Network and Practical

Introduction to ML for Natural Language Processing

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Multi-class Classification of European Laws in Five Languages How do Neural Networks and Statistical Models Behave?



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1 Introduction

Natural Language Processing (NLP) originated in the 1950s as the intersection of artificial intelligence and linguistics [12]. It arose in response to the many challenges presented by natural language.

The ambiguity of human language makes it extremely difficult to write software that accurately determines the meaning of text or speech data. Creating hand-written algorithms and rules that can work on any text is very complex, given the presence of homonyms, homophones, sarcasm, unspoken content, idioms, metaphors, grammar and usage exceptions in virtually any text.

NLP combines computational linguistics with statistical models and machine learning in order to successfully process human language in the form of text or speech, understanding its meaning in ways that traditional algorithms are unable to do [10].

Natural Language Processing is used to solve a wide range of tasks: in this work we are interested in the one of multi-class text classification. Furthermore, our attention will be drawn to a particularly problematic aspect of the NLP field: the fact that most algorithms are designed and implemented for only 7 of the more than 7000 languages spoken in the world. These are English, Chinese, Urdu, Farsi, Arabic, French, and Spanish [17].

Our objective aims to test whether different languages can influence architectures such as neural networks or statistical models. To this end, we used a dataset containing 65000 European laws, officially translated into several languages, on which we were able to compare the behaviour of various architectures, both with regard to fine-tuning and their performance.

The project code is public and available on Github.

2 Data

The chosen dataset is called *MultiEURLEX* and is available on HuggingFace. It comprises 65000 European laws officially translated into 23 languages.

Of these available languages, five were selected, based on their adoption at European level:

- English, with 51% EU speakers
- German, with 32% EU speakers
- *Italian*, with 16% EU speakers
- Polish, with 9% EU speakers
- Swedish, with 3% EU speakers

For each text in the dataset, there are one or more labels annotated. However, as can be understood by reading the paper on the dataset, these annotations are *granular* [8]. This allowed us to keep only the first level of labels, transforming the problem from a multi-label task to a multi-class one. This operation both simplified the classification and facilitated the evaluation of the architectures results.

This resulted in 21 different classes being initially present, shown in Figure 1 together with the number of occurrences (texts) available for each of them.

A preliminary analysis of the data, however, revealed that each law averaged around 1000 words, with slight fluctuations depending on the considered language. Being aware that this factor would be computationally burdensome during the training phase of the models, it was decided to choose 2000 random laws from each of the three classes with the most observations, i.e. class 2, 3 and 18.

Number of Texts for each Label

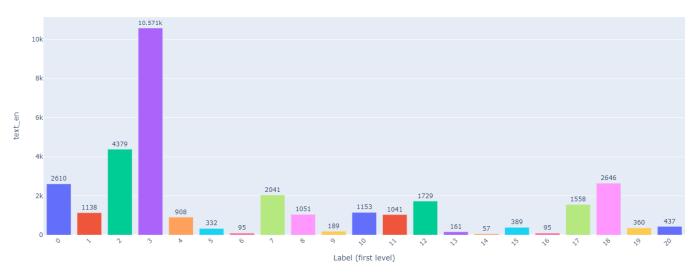


Figure 1: MultiEURLEX Dataset - Distribution of Classes and Texts Occurrences

Ultimately, the final dataset consists of 6000 laws, officially translated into 5 languages and belonging to 3 distinct classes.

3 Architectures

In order to test the impact of different languages, various architectures were chosen, belonging both to the world of neural networks and statistics.

For reasons of limited space, one architecture per typology is presented in detail: for neural networks, the functioning of an LSTM is explained in depth; for statistical models, the Linear Support Vector Classifier is illustrated. Nevertheless, the CNN architecture is briefly introduced.

3.1 Long Short Term Memory

The Long Short Term Memory (LSTM) is a kind of recurrent neural network (RNN). RNNs are a type of networks particularly suited to the use of sequential data, or time series.

A distinguishing characteristic from the traditional neural networks is their capacity to have a *memory*: they use information obtained from previous inputs to modify their behaviour on the current input and output. They keep the information of the former inputs in memory for an unspecified time, which depends on the weight of the network nodes and the specific data. Recurrent neural networks do not assume that input and output are independent, as traditional networks do, but make their output depend on the previous elements in the sequence. Figure 2 shows the typical structure of a feedforward neural network on the left, and that of a recurrent neural network on the right.

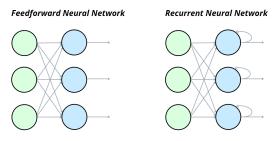


Figure 2: Structure of Feedforward and Recurrent Neural Networks

Recurrent neural networks use an algorithm of backpropagation through time to determine gradients. Backpropagation is at the basis of neural networks training, making possible the fine tuning of their weights, based on the results obtained in the previous epochs. The gradient refers to the derivative of a function that has more than one input variable; it measures the change in all the weights of the network with respect to the change error.

During this operation, recurrent neural networks tend to encounter two types of problems, both arising from the size of the gradient:

• Vanishing Gradients

In some circumstances, the gradient is very small and continues to decrease until it becomes extremely close to 0. When this happens, the model stops learning, as its weights are no longer updated.

• Exploding Gradients

Conversely, the gradient can become extremely large, altering the network weights unreasonably. This causes the model to become unstable and consequently underperforming.

Long Short Term Memory networks were conceived in response to the problems of vanishing and exploding gradient, proposing a possible solution to these issues [4]. A clear description of the inner workings of LSTMs can be found in the paper by Graves et al. [9] where it is detailed that:

"An LSTM layer consists of a set of recurrently connected blocks, known as memory blocks. These blocks can be thought of a differentiable version of the memory chips in a digital computer. Each one contains one or more recurrently connected memory cells and three multiplicative units - the input, output and forget gates - that provide continuous analogues of write, read and reset operations for the cells."

In order, what happens is:

- 1. The cell input is multiplied by the activation of the input gate;
- 2. The model output is multiplied by the output gate activation;
- 3. Finally, the previous cell values are multiplied by the forget gate values.

The network is capable of interacting with its cells exclusively through the use of gates.

LSTMs achieve remarkably high performance on major and complex tasks; these include language modelling, speech recognition and machine translation [18].

3.2 Convolutional Neural Network

Convolutional neural networks (CNN) are one of the most renowned algorithms of deep learning. Under the deep learning category fall all those artificial neural networks that have more than one layer (and are thus called multi-layer architectures).

The layers of CNNs are usually the following: Convolutional Layer, Non-Linearity Layer, Pooling Layer and Fully Connected Layer. We will explain part of these layers in subsection 4.3, concerning the implementation of architectures.

The models that use convolutional neural networks have shown over the years that they are able to achieve state of the art results in many Machine Learning problems. CNNs perform particularly well in image classification and computer vision tasks, but they have also proven powerful in problems of Natural Language Processing [1].

3.3 Linear Support Vector Classifier

In order to fully understand the idea behind the Linear Support Vector Classifier, it is first necessary to introduce some concepts that form its foundation.

First, let us introduce the notion of *hyperplane*. Intuitively, it is possible to think of the hyperplane as something that divides space into two sides, where it is straightforward to determine in which of the two sides a point is located. An example of hyperplane in two-dimensional space can be observed in Figure 3.

The visualisation also suggests how a hyperplane can be used to classify different observations. Let's suppose that the points in the figure represent the training set, i.e. the data initially available for training a model. It is then possible to construct a classifier in a very straightforward way: new points will be classified into green or yellow observations according to their position in space, given by X_1 and X_2 .

Depending on how close the new observation is to the hyperplane it is also possible to have a certain degree of confidence in the correctness of the prediction. For points close to the separating hyperplane we are less certain about their class than for those further away.

There can exist an infinite number of different hyperplanes able to separate the green observations from the yellow ones. For this reason it is necessary to decide how to choose the *best* hyperplane. An appropriate choice is to use the *maximal margin hyperplane*, which is the hyperplane most distant from the training observations. It depends entirely on the so-called *support vectors*, i.e. the points closest to it that influence its position and orientation. See Figure 4 for an example.

Until now we have discussed a binary case in which the two classes were linearly separable. In the event that they are not, as it happens in many real world cases, it is no longer possible to find the maximum margin hyperplane. Even if it were possible to find it, using it is not always the best solution: trying to perfectly separate the observations of the two classes makes the maximum margin hyperplane very sensitive to individual observations. It may be a better solution, in many cases, to consider a hyperplane that does not perfectly separate the training observations, but that allows to obtain a higher robustness regarding the individual observations and that correctly classifies most of the training observations.

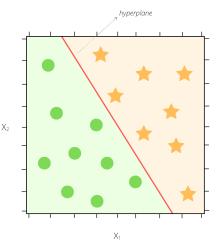


Figure 3: Hyperplane in 2D case

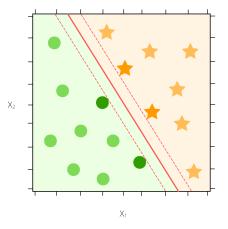


Figure 4: Maximum Margin Hyperplane and Support Vectors (highlighted points)

This is indeed the objective of the Linear Support Vector Classifier. The model purposely allows the misclassification of certain observations, in order to correctly classify as many points as possible, while maximising the width of the margin.

Since in our specific case we are working with 3 classes, we are not in a binary classification scenario. We have therefore extended the binary approach to a multiclass approach using the scheme *One-VS-Rest*. It is a heuristic that allows us to split our multiclass dataset into multiple binary classification problems.

4 Experimental Setup

4.1 Data Cleaning and Preprocessing

The majority of data is prone to the presence of errors and noise. This is particularly true in the field of Natural Language Processing. Therefore, textual data must be subjected to pre-processing procedures before being given to the models.

The first critical step is data cleaning. In addition to the usual procedures such as removing duplicates, missing data, etc., when working with textual data, one often has to deal with texts that contain spelling mistakes, the use of jargon and symbols that are not useful for learning models. In our specific case, using a dataset of officially translated European laws, we do not have to worry about these issues.

Furthermore, texts always contain *stopwords*, commonly used terms that do not provide information, and which it is common practice to remove. However, these removal procedures use lists of words considered to be stopwords, but there are no universal rules. In order not to give any advantages to the most frequently used languages, such as English and German, we have decided not to remove them from the texts. Furthermore, working with recurrent neural networks like LSTMs, stopwords could increase the contextual sense of the networks and, contrary to what usually happens in other tasks, improve the performance of the models.

Another technique used when working with texts and natural language is to apply stemming or lemmatization procedures. These are described below.

• Stemming

As can be found in the article Development of a stemming algorithm [11]:

"A stemming algorithm is a computational procedure which reduces all words with the same root (or, if prefixes are left untouched, the same stem) to a common form, usually by stripping each word of its derivational and inflectional suffixes."

• Lemmatization

It is a procedure that removes inflectional endings and returns the base or dictionary form of a word.

This technique ensures that similar words are reduced to the same word, which often leads to higher performance in models [2].

However, the same logic of stopwords applies to stemming and lemmatization procedures, which are therefore not applied to texts.

It is only fair to mention that the initial dataset has a very high quality: it was thus only necessary to lowercase all words, remove symbols and non-unicode characters, in order to ensure the proper functioning of future architectures.

4.2 Feature Extraction

Feature extraction describes the process of transforming raw data into numerical features that can be processed while retaining the information in the original dataset.

There are several ways of extracting information from data, and they change depending on the type of input that is being handled. In the context of this work, the primary focus is on the extraction of features from natural language. The procedures employed are outlined below.

Feature Extraction for Neural Networks

With regard to the creation of features for the neural networks used, namely long short term memory and convolutional neural network, we performed the following procedures:

• Tokenization

It involves breaking up a raw text into smaller pieces. In our case, we break down each law text into words, the tokens.

• Count Token Occurrences

We count the number of occurrences of each token (word) in our five corpuses.

This is where we noticed the first difference between the languages: the number of different unique words used changes considerably, as can be seen in the Table 1. The German language, for example, uses almost three times as many words as English to describe the same laws.

Language	Unique	Words
English	31454	
German	87084	
Italian	44570	
Polish	74691	
Swedish	79973	

Table 1: Total number of unique words in each language corpus.

• Removal of Uncommon Words

The number of different words in the texts is very high. This factor would certainly negatively affect the performance of the architectures or, at the very least, make training times extremely long.

In order to avoid these possible drawbacks, all words that did not appear in the corpus at least 100 times were removed. Table 2 shows the number of words left for each language after this operation.

Language	Unique Words
English	3506
German	4216
Italian	4180
Polish	5255
Swedish	4010

Table 2: Total number of unique words in each language corpus that appears at least 100 times.

After this adjustment, the disparity between languages in the number of different words has softened, but is still present.

• Encoding

In this last step, we created a mapping between vocabulary and index, in order to use it in the encoding of the texts. Here, we had to set a maximum token number for each text. Since the average number of words per law is about 1000, we used this value as an upper limit.

Feature Extraction for Linear Support Vector Classifier

Text Vectorization is a technique that consists of transforming textual data into numerical vectors that can be understood by machine learning architectures.

One of the most widely adopted techniques for text vectorization is *Term Frequency - Inverse Document Frequency* (TF-IDF), a statistical measure that assesses how relevant a word is to a document in a collection of documents.

In the case of tickets classification, the text of each ticket represents one document.

The TF-IDF score for a word is obtained by multiplying the two metrics that compose the algorithm:

• Term Frequency (TF)

In its simplest form, it refers to the number of times a word appears in a document.

• Inverse Document Frequency

It refers to how common a word is in the entire set of sentences. The more frequent a word is, the closer its value is to 0.

It is possible to calculate the IDF of a word by obtaining the total number of documents, dividing it by the number of documents containing that given word and computing the logarithm.

Multiplying these two values makes it possible to obtain the TF-IDF of a word in a document. The higher the value, the greater the relevance of the word within a document.

The TF-IDF is an efficient solution to the problem of *stopwords*: since they are frequent words in all documents, they have a value very close to zero and do not become influential for the models.

If a word is repeated many times within the same document, on the contrary, it might indicate its importance. The algorithm is able to detect these subtleties and is a particularly effective way of preparing the input for future architectures.

4.3 Implementing the Architectures

A brief description of the implementation of the architectures presented in Section 3 follows. In the case of neural networks, we focus on the employed layers.

Long Short Term Memory

The architecture was implemented using the PyTorch framework. In the development of the network, the following layers were used:

• Embedding Layer

This layer is often used to store and retrieve word embeddings using indexes. The input of the module is a list of indexes and the output is the corresponding word embedding [14].

• LSTM Layer

As one can easily deduce from its name, this module is the heart of the network. It applies a multi-layer long short-term memory to an input sequence [16].

• Linear Layer

Its purpose is to apply a linear transformation to the input data [15].

• Dropout Layer

During the training phase, some elements of the input tensor are zeroed with probability p, using samples of a Bernoulli distribution.

This technique has proven to be effective in regularising and preventing co-adaptation of neurons [13].

Convolutional Neural Network

This architecture, like the illustrated one above, has also been implemented using the PyTorch framework. Note that some layers are the same as those used for the LSTM and will thus not be explained again.

- Embedding Layer
- Convolutional 1D Layer

Convolutional layers are one of the main building blocks used in convolutional neural networks. The term *convolution* refers to the application of a filter to an input, which then produces an activation. A repeated application of the same filter to an input produces an activation map, also called a feature map, which indicates the position and strength of a detected feature in an input.

It is a linear operation involving the multiplication of a set of weights with the input, a common operation in neural networks [6].

• Max Pooling 1D Layer

Pooling layers make it easy to down-sample feature maps. They summarise the presence of features in patches of the feature map.

One of the most widely used pooling levels is max pooling, partly due to its high performance. In this instance, the pooling operation calculates the maximum value in each patch of each feature map [5].

- Linear Layer
- Dropout Layer

Linear Support Vector Classifier

This architecture was developed using the scikit-learn framework. Since it is a more high-level library than Pytorch, the implementation was simpler and more straightforward. Most of the work consisted in obtaining suitable features for the model, which was performed via the TF-IDF algorithm presented in subsection 4.2.

4.4 Training Regime

During the course of the project, particular emphasis was placed on fine-tuning the models. A detailed explanation of what was performed for each architecture follows.

Dataset Splits

Prior to proceeding with the model training phase, each dataset was divided upstream into three sets:

- training set (3840 texts)
- validation set (960 texts)
- *test set* (1200 texts)

Saving the Best Model

<u>Neural Networks</u> During each epoch the models were trained on the training set and validated on the validation one. As loss we have opted for *cross entropy*, which is considered suitable for cases of multiclass classification, both when working with statistical models and NNs [3].

We implemented an automatic mechanism capable of saving the model weights of the best epoch based on the average accuracy of the three classes, along with a dataframe containing the training results. The latter was employed to create data visualizations and better understand what occurred during the training phase, thus making it possibile to observe the differences between languages. We will further discuss training results in Section 5.

<u>Statistical Models</u> Saving the results of the Linear Support Vector Classifier was straightforward, as we did not have to handle the different epochs. The *squared hinge loss*, proposed as default by the **sklearn** framework, was chosen.

Long Short Term Memory

For each of the five available languages, 16 models were trained, i.e. the combination of a grid search performed on the hyperparameters shown in Table 3.

Hyperparameter	Employed By	Values Tested	
Embedding Dim	Embedding Layer	1024, 2048	
$Hidden\ Dim$	LSTM Layer	1024, 2048	
$Dropout\ Probability$	Dropout Layer	0.0001,0.001	
Learning Rate	Adam Optimizer	0, 0.1	

Table 3: Hyperparameters tested for LSTM networks

Each model was trained for a total of 50 epochs, using a batch size of 64. As a note, we are aware that in order to obtain more robust and reliable results it would have been ideal to increase the number of hyper-parameter values tested, along with the number of epochs. Training times, however, would have risen exponentially. Each complete search grid, which trains 16 models over 50 epochs, takes between 14 and 16 hours on an RTX2070 Super GPU, varying according to the language of the texts.

Convolutional Neural Network

Convolutional neural networks proved to be less computationally demanding than LSTMs. This allowed us to test a higher number of hyperparameters, with a total of 96 models per language (Table 4).

Hyperparameter	Employed By	Values Tested		
Embedding Dim	Embedding Layer	1024, 2048		
$Kernel\ Size$	CNN Layer, Max Pooling Layer	3, 5, 7		
Stride	CNN Layer, Max Pooling Layer	1, 3		
Padding	CNN Layer	1, 3		
$Dropout\ Probability$	Dropout Layer	0.0001,0.001		
Learning Rate	Adam Optimizer	0, 0.1		

Table 4: Hyperparameters tested for Convolutional Neural Networks

As with the LSTMs, the CNNs were also trained for 50 epochs, using a batch size of 64.

Linear Support Vector Classifier

For this architecture, there are no performance concerns. We therefore trained 7 different models for each language, each with a different value of C, the model's only hyperparameter.

C is the parameter responsible for regularization, which is inversely proportional to its value. Thus, for high values of C the optimiser will choose a hyperplane with a narrower margin if it correctly classifies all points in the training set. Conversely, a small value of C will cause the optimiser to look for a separating hyperplane with a larger margin, even if this hyperplane incorrectly classifies a greater number of points.

The values that we tested for the C hyper-parameter are the following: 0.1, 1, 5, 10, 50, 100, 1000.

5 Results

5.1 Training Times

Throughout this project, we have always kept track of the training times of the models. We believe that this can provide useful insights regarding the differences that are present due to the languages, as well as those present due to the different architectures. A table of training times is shown in Figure 5.

	[LSTM] Training Times									
Language	Fine-Tuning Time	Mean Time per Model	Mean Time per Epoch							
English	15h 44m (944m)	0h 59m	1m 11s							
German	16h 36m (996m)	1h 02m	1m 14s							
Italian	14h 20m (860m)	0h 54m	1m 05s							
Polish	15h 38m (938m)	0h 59m	1m 10s							
Swedish	16h 12m (972m)	1h 01m	1m 13s							

[CNN] Training Times									
Language Fine-Tuning Time Mean Time per Model Mean Time per Epoch									
English	2h 06m	1m 19s	1,58s						
German	2h 12m	1m 23s	1,66s						
Italian	2h 06m	1m 19s	1,58s						
Polish	2h 10m	1m 21s	1,62s						
Swedish	1h 58m	1m 14s	1,48s						

[LSVC] Training Times							
Language	Fine-Tuning Time	Mean Time per Model					
All	2m 43s	5s					

Figure 5: Training Times of the Architectures

As can be observed from the illustration, the differences in time due to architectures are significantly greater than those due to the language used. There are, however, considerable time fluctuations in CNNs when training a model with German or Swedish texts, compared to other languages.

It is worth noting, however, that although we have attempted to train the models consistently under the same conditions, the computer may autonomously launch background processes beyond our control that affect the training times.

5.2 Training Results

Fare una piccola riflessione sulla fase di training dei modelli e le differenze trovate tra le diverse lingue.

Prima immagine in Figure 6.

	[LSTM] Best Hyperparameters, Best Epoch and Val Accuracy												
Language Batch Size Vocab Size Emb. Dim Hidden Dim Learn. Rate Dropout Best Epoch Val Accuracy (
English	64	3506	1024	2048	0,001	0,0	31	0,8488					
German	64	4216	1024	1024	0,001	0,0	42	0,7774					
Italian	Italian 64 4180			2048	0,001	0,0	34	0,8538					
Polish	64	5255	1024	1024	0,001	0,0	13	0,8462					
Swedish	64	4010	1024	1024	0,001	0,0	48	0,8459					

	[CNN] Best Hyperparameters, Best Epoch and Val Accuracy										
Language Batch Size Vocab Size Emb. Dim Kernel Size Stride Padding Learn. Rate Di								Dropout	Best Epoch	Val Accuracy (mean)	
English	64	3506	1024	5	1	3	0,001	0,1	46	0,8874	
German	64	4216	2048	5	1	1	0,001	0,1	31	0,8731	
Italian	64	4180	2048	7	1	1	0,001	0,0	20	0,8893	
Polish	64	5255	2048	5	1	3	0,001	0,0	12	0,8929	
Swedish	64	4010	1024	7	1	1	0.001	0.1	15	0.8698	

[LSVC] Best C and Val Accuracy							
Language C Val Accuracy (mean							
English	1,0	0,9112					
German	1,0	0,9121					
Italian	1,0	0,9096					
Polish	1,0	0,9123					
Swedish	1.0	0,9090					

Figure 6: Best Hyperparameters, Best Epoch and Validation Accuracy for each Architecture

Risultati del training in Figure 8 and 9.

5.3 Model Performances

Figure 7 shows the results obtained from the best architectures for each language on the test set. Colours are used in the visualisation to highlight the differences in performance of the models.

LSTM LSTM networks have the worst performance despite the highest training times.

It is interesting to observe how they have the most difficulties classifying German and Swedish texts, which are characterised by a higher loss compared to the other languages and, in the case of German, significantly lower performance.

It can also be seen that, with the exception of just one language, class 1 is always the one with the highest accuracy.

<u>CNN</u> TODO: attendere risultati completi

<u>LSVC</u> It is very interesting to observe how a statistical model that trains in five seconds on any CPU can compete and outperform neural networks that require hours of training on a dedicated GPU. According to the literature, this can happen thanks to TF-IDF, which is very reliable and leads to higher results than the words embedding employed by neural networks [7].

Unlike the LSTMs, this model predicts texts belonging to class 0 more accurately.

	[LSTM] Test Set Results										
Language Loss (global) Accuracy (global) Accuracy (mean) Accuracy [0] Accuracy [1]											
English	0,5561	0,8217	0,8213	0,8307	0,8474	0,7859					
German	0,9223	0,7575	0,7601	0,7025	0,7632	0,8146					
Italian	0,5415	0,8133	0,8138	0,8032	0,8263	0,8120					
Polish	0,6351	0,8075	0,8086	0,7849	0,8237	0,8172					
Swedish	0,9211	0,8192	0,8196	0,8124	0,8500	0,7900					

[CNN] Test Set Results									
Language	Loss (global)	Accuracy (global)	Accuracy (mean)	Accuracy [0]	Accuracy [1]	Accuracy [2]			
English	0,7085	0,8425	0,8415	0,8627	0,8237	0,8381			
German	0,7286	0,8208	0,8211	0,8146	0,8368	0,8120			
Italian	0,6919	0,8525	0,8517	0,8696	0,8342	0,8512			
Polish	0,7137	0,8375	0,8381	0,8261	0,8737	0,8146			
Swedish	0,7139	0,8308	0,8291	0,8696	0,8395	0,7781			

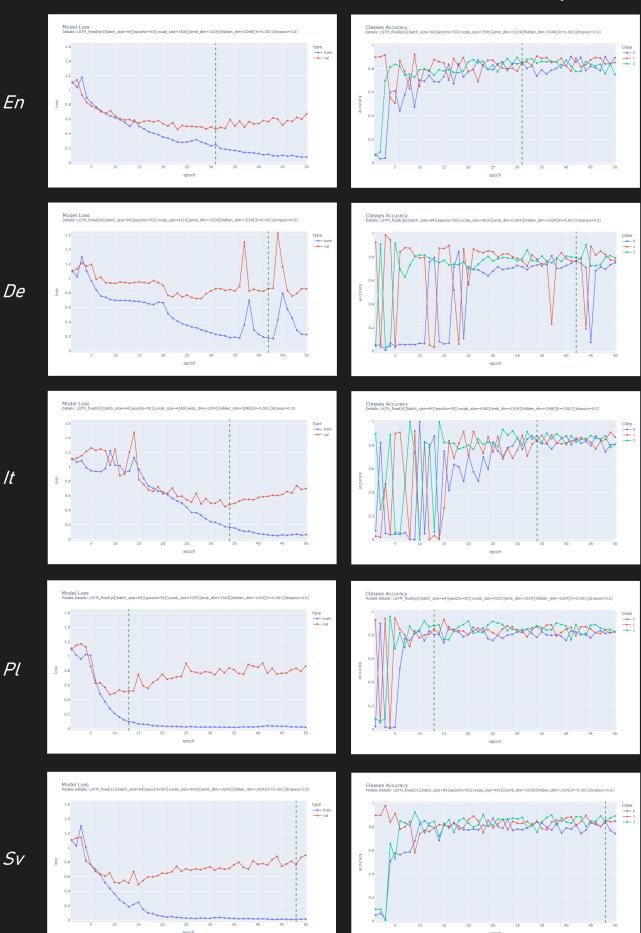
[LSVC] Test Set Results									
Language	Loss (global)	Accuracy (global)	Accuracy (mean)	Accuracy [0]	Accuracy [1]	Accuracy [2]			
English	-	0,8792	0,8783	0,9078	0,8601	0,8670			
German	=	0,8758	0,8750	0,9117	0,8594	0,8539			
Italian	-	0,8775	0,8768	0,9074	0,8528	0,8701			
Polish	-	0,8800	0,8792	0,9141	0,8668	0,8568			
Swedish	-	0,8850	0,8842	0,9125	0,8590	0,8811			

Figure 7: Training Times of the Architectures

LSTM Training Results



Classes Accuracy



CNN Training Results

Models Loss Classes Accuracy En De /t Pl Sv

6 Discussion

6.1 About the Results

TODO: scrivere una volta ottenuti tutti i risultati

Riflessione su come il LinearSVC sia competitivo verso modelli che impiegano ore ed ore per essere addestrati.

6.2 Ideas for Expanding the Project

During the course of the project, choices were made to simplify the task and the training time required by the models. This was mainly done for reasons of limited time and means at our disposal: having only one graphics card available, certain choices proved necessary.

The following is a list of suggestions and ideas through which the project could be expanded, if we had the means to do so.

- It would first be interesting to use all the texts of the laws at our disposal and repeat the experiment.
- Furthermore, it would also be worthwhile to consider the multi-label task and observe how the architectures behave in different languages.
- Having multiple GPUs avalaible, the experiment could be repeated using all 23 languages from the initial dataset.
- A further idea is to try applying stopword removal, lemmatization and/or stemming procedures to texts. The aim here is to see if and how the performance of the models changes and if the procedures favour the most common languages.
- With regard to the fine-tuning phase, as already mentioned in the relevant subsection 4.4, it would be advisable to extend the range of values used for each hyper-parameter in order to obtain more robust results.

References

- [1] Saad Albawi, Tareq Abed Mohammed, and Saad Al-Zawi. Understanding of a convolutional neural network. 2017 International Conference on Engineering and Technology (ICET), pages 1–6, 2017.
- [2] Vimala Balakrishnan and Ethel Lloyd-Yemoh. Stemming and lemmatization: a comparison of retrieval performances. 2014.
- [3] Jason Brownlee. A Gentle Introduction to Cross-Entropy for Machine Learning. https://machinelearningmastery.com/cross-entropy-for-machine-learning/. Accessed: 2705-2022.
- [4] Jason Brownlee. A Gentle Introduction to Long Short-Term Memory Networks by the Experts. https://machinelearningmastery.com/gentle-introduction-long-short-term-memory-networks-experts/.

 Accessed: 23-05-2022.
- [5] Jason Brownlee. A Gentle Introduction to Pooling Layers for Convolutional Neural Networks. https://machinelearningmastery.com/pooling-layers-for-convolutional-neural-networks/.
 Accessed: 27-05-2022.
- [6] Jason Brownlee. How Do Convolutional Layers Work in Deep Learning Neural Networks? https://machinelearningmastery.com/convolutional-layers-for-deep-learning-neural-networks/. Accessed: 27-05-2022.
- [7] Denis Eka Cahyani and Irene Patasik. Performance comparison of tf-idf and word2vec models for emotion text classification. *Bulletin of Electrical Engineering and Informatics*, 10(5):2780–2788, 2021.
- [8] Ilias Chalkidis, Manos Fergadiotis, and Ion Androutsopoulos. MultiEURLEX A multi-lingual and multi-label legal document classification dataset for zero-shot cross-lingual transfer. In *EMNLP*, 2021.
- [9] Alex Graves and Juergen Schmidhuber. Framewise phoneme classification with bidirectional lstm networks. Proceedings. 2005 IEEE International Joint Conference on Neural Networks, 2005., 4:2047– 2052 vol. 4, 2005.
- [10] IBM. Natural Language Processing (NLP). https://www.ibm.com/cloud/learn/natural-language-processing. Accessed: 21-05-2022.
- [11] Julie Beth Lovins. Development of a stemming algorithm. *Mech. Transl. Comput. Linguistics*, 11(1-2):22–31, 1968.
- [12] Prakash M Nadkarni, Lucila Ohno-Machado, and Wendy W Chapman. Natural language processing: an introduction. *Journal of the American Medical Informatics Association*, 18(5):544–551, 2011.
- [13] PyTorch. Dropout. https://pytorch.org/docs/stable/generated/torch.nn.Dropout.html. Accessed: 24-05-2022.
- [14] PyTorch. Embedding. https://pytorch.org/docs/stable/generated/torch.nn.Embedding.html. Accessed: 24-05-2022.

[15] PyTorch. Linear.

https://pytorch.org/docs/stable/generated/torch.nn.Linear.html.

Accessed: 24-05-2022.

[16] PyTorch. LSTM.

https://pytorch.org/docs/stable/generated/torch.nn.LSTM.html.

Accessed: 24-05-2022.

[17] Towards Data Science. The Importance of Natural Language Processing for Non-English Languages. https://towardsdatascience.com/the-importance-of-natural-language-processing-for-non-english-languages-ada463697b9d.

Accessed: 21-05-2022.

[18] Wojciech Zaremba, Ilya Sutskever, and Oriol Vinyals. Recurrent neural network regularization. ArXiv, abs/1409.2329, 2014.