

INVERSE PROBLEMS

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**M2 AI — SIGNAL PROCESSING**

## DIRECT PROBLEM

- ▶ Let
  - ▶  $x \in \mathbb{R}^N$  be the signal of interest (clean image, music, etc.)
  - ▶  $A \in \mathbb{R}^{MN}$  be a known linear operator (sensing matrix, mixing matrix, diffusion matrix, etc.)
  - ▶  $y \in \mathbb{R}^M$  be the (noisy) observed/measured signal
  - ▶  $e \in \mathbb{R}^M$  be some noise (assumed to be white Gaussian noise)
- ▶ The direct problem is :

$$y = Ax + n$$

# INVERSE PROBLEM

- ▶ The goal of the inverse problem is to estimate the original signal  $x$  from the measurement

$$y = Ax + n$$

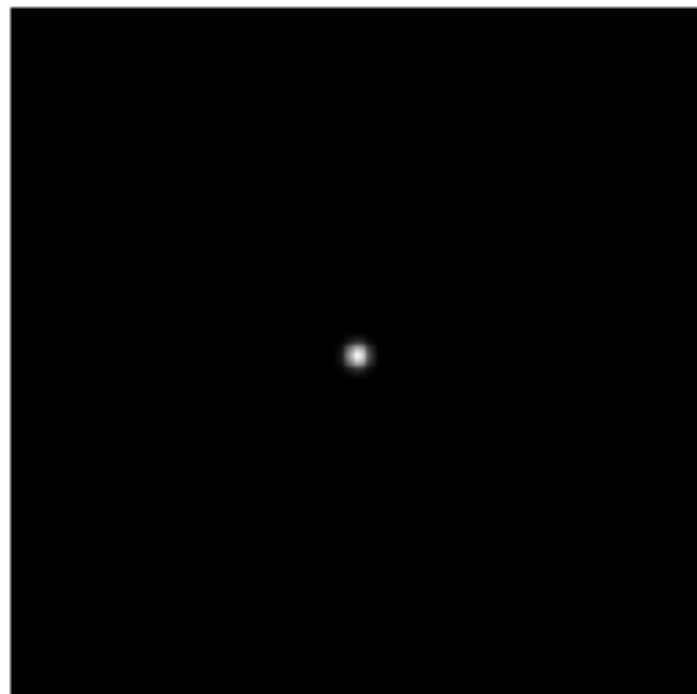
- ▶ If  $M \geq N$  the problem is said (over)-determined
- ▶ If  $M < N$  the problem is said under-determined

# INVERSE PROBLEM: EXAMPLES

- ▶ Denoising:  $y = x + n$

- ▶ Deconvolution:  $y = h * x + n = Hx + n$  or  $\hat{y} = \hat{h} \odot \hat{x} + \hat{n}$

Filter



Fourier transform

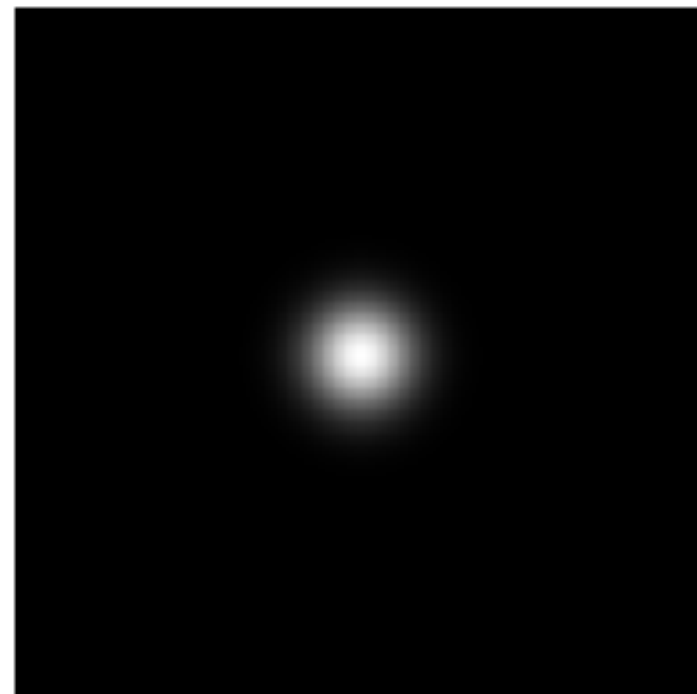
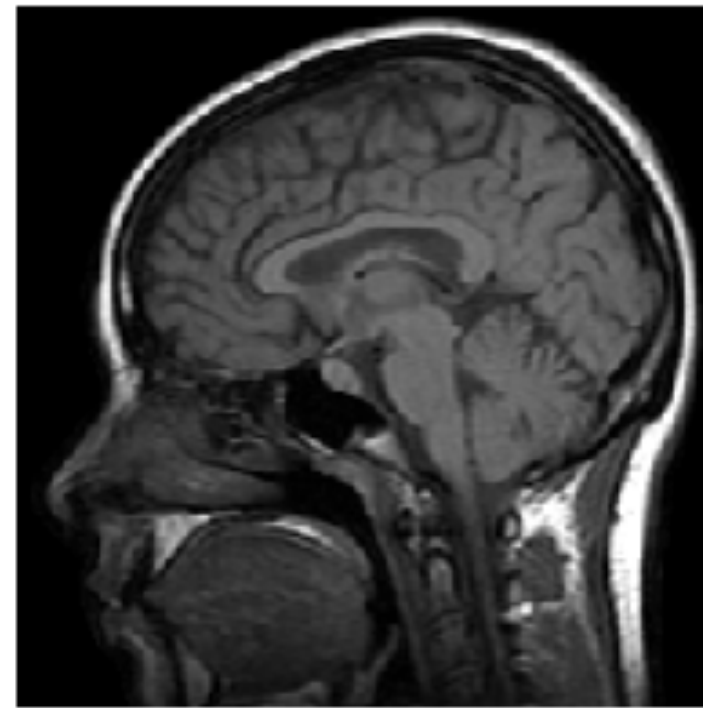
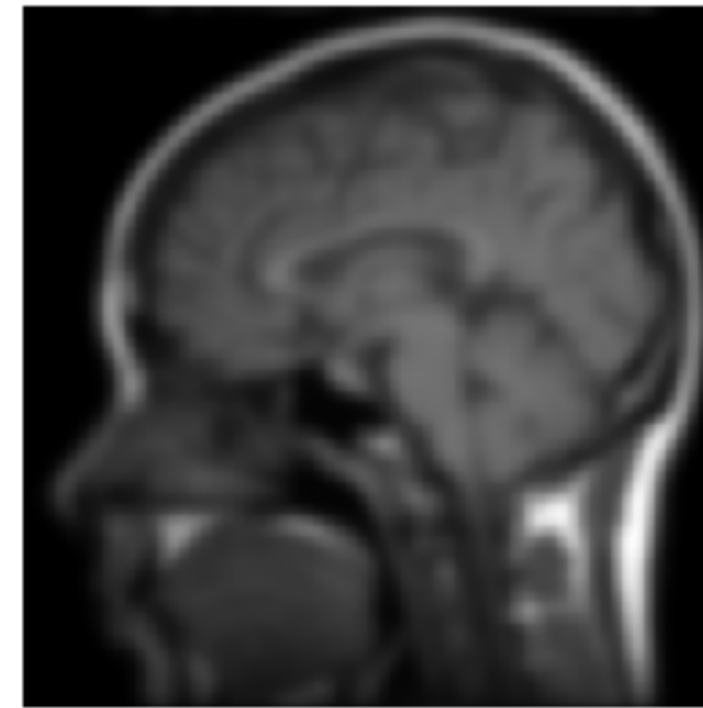


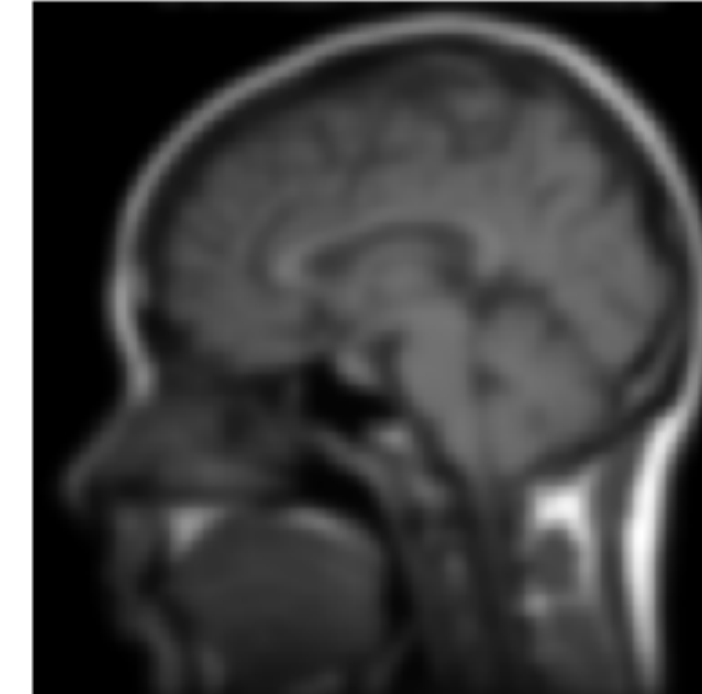
Image t0



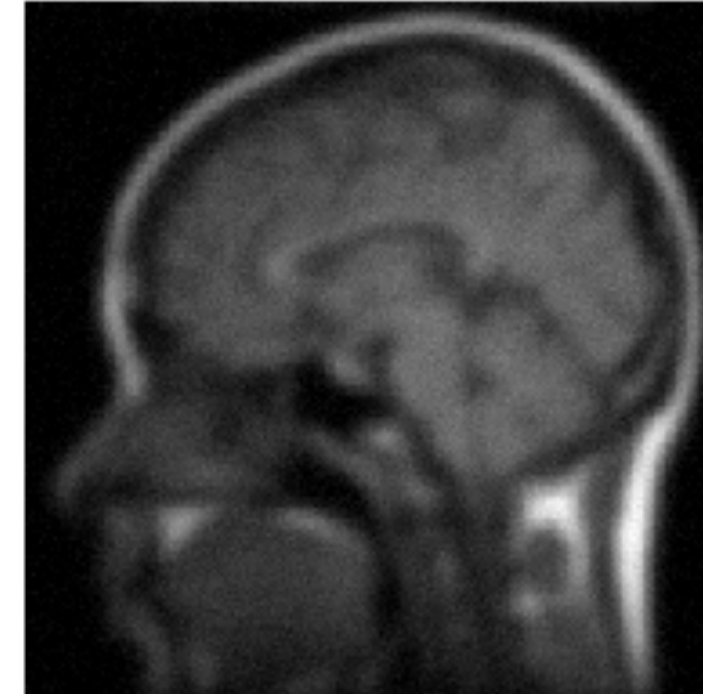
Observation without noise



Observation without noise



Observation with noise



## INVERSE PROBLEM: EXAMPLES

- ▶ Compressive sensing:  $A$  is a random matrix with  $M < N$
- ▶ Inpainting:  $A$  is a binary mask

Image  $f_0$ Observations  $y$ 

# OPTIMIZATION FRAMEWORK

- ▶ We seek an estimation of  $x$  by

$$x = \operatorname{argmin}_x \mathcal{L}(y, A, x) + \lambda \mathcal{R}(x)$$

- ▶ where
  - ▶  $\mathcal{L}(y, A, x)$  is the loss: models the link between the signal  $x$  and the observation  $y$  through the operator  $A$
  - ▶  $\mathcal{R}(x)$  is the loss: models the prior on the signal  $x$
  - ▶  $\lambda > 0$  is some hyper-parameter

# OPTIMIZATION FRAMEWORK

- We seek an estimation of  $x$  by

$$x = \operatorname{argmin}_x \mathcal{L}(y, A, x) + \lambda \mathcal{R}(x)$$

- Loss:
  - $\frac{1}{2} \|y - Ax\|_2^2$ : energy of the residual, adapted to white Gaussian noise
  - $\|y - Ax\|_1$ : robust regression
- Regularization:
  - $\frac{1}{2} \|x\|_2^2$ : energy of the signal
  - $\frac{1}{2} \|\nabla x\|_2^2$ : energy of the derivative
  - $\|x\|_1$ : sparsity of the signal
  - $\|\nabla x\|_1$ : sparsity of the derivative (total variation)

## USE OF A DICTIONARY

- $\mathcal{R}(x)$  can be difficult to choose
- Idea: use a dictionary (such as Wavelets or time-frequency), where the signal is known to be sparse (well represented by few coefficients)
- Let  $\Phi \in \mathbb{R}^{NK}$  be such a dictionary, with  $x = \Phi\alpha$ .  $\alpha$  are called the synthesis coefficients

- The direct problem writes

$$y = Ax + n = A\Phi\alpha + n$$

- The minimization becomes

$$\alpha = \operatorname{argmin} \frac{1}{2} \|y - A\Phi\alpha\|_2^2 + \lambda \|\alpha\|_1$$

- and  $x = \Phi\alpha$



# INVERSE PROBLEM: ALGORITHM

- ▶ How to minimize

$$\alpha = \operatorname{argmin} \frac{1}{2} \|y - A\Phi\alpha\|_2^2 + \lambda \|\alpha\|_1$$

- ▶ It is a non-smooth convex problem
  - ▶ Known as the Lasso or Basis-Pursuit Denoising
- ▶ Consider the "simple" denoising problem (e.g.  $\Phi$  is orthogonal)

$$\alpha = \operatorname{argmin} \frac{1}{2} \|y - \alpha\|_2^2 + \lambda \|\alpha\|_1$$

- ▶ We can show that the solution is given by the so-called Soft-Thresholding operator :

$$\alpha = \mathcal{S}_\lambda(y) = y \left(1 - \frac{\lambda}{|y|}\right)^+$$

# INVERSE PROBLEM: ALGORITHM

- ▶ How to minimize

$$\alpha = \operatorname{argmin} \frac{1}{2} \|y - A\Phi\alpha\|_2^2 + \lambda \|\alpha\|_1$$

- ▶ FISTA (Fast Iterative Shrinkage/Thresholding Algorithm)

- ▶ Initialization:  $\alpha^{(0)} \in \mathbb{R}^N$ ,  $z^{(0)} \in \mathbb{R}^N$ ,  $L \leq \|A\Phi\|^2$ ,  $t = 0$

- ▶ Do until convergence :

- ▶  $\alpha^{(t+1)} = \mathcal{S}_{\lambda/L} \left( z^{(t)} + \frac{1}{L} \Phi^* A^* (y - A\Phi z^{(t)}) \right)$

- ▶  $z^{(t+1)} = \alpha^{(t+1)} + \frac{t}{t+5} (\alpha^{(t+1)} - \alpha^{(t)})$

- ▶ Remark: we have  $\|A\Phi\|^2 \leq \|A\|^2 \|\Phi\|^2$ , then if  $\Phi$  is a Parseval frame, we simply have  $L \leq \|A\|^2$ .

## FISTA WITH WARM RESTART

- ▶ In practice, the algorithm must be run with various values of lambda.
- ▶ When  $\lambda \rightarrow 0$  we have  $\|y - A\Phi\alpha\| \rightarrow 0$ .
- ▶ When  $\lambda = \|\Phi^*A^*y\|_\infty$  the solution is  $\alpha = 0$
- ▶ One can choose these values distributed on a log scale  $\in [10^{-4}\|\Phi^*A^*y\|_\infty, \|\Phi^*A^*y\|_\infty]$ , with a fixed number of  $\lambda$ , such as we have  $\lambda_1 < \lambda_2 < \dots < \lambda_I$
- ▶ The idea is to initialize the algorithm for  $\lambda_i$  with the results get from  $\lambda_{i-1}$

## FISTA WITH THRESHOLDING RULES

- ▶ The Soft-thresholding can be replaced by any thresholding rules

- ▶ Some examples:

- ▶ Hard Thresholding:  $\mathcal{H}_\lambda(\alpha) = \alpha \mathbf{1}_{|\alpha| < \lambda}$

- ▶ Empirical Wiener:  $\mathcal{S}_\lambda^{EW}(\alpha) = \alpha \left( 1 - \frac{\lambda^2}{|\alpha|^2} \right)^+$

# TO DO (1/2): INPAINTING (DEADLINE: 31/12)

- Data
  - Image or signal you want
- Todo
  - Simulate various inpainting problems as follows
    - Generate a random binary matrix  $A$  of the same size as the signal, with a parameter  $p$  controlling the Bernoulli law
    - Add some white Gaussian noise (at various levels)
    - Generate the direct problem  $y = Ax + b$  (where  $x$  is the original signal)
    - Estimate  $x$  using the sparse approach (reminder: an audio signal (resp. image) is sparse in the time-frequency (resp. Wavelet) domain)
  - Discuss the results obtained by changing:
    - the sparse representation (various wavelets, various STFT parameters...)
    - the thresholding rules (soft, hard, empirical Wiener)
    - the choice of the  $\lambda$  parameter
  - Discussion should be made concerning the value of  $p$  and the level of noise

## TO DO (2/2): COMPRESS SENSING (DEADLINE: 31/12)

- Data
  - Image or signal you want of size  $N$
- Todo
  - Simulate various compress sensing problems as follows
    - Generate a random normal matrix  $A$  of size  $M \times N$ ,  $M < N$ , with a given mean and a given covariance matrix (start with a standard normal pdf)
    - Add some white Gaussian noise (at various levels)
    - Generate the direct problem  $y = Ax + b$  (where  $x$  is the original signal)
    - Estimate  $x$  using the sparse approach (reminder: an audio signal (resp. image) is sparse in the time-frequency (resp. Wavelet) domain)
  - Discuss the results obtained by changing:
    - the sparse representation (various wavelets, various STFT parameters...)
    - the thresholding rules (soft, hard, empirical Wiener)
    - the choice of the  $\lambda$  parameter
  - Discussion should be made concerning the size of  $A$ , its mean and its covariance matrix, as well as the level of noise.