

Computational Mathematics for Learning and Data Analysis

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Abstract

Assigned project: ML project 6

(M1) is a neural network with topology of your choice, but mandatory piecewise-linear activation function (of your choice); any regularization is allowed.

(M2) is a standard L_2 linear regression (min least squares).

(A1) is a standard momentum descent approach applied to (M1).

(A2) is an algorithm of the class of deflected subgradient methods applied to (M1).

(A3) is a basic version of the direct linear least squares solver of your choice (normal equations, QR, or SVD) applied to (M2).

1 Introduction

This report is about the project assigned for the course of Computational Mathematics for Learning and Data Analysis. All the work in this project is the result of the knowledge gathered from the courses of **ML** and **CM**. The report contents that are not directly work of the authors is referenced and, as requested, we point to the references down to chapter and number of page (when necessary).

We start by giving a short description of the problem at hand and the methods used to solve it, including all the mathematical derivation needed to adapt the chosen methods to the problem. Next, we give a brief recap of the expected results for the experiments, properties of the problem that suits our methods and details about the solvability of our problem with the used methods. In the end, we show the achieved results, comparing them with the expected one describing which are the factors that determined a difference in the results.

The models to be implemented are:

- **M1**: a neural network, *ANN* in the following, with piecewise-linear activation function, with possible regularization;
- **M2**: standard L_2 linear regression

The methods to be applied to the models are:

- **A1**: standard momentum descent approach applied to **M1**;
- **A2**: deflected subgradient methods applied to **M1**;
- **A3**: basic version of one of the direct linear least squares solvers (i.e. normal equations, QR, SVD) applied to **M2**.

In the following we describe the main implementation choices and introduce some of the notation used in the rest of the report. The detailed description of the implemented methods is given in the related sections of this document.

2 Artificial Neural Network

The implemented *ANN* can be seen as a *fully connected multilayer Perceptron*. An *ANN* is composed by an interconnection of units, each one of them can be represented as the composition of two functions that determines the output given the fixed weight vector and the input from the previous layer. The two functions are referred as the *network function* and the *activation function*, where the former computes the scalar product of the input vector with the weight vector of the current unit, the latter is the function that directly determines the output of the current unit. In our particular case the activation function is required to be a *piecewise-linear activation function*.

The *ANN* will be structured with multiple layers, each layer will have all the units fully connected with the adjacent layers and, as convention, we refer to the first layer as *input layer* and to the last layer as *output layer*. The others are referred as *hidden layers*. Another important aspect when implementing an *ANN* is the choice of the number of units. Later in this report we show how the exact number of units in each layer is chosen, but we can already describe the structure of the input and the output layer. The *input layer* will contain a number of units that is the same as the number of features contained in the data that will be fed up to the *ANN*, instead the *output layer* will depend on the task to perform.

To simplify the development of the *ANN* we plan to fix the number of hidden layer and the number of units per layer, changing only the input/output layers depending on the task to be performed. The process involved in the determination of this characteristic of the *ANN* will be described in the testing section.

In the following sections we describe in more details which are the main aspects of the implemented *ANN* like the network structure, the functions used to compute the output of each unit and the algorithm used to let the network learn the task at hand.

2.1 Activation function

The choice of the activation function is a crucial step for the construction of the *ANN*. This function directly determines what is the output of each of the units in the network, depending on the result of the scalar product of the received input vector and the unit weights vectors.

The activation function, for this project, is required to be a *piecewise-linear function*, so the choice can be restricted between the two most popular among them:

- **ReLU:**

- defined as: $f(x) = \max(0, x)$;

- we can impose the derivative to be: $f'(x) = \begin{cases} 0 & x < 0 \\ 1 & x \geq 0 \end{cases}$

- **Leaky ReLU:**

- defined as: $f(x) = \begin{cases} \alpha x & x \leq 0 \\ x & x > 0 \end{cases}$

- also in this case we can impose the derivative to be: $f'(x) = \begin{cases} \alpha & x < 0 \\ 1 & x \geq 0 \end{cases}$

Where α determines the slope of the negative part of the function, and usually is chosen as $\alpha = 0.01$.

The main characteristics of these functions are:

- Sparse activation;
- Avoid vanishing gradient;
- Efficient computation;

These functions are widely used to deal with the problem of *vanishing gradients* when using the backpropagation algorithm described in §2.2. Our main choice is using the **Leaky ReLU** activation function due to its simplicity and to the fact that it can help avoiding the dying ReLU problem related to the *ReLU* activation function. The only problem is that it is not entirely differentiable, in particular there is only one point (i.e. $x = 0$) where the derivative is not defined. As shown in [1, Chap. 6.3] the slight modification of the derivative with $f'(0) = 1$ leads to good results and does not impair the convergence of the learning algorithm from a practical point of view. As we point out in §4.1.1, this modification is not enough to guarantee the optimization algorithm to theoretically converge.

2.2 Backpropagation algorithm

The backpropagation algorithm [see 1, Chap. 6.5.4], in a multi-layer neural network, is used to compute the gradient of the cost function. The resulting gradient is used by the learning algorithm to minimize the squared difference between the network output values $\hat{\mathbf{y}}$ and the target values \mathbf{y} associated to these outputs. The backpropagation algorithm can then be used to efficiently compute the derivative of the *ANN* seen as a composition of functions.

This algorithm is also described in [2], [3] and is composed by two main parts:

- **Forward phase:** data traverse the network from the input units to the output units, in such a way the network's output value is generated and used to compute the cost function. The procedure is shown in **Algorithm 1**.
- **Backward phase:** the error is computed by comparing the network's output with the expected one. The computed error is then propagated back to all the network's layers. At each backward step the *Chain Rule of Calculus* is used to compute the partial derivative of the unit's function related to the current layer's weights. The gradient at each layer represents how much the current units are responsible for the total error and the result of the backward phase is used by the optimization algorithm to update the weight vector of each layer. This phase is defined in **Algorithm 2**.

Algorithm 1 Forward propagation

```

1:  $\mathbf{h}_0 = \mathbf{x}$ 
2: for  $k = 1, \dots, l$  do
3:    $\mathbf{net}_k = \mathbf{b}_k + \mathbf{W}_k \mathbf{h}_{k-1}$ 
4:    $\mathbf{h}_k = f(\mathbf{net}_k)$ 
5: end for
6:  $\hat{\mathbf{y}} = \mathbf{h}_l$ 
7:  $J = L(\hat{\mathbf{y}}, \mathbf{y}) + \lambda \Omega(\theta)$ 

```

Algorithm 1 proceeds to compute the composition of functions that represents the network. The network's output value $\hat{\mathbf{y}}$ is produced by the *output layer*. Each of the \mathbf{h}_i represents the output vector coming from the layer i . Once the predicted output is produced, the algorithm proceeds into computing the loss function $L(\hat{\mathbf{y}}, \mathbf{y})$ that estimates the error for the given output vector and, by adding this value to a regularizer $\Omega(\theta)$ we obtain the total cost J .

The gradient of the cost function is computed by the next algorithm and passed to the optimizer.

Algorithm 2 Backward computation

```

1:  $\mathbf{g} \leftarrow \nabla_{\hat{\mathbf{y}}} J = \nabla_{\hat{\mathbf{y}}} L(\hat{\mathbf{y}}, \mathbf{y})$ 
2: for  $k = l, l-1, \dots, 1$  do
3:    $\mathbf{g} \leftarrow \nabla_{\mathbf{net}_k} J = \mathbf{g} \odot f'(\mathbf{net}_k)$ 
4:    $\nabla_{\mathbf{b}_k} J = \mathbf{g} + \lambda \nabla_{\mathbf{b}_k} \Omega(\theta)$ 
5:    $\nabla_{\mathbf{W}_k} J = \mathbf{g} \mathbf{h}_{k-1}^T + \lambda \nabla_{\mathbf{W}_k} \Omega(\theta)$ 
6:    $\mathbf{g} \leftarrow \nabla_{\mathbf{h}_{k-1}} J = \mathbf{W}_k^T \mathbf{g}$ 
7: end for

```

The gradient produced as a result for the *backward phase* is used by the optimization algorithm to minimize the error function via automatic fine-tuning of the weight vector. Each layer proceeds to update its weight vector depending on the layer's contribute on the total error. The way the weight vectors are updated is determined by the type of optimization methods used. Detailed informations about the way the optimization algorithms work are given in the related section.

2.3 Loss function

The loss function is used to estimate the error at the output of the network. In a supervised learning approach the main aim is the minimization of the *Loss Function* via automatic tuning of the weight w_k at each layer k . In our case this is done via *subgradient method* and *standard momentum descent*.

We use the *MSE* as a measure of error, this is the averaged sum, over all available data, of the squared differences between the predicted value and the desired one.

This is obtained by:

$$MSE = \frac{1}{2N} \sum_{p=1}^N \sum_{k=1}^K (y_k - \hat{y}_k)_p^2 \quad (1)$$

where N is the total number of examples our network is trained on and K is the total number of output units.

As described by [3, Chap. 4.3], the equation (1) changes based on the specific algorithm used to train the network. In particular for *batch learning* we have as *Loss Function* the one described above, instead for *online learning* the weights are updated on an *example-by-example* basis, so the function to be minimized is the instantaneous error computed for each pattern that flows into the network. In the following we show the derivation of the *Loss Function*, noting that for the *Stochastic learning* the minimization is performed only on the instantaneous error and for *Batch learning* the minimization is performed on the averaged error over the used patterns.

2.3.1 Derivation of the loss function

We start by defining:

$$E_{tot} = \frac{1}{N} \sum_{p=1}^N E_p$$

$$E_p = \frac{1}{2} \sum_{k=1}^K (y_k - \hat{y}_k)^2$$

where E_{tot} is the average error over all the used samples and E_p is the instantaneous error for a given pattern p .

The ∇w to be used by the optimization algorithm is equal to:

$$\nabla w = -\frac{\partial E_{tot}}{\partial w} = -\frac{1}{N} \sum_{p=1}^N \frac{\partial E_p}{\partial w} = \frac{1}{N} \sum_{p=1}^N \nabla_p w$$

The $\nabla_p w$ for a generic unit t with inputs coming from unit i is equal to:

$$\nabla_p w_{t,i} = -\frac{\partial E_p}{\partial w_{t,i}} = -\frac{\partial E_p}{\partial o_t} * \frac{\partial o_t}{\partial net_t} * \frac{\partial net_t}{\partial w_{t,i}}$$

by defining:

$$\begin{cases} o_t = f_t(net_t), \\ net_t = \sum_{j \in C} w_{t,j} o_j \end{cases}$$

where, for a generic unit t , o_t is the output of the unit, f_t is the activation function, net_t is the network function and C is the set of all the units that are giving an input to the current one, we have that:

$$\frac{\partial o_t}{\partial net_t} = f'_t(net_t), \text{ and } \frac{\partial net_t}{\partial w_{t,i}} = \frac{\partial \sum_{j \in C} w_{t,j} o_j}{\partial w_{t,i}} = o_i$$

By defining:

$$\delta_t = -\frac{\partial E_p}{\partial net_t} = -\frac{\partial E_p}{\partial o_t} * \frac{\partial o_t}{\partial net_t}$$

We have to study two different cases for $-\frac{\partial E_p}{\partial o_t}$, depending on whether o_t is the output coming from an output unit or an hidden unit.

Case t output unit:

$$-\frac{\partial E_p}{\partial o_t} = -\frac{1}{2} * \frac{\sum_{k=1}^K \partial((y_k - \hat{y}_k)^2)}{\partial o_t} = (y_t - \hat{y}_t), \text{ and}$$

$$\delta_t = -\frac{\partial E_p}{\partial net_t} = (y_t - \hat{y}_t) * f'_t(net_t)$$

We finally have that:

$$\nabla_p w_{t,i} = \delta_t * o_i = (y_t - \hat{y}_t) * f'_t(net_t) * o_i$$

Case t hidden unit: Since a generic hidden unit t contributes to the output generated by all the units k in the layer to the immediate right of t , to estimate its contribution on the network error we use the propagated error δ_k :

$$-\frac{\partial E_p}{\partial o_t} = \sum_{k=1}^K -\frac{\partial E_p}{\partial net_k} \frac{\partial net_k}{\partial o_t} = \sum_{k=1}^K \delta_k w_{k,t}, \text{ and}$$

$$\delta_t = \sum_{k=1}^K \delta_k w_{k,t} * f'_t(net_t)$$

where each δ_k is the exact result obtained in the previous step of backward computation for the units connected to t . This represents a backward step and is the core of the backpropagation algorithm.

We finally have that:

$$\nabla_p w_{t,i} = \sum_{k=1}^K \delta_k w_{k,t} * f'_t(net_t) * o_i$$

2.3.2 Properties of loss function

In this section we describe general properties of the chosen loss function, in particular discussing continuity, differentiability and convexity. We have that:

- **Continuity:** the loss function used is represented by the sum of square functions composed with the *ANN* function. Given that the *ANN* can be represented as a composition of Lipschitz continuous functions, in particular *ReLU* activation function and the linear function $W_k \hat{y}_{k-1} + b_k$ at each layer k , using [4, Claim 12.7] we can say that the network function is a Lipschitz continuous function. Considering the fact that a square function is Lipschitz continuous only if the input set is bounded, noting that the *ReLU* function output is not bounded, we can conclude that our loss function is not Lipschitz continuous unless we provide a bound on the output values of the different *ReLU* activation functions.
- **Convexity:** we first study convexity of the *ANN* by representing it as a composition of functions. The functions that builds up the network are composition of the *ReLU* activation functions, which are convex, with the linear function $W_k \hat{y}_{k-1} + b_k$ for each layer k . As seen in [5, Chap. 3.2.4], given that the composition $f = h \circ g : \mathbb{R}^n \rightarrow \mathbb{R}$ of two functions $h : \mathbb{R} \rightarrow \mathbb{R}$ and $g : \mathbb{R}^n \rightarrow \mathbb{R}$ is convex if:
 - h is convex;
 - h is increasing;
 - g is convex.

In our case, the *ANN* is convex. However, the MSE is the composition of convex functions, but noting that the square function is increasing only for positive values, it means that in our case the loss function is not convex.

- **Differentiability:** using *piecewise-linear functions* as activation functions for the *ANN* leads the loss function to be non-differentiable, however using the assumption made in §2.1, i.e. fixing the value of the activation function's derivative in the points where this is not differentiable, from a practical point of view, does not impair convergence of the algorithm and allows to use the backpropagation algorithm to compute the gradient of the network. The requirement on the differentiability of the loss functions is not needed for the subgradient method as it will be described in a later section of this document.

Further discussions about the properties of the loss function are needed for the implementation of the subgradient methods, in fact we know that it requires the optimized function to be convex and Lipschitz continuous.

3 Least Square

The *Least Square problem* is described in [6, Lecture 11] and [7, Chap. 3] as the problem of finding a solution of an overdetermined system of equations $Ax = b$ by finding a vector x that minimizes the 2-norm of the residual vector defined as $r = b - Ax$.

The *Least Square problem*, given $A \in \mathbb{R}^{m \times n}$, $m \geq n$, $b \in \mathbb{R}^m$, has the following form:

$$\text{find } x \in \mathbb{R}^n \text{ that solves the minimization problem } \min_x \|b - Ax\|_2. \quad (2)$$

We describe in section §4.3, related to the implemented direct solver, that this kind of problems have a unique solution if the matrix A has *full column rank*.

4 Methods

This section gives detailed information about the required methods that will be implemented and applied to the models described in §2 and §3. The main methods to be implemented are:

- Standard momentum descent approach applied to the *ANN*;
- Deflected subgradient method applied to the *ANN*;
- Direct linear least square solver applied to the *Least Square problem*.

4.1 Momentum method

The momentum approach is a technique that accelerates the gradient descent accumulating a velocity vector in directions of persistent reduction [8].

We can define *Classical Momentum* (CM) as:

$$v_{t+1} = \mu v_t - \epsilon \nabla f(\Theta_t) \quad (3)$$

$$\Theta_{t+1} = \Theta_t + v_{t+1} \quad (4)$$

where $\epsilon > 0$ is the learning rate, $\mu \in [0, 1)$ is the momentum coefficient and $\nabla f(\Theta_t)$ is the gradient at Θ_t . The hyperparameter of momentum determines how quickly the contributions of previous gradients exponentially decay. Another important aspect is that the larger μ is, relative to ϵ , the more previous gradients affect the current direction. At the moment, no prior choices of the momentum coefficient and learning rate can be done, these will be studied more in the detail in a later phase of the project where we implement and test these models, but as suggested by [1, Chap. 8], common values for μ are 0.5, 0.9, 0.99.

The next algorithm is the pseudocode relative to the *Gradient descent with momentum*, given the learning rate and the momentum coefficient, performs a descent approach until termination conditions are met.

Algorithm 3 Gradient descent with momentum. Termination conditions, learning rate and momentum coefficients have to be determined by testing as it will be shown in a later phase of the project. For the moment we assume that they are given.

```

1: Initialize  $\Theta$  and  $\mathbf{v}$ 
2: while termination conditions not met do
3:   Sample  $m$  examples  $(x_1, y_1), (x_2, y_2) \dots (x_m, y_m)$ 
4:    $\tilde{\Theta} \leftarrow \Theta$ 
5:   if Nesterov then
6:      $\tilde{\Theta} \leftarrow \tilde{\Theta} + \mu \mathbf{v}$ 
7:   end if
8:   Compute gradient estimate:  $\mathbf{g} \leftarrow \frac{1}{m} \nabla_{\tilde{\Theta}} \sum_i (L(f(x_i, \tilde{\Theta}), y_i))$ 
9:   Compute velocity update:  $\mathbf{v} \leftarrow \mu \mathbf{v} - \epsilon \mathbf{g}$ 
10:  Apply update:  $\Theta \leftarrow \Theta + \mathbf{v}$ 
11: end while

```

In Algorithm 3 the amount of samples taken at *line 3* determines the type of *Gradient descent algorithm*, such as:

- $\mathbf{m} = \mathbf{1}$: *stochastic gradient descent (SGD)*;
- $\mathbf{m} < \mathbf{n}$, where \mathbf{n} is the total number of examples: *batch stochastic gradient descent*;
- $\mathbf{m} = \mathbf{n}$: *standard gradient descent (GD)*.

A further improvement is given by a modification of the *CM* approach, called *Nesterov's Accelerated Gradient (NAG)* that, seen as a momentum approach, can be defined as:

$$v_{t+1} = \mu v_t - \epsilon \nabla f(\Theta_t + \mu v_t) \quad (5)$$

$$\Theta_{t+1} = \Theta_t + v_{t+1} \quad (6)$$

The only difference from *CM* is relative to the point where the gradient is computed, in fact NAG performs a partial update to Θ_t and uses this update to compute the gradient at step t . After computing the gradient, the update rule is the same, but in this way NAG allow changing v in a more responsive way. This modification is implemented in Algorithm 3 at *line 6* where an update to the point of evaluation of the gradient is performed. Differences between *CM* and *NAG* can be seen in **Figure 1**.

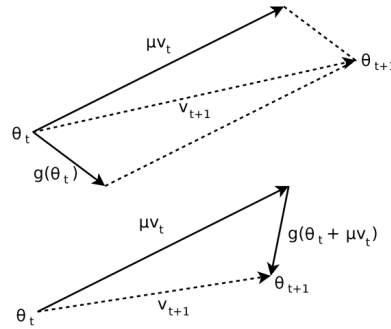


Figure 1: *CM* (on top) shows how the update of the vector v_t using the gradient at Θ_t differs from the *NAG* (on bottom). When μv_t is a poor update, we can see how *NAG* points v_{t+1} back towards Θ_t more strongly than *CM*.

4.1.1 Convergence

As shown in [8], *NAG* can help avoiding oscillations in the path taken by *CM*. In **Figure 2** we can see the comparison between the two momentum methods with same momentum and learning rate coefficients. This shows how the correction rule for a poor update, as shown in Equation 5, over multiple iterations, can help *NAG* to be more effective than *CM* at decelerating over time. This also helps *NAG* to be more tolerant to larger values of μ compared to *CM*.

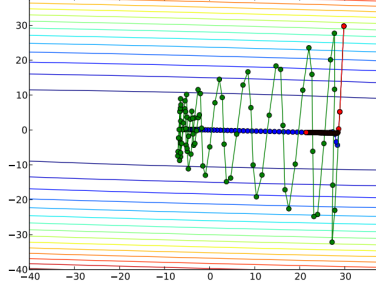


Figure 2: Given the global minimizer of the objective function in $(0,0)$, the red, blue and green curves show, respectively, the trajectories of gradient descent, *NAG* and *CM*. It's clearly visible the difference in oscillations between *CM* and *NAG* approach.

To study convergence of the *SGD* involving *CM* and *NAG*, applied to a non-convex objective function, we refer to the results obtained in [9] that describes a unified framework for both momentum methods. The unified framework uses the constant $s = 0$ and $s = 1$ to refer, respectively, to *CM* and *NAG*. By defining $C > 0$ a positive constant and $G_k = G(x_k; \xi_k)$ a stochastic gradient of $f(x)$ at x_k , depending on a random variable ξ_k , such that $\mathbb{E}[G(x_k; \xi_k)] = \nabla f(x_k)$, the following theorem shows the convergence of *SGD* with momentum for a non-convex objective function $f(x)$.

Theorem 4.1. Suppose $f(x)$ is a non-convex and L -smooth function, $\mathbb{E}[\|G(x; \xi) - \nabla f(x)\|_2^2] \leq \delta^2$ and $\|\nabla f(x)\|_2 \leq B$ for any x . Let the stochastic gradient descent method run for t iterations. By setting $\epsilon = \min\{\frac{1-\mu}{2L[1+((1-\mu)s-1)^2]}, \frac{C}{\sqrt{t+1}}\}$ we have:

$$\begin{aligned} \min_{k=0, \dots, t} \mathbb{E}[\|\nabla f(x_k)\|_2^2] &\leq \frac{2(f(x_0) - f^*)(1-\mu)}{t+1} \max\left\{\frac{2L[1+((1-\mu)s-1)^2]}{1-\mu}, \frac{\sqrt{t+1}}{C}\right\} \\ &+ \frac{C}{\sqrt{t+1}} \frac{L\mu^2(B^2 + \delta^2) + L\delta^2(1-\mu)^2}{(1-\mu)^3} \end{aligned}$$

This theorem shows that the gradient norm converges in expectation at $\mathcal{O}(\frac{1}{\sqrt{t}})$ [9]. Additionally, **Theorem 4.1** suggests that for *NAG* ($s = 1$) we can set a larger initial learning rate than that of *CM* ($s = 0$), as also shown in [9, Section 4], which leads to a faster convergence in training error.

However, noting that in our case the assumptions made for **Theorem 4.1** do not hold, in particular our objective function is not continuously differentiable and not Lipschitz continuous, we can't use this result to prove convergence for our setting. At the moment we are not able to state anything about the converge, in practice, of our algorithm with our specific setting (i.e. non-convex, non-differentiable objective function), so we postpone this discussion in the testing phase.

4.2 Subgradient Method

The *subgradient method* (*SM*) is a minimization method used to minimize non-differentiable convex objective functions. It is not a descent method, the value of the function is not decreasing at every step, in fact the direction negative to a subgradient is not necessarily a direction of descent of the function $f(\cdot)$ [10].

Given $f : \mathbb{R}^n \rightarrow \mathbb{R}$ a convex function not necessarily smooth, to minimize f the *SM* constructs a sequence of iterates $\{x_k\}$ by the iterative formula:

$$\mathbf{x}_{k+1} = \mathbf{x}_k - \alpha_k g_k$$

where $g_k \in \partial f(x_k)$ is any subgradient of $f(\cdot)$ at x_k , $\alpha_k > 0$ is the k th stepsize and x_k is the iterate at the step k . It is worthwhile to recall that a subgradient of f at x is any vector g that satisfies the inequality $f(y) \geq f(x) + g^T(y - x)$ for all y .

We show a pseudocode for the *SM*:

Algorithm 4 Basic subgradient method. Assuming starting point x_1 and subgradient at each point are given.

```

1: Initialize  $x_1$ 
2: Starting upper bound:  $UB_1 \leftarrow f(x_1)$ 
3: Starting optimal point:  $x^* \leftarrow x_1$ 
4:  $k \leftarrow 1$ 
5: while termination conditions not met do
6:   Find a subgradient of  $f$  in  $x_k$ :  $g_k \in \partial f(x_k)$ 
7:   if  $g_k = 0$  then
8:     Terminate with  $x^* = x_k$ 
9:   end if
10:  Select a direction:  $d_k \leftarrow -g_k / \|g_k\|_2$ 
11:  Select a step size:  $\alpha_k > 0$ 
12:   $x_{k+1} \leftarrow x_k + \alpha_k d_k$ 
13:  if  $f(x_{k+1}) < UB_k$  then
14:     $UB_{k+1} \leftarrow f(x_{k+1})$ 
15:     $x^* \leftarrow x_{k+1}$ 
16:  else
17:     $UB_{k+1} \leftarrow UB_k$ 
18:  end if
19:   $k \leftarrow k + 1$ 
20: end while

```

As shown in [11, Chap. 8.9], however, the stopping criterion $g_k = 0$ may never be realized because the algorithm selects the subgradient g_k arbitrarily. Usually, a stopping criterion is imposing a limit on the number of iterations performed by the algorithm. If we know the optimal value, which in general is unknown, we can impose the algorithm to stop when we reach a desired accuracy $UB_k < f^* + \epsilon$.

4.2.1 Convergence

We start with the assumption that *SMs* are feasible only for those problems that do not require a high accuracy, as shown in [12]. In fact, *SM* requires $\Theta(1/\epsilon^2)$ iterations to attain an absolute error up to ϵ . We can also note that the complexity does not depend on the size of the problem.

To study converge we first look for a bound on the distance to the optimal set, assuming that there exist an optimal solution, by [10, Theorem 7.4] we have:

Theorem 4.2. Let the subgradient method use non-negative step sizes $\{\alpha_k\}$ such that

$$\sum_{k=1}^{\infty} \alpha_k = \infty \text{ and, } \sum_{k=1}^{\infty} \alpha_k^2 < \infty. \quad (7)$$

Then the sequence $\{x_k\}$ generated by the subgradient method is convergent to a solution of the problem.

Proof. By assuming x^* is an optimal solution and considering that $f(x_k) - f(x^*) \geq 0$, then:

$$\begin{aligned} \|x_{k+1} - x^*\|_2^2 &= \|x_k - \alpha_k g_k - x^*\|_2^2 \\ &= \|x_k - x^*\|_2^2 - 2\alpha_k \langle g_k, x_k - x^* \rangle + \alpha_k^2 \|g_k\|_2^2 \\ &\leq \|x_k - x^*\|_2^2 - 2\alpha_k (f(x_k) - f(x^*)) + \alpha_k^2 \|g_k\|_2^2 \\ &\leq \|x_k - x^*\|_2^2 + \alpha_k^2 \|g_k\|_2^2, \end{aligned} \quad (8)$$

where the third equation of (8) comes from $g_k^T(x_k - x^*) \geq f(x_k) - f(x^*)$, that follows from the definition of subgradient. By induction on k we have:

$$\|x_k - x^*\|_2^2 \leq \|x_0 - x^*\|_2^2 - 2 \sum_{l=0}^{k-1} \alpha_l (f(x_l) - f^*) + \sum_{l=0}^{k-1} \alpha_l^2 \|g_l\|_2^2 \quad (9)$$

$$\leq \|x_0 - x^*\|_2^2 + \sum_{l=0}^{k-1} \alpha_l^2 \|g_l\|_2^2 \quad (10)$$

$$\leq \|x_0 - x^*\|_2^2 + \sum_{l=0}^{\infty} \alpha_l^2 \|g_l\|_2^2 \quad (11)$$

By (7), the sum in (11) is bounded, thus the sequence $\{x_k\}$ is bounded. Using the result from [10, Theorem 7.2] with *learning rate* $\bar{\tau} = 0$, there exists an infinite set of iterations \mathcal{K} such that for $k \in \mathcal{K}$, as $k \rightarrow \infty$, we have $f(x_k) \rightarrow f(x^*)$. We can choose an infinite set $\mathcal{K}_1 \subset \mathcal{K}$ such that the subsequence $\{x_k\}$, with $k \in \mathcal{K}_1$, is convergent to \hat{x} which must be an optimal solution and can be substituted to x^* . Choosing $l \in \mathcal{K}_1$, adding inequalities (8) from $k = l$ to m we obtain:

$$\|x_{m+1} - \hat{x}\|_2^2 \leq \|x_l - \hat{x}\|_2^2 + \sum_{k=l}^{\infty} \alpha_k^2 \|g_k\|_2^2 \quad m = l+1, l+2, \dots$$

For each $\epsilon > 0$ we can choose $l \in \mathcal{K}_1$ such that $\|x_l - \hat{x}\|_2^2 \leq \epsilon$, and $\sum_{k=l}^{\infty} \alpha_k^2 \|g_k\|_2^2 \leq \epsilon$. Then $\|x_{m+1} - \hat{x}\|_2^2 \leq 2\epsilon$ for all $m \geq l$, which proves that the entire sequence $\{x_k\}$ is convergent to \hat{x} . \square

To study convergence rate we refer to [13, Theorem 3.1]:

Theorem 4.3. When $f : \mathbb{R}^n \rightarrow \mathbb{R}$ is convex and its subgradients are bounded by L , for any $g \in \partial f(x)$ at any x , subgradient descent starting at x_0 s.t. $\|x_0 - x^*\|_2 \leq R$ with step size $\alpha_l \leftarrow \frac{R}{\sqrt{k} \|g_l\|_2}$ satisfy $f_k^* - f^* \leq \frac{LR}{\sqrt{k}}$ after k iterations.

Proof. Rearranging (9) we have:

$$\begin{aligned} 2 \sum_{l=0}^{k-1} \alpha_l (f(x_l) - f^*) &\leq \|x_0 - x^*\|_2^2 - \|x_k - x^*\|_2^2 + \sum_{l=0}^{k-1} \alpha_l^2 \|g_l\|_2^2 \\ &\leq \|x_0 - x^*\|_2^2 + \sum_{l=1}^{k-1} \alpha_l^2 \|g_l\|_2^2 \end{aligned} \quad (12)$$

Let $\beta_l = \alpha_l \|g_l\|_2$, $f_k^* = \min_{l < k} f(x_l)$. By the result obtained in (12) at the k -th iteration, we have:

$$2(f_k^* - f^*) \sum_{l=0}^{k-1} \frac{\beta_l}{\|g_l\|_2} \leq \|x_0 - x^*\|_2^2 + \sum_{l=0}^{k-1} \beta_l^2 \leq R^2 + \sum_{l=0}^{k-1} \beta_l^2.$$

Since the subgradients are bounded by L :

$$\frac{2}{L}(f_k^* - f^*) \sum_{l=0}^{k-1} \beta_l \leq R^2 + \sum_{l=0}^{k-1} \beta_l^2$$

By rearranging, we obtain

$$f_k^* - f^* \leq \frac{R^2 + \sum_{l=1}^k \beta_l^2}{\frac{2}{L} \sum_{l=1}^k \beta_l}$$

Since the bound is symmetric in $\{\beta_l\}$, the bound is minimized when all the β_{lS} are equal. For a given k , the bound is minimized at $\frac{R}{\sqrt{k}}$. The optimized bound is $f_k^* - f^* \leq \frac{LR}{\sqrt{k}}$. \square

Thus, *subgradient method* attains $\mathcal{O}(\frac{1}{\sqrt{k}})$ -suboptimality after k iterations, that means it obtains an ϵ -suboptimal point after at most $\mathcal{O}(\frac{1}{\epsilon^2})$ iterations.

However, as seen in §2.3.2, our loss function is not convex, so we can't state anything at this point about convergence of our setting, given that it doesn't respect the assumptions made for **Theorem 4.2** and **Theorem 4.3**. We postpone the discussion about practical convergence of the algorithm in the testing section.

4.3 Direct solver for Linear Least Square

We have chosen to implement the direct solver via *QR factorization*. This section gives a description of all the properties needed for a *least square problem* to be solved via this method and all the expected results for this kind of implementation. In a successive section we plan to insert the comparison between the theoretical results shown in this section and the actual result obtained in testing the implemented algorithm.

4.3.1 QR factorization

As described in [7, Chap. 5], *QR decomposition* (or factorization) is a factorization of a matrix A in a product of an orthogonal matrix and a triangular matrix obtained via successive orthogonal transformations.

Theorem 4.4. *Any matrix $A \in \mathbb{R}^{m \times n}$, $m \geq n$, can be transformed to upper triangular form by an orthogonal matrix. The transformation is equivalent to a decomposition*

$$A = Q \begin{bmatrix} R \\ 0 \end{bmatrix},$$

where $Q \in \mathbb{R}^{m \times m}$ is orthogonal and $R \in \mathbb{R}^{n \times n}$ is upper triangular. If the columns of A are linearly independent, then R is nonsingular.

If we partition $Q = (Q_1 \ Q_2)$ where $Q_1 \in \mathbb{R}^{m \times n}$, noting that in the multiplication Q_2 is multiplied by zero, we can write:

$$A = [Q_1 \ Q_2] \begin{bmatrix} R \\ 0 \end{bmatrix} = Q_1 R, \quad (13)$$

where equation (13) refers to the **thin QR factorization**. This form of the *QR factorization* is the one used from now on to solve the *linear least square problem*.

To implement this factorization we make use of the *Householder transformations* described in [7, Chap. 4.2.2].

Lemma 4.5. *For every $\mathbf{v} \in \mathbb{R}^m$, the matrix $\mathbf{H} = I - \frac{2}{v^T v} v v^T = I - \frac{2}{\|v\|_2^2} v v^T = I - 2u u^T$, (where $u = \frac{1}{\|v\|_2} v$ has norm 1) is orthogonal. We call \mathbf{H} an *Householder transformation*.*

Lemma 4.6. *Let x, y be two vectors such that $\|x\|_2 = \|y\|_2$. If one chooses $v = x - y$, then $H = I - \frac{2}{v^T v} v v^T$ is such that $Hx = y$.*

By choosing $y = \|x\|_2 e_1 = \begin{bmatrix} \|x\|_2 \\ 0 \\ \vdots \\ 0 \end{bmatrix}$ we can build a procedure to find the householder vector \mathbf{u} of

a generic vector \mathbf{x} . The pseudocode for this procedure is shown in Algorithm 5.

Algorithm 5 Householder vector

```

1:  $\mathbf{s} \leftarrow \text{norm}(x)$ 
2:  $\mathbf{v} \leftarrow x$ 
3:  $\mathbf{v}[1] \leftarrow \mathbf{v}[1] - \mathbf{s}$ 
4:  $\mathbf{u} \leftarrow \mathbf{v} / \text{norm}(\mathbf{v})$ 

```

Now we illustrate the method used to compute the QR factorization through the *Householder transformation*. By a sequence of orthogonal transformation we can transform any matrix $A \in \mathbb{R}^{m \times n}$, $m \geq n$,

$$A \rightarrow Q^T A = \begin{bmatrix} R \\ 0 \end{bmatrix}, R \in \mathbb{R}^{n \times n}$$

where R is upper triangular and $Q \in \mathbb{R}^{m \times m}$ is orthogonal. As shown in [7, Chap. 5.1] we can illustrate the procedure using a smaller matrix $A \in \mathbb{R}^{5 \times 4}$. Basically the algorithm proceeds by zeroing the elements under the main diagonal, where at each step i the elements below the element $a_{i,i}$ are zeroed by left-multiplying the current matrix A_i to a matrix H_{i+1} .

In the first step we zero the elements below the main diagonal in the first column:

$$H_1 A = H_1 \begin{pmatrix} x & x & x & x \\ x & x & x & x \\ x & x & x & x \\ x & x & x & x \\ x & x & x & x \end{pmatrix} = \begin{pmatrix} + & + & + & + \\ 0 & + & + & + \\ 0 & + & + & + \\ 0 & + & + & + \\ 0 & + & + & + \end{pmatrix} = A_1,$$

where $+$ denotes an element that has changed in the transformation. The orthogonal matrix H_1 can be taken equal to a *Householder transformation*. In the second step we use an embedded *Householder transformation* to zero the elements below the diagonal of the second column of matrix A_1 :

$$H_2 A_1 = H_2 \begin{pmatrix} x & x & x & x \\ 0 & x & x & x \\ 0 & x & x & x \\ 0 & x & x & x \\ 0 & x & x & x \end{pmatrix} = \begin{pmatrix} x & x & x & x \\ 0 & + & + & + \\ 0 & 0 & + & + \\ 0 & 0 & + & + \\ 0 & 0 & + & + \end{pmatrix} = A_2$$

And so on, after the fourth step we have computed the upper triangular matrix R . The sequence of transformations is summarized as:

$$Q^T A = \begin{bmatrix} R \\ 0 \end{bmatrix}, \quad Q^T = H_4 H_3 H_2 H_1.$$

Assuming $A \in \mathbb{R}^{m \times n}$ the matrices H_i have the following structure:

$$\begin{aligned} H_1 &= I - 2u_1 u_1^T, \quad u_1 \in \mathbb{R}^m \\ H_2 &= \begin{pmatrix} 1 & 0 \\ 0 & P_2 \end{pmatrix}, \quad P_2 = I - 2u_2 u_2^T, \quad u_2 \in \mathbb{R}^{m-1} \\ H_3 &= \begin{pmatrix} 1 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & 0 & P_3 \end{pmatrix}, \quad P_3 = I - 2u_3 u_3^T, \quad u_3 \in \mathbb{R}^{m-2} \end{aligned}$$

Thus vectors \mathbf{u}_i , obtained with the procedure defined in Algorithm 5, become shorter at each step and we embed *Householder transformations* of increasingly smaller dimensions in identity matrices.

4.3.2 Solving Least Square via QR factorization

In this section we show how *QR factorization*, shown in §4.3.1, can be used to solve the *linear least square problem* defined in equation (2).

In the following we use the fact that the Euclidean vector norm is invariant under orthogonal transformations, i.e. $\|Qy\|_2 = \|y\|_2$.

Theorem 4.7. *Let the matrix $A \in \mathbb{R}^{m \times n}$ have full column rank and thin QR decomposition $A = Q_1 R$. Then the least squares problem $\min_x \|b - Ax\|_2$ has the unique solution*

$$x = R^{-1} Q_1^T b.$$

Proof. Introducing the QR decomposition of A in the residual vector, we get

$$\|r\|_2^2 = \|b - Ax\|_2^2 = \left\| b - Q \begin{bmatrix} R \\ 0 \end{bmatrix} x \right\|_2^2 = \left\| Q^T b - Q^T Q \begin{bmatrix} R \\ 0 \end{bmatrix} x \right\|_2^2 = \left\| Q^T b - \begin{bmatrix} R \\ 0 \end{bmatrix} x \right\|_2^2$$

Then we partition $Q = (Q_1 \ Q_2)$, where $Q_1 \in \mathbb{R}^{m \times n}$, so we can write

$$\|r\|_2^2 = \left\| \begin{bmatrix} Q_1^T b \\ Q_2^T b \end{bmatrix} - \begin{bmatrix} Rx \\ 0 \end{bmatrix} \right\|_2^2 = \left\| \begin{bmatrix} Q_1^T b - Rx \\ Q_2^T b \end{bmatrix} \right\|_2^2 \quad (14)$$

Under the assumption that the columns of A are linearly independent and since the bottom block of equation (14) is independent from vector x , we can solve $Rx = Q_1^T b$ and minimize $\|r\|_2^2$ by making the upper block in equation (14) equal to zero. \square

As shown in [6, Lecture 10] and [7, Chap. 5.3], in the solution of (14) there is no need in computing Q because we only need the product $Q_1^T b$ to solve the reduced problem $Rx = Q_1^T b$ as described by **Theorem 4.7**. We can exploit this fact and apply the following two algorithms to obtain the upper triangular matrix R (Algorithm 6) and the product $Q_1^T b$ (Algorithm 7).

Algorithm 6 Householder QR factorization

```

1: for  $k = 1, 2, \dots, \min(m, n)$  do
2:    $[\mathbf{u}, \mathbf{s}] \leftarrow \text{householder\_vector}(\mathbf{A}_{k:\text{end}, k})$ 
3:    $\mathbf{A}_{j,j} \leftarrow \mathbf{s}$ 
4:    $\mathbf{A}_{j+1:\text{end}, j} \leftarrow 0$ 
5:    $\mathbf{A}_{j:\text{end}, j+1:\text{end}} \leftarrow \mathbf{A}_{j:\text{end}, j+1:\text{end}} - 2u(u^T(\mathbf{A}_{j:\text{end}, j+1:\text{end}}))$ 
6: end for
```

Algorithm 7 Implicit computation of $Q_1^T b$

```
1: for  $k = 1, 2, \dots, n$  do  
2:    $\mathbf{b}_{k:m} \leftarrow \mathbf{b}_{k:m} - 2\mathbf{v}_k(\mathbf{v}_k^T \mathbf{b}_{k:m})$   
3: end for
```

As a final step we give the total amount of operations needed (for a *thin QR factorization*) to solve the *Least Square problem* via QR factorization using Householder transformations. As shown in [7, Chap. 5.4], we can estimate the number of flops for computing R approximately with:

$$4 \sum_{k=0}^{n-1} (m-k)(n-k) \approx 2mn^2 - \frac{2n^3}{3} + \mathcal{O}(mn),$$

and we can state the behaviour in two common regimes:

- Square matrices ($\mathbf{m} = \mathbf{n}$): $\frac{4}{3}n^3$
- $\mathbf{m} \gg \mathbf{n}$: scales like $2mn^2$

5 Experiments

This section provides the detailed description of the datasets used to test the implemented methods together with a comparison between the performance achieved by the different models. We provide all the configuration tested and the listing of the values of different parameters in order to allow the reproducibility of the performed tests. We conclude each section showing the achieved results for all the models, comparing them to the expected ones.

In §5.1 we describe the general structure of each dataset, the main characteristics and the pre-processing operations we performed before executing the tests. We used a standardized approach for all the performed test, in particular we implemented the NN in such a way to be compatible with the `GridSearchCV` model selection class provided by the `sklearn` library. Following the same approach adopted in the *ML* course, we selected the best model by performing a *k-fold cross validation* by holding out, at each iteration, 20% of the data, resulting in a *5-fold cross validation*. The best model selected in this way is the one used in the different tests for comparison. This allowed us to discriminate between different models without the need to manually fine-tune the different hyperparameters of the different models. However, since the main goal of this project is not the one of selecting the best model on the out-of-sample data, we only used the results obtained in the first gridsearch approach to better understand which were the models that better suited our needs. We also notice, as a confirmation of the goodness of the approach and the basic implementation of our models, that the achieved results are in line with the ones we achieved during the *ML* competition.

All the tests were performed on a AMD Ryzen 7 Pro 4750u with 8 physical cores and 16 threads, with a base clock of 1.7GHz.

5.1 Datasets

All the implemented methods for the (M1) model are tested on a set of datasets that were also used for testing the implemented models during the *ML* course. We refer to the datasets as the **MONK datasets** [14]. The choice of this kind of dataset comes from the possibility to achieve good results with fairly small networks in a short amount of time, allowing us to test various configuration. Moreover, good results on these datasets can be achieved in a smaller training time with respect to other well known datasets. Moreover, in order to compare model (M1) and

(M2), described in §1, we refer to the **CUP** dataset provided for the final competition of the *ML* course.

5.1.1 MONK

The MONKS datasets are described as "*A set of three artificial domains over the same attribute space; Used to test a wide range of induction algorithms.*" in the original webpage [14].

The main characteristics of the datasets are:

- **attributes:** categorical, with a total of 7 attributes;
- **task:** binary classification task;
- **records:** total of 432 possible examples, the datasets contains:
 - **Monk1:** 124 randomly selected records;
 - **Monk2:** 169 randomly selected records where **exactly** two attributes have their *first* value (in the set of possible values);
 - **Monk3:** 122 randomly selected records, with 5% of misclassifications.
- **noise and missing values:** no missing values are present in all the datasets, only the **Monk3** dataset contains noise.

The only transformation needed to work with the given datasets is the *1-of-k encoding* that transforms each categorical attribute in a vector of 0s and 1s with a 1 in the position of the given attribute value. In such a way, the resulting inputs will be $x_i \in \mathbb{R}^{17}$. This pre-processing method is embedded in the loading function `load_MONK` in the `utils.py` file.

The adopted validation schema for these datasets makes use of the *stratification* technique, implemented by the `StratifiedShuffleSplit` class provided by the `sklearn` framework, that allows to split data for the *k-fold* schema maintaining the proportion of the different classes in all the splits over the validation phases. This allows better generalization, also considering the low amount of data available for training.

5.1.2 CUP

The CUP dataset is the dataset provided by the *ML* course of the A.Y. 2020/2021 and refers to a multi-output regression task, where each target represents coordinates in a 2D space.

The characteristics of the training dataset are:

- **task:** regression task on two real valued output variables;
- **attributes:** real valued variables, with a total of 10 attributes;
- **records:** total of 1524 training examples;
- **noise and missing values:** no missing values are present, no prior information is given about noisy data.

Given the type of variables involved in this dataset, no pre-processing of the type *1-hot* is needed. Instead, it may be useful to perform some kind of normalization or standardization of the data, but our prior knowledge coming from the *ML* competition, allows us to state that normalization and standardization approaches did not show improvements in the achieved results. Since the aim of the project was not to test the generalization capability of the selected model, we did not hold out a percentage of data as an internal test set. In this way, after selecting the best model via a *gridsearch* approach, we performed the training over the whole dataset.

In this case, for the validation schema, we used a simple *5-fold* cross validation directly implemented by the `GridSearchCV` class. No further operations are performed for the fine-tuning of the hyperparameters.

5.2 Model selection

All the models used for the performance evaluation and comparison were selected following a gridsearch cross-validation schema, specifically a 5-fold cross validation for the **CUP** dataset and a *StratifiedKFold* cross validation schema for the **MONK**s datasets. In this paragraph we show the range of tested hyper-parameters used in order to select the best model for each dataset.

Task	batch_size	ϵ	μ	λ	sizes
MONK	{10, 32, None}	[0.01, 0.1]	[0, 0.9]	[0, 0.1]	{2, 3, 5}
CUP	{32, None}	[0.01, 0.1]	[0, 0.9]	[0, 0.01]	[16, 50]

(a) SGD gridsearch ranges.

Task	batch_size	ϵ	λ	sizes
MONK	{10, 32, None}	[0.01, 0.1]	[0, 0.1]	{2, 3, 5}
CUP	{32, None}	[0.01, 0.1]	[0, 0.01]	[16, 50]

(b) SGM gridsearch ranges.

Table 1: Gridsearch ranges for both SGD (a) and SGM (b) over **MONK** and **CUP** datasets. The search was performed imposing as stopping condition, for both optimizers, a number of epochs equal to **1000** and a precision computed as norm of the gradient of **1e−4**. *Note: the sizes specified in both (a) and (b), for the CUP row, refers to the amount of units used in each of the hidden layers of the tested network.*

The best models resulting from the searches specified in **Table 1** where used to perform the comparisons in §5.3. SGD models are tested also with the *NAG* variant, as described in **Eq. 5**.

5.3 Results

In this section we show, for each implemented model, the achieved results and we compare them with the expected ones. We also show a comparison with the two models (**M1**) and (**M2**) on the **CUP** dataset.

For each tested model, we consider the achieved gradient norm and the score in terms of *MSE* as performance metrics. For all the models based on a NN implementation, we imposed a different stopping condition for the two datasets, in particular:

- **MONK**: stopping condition only based on the norm of the gradient, without selecting a particular amount of epochs;
- **CUP**: stopping condition based also on the number of epochs, this was due to the fact that a given accuracy on the norm of the gradient can't be reached by all the used models.

5.3.1 MONK

In this section we show the comparison of the different models over the three **MONK** datasets. In particular we compare the achieved losses and gradient, considering the number of epochs and the amount of time necessary to achieve the desired precision.

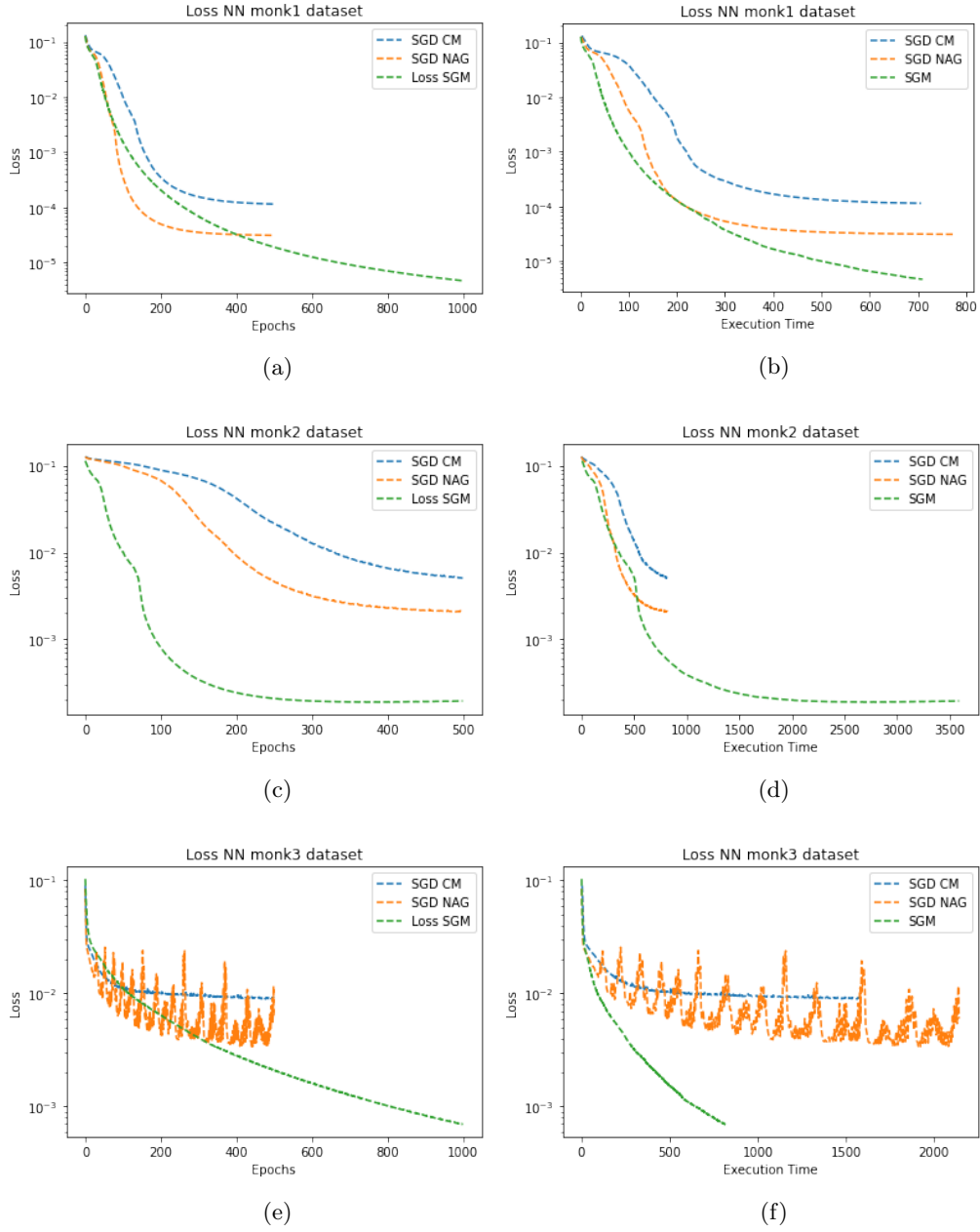


Figure 3: Comparison between the three tested models **SGD** (both with *CM* and *NAG*) and **SGM**. On each row we show the comparison on a different **MONK** dataset and the two columns represent, from left to right, the loss w.r.t. the number of epochs and the required time in milliseconds.

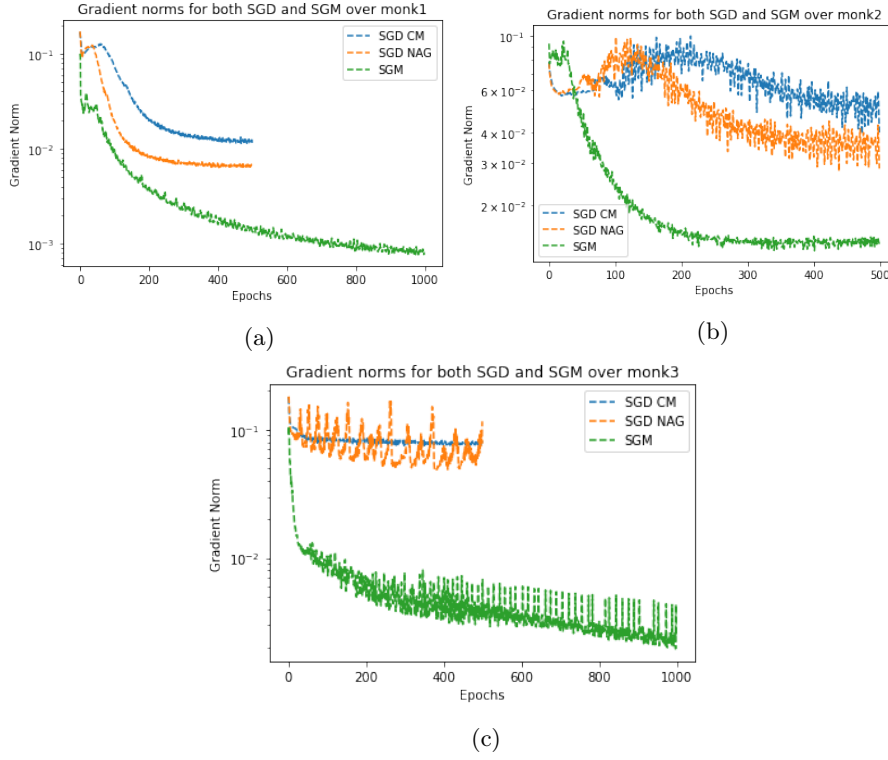


Figure 4: Gradient norm comparison over the three **MONK** datasets for the three tested models.

Task	optimizer	batch_size	ϵ	μ	λ	sizes	∇f_*	f_*
MONK1	CM	32	0.1	0.9	0.1	5	1.271e-2	1.145e-4
MONK1	NAG	32	0.1	0.9	0.01]	5	6.801e-3	3.088e-5
MONK1	SGM	None	0.1	-	0.01	5	8.020e-4	4.662e-6
MONK2	CM	32	0.1	0.5	0.01	3	5.328e-2	5.079e-3
MONK2	NAG	32	0.1	0.5	0.01	3	4.719e-2	2.090e-3
MONK2	SGM	10	0.1	-	0.01	3	1.935e-2	1.898e-4
MONK3	CM	10	0.1	0.9	0.01	5	1.635e-2	8.945e-3
MONK3	NAG	10	0.1	0.9	0.01	5	7.106e-3	3.391e-3
MONK3	SGM	None	0.1	-	0.01	5	2.449e-3	6.921e-4

(a) Test results over **MONK** datasets. Models chosen via gridsearch as shown in §5.3

Task	optimizer	epochs	total	BP	EP
MONK1	CM	500	601.93	0.078	1.203
MONK1	NAG	500	497.99	0.061	0.995
MONK1	SGM	1000	604.02	0.228	0.604
MONK2	CM	500	693.86	0.062	1.387
MONK2	NAG	500	924.36	0.078	1.848
MONK2	SGM	500	3496.17	0.210	6.992
MONK3	CM	500	1360.03	0.059	2.720
MONK3	NAG	500	1772.94	0.074	3.545
MONK3	SGM	1000	751.25	0.277	0.751

(b) Execution statistics relative to the models used for testing in table 2a.

Table 2: Execution results and performances for the models selected via gridsearch.

In **Table 2a** we clearly see an increased cost for the *backpropagation* computation for the **SGM** model. This was due to the fact that this model requires to update the stepsize at the end of each iteration in order to converge to a solution. This inevitably introduces a bottleneck in the computation of the backpropagation algorithm compared to the **SGD** model.

5.3.2 CUP

In this section we show the comparison of different models over the CUP dataset. We show the achieved loss and gradient norms for all the tested models.

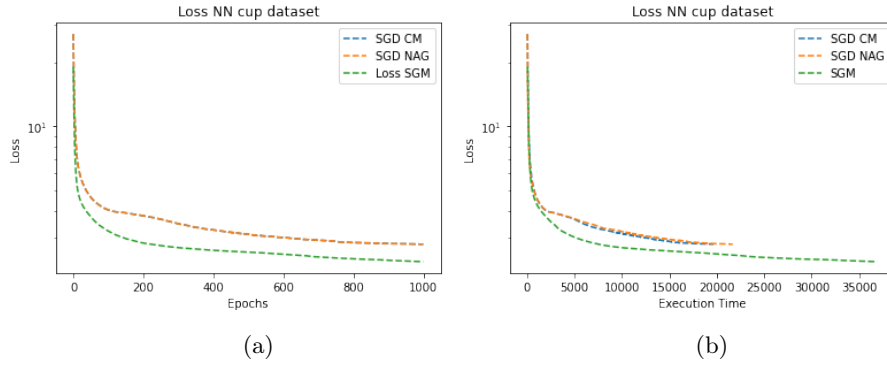


Figure 5: Comparison between the three tested models **SGD** (both with *CM* and *NAG*) and **SGM** over the CUP dataset. Image **(a)** and **(b)** show, respectively, the loss w.r.t. the number of epochs and the required time in milliseconds.

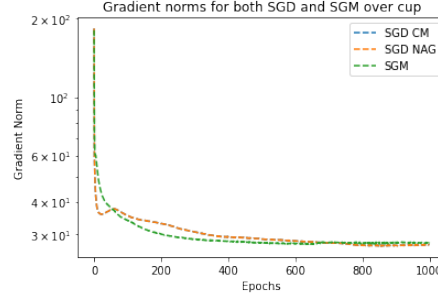


Figure 6: Gradient norm comparison over the **CUP** dataset for the three tested models.

Task	optimizer	batch_size	ϵ	μ	λ	sizes	∇f_*	f_*
CUP	CM	32	0.1	0.9	0.1	5	1.271e-2	1.145e-4
CUP	NAG	32	0.1	0.9	0.01]	5	6.801e-3	3.088e-5
CUP	SGM	None	0.1	-	0.01	5	8.020e-4	4.662e-6

(a) Test results over **CUP** dataset. Models chosen via gridsearch as shown in §5.3

Task	optimizer	iterations	total	BP	EP
CUP	CM	1000	20059	0.187	20.059
CUP	NAG	1000	20061	0.182	20.061
CUP	SGM	1000	33282	0.526	33.282

(b) Execution statistics relative to the models used for testing in table 2a.

Table 3: Execution results and performances for the models selected via gridsearch.

5.4 Direct Solver

In this section we show the main results obtained with the implemented direct solver and we compare them with the ones from the out-of-the-box solver from `Numpy` library. As we show in the following paragraphs, our solver is able to achieve the expected precision and the achieved results are comparable with the ones obtained with the `Numpy` ones. However, we can't compare our solver to the `Numpy` one under an execution efficiency standpoint, since our solver lacks all the necessary optimizations needed to achieve the same performances as the library we used to compare our model with.

5.4.1 QR implementation comparison

In this section we show the comparison between the implemented algorithm **(A3)**, needed as an intermediate step for the LS solution as specified in §4.3, and the `Numpy` out-of-the-box solver for the QR computation.

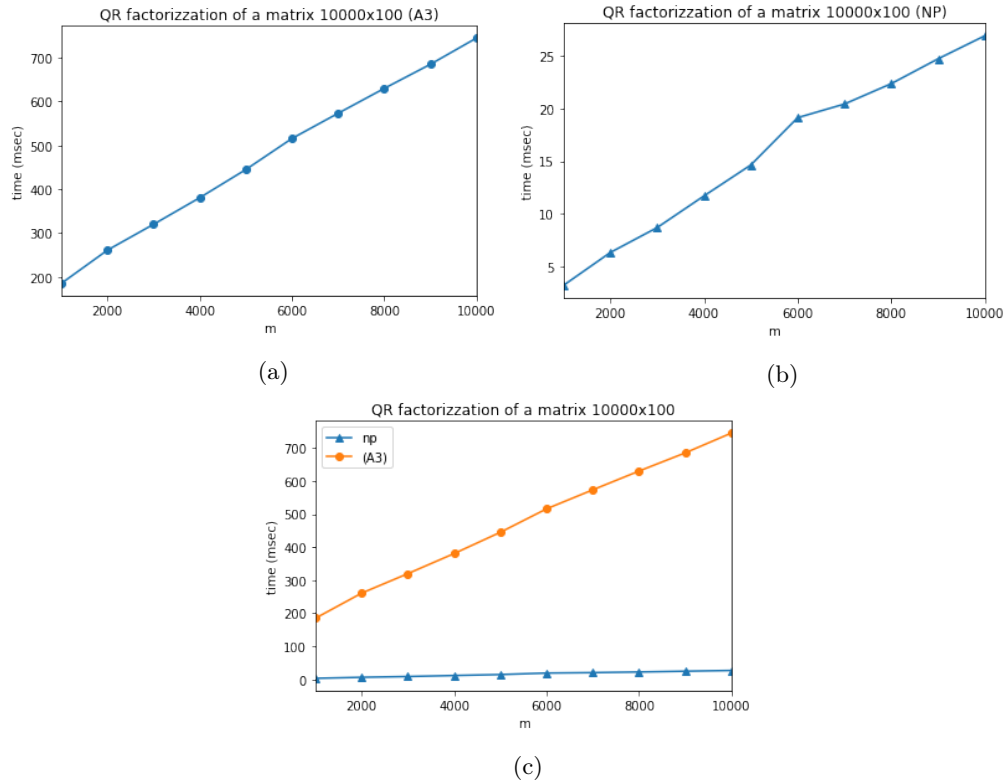


Figure 7: Comparison between the implemented **(A3)** algorithm with the `Numpy` off-the-shelf solver. In figure **8a** and **8b** there are, respectively, the performances in QR computation for the implemented **(A3)** solver and the `Numpy` one over datasets with increasing dimensions, up to 10000x100. In **8c** the same plots are put together to highlight the difference in performance.

m	time (A3)	delta (A3)	time (NP)	delta (NP)
1000	185.1129	-	3.1858	-
2000	260.9890	75.8761	6.3108	3.1250
3000	319.9584	58.9694	8.6793	2.3686
4000	380.6262	60.6678	11.6935	3.0141
5000	444.9249	64.2987	14.6366	2.9432
6000	515.8109	70.8860	19.1323	4.4956
7000	573.2926	57.4817	20.4345	1.3022
8000	630.0169	56.7243	22.3729	1.9384
9000	685.0543	55.0375	24.7365	2.3636
10000	745.0878	60.0335	26.9332	2.1967

Table 4: Scaling performances of the implemented algorithm (**A3**) and the **Numpy** one. Time is expressed in milliseconds.

As we can see in both **Figure 7** and **Table 4**, the implemented algorithm for QR computation scales linearly with the m dimension of the matrix, as expected theoretically and as shown at the end of §4.3.2 paragraph. However, given that our algorithm does not make use of extensive optimizations, this reflects on the completion time required for the computation of the QR factorization.

5.4.2 LS implementation comparison

In this paragraph we show the comparison between the implemented algorithm (**A3**) and the **Numpy** solver. We used a sequence of randomly generated matrix with increasing dimension m in $[1000, 10000]$, in order to highlight the scaling performances and show that they are in line with the theoretically expected ones. We decided to fix the n dimension to 100 in order to allow the algorithm to terminate in a reasonable amount of time on the machine at our disposal.

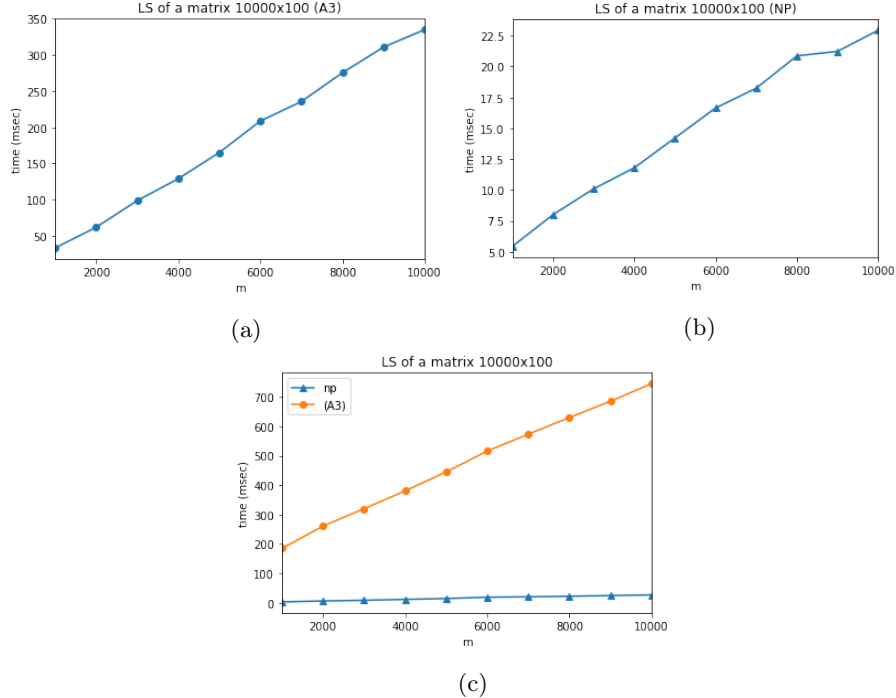


Figure 8: Comparison between the implemented (**A3**) algorithm with the **Numpy** off-the-shelf solver. In figure **8a** and **8b** there are, respectively, the performances in LS computation for the implemented (**A3**) solver and the **Numpy** one. In **8c** the same plots are put together to highlight the difference in performance.

m	time (A3)	delta (A3)	time (NP)	delta (NP)
1000	33.2898	-	5.4329	-
2000	61.8065	28.5167	8.0054	2.5725
3000	98.6733	36.8669	10.0943	2.0889
4000	128.8937	30.2204	11.8025	1.7082
5000	165.1421	36.2484	14.2208	2.4183
6000	208.6527	43.5106	16.6491	2.4283
7000	235.6089	26.9562	18.2518	1.6028
8000	275.4459	39.8371	20.8711	2.6193
9000	310.8279	35.3820	21.2367	0.3656
10000	334.9303	24.1024	22.9452	1.7085

Table 5: Scaling performances of the implemented algorithm (A3) and the Numpy one. Time is expressed in milliseconds.

We can clearly see, both in **Figure 8** and **Table 5**, that the algorithm linearly scale with the m dimension, as expected. Once again, the comparison under a computational efficiency standpoint can't be done properly, since our algorithm requires 10 times more with respect to the Numpy one.

Finally, in table **Table 6** we can see the results comparison in the Least Square solution for the CUP dataset by using the implemented solver and the one offered by the Numpy library. We also show the error in the reconstruction of the original matrix via QR factorization and we can clearly see the goodness of the implementation with reference to highly optimized solvers.

Task	residual (A3)	residual (NP)	QR to A (A3)	QR to A (NP)
CUP	1.0538	0.9963	4.93339e-16	3.28932e-16
RANDOM	1.0047	0.9953	7.89014e-16	3.92280e-16

Table 6: Residual and reconstruction relative errors for the (A3) algorithm and the Numpy one.

However, even if the reconstruction error is up to machine precision, we can't state the same for what concerns the residual error for both of the compared algorithms. Since the problems are far from being linear, the solution found by the LS solver we used is very poor and it is close to the one achieved on a random dataset generated with a gaussian distribution.

As a final test, we checked if the precision in the results was maintaining at a stable and reasonable level also for bigger matrices. In **Table 7** we show the relative errors of the residual and the reconstruction error for datasets with the m dimension in $[10000, 50000]$ with scaling factor of 10000 while keeping the n dimension once again fixed to 100.

m	residual (A3)	residual (NP)	QR to A (A3)	QR to A (NP)
10000	1.004136	0.995936	1.148071e-15	5.144102e-16
20000	1.001552	0.997711	7.093753e-16	5.028219e-16
30000	1.001714	0.998399	6.781397e-16	4.979521e-16
40000	1.001585	0.998724	7.984209e-16	4.945500e-16
50000	1.001005	0.999045	8.001721e-16	4.950132e-16

Table 7: Scaling performances of the implemented algorithm (A3) and the Numpy one.

As we can see, also for this kind of test, the implemented algorithm attains a similar precision to the one provided by the Numpy library.

6 Conclusions

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Todo list