A Fast Iterative Algorithm for Demixing Sparse Signals from Nonlinear Observations

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Demixing problem — Motivation

• What is demixing and why do we care?



Image credits: NASA

• Another example



Demixing problem — Examples

- *Demixing* problem is special of interest of the numerous applications ranging from
 - signal processing, astronomy, computer vision, and machine learning
- Examples
 - Morphological Component Analysis (MCA)
 - Separation of foreground and background in video



8 Robust PCA



Demixing problem — Definition

 In simple form, demixing involves disentangling two (or more) constituent signals from observations of their linear superposition:

$$x = \Phi w + \Psi z$$

- Φ and Ψ are *incoherent* orthonormal bases of \mathbb{R}^n ,
- $w, z \in \mathbb{R}^n$ are the corresponding basis coefficients



• Goal: reliably recover the constituent signals (equivalently, their basis representations *w* and *z*) from the superposition signal *x*.

Four problems

 $y \in \mathbb{R}^{m \times 1}$, $A \in \mathbb{R}^{m \times n}$, $m \ll n$

• Compressive sensing (Linear inverse problem)

$$y = Ax$$

where $x \in \mathbb{R}^n$ s.t. $||x||_0 \leq s$.

Linear demixing problem

$$y = A \left(\Phi w + \Psi z \right)$$

where Φ and Ψ are *incoherent* bases in \mathbb{R}^n , and $w, z \in \mathbb{R}^n$ are the basis coefficients s.t. $||w||_0 \leq s$ and $||z||_0 \leq s$.

Nonlinear signal recovery

$$y = g(Ax)$$

where g denotes an element wise nonlinear *link* function.

Nonlinear demixing problem – Our focus in this talk

$$y = g(A(\Phi w + \Psi z))$$

Challenges in (nonlinear) demixing Problem

Fundamental identifiability issue (Linear demixing)

- number of unknowns (2*n*) is greater than the number of observations (*n*).
- remedy: some type of *incoherence* between the constituent signals (or more specifically, between the corresponding bases Φ and Ψ).

Limited number of measurements

- y = Ax, x is superposition signal and $A \in \mathbb{R}^{m \times n}$ denotes the measurement operator with $m \ll n$.
- remedy: structural assumptions on the constituent signals.

Nonlinear observation model

- y = g(Ax) + e, x is superposition signal and $e \in \mathbb{R}^m$ denotes the additive noise.
- remedy: The subject of this talk.

• Formulation with additive noise

$$y = g(Ax) + e$$

where g denotes an element wise nonlinear *link* function and $e \in \mathbb{R}^m$ represents the noise.

- Nonlinear Signal recovery
 - **1**-bit compressive sensing \rightarrow modeling high speed ADC's
 - 2 phase retrieval \rightarrow modern astronomical imaging systems



Nonlinear Demixing Problem — Formulation

General formulation

$$y = g(A(\Phi w + \Psi z)) + e \tag{1}$$

- Φ and Ψ are *incoherent* bases in \mathbb{R}^n
- $w, z \in \mathbb{R}^n$ are the basis coefficients such that. $\|w\|_0 \leq s$ and $\|z\|_0 \leq s$
- $e \in \mathbb{R}^m$ denotes the additive noise.
- The goal is to recover w and z (constituent signals)

Definition (ε -incoherence)

The orthonormal bases Φ and Ψ are said to be ε -incoherent if:

$$\varepsilon = \sup_{\substack{\|u\|_0 \le s, \|v\|_0 \le s \\ \|u\|_2 = 1, \|v\|_2 = 1}} |\langle \Phi u, \Psi v \rangle|.$$

Nonlinear Demixing Problem — First Scenario

• We consider two scenarios:

1 In the first scenario, the model is given by

$$y = g(A(\Phi w + \Psi z))$$

- *g* is <u>unknown</u> and odd function. It can be non-smooth and non-invertible
- A with i.i.d standard normal entries
- no additive noise
- We introduced an algorithm called ONESHOT¹.
- Sample complexity result

Suppose we fix $\kappa > 0$ as a small constant, and suppose that the incoherence parameter $\varepsilon = c\kappa$ for some constant c, and that the number of measurements scales as:

$$m = \mathcal{O}\left(rac{s}{\kappa^2}\lograc{n}{s}
ight).$$

¹M. Soltani and C. Hegde, Demixing, Sparse Signals from Nonlinear Observations, Asilomar Conference on Signals, Systems, and Computers, November 2016.

- Disadvantages of ONESHOT:
 - sparse components are recovered only up to an arbitrary scale factor
 - leading to high estimation errors in practice
 - 3 its sample complexity is inversely dependent on the estimation error
- To solve these problems, we propose a different, iterative algorithm for recovering the signal components
 - Demixing with Hard Thresholding (DHT)

• In the second scenario, the model is given by

$$y = g(A(\Phi w + \Psi z)) + e$$

- g is **known** and its derivative is strictly bounded either within a positive, or within a negative interval
- A with independent isotropic rows
- A with independent subgaussian isotropic rows
- additive noise is assumed
- By defining Γ = [Φ Ψ] and t = [w; z], and Θ(x) = ∫_∞^x g(u)du, DHT tries to solve the following optimization problem (F(t) : ℝ²ⁿ → ℝ):

$$\min_{t \in \mathbb{R}^{2n}} F(t) = \frac{1}{m} \sum_{i=1}^{m} \Theta(a_i^T \Gamma t) - y_i a_i^T \Gamma t$$

s. t. $\|t\|_0 \le 2s$.

Algorithm 1 Demixing with Hard Thresholding DHT

```
Inputs: Bases \Phi and \Psi, measurement matrix A, link function g, measurements
y, sparsity level s, step size \eta'.
Outputs: Estimates \hat{x} = \Phi \hat{w} + \Psi \hat{z}, \hat{w}, \hat{z}
Initialization:
(x^0, w^0, z^0) \leftarrow \text{ARBITRARY INITIALIZATION}
k \leftarrow 0
while k < N do
    t^k \leftarrow [w^k; z^k]
                                           {Forming constituent vector}
   t_1^k \leftarrow \frac{1}{T} \Phi^T A^T (g(Ax^k) - y)
t_2^k \leftarrow \frac{1}{T} \Psi^T A^T (g(Ax^k) - y)
   \nabla F^k \xleftarrow{m} [t_1^k; t_2^k] \qquad \{\text{Forming gradient}\}
   \tilde{t}^k = t^k - \eta' \nabla F^k
                                           {Gradient update}
   [w^k; z^k] \leftarrow \mathcal{P}_{2s}(\tilde{t}^k) {Projection}
    \mathbf{x}^k \leftarrow \Phi \mathbf{w}^k + \Psi \mathbf{z}^k
                                              {Estimating \hat{x}}
    k \leftarrow k + 1
end while
Return: (\widehat{w}, \widehat{z}) \leftarrow (w^N, z^N)
```

Definition (Subgaussian random variable)

A random variable X is called subgaussian if it satisfies the following:

$$\mathbb{E}\exp\left(\frac{cX^2}{\|X\|_{\psi_2}^2}\right) \leq 2,$$

where c > 0 is an absolute constant and $||X||_{\psi_2}$ denotes the ψ_2 -norm which is defined as follows:

$$\|X\|_{\psi_2} = \sup_{p \ge 1} \frac{1}{\sqrt{p}} (\mathbb{E}|X|^p)^{\frac{1}{p}}.$$

Definition (Isotropic random vectors)

A random vector-valued variable $v \in \mathbb{R}^n$ is said to be isotropic if $\mathbb{E}vv^T = I_{n \times n}$.

Some definitions — Second Scenario

Definition (Cross-coherence)

The cross-coherence parameter between the measurement matrix A and the dictionary $\Gamma = [\Phi \ \Psi]$ is defined as follows:

$$\vartheta = \max_{i,j} \frac{a_i^T \Gamma_j}{\|a_i\|_2},$$

where a_i and Γ_i denote the i^{th} row of A and the j^{th} column of Γ .

Definition (*Restricted Strong Convexity/Smoothness*)

A loss function f satisfies (RSC/RSS) if:

$$m_{4s} \leq \|\nabla_{\xi}^2 f(t)\| \leq M_{4s}, \quad t \in \mathbb{R}^{2n},$$

where $\xi = \text{supp}(t_1) \cup \text{supp}(t_2)$, for all $||t_i||_0 \le 2s$ and i = 1, 2. Also, m_{4s} and M_{4s} are (respectively) called the RSC and RSS constants.

Theorem (Performance of DHT)

Consider the model $y = g(A(\Phi w + \Psi z)) + e$. Suppose that the corresponding objective function F satisfies the RSS/RSC properties with constants M_{6s} and m_{6s} on the set J with $\|J\|_0 \leq 6s$ such that $1 \leq \frac{M_{6s}}{m_{6s}} \leq \frac{2}{\sqrt{3}}$. Choose a step size parameter η' with $\frac{0.5}{M_{6s}} < \eta' < \frac{1.5}{m_{6s}}$. Then, DHT outputs a sequence of estimates (w^k, z^k) such that the estimation error of the true constituent vector, $t^* = [w^*; z^*]$ satisfies the following upper bound (in expectation) for any $k \geq 1$:

$$\|t^{k+1} - t^*\|_2 \le (2q)^k \|t^0 - t^*\|_2 + C\tau \sqrt{\frac{s}{m}},$$
 (2)

where $q = 2\sqrt{1 + {\eta'}^2 M_{6s}^2 - 2\eta' m_{6s}}$ and C > 0 is a constant that depends on the step size η' and the convergence rate q.

Theorem (Sample complexity when the rows of A are isotropic)

Suppose that the rows of A are independent isotropic random vectors. In order to achieve the requisite RSS/RSC properties of Theorem of DHT, the number of samples needs to scale as: $m = O(s \log n \log^2 s \log(s \log n))$, provided that the bases Φ and Ψ are incoherent enough.

Theorem (Sample complexity when the elements of *A* are subgaussian)

Assume that all assumptions and definitions in Theorem of DHT holds except that the rows of matrix A are independent subgaussian isotropic random vectors. Then, in order to achieve the requisite RSS/RSC properties of Theorem DHT, the number of samples needs to scale as: $m = O\left(s \log \frac{n}{s}\right)$, provided that the bases Φ and Ψ are incoherent enough.

Proof sketch

- Assuming the defined objective function satisfies RSC/RSS.
- Establishing linear convergence of DHT in expectation using *Khintchine inequality*

$$\|t^{k+1} - t^*\|_2 \le (2q)^k \|t^0 - t^*\|_2 + C\tau \sqrt{\frac{s}{m}}$$

- Verifying the objective function satisfies RSC/RSS in two cases:
 - A with independent isotropic rows
 - using Uniform Rudelson's inequality and Uniform symmetrization
 - A with independent subgaussian isotropic rows
 - using *D-RIP* argument

Please see the following for more details:

"M. Soltani and C. Hegde, Fast Algorithms for Demixing Sparse Signals from Nonlinear Observations, arXiv:1608.01234."

Table: Summary of our contributions, and comparison with existing methods for the concrete case where Φ is the identity and Ψ is the DCT basis. Here, s denotes the sparsity level of the components, n denotes the ambient dimension, m denotes the number of samples, and κ denotes estimation error.

Algorithms	Sample complexity	Running time	Measurements	Link function
LASSO[1]	$\mathcal{O}(\frac{s}{\kappa^2}\log\frac{n}{s})$	poly(n)	Gaussian	unknown
OneShot	$\mathcal{O}(\frac{s}{\kappa^2}\log\frac{n}{s})$	$\mathcal{O}(mn)$	Gaussian	unknown
DHT	$\mathcal{O}(s \text{ polylog } n)$	$\mathcal{O}(mn \log \frac{1}{\kappa})$	Isotropic rows	known
DHT	$\mathcal{O}(s \log \frac{n}{s})$	$\mathcal{O}(mn\log\frac{1}{\kappa})$	Subgaussian	known

[1]. Y. Plan, R. Vershynin, and E. Yudovina. High-dimensional estimation with geometric constraints. arXiv preprint arXiv:1404.3749, 2014.

We compare $\operatorname{ONESHOT}$ and DHT with two other algorithms:

• *Nonlinear convex demixing with LASSO* or NLCDLASSO. This algorithm solves the following optimization problem:

$$\min_{\substack{z,w\\ \text{subject to}}} \|\widehat{x}_{\text{lin}} - (\Phi z + \Psi w)\|_2$$

$$\|w\|_1 \le \sqrt{s}, \quad \|z\|_1 \le \sqrt{s}.$$

$$(3)$$

Operation of Demixing with Soft Thresholding or DST. This algorithm solves the following optimization problem:

$$\min_{t} \quad \frac{1}{m} \sum_{i=1}^{m} \Theta(a_i^T \Gamma t) - y_i a_i^T \Gamma t + \beta \|t\|_1, \tag{4}$$

Experimental Results — Synthetic data

• Second Scenario — Link function, g is known



• Phase transition plots with cosine similarity as the criterion. Link function is defined as g(x) = 2x + sin(x).

• Second Scenario — Link function, g is known



Image credits: NASA and Convexity in Source Separation

• Parameters:

 $n = 512 \times 512, s = 1000, m = 15000, g(x) = \frac{1}{2} \frac{1 - e^{-x}}{1 + e^{-x}}.$

- Considering the problem of demixing sparse signals from their nonlinear measurements
- Specifically, studying the more challenging scenario with a limited number of nonlinear measurements
- As our contribution:
 - I proposing a fast algorithm for recovery of the constituent signals
 - **2** supporting the proposed algorithm with the rigorous theoretical analysis
 - 6 deriving nearly-tight upper bounds on their sample complexity
 - verifying experimentally the superiority of the proposed algorithms compared to existing convex demixing methods both on synthetic and real data