

Blind Signal Separation (BSS) Problem solved with NMF.

In this exercise, you are given a file named `SoundSourceData.mat`, contained into the folder `data`. You are asked to apply NMF to that problem.

We want to use NMF to solve the so-called Blind Signal Separation (BSS) Problem. In particular, we will consider the two dimensional case in which we have m sound sources inside a unit disc. The position of the sources are given by the two dimensional vectors

$$y^{(1)} = (y_1^{(1)}, y_2^{(1)}), y^{(2)} = (y_1^{(2)}, y_2^{(2)}), \dots, y^{(m)} = (y_1^{(m)}, y_2^{(m)})$$

The emitted sound is collected by a series of n microphones, located at certain coordinates

$$x^{(1)} = (x_1^{(1)}, x_2^{(1)}), x^{(2)} = (x_1^{(2)}, x_2^{(2)}), \dots, x^{(n)} = (x_1^{(n)}, x_2^{(n)})$$

positioned at the boundary of the unit disc.

Each microphone measures the sound it receives at p different amount of time, and the sound collected by the l -th microphone at time t_i is named $b_l(t_i)$.

Consider the data matrix X such that $X_{i,j} = b_i(t_j)$, of shape $n \times p$. We will suppose that the sound measured by each microphone is corrupted by noise.

Both the data matrix X of dimension $n \times p$ and the true sound source (in our example, there are $m = 4$ sound sources), are collected into the file `SoundSourceData.mat`.

1. Import the needed libraries (remember the function `scipy.io.loadmat` to load `.mat` files).
2. Use the functions of `scipy.io` to load the content of `SoundSourceData.mat`, and use it to identify the dataset X and the true sound source matrix F . Use those matrix to find the values for n and p in this case (remember that $n \times p$ is the shape of X).
3. To solve convergence problems, set equals to 0 the values contained in X which are strictly negative (those values exists because of the presence of noise).
4. Import the utility function NMF contained into the folder `./utils/NMF.py` and use it to compute the Non-Negative Matrix Factorization for the matrix X . Try different values for the number of iterations. Remember that the NMF of a positive matrix $X \in \mathbb{R}_+^{n \times p}$ is the matrix decomposition

$$X = WH \quad \text{where } W \in \mathbb{R}_+^{n \times m}, H \in \mathbb{R}_+^{m \times p}$$

that minimizes the error measured in Frobenius norm

$$\|X - WH\|_F$$

among the set of positive matrices.

Below you will find the documentation on how to use the utility function `NMF` of `NMF.py`.

The `NMF` function takes as input:

`X`: the data matrix.

`m`: the number of source (known in advance).

`T` (optional): the number of iterations of the algorithm. The Default value is `T=1000`

`tau` (optional): the stopping criterion. The Default value is `tau=1e-2`.

`return_error` (optional): a boolean variable. if `True`, also returns the error vector.
The Default value is `False`.

And returns:

`(W, H)`: tuple containing the NMF decomposition of `X`,

where the shape of `W` is `n x m`, while the shape of `H` is `m x p`.

`err` (only if `return_error = True`): an array of length `T` (the number of iterations)
that contains the behavior of the error during the iterations.

5. Using a subplot, show a 2×4 table of plots, where in the first line the rows of F are shown (the true sound sources), while in the second one, you show the rows of H (the approximated sound sources).
6. Comment the obtained results.