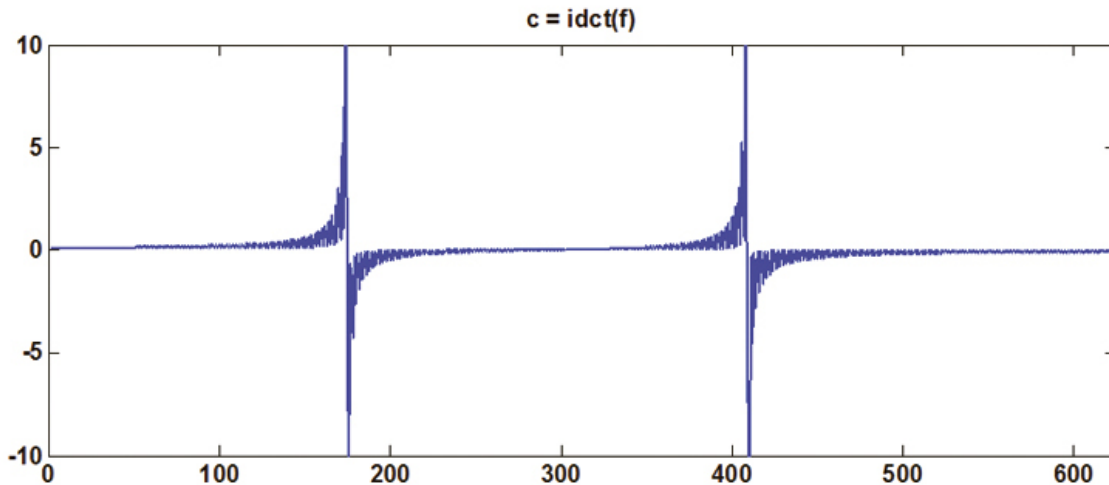


1/8 of second at 4 kHz

$$n = 5000$$

$m = 500$  random samples

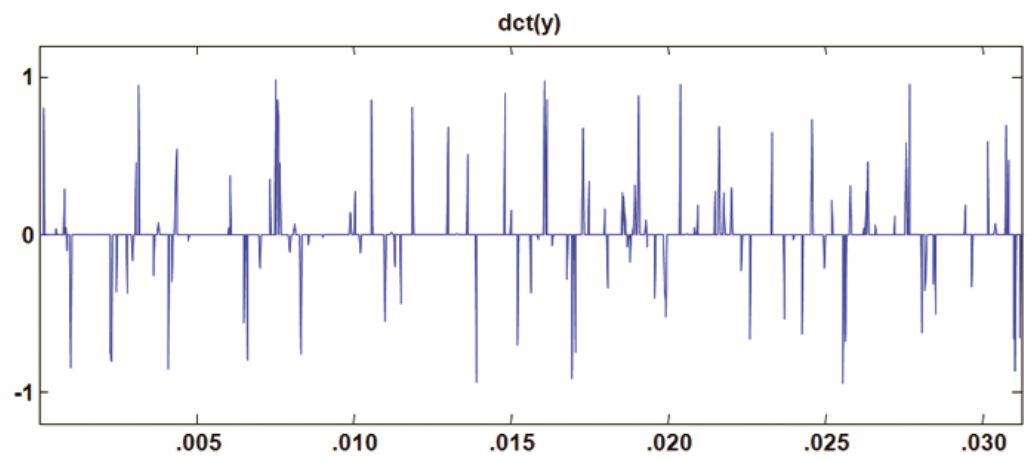
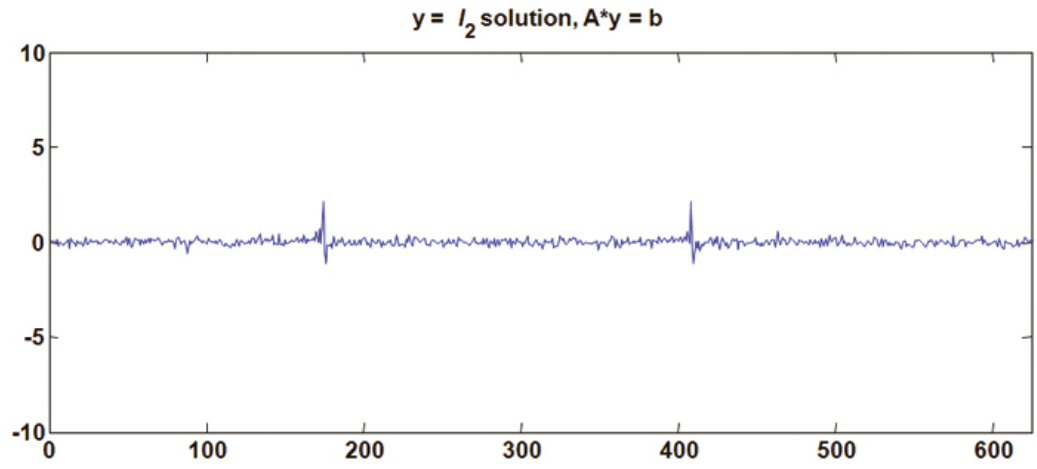
A DCT based dictionary is used



Top: Random samples of the original signal generated by the “A” key on a touch-tone phone. Bottom: The inverse discrete cosine transform of the signal.



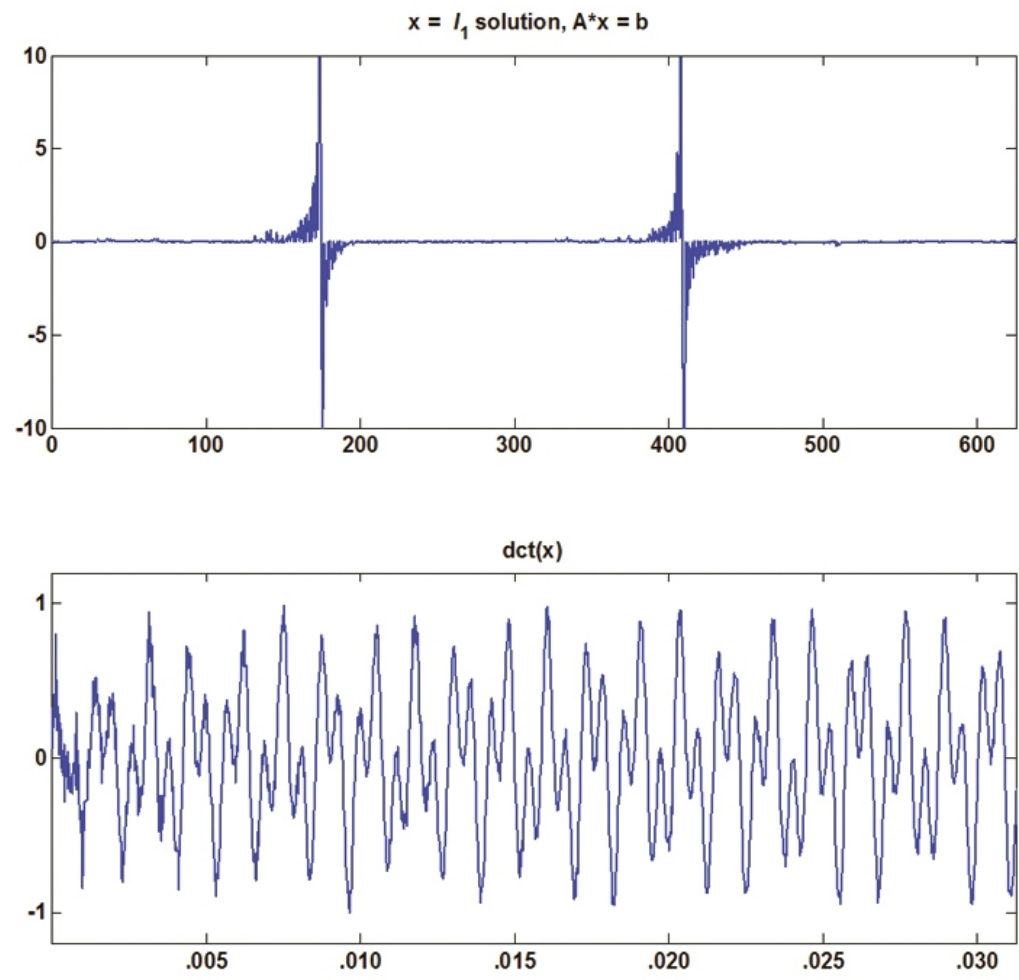
# Introduction



Results by using L<sub>2</sub> norm



# Introduction



Results by using L<sub>1</sub> norm



# Optimization algorithms

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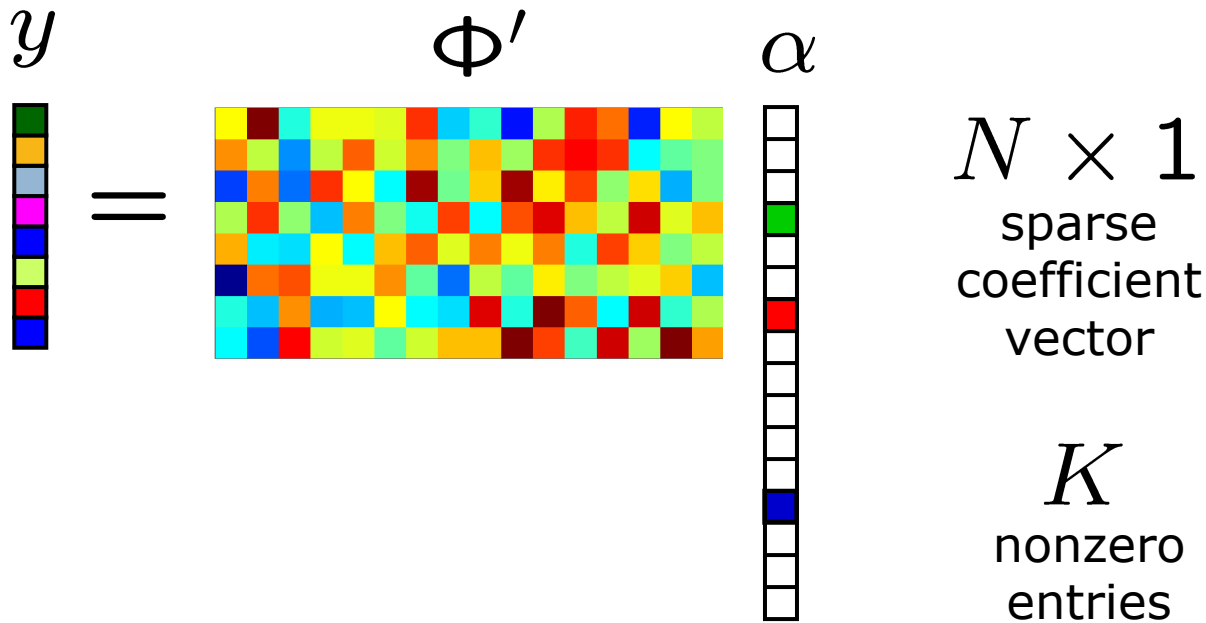
- Signal recovery via iterative greedy algorithm
  - (orthogonal) matching pursuit
  - iterated thresholding
  - CoSaMP





# Universality

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



Random measurements can be used for signals sparse in *any* basis: DCT/FFT/Wavelet/Learned Dictionary



# Dictionary learning

- Goal

- Given training data

$$x_1, x_2, \dots, x_T \quad x_i \in \mathbb{R}^N$$

- learn a dictionary  $D$

$$x_i = D s_i \quad \begin{array}{l} D \in \mathbb{R}^{N \times Q} \\ s_i \in \mathbb{R}^Q \end{array}$$

where  $s_i$  are sparse



# Dictionary learning

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- Optimization approach

$$\min_{D, S} \|X - DS\|_F$$

s.t

$$\forall i, \|s_i\|_0 \leq K$$

Non-convex constraint

Bilinear in  $D$  and  $S$



# Dictionary learning

## ■ Optimization approach

$$\min_{D,S} \|X - DS\|_F$$

s.t

$$\forall i, \|s_i\|_0 \leq K$$

Non-convex constraint

Bilinear in  $D$  and  $S$

## ■ Biconvex in $D$ and $S$

$$\min_{D,S} \|X - DS\|_F + \lambda \sum_k \|s_k\|_1$$

Given  $D$ , the optimization problem is convex in  $s_k$

Given  $S$ , the optimization problem is a least squares problem



# Dictionary learning

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## ■ K-SVD

### ■ Solve using alternate minimization techniques

■ Start with  $D =$  wavelet or DCT bases

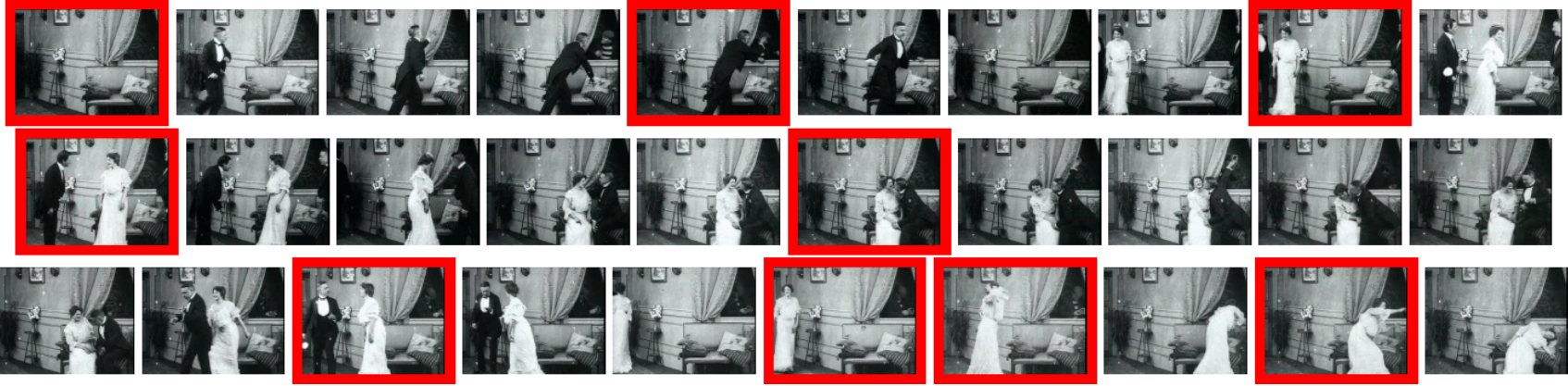
■ Additional pruning steps to control size of the dictionary

## ■ Sparse Modeling for Finding Representative Objects

$$\min \|Y - YC\|_F^2 \quad \text{s.t.} \quad \|C\|_{1,q} \leq \tau, \quad \mathbf{1}^\top C = \mathbf{1}^\top$$



# Finding Representative Objects



# Deblurring



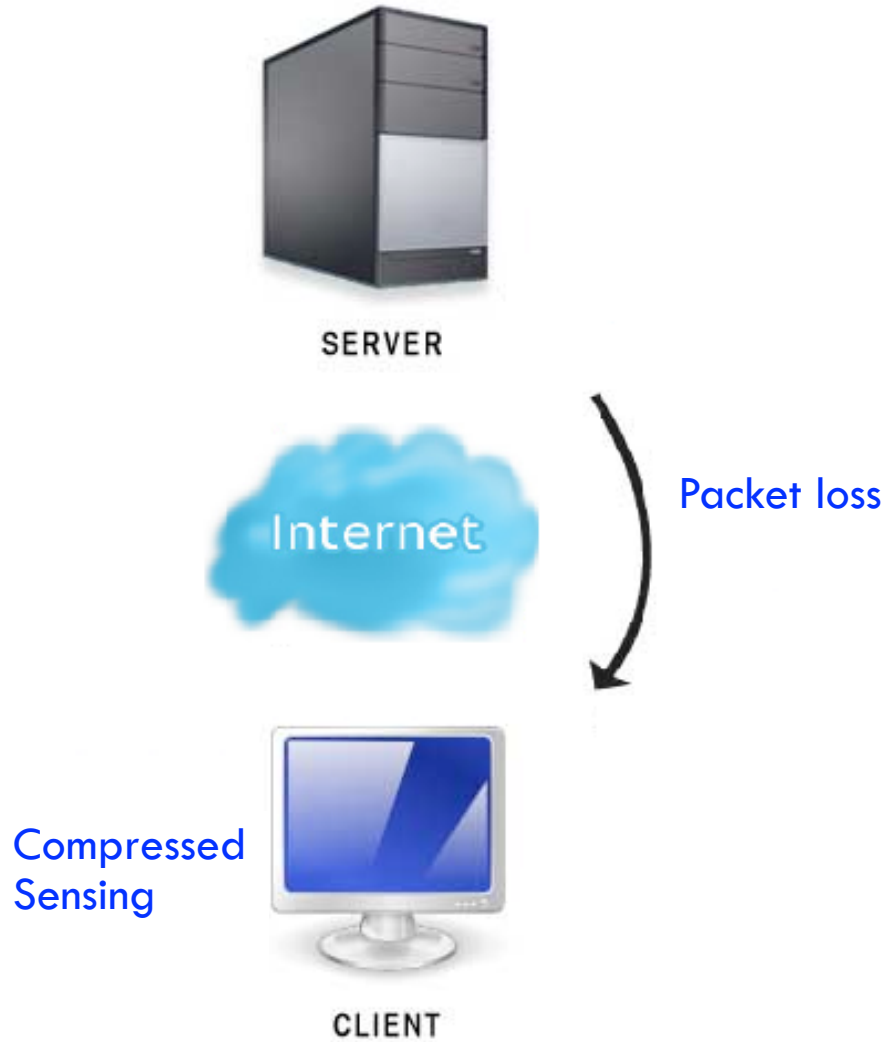
Blurred Photos



Deblurred Result

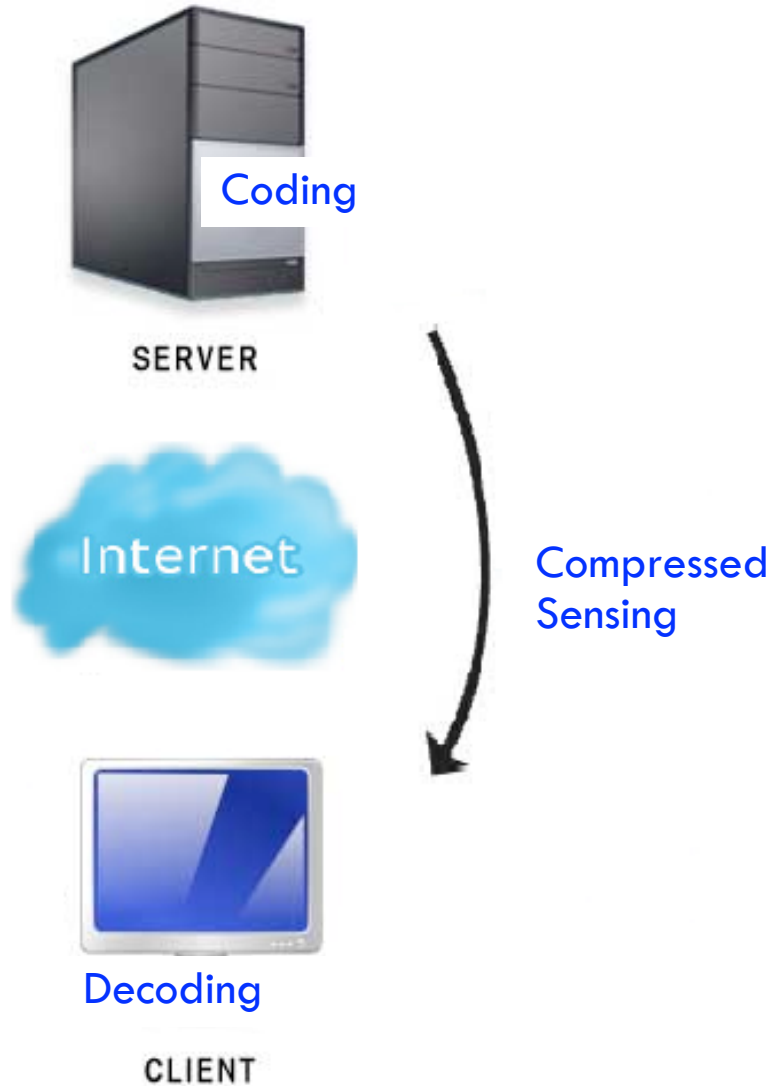


# First scenario



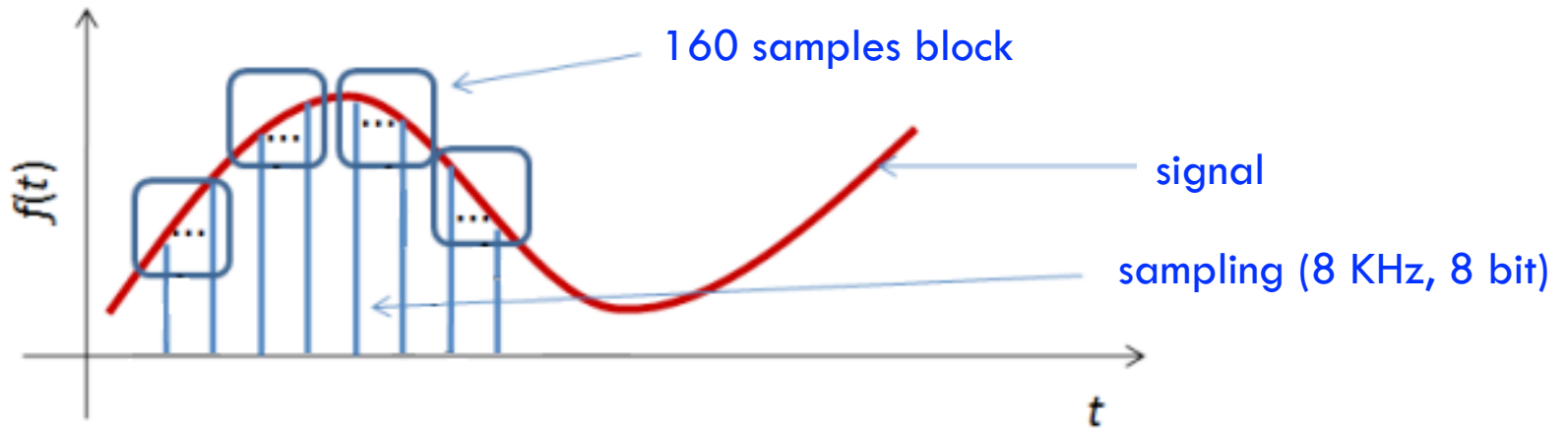


# Second Scenario

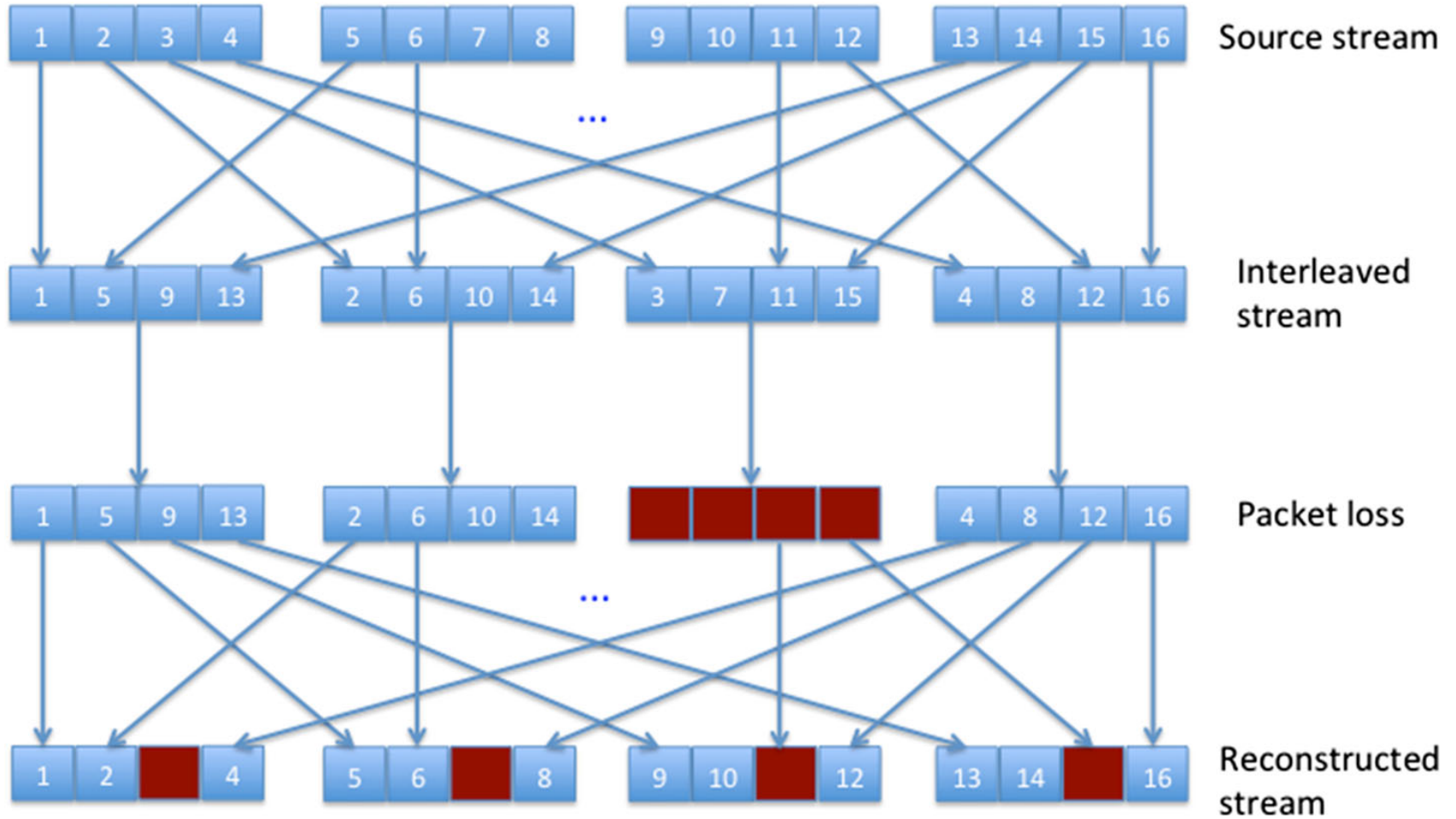


# Packet loss

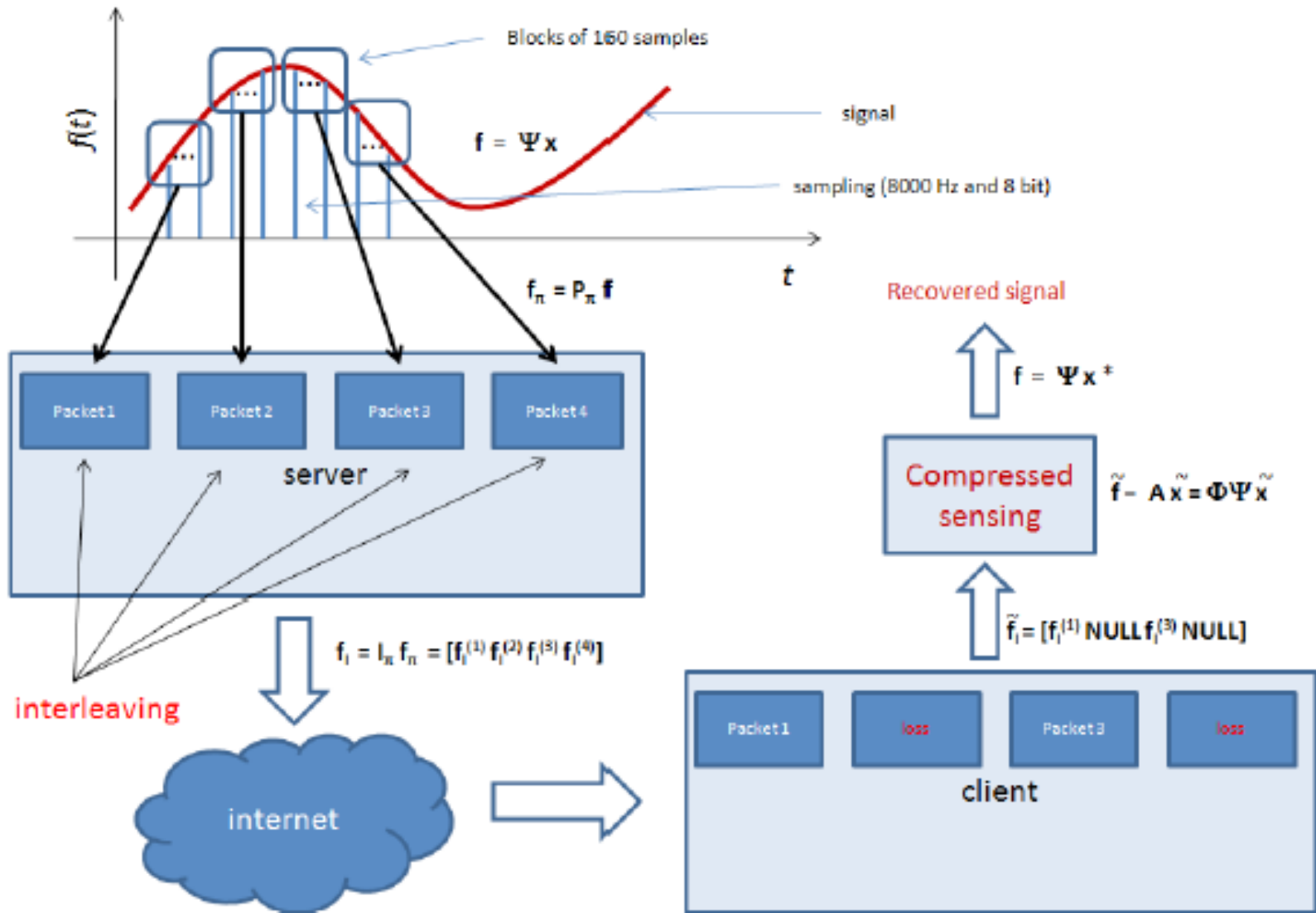
1 sec = 64kbit/sec  $\rightarrow$  1 msec = 64bit  $\rightarrow$  20msec = 1280bit  $\rightarrow$  160 byte



# Interleaving



# Reconstruction scheme



# Experimental results

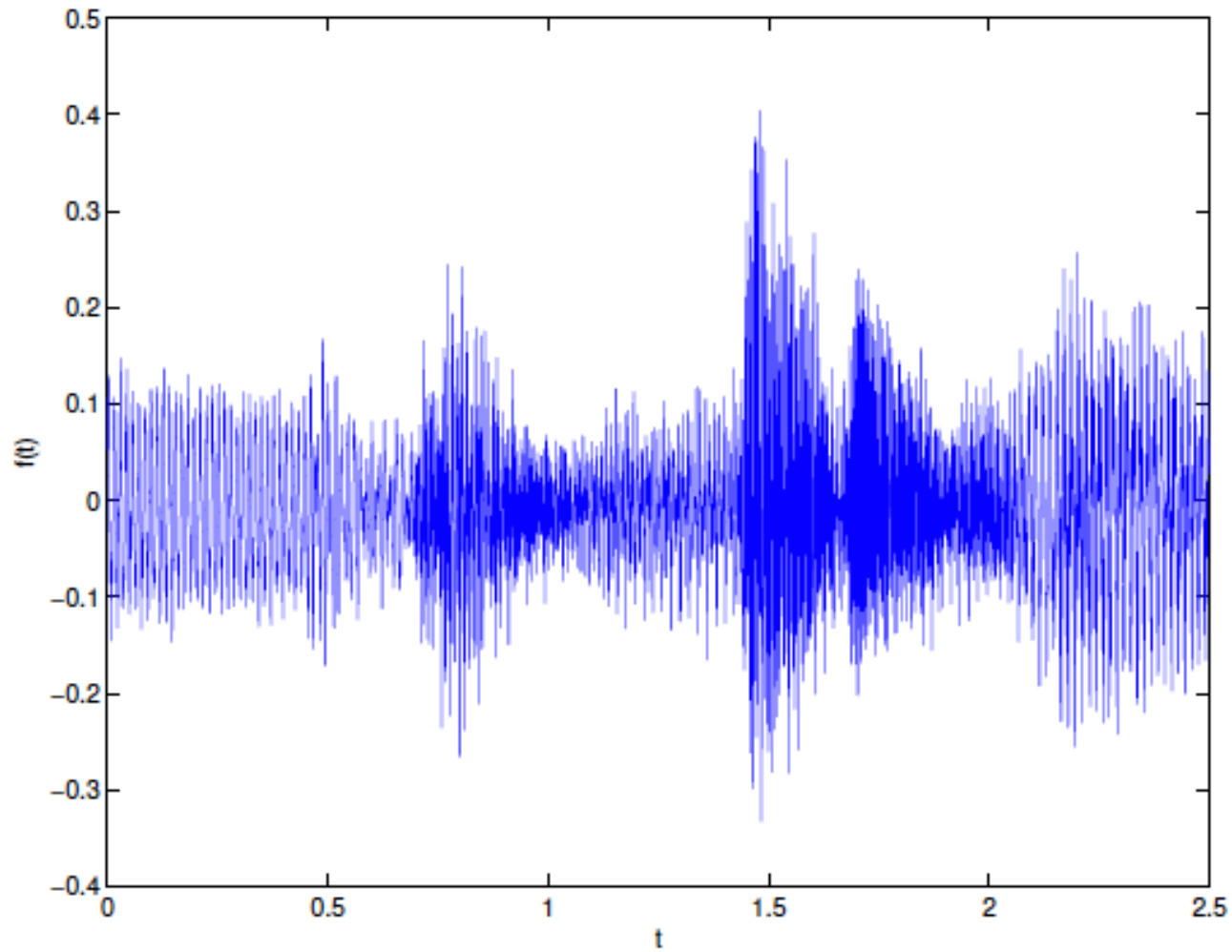


Fig. 4. Audio signal of a female speaker.



# Experimental results

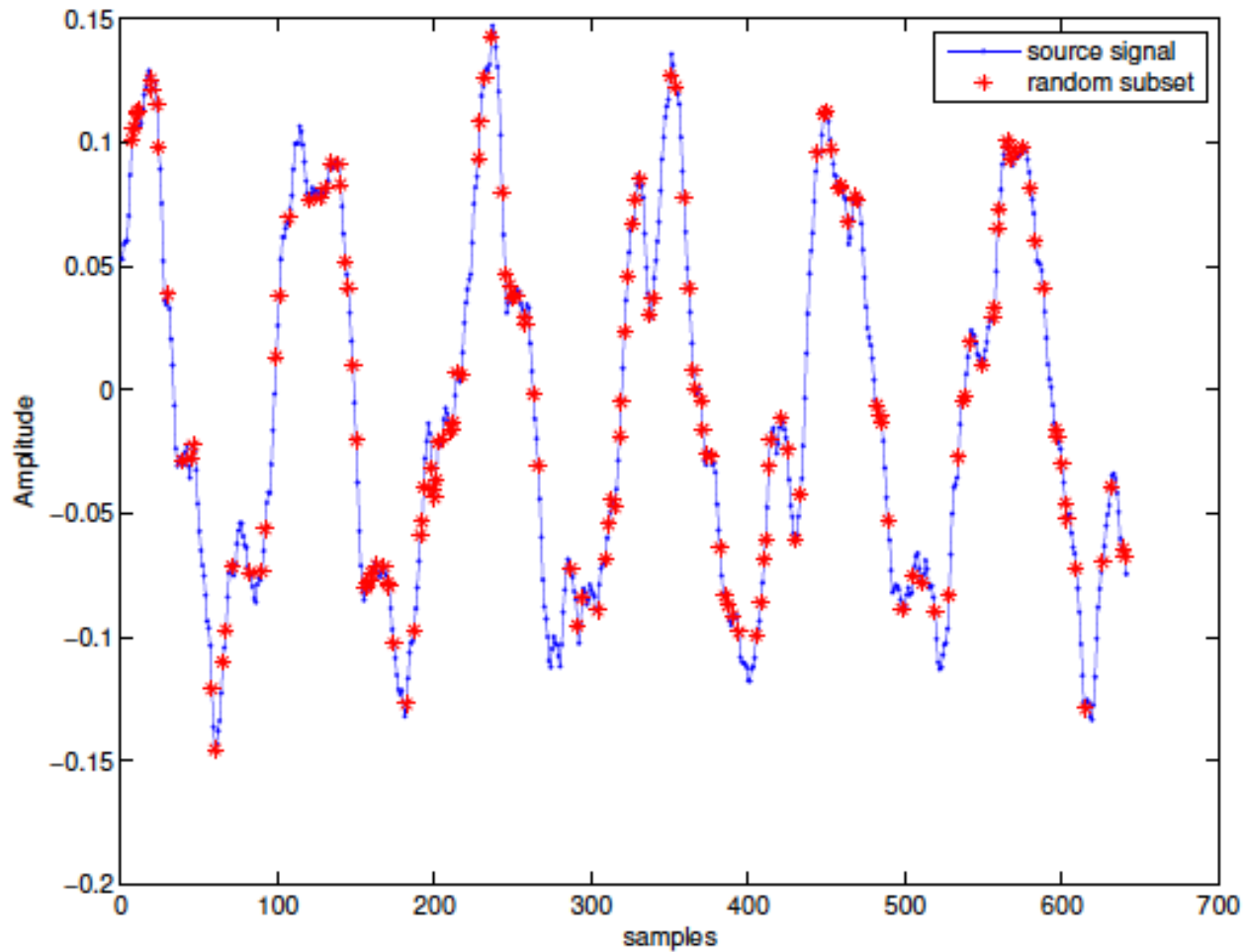


Fig. 5. Frame information after 3 packets lost.



# Experimental results

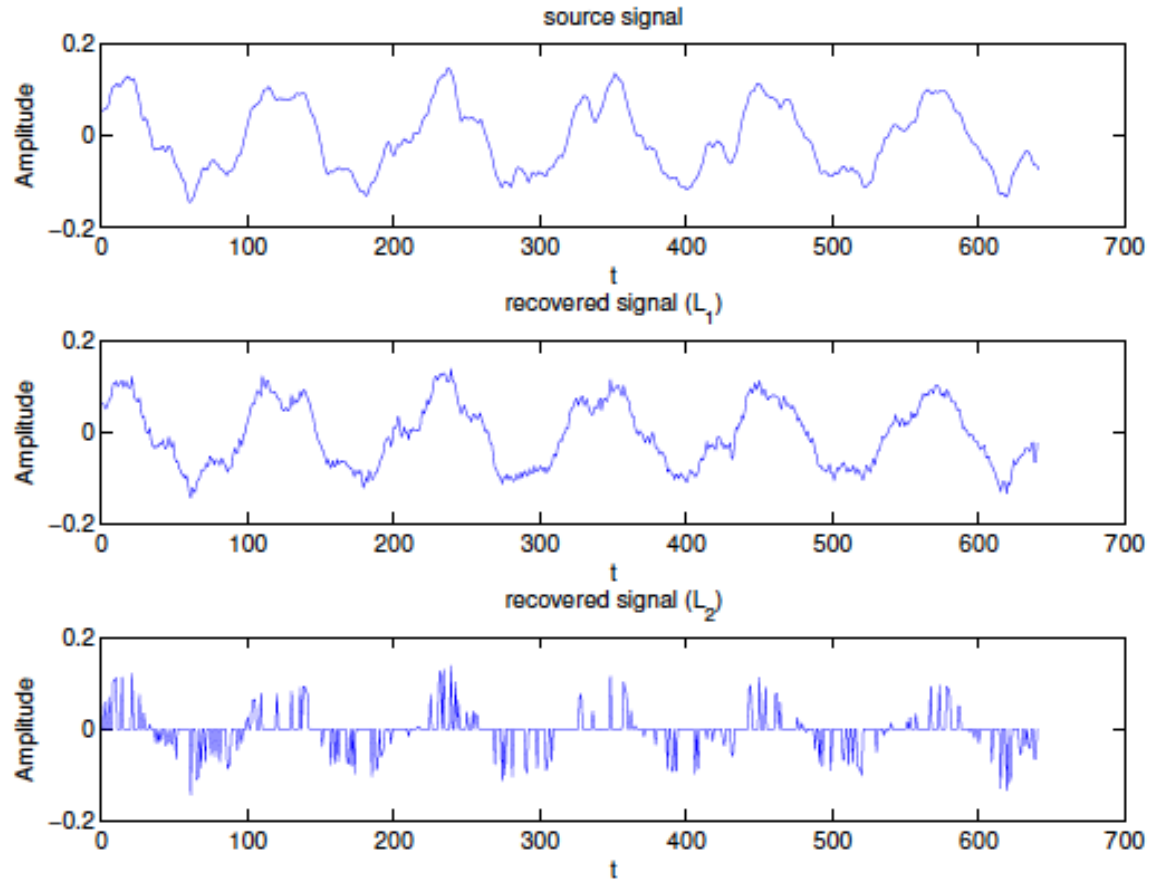


Fig. 6. Comparison between a frame of the source signal and those of the recovered signals by using  $L_1$  and  $L_2$  norms.



# Experimental results

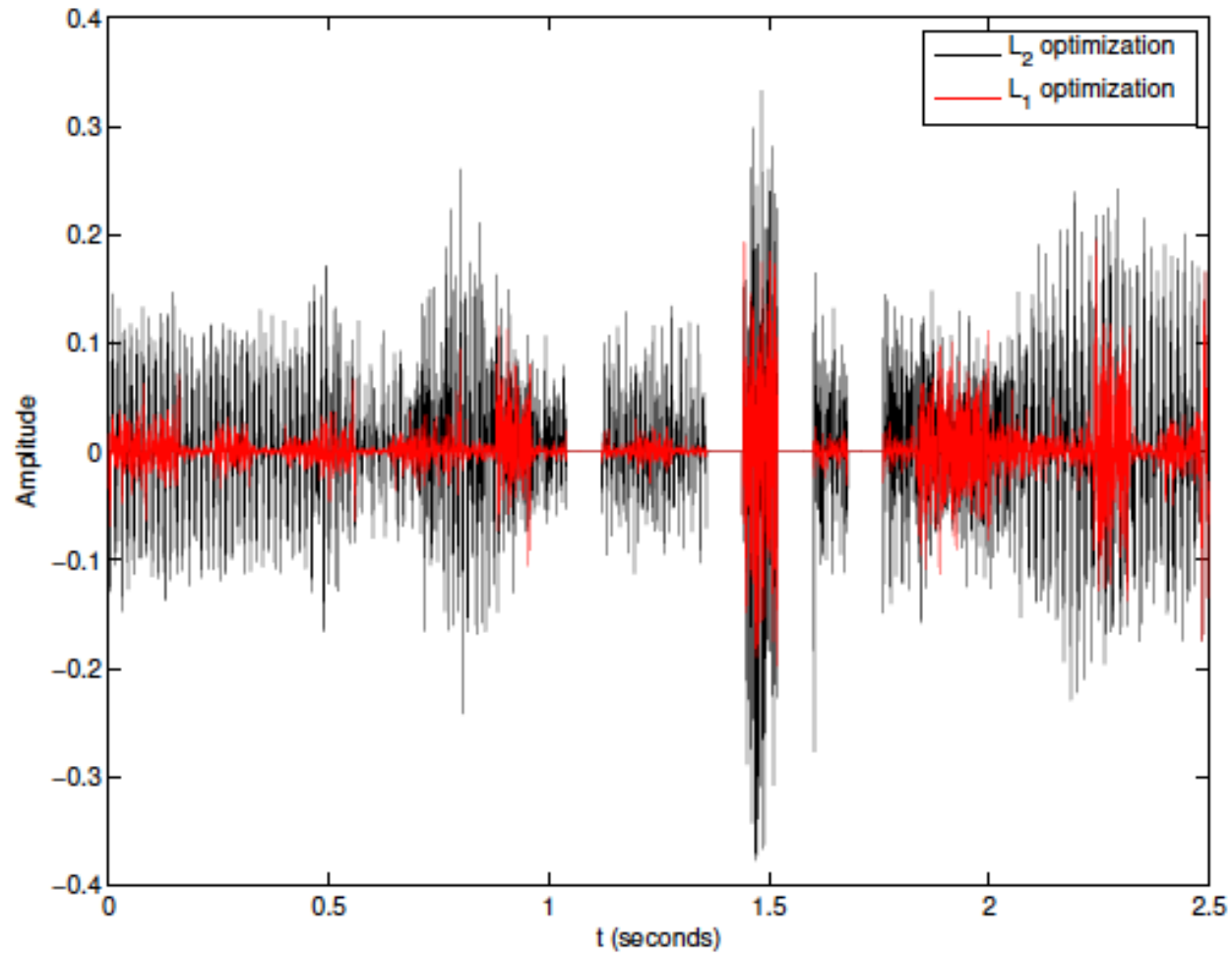


Fig. 7. Residua between the source signal and the recovered signals by using  $L_1$  and  $L_2$  norms.





# Experimental results

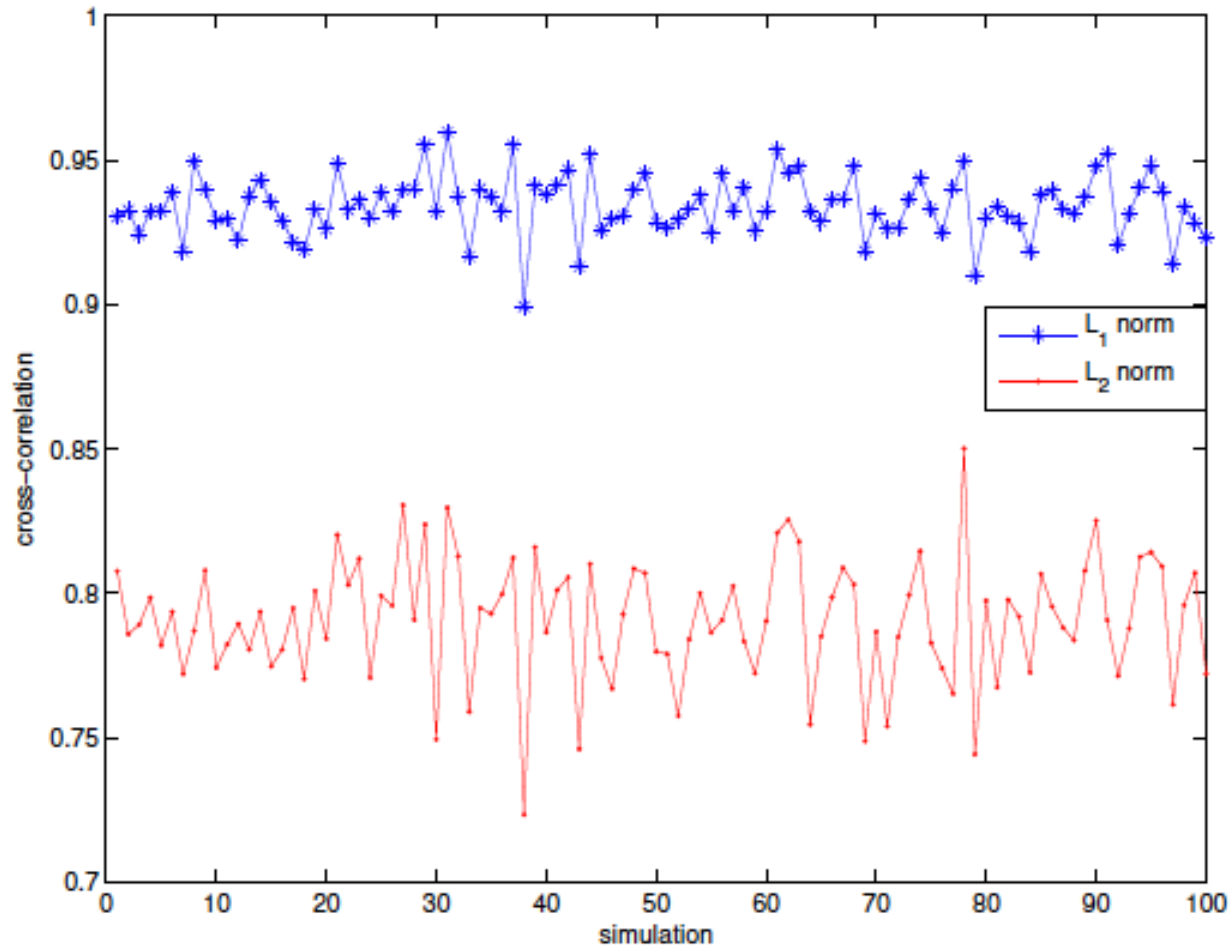


Fig. 8. Cross-correlation coefficients after 100 simulations: audio female speaker.



# Experimental results

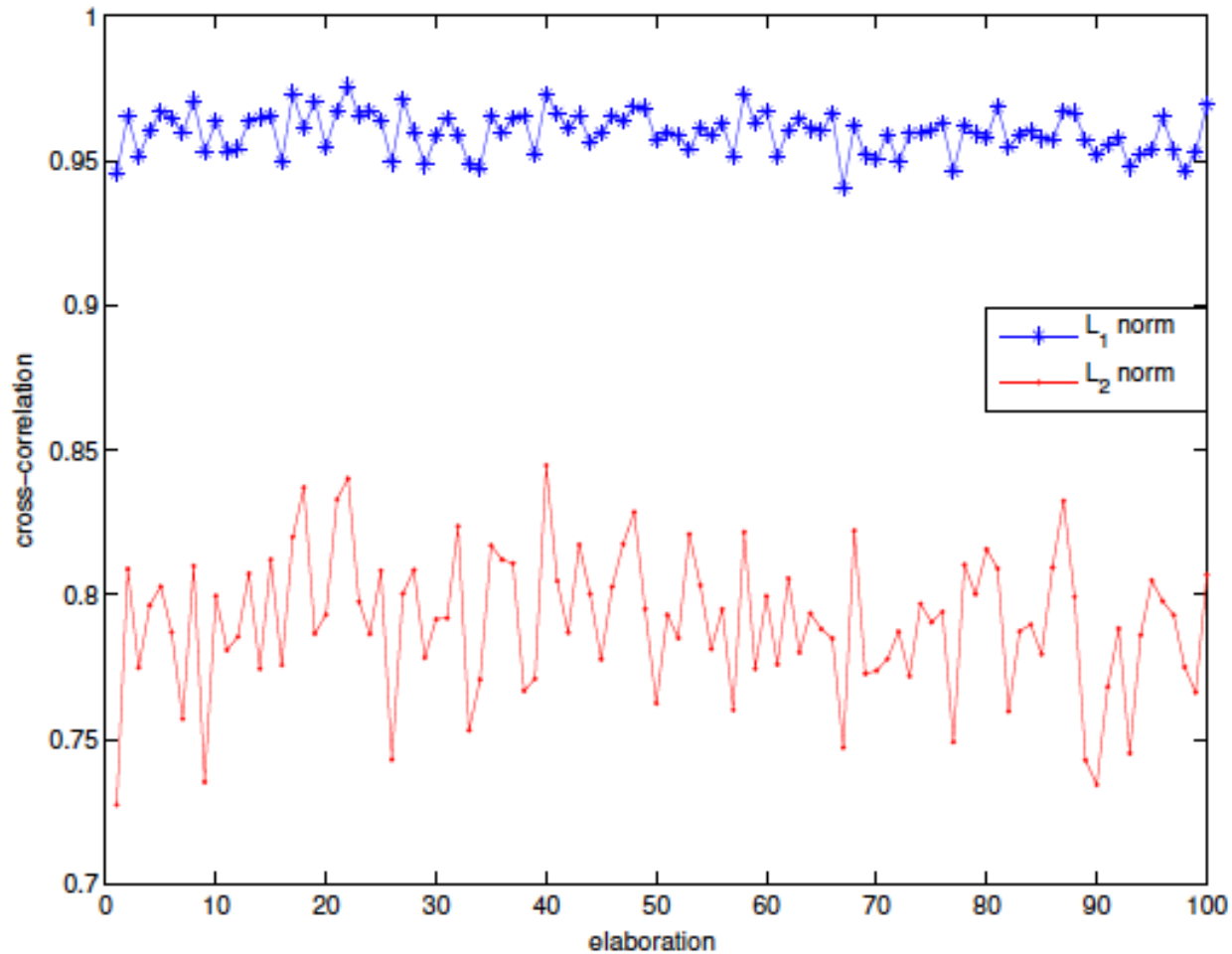


Fig. 9. Cross-correlation coefficients after 100 simulations: audio male speaker.



# Experimental results

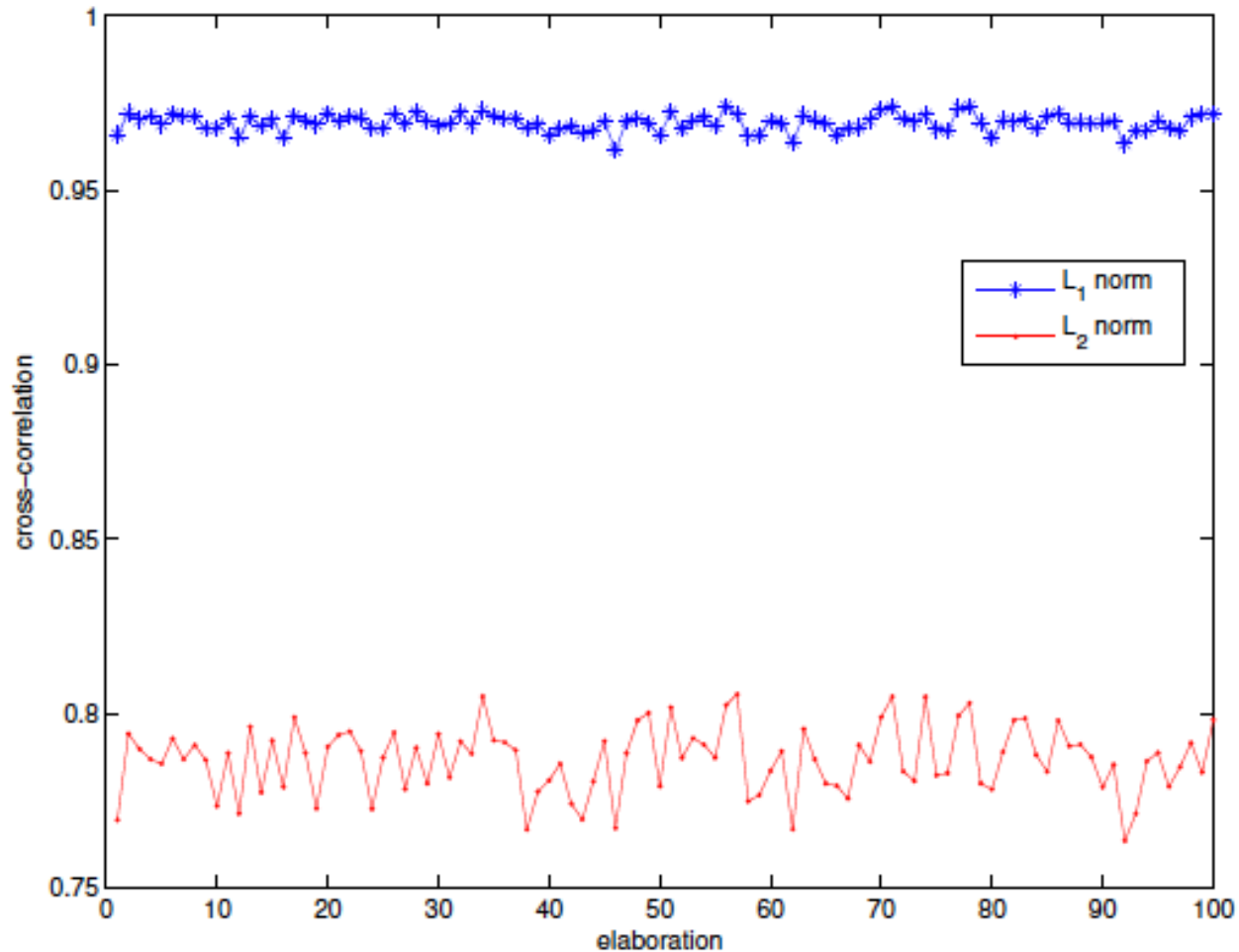


Fig. 10. Cross-correlation coefficients after 100 simulations: audio song.



# Compression scheme

