A CONVOLUTIONAL NEURAL NETWORK FOR THE MNIST DATASET USING TENSORFLOW

Project report

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ABSTRACT

Qua scriveremo l'abstract...

We have seen that computer programming is an art,
because it applies accumulated knowledge to the world,
because it requires skill and ingenuity, and especially
because it produces objects of beauty.

— Donald E. Knuth [Knuth 1974]

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LISTINGS

ACRONYMS

CNN	Convolutional Neural Network
MLP	Multylayer Perceptron
ReLU	Rectified Linear Units
NLP	Natural Language Processing
TF	TensorFlow
CPU	Central Processing Unit
GPU	Graphical Processing Unit
CUDA	Compute Unified Device Architecture

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INTRODUCTION

2 CONVOLUTIONAL NEURAL NETWORKS

Convolutional Neural Networks (CNNs) are a kind of deep artificial feed-forward neural network in which the organization of the connections between the neurons is inspired by the animal visual cortex. Actually, individual cortical neurons respond to stimuli in a restricted region of space known as **receptive field**. The receptive fields of different neurons partially overlap such that they tile the whole visual field. Similarly, in CNNs each neuron is connected with only a small subset of inputs from the previous layer.

These networks are extremely useful when dealing with data with a grid-like topology, such as time-series data or images.

The name "convolutional neural network" indicates that the network employs a specialized kind of linear operation called **convolution**. In short, as stated in [Goodfellow, Bengio, and Courville 2016], *convolutional networks are simply neural networks that use convolution in place of general matrix multiplication in at least one of their layers*.

This chapter starts with a list of the main features of a convolutional neural network. Then, the building blocks of the CNN architecture are analysed with particular attention to the tuning of hyperparameters. In the last part, some applications of this model are proposed.

2.1 MAIN FEATURES

While traditional Multylayer Perceptron (MLP) models were successfully used in the past for image recognition, due to the full connectivity between nodes they suffer from the curse of dimensionality and, thus, they do not scale well to higher resolution images.

For instance, in the CIFAR-10 dataset, images are of size 32x32x3 (32 wide, 32 high, 3 color channels), so a single fully connected neuron in the first hidden layer of a regular MLP would have $32 * 32 * 3 = 3,072$ weights. A 200x200 image, however, would lead to neurons that have $200 * 200 * 3 = 120,000$ weights. Such network architecture does not take into account the spatial structure of data, treating input pixels which are far apart or close together exactly in the same way. The full connectivity of neurons is wasteful in the framework of image recognition, and the huge number of parameters quickly leads to overfitting.

As said before, convolutional neural networks are biologically inspired variants of multilayer perceptrons, designed to emulate the behaviour of the visual cortex. These models mitigate the challenges posed by the MLP architecture by exploiting the strong spatially lo-

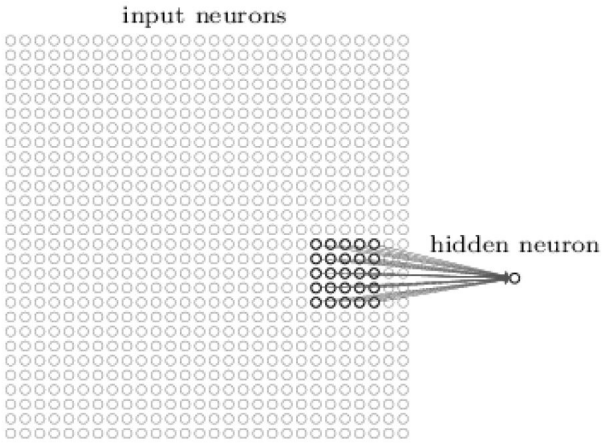

Figure 1: Implementation of a 5x5 receptive field in a CNN

cal correlation present in natural images. In particular, CNNs have the following distinguishing features:

3D VOLUMES OF NEURONS The layers of a CNN have neurons arranged in 3 dimensions: width, height and depth. Neurons inside a layer are only connected to a small region of the layer before it, called a receptive field. Figure 1, for instance, shows a 5x5 receptive field from the input neurons to the first hidden layer, with a 28x28 input. Distinct types of layers, both locally and completely connected, are stacked to form the CNN architecture.

LOCAL CONNECTIVITY Following the concept of receptive field, CNNs exploit spatially local correlation by enforcing a local connectivity pattern between neurons of adjacent layers. The architecture, thus, ensures that every **filter** (i. e. weight patch) learnt produces the strongest response to a spatially local input pattern. Stacking many of such layers leads to non-linear filters that become increasingly "global" (i. e. responsive to a larger region of input space). This allows the network to first create good representations of small parts of the input, then assemble representations of larger areas from them.

SHARED WEIGHTS In CNNs, each filter is replicated across the entire visual field. These replicated units share the same parameterization (weight vector and bias) and produce a feature map. This means that all the neurons in a given convolutional layer detect exactly the same features. Replicating units in this way allows for features to be detected regardless of their position in the visual field, thus constituting the property of translation invariance.

Together, these properties allow convolutional neural networks to achieve better generalization performances in vision problems. Moreover, the weight sharing helps by dramatically reducing the number

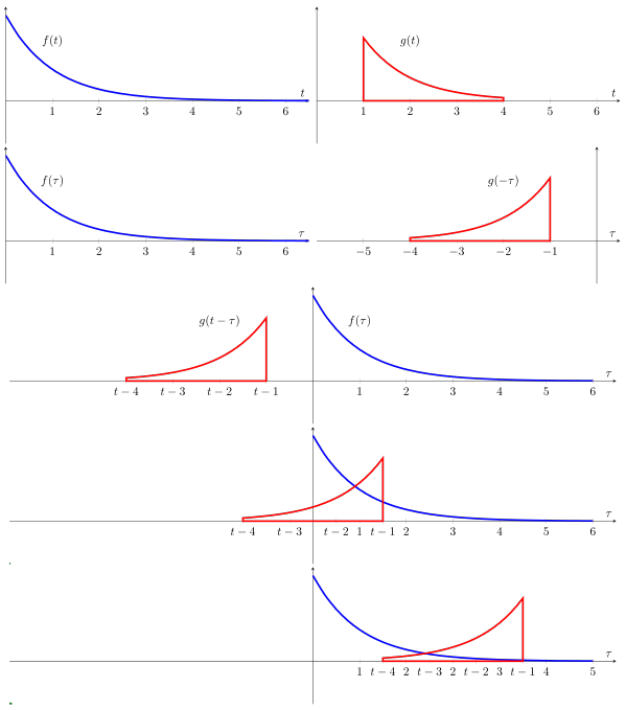

Figure 2: Visual explanation of the convolution operation

of free parameters being learnt, thus lowering the memory requirements for running the network. Decreasing the memory footprint allows the training of larger and more powerful networks.

2.2 THE CONVOLUTION OPERATION

In its most general form, convolution is an operation on two functions of a real-valued argument. It produces a third function, that is typically viewed as a modified version of one of the original functions, giving the integral of the point-wise multiplication of the two functions as a function of the amount that one of the original functions is translated. It is typically denoted with an asterisk and it is defined as:

$$s(t) = (f * g)(t) = \int_{-\infty}^{\infty} f(\tau)g(t-\tau)\,d\tau$$

The convolution formula can be described as a weighted average of the function $f(\tau)$ at moment t where the weighting is given by $g(-\tau)$ simply shifted by amount t. As t changes, the weighting function emphasizes different parts of the input function. The output is sometimes referred to as **feature** or **activation map**.

Figure 2 shows a visual explanation of the operation. Wherever the two functions intersect, the integral of their product is found. In other words, it computes a sliding, i. e. a weighted-sum of function $f(\tau)$, where the weighting function is $g(-\tau)$.

If we now assume that f and g are defined only on integer t, we can define the discrete convolution as:

$$s(t) = (f * g)(t) = \sum_{-\infty}^{\infty} f(\tau)g(t-\tau)$$

In convolutional networks terminology, the first argument of the convolution is often referred to as input and the second as **kernel** or **filter**. In machine learning applications, the input is usually a multidimensional array (i. e. a tensor) of data and the kernel is usually a multidimensional array of parameters that are adapted by the learning algorithm.

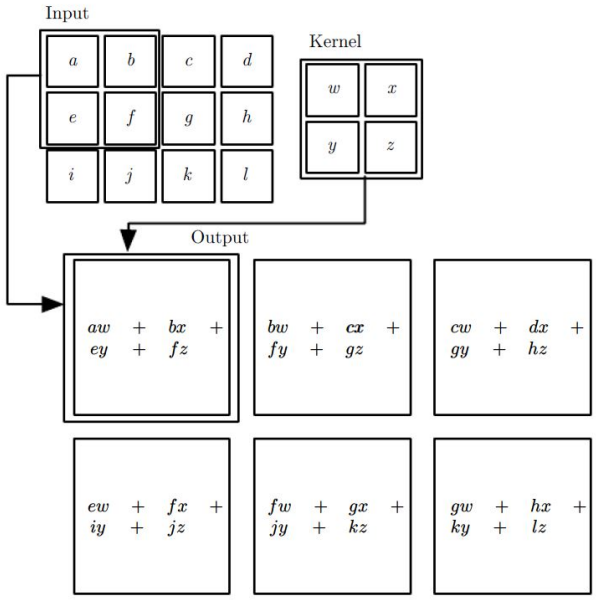

Figure 3: Example of -D convolution without kernel-flipping, as reported in [Goodfellow, Bengio, and Courville 2016]

Figure 3 shows an example of convolution (without kernel flipping) applied to a 2D tensor. In this case, the output is restricted to only positions where the kernel lies entirely within the image.

2.3 ARCHITECTURE

A CNN architecture is formed by a stack of distinct layers that transform the input volume into an output volume through a differentiable function. A few distinct types of layers are commonly used and they are described below.

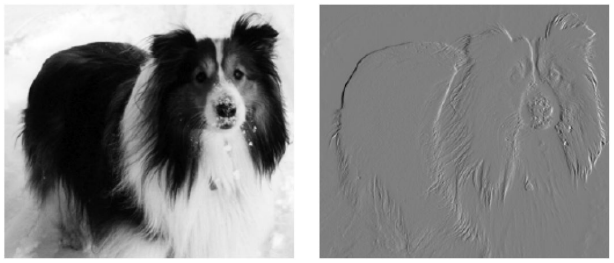

Figure 4: Application of a filter for edge detection (Photo credit: Paula Goodfellow)

2.3.1 *Convolutional layer*

The **convolutional layer** is the core building block of a CNN. The layer's parameters consist of a set of learnable filters (or kernels), which have a small receptive field, but extend through the full depth of the input volume. During the forward pass, each filter is convolved across the dimensions of the input volume, computing the dot product between the entries of the filter and the input and producing a feature map of that filter.

The amount of units by which the filter shifts is called **stride** and it controls how the filter convolves around the input volume.

As a result of this process, the network learns filters that activate when they detect some specific type of features in a given spatial position in the input.

It is evident that different filters will produce different feature maps for the same input. Figure 4 shows the result of the application of a filter for edge detection to a given image. The image on the right was formed by taking each pixel in the original image and subtracting the value of its neighboring pixel on the left. This shows the strength of all of the vertically oriented edges in the input image, which can be a useful operation for object detection.

Stacking the activation maps for all filters along the depth dimension forms the full output volume of the convolution layer. Every entry in the output volume can thus also be interpreted as an output of a neuron that looks at a small region in the input and shares parameters with neurons in the same activation map.

In fact, in order to recognize an image, we'll need more than one filter. For this reason, the output of a full convolutional layer will be a set of feature maps with a structure similar to the one in figure 5.

Sometimes it is convenient to have an output of the same size of the input. In this case it is necessary to add some additional pixels to the input image; this operation is called **padding**. For instance, a zero padding like the one in figure 6 pads the input volume with zeros all around the border. The size of this zero-padding is an optional

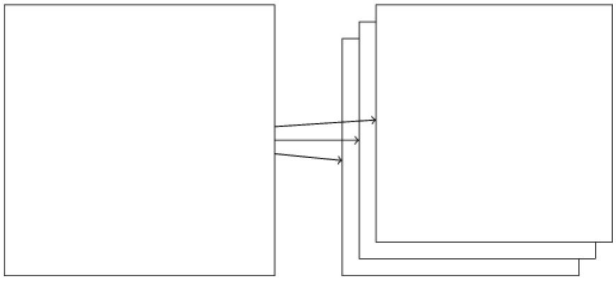

Figure 5: Application of three filters to a given input producing three different feature maps

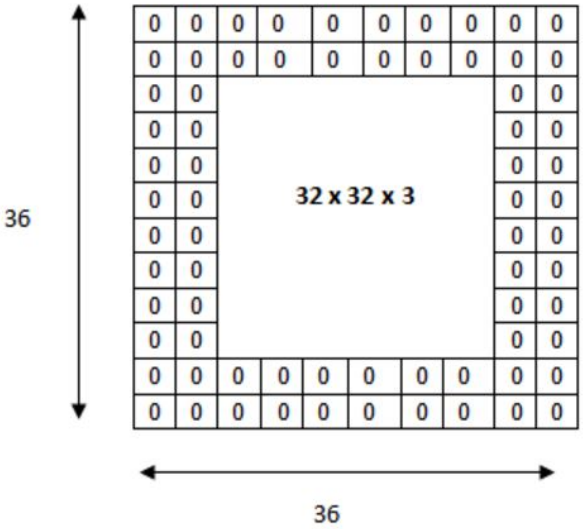

Figure 6: Zero padding of two applied to a 32x32x3 picture

hyperparameter. Zero padding, in particular, provides control of the output volume spatial size.

In general, the formula for calculating the output size for any given convolutional layer is:

$$O = \frac{W - K + 2P}{S} + 1$$

where O is the output height/length, W is the input height/length, K is the filter size, P is the padding, and S is the stride. If this number is not an integer, then the strides are set incorrectly and the neurons cannot be tiled to fit across the input volume in a symmetric way.

Setting the padding to

$$P = \frac{K - 1}{2}$$

when the stride is $S = 1$ ensures that the input volume and output volume will have the same size spatially.

2.3.2 *ReLU layer*

After a convolutional layer, it is convention to apply a non-linear layer (or activation layer) immediately afterward. The purpose of this layer is to introduce non-linearity to a system that basically has just been computing linear operations during the convolution operation (just element wise multiplications and summations). This is a layer of neurons that applies the non-saturating activation function:

$$f(x) = \max(0, x)$$

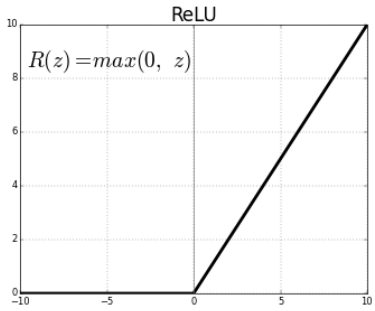

Figure 7: ReLU function

This increases the non-linear properties of the decision function and of the overall network without affecting the receptive fields of the convolution layer. The output of this layer known as **rectified feature map**.

In the past, non-linear functions like $\tanh$ and sigmoid were used, but researchers found out that ReLU layers work far better because the network is able to train a lot faster (because of the computational efficiency) without significant reductions in the accuracy [Nair and Hinton 2010]. Using a ReLU function also helps to alleviate the vanishing gradient problem, which is the issue where the lower layers of the network train very slowly because the gradient decreases exponentially through the layers.

The ReLU layer applies the function $f(x) = \max(0, x)$ to all of the values in the input volume. In basic terms, this layer just changes all

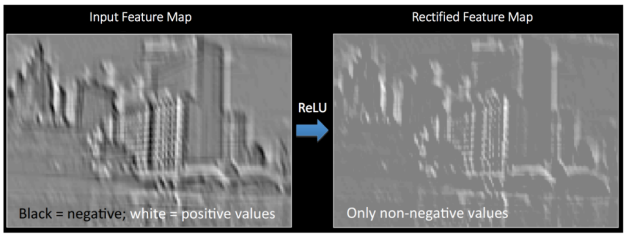

Figure 8: ReLU operation. Source http://mlss.tuebingen.mpg.de/2015/slides/fergus/Fergus_1.pdf

the negative activations to 0. Figure 8 shows the input of a ReLU layer and the resulting output.

2.3.3 *Pooling layer*

Another important concept of CNNs is **pooling**, which is a form of non-linear down-sampling. There are several non-linear functions to implement pooling among which **max pooling** is the most common. It partitions the input image into a set of non-overlapping regions and, for each of them, outputs the maximum value.

The intuition is that once a feature has been found, its exact location isn't as important as its rough location relative to other features. The function of the pooling layer is to progressively reduce the spatial size of the representation to reduce the amount of parameters and computation in the network, and hence to control overfitting, too. It is common to periodically insert a pooling layer in-between successive convolutional/ReLU layers in a CNN architecture.

The pooling operation also provides a form of translation invariance. This is very powerful since we can detect objects in an image no matter where and how they are located.

The pooling layer operates independently on every depth slice of the input and resizes it spatially. This operation is also known as **subsampling** or **downsampling**. The most common form is a pooling layer with filters of size 2x2 applied with a stride of 2. Every operation would in this case be taking a max over 4 numbers. Figure 9 shows an example of this operation.

In addition to max pooling, the pooling units can also perform other functions, such as average pooling and even L2-norm pooling. Average pooling was often used historically but has recently fallen out of favor compared to the max pooling operation, which has been found to work better in practice, as reported in [Scherer, Müller, and Behnke 2010].

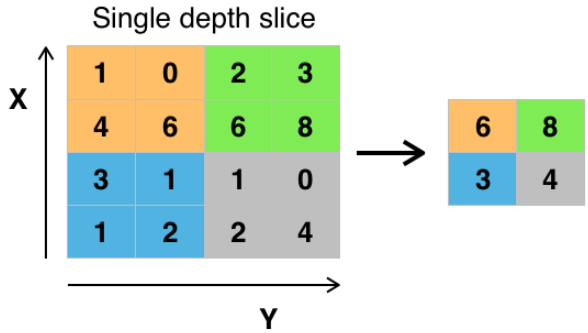

Figure 9: Max pooling with a 2x2 filter and stride $S = 2$

Due to the aggressive reduction in the size of the representation (which is helpful only for smaller datasets to control overfitting), the current trend in the literature is towards using smaller filters [Graham 2014] or completely discarding the pooling layer [Springenberg et al. 2014].

Obviously, the pooling operation is applied separately to each feature map.

2.3.4 *Fully connected layer*

Finally, after several convolutional/ReLU and max pooling layers, the high-level reasoning in the neural network is done via **fully connected layers**. Neurons in a fully connected layer have full connections with all the activations in the previous layer, as in traditional MLPs. Their activations can hence be computed with a matrix multiplication followed by a bias offset.

The output from the convolutional and pooling layers represent high-level features of the input image. The purpose of the fully connected layer is usually to use these features for classifying the input into various classes based on the training dataset. For instance, the image classification task aims to put a given image into a precise category.

Apart from classification, adding a fully-connected layer is also a cheap way of learning combinations of these features. Most of the features from convolutional and pooling layers may be good for the classification task, but combinations of those features might be even better.

2.3.5 *Dropout layer*

Dropout layers have a very specific function in neural networks; they reduce overfitting. The problem of overfitting is very important in machine learning. It happens when after training, the weights of the network are so tuned to the training examples they are given that the network doesn't perform well when given new examples (i. e. it doesn't generalize).

The idea of dropout is simplistic in nature. This layer "drops out" a random set of activations in that layer by setting them to zero in the forward pass. At each training stage, individual nodes are either "dropped out" of the net with probability $1 - p$ or kept with probability p, where p is a user-defined parameter, so that a reduced network is left. The removed nodes are then reinserted into the network with their original weights.

This pruning, in a way, forces the network to be redundant; the network should be able to provide the right classification or output for a specific example even if some of the activations are dropped out.

This makes sure that the network isn't getting too "fitted" to the training data and, thus, helps alleviating the overfitting problem. An important note is that this layer is only used during training.

The method also significantly improves the speed of training. This makes model combination practical, even for deep neural networks. The technique seems to reduce the complex, tightly fitted interactions between nodes, leading them to learn more robust features which better generalize to new data. Dropout has been shown to improve the performance of neural networks on tasks in vision, speech recognition, document classification, and computational biology.

An interesting analysis of the use of dropout in order to manage overfitting can be found in [Srivastava et al. 2014]

2.3.6 *Loss layer*

Finally, a loss layer specifies how the network training penalizes the deviation between the predicted and true labels. This is normally the last layer in the network. Various loss functions appropriate for different tasks may be used there.

Softmax loss is used for predicting a class of K different classes (i. e. the digit represented by an image from the MNIST dataset of handwritten digits, see chapter **??** for further details).

The function

$$f_j(z) = \frac{e^{z_j}}{\sum_k e^{z_k}}$$

is called the softmax function: it takes a vector of arbitrary real-valued scores and squashes it to a vector of values between zero and one that sum to one.

Moreover, sigmoid cross-entropy loss is used for predicting K independent probability values in $[0, 1]$. Euclidean loss is used for regressing to real-valued labels in $[-\infty, \infty]$.

These layers are the basic building blocks of any CNN. They can be stacked in order to produce complex architectures. The number of layers depends on the addressed task and on the complexity of the input. A typical convolutional architecture is composed by one or more sets of convolutional, ReLU and pooling layers and ends with a fully connected, a dropout and a loss layer. Again, the loss function to use depends on the task to be solved.

Figure 10 shows an example of architechture for image recognition with two convolutional layers and two pooling layers.

Many different architectures have been proposed. Among them, LeNet [LeCun et al. 1998] was one of the very first convolutional neural networks, for the recognition of handwritten digits, which helped propel the field of deep learning. In 2012, Alex Krizhevsky and others released AlexNet [Krizhevsky, Sutskever, and Hinton 2012] which was

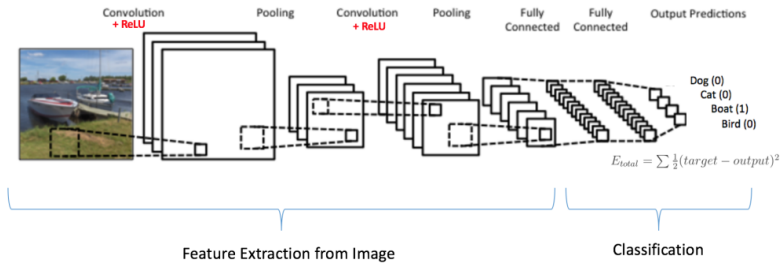

Figure 10: Convolutional neural network for the classification of an image

a deeper and much wider version of the LeNet and won by a large margin the difficult ImageNet Large Scale Visual Recognition Challenge (ILSVRC) in 2012. The ILSVRC 2014 winner was GoogleNet, a convolutional network from [Szegedy et al. 2014] from Google. Its main contribution was the development of an Inception Module that dramatically reduced the number of parameters in the network. In 2016 GoogleNet (now called Inception) is at its third release.

2.4 HYPERPARAMETERS

CNNs use more hyperparameters than a standard **mlp!** (**mlp!**). While the usual rules for learning rates and regularization constants still apply, the following should be kept in mind when optimising a convolutional network[].

NUMBER OF FILTERS Since feature map size decreases with depth, layers near the input layer will tend to have fewer filters while layers higher up can have more. To equalize computation at each layer, the product of the number of features and the number of pixel positions is typically picked to be roughly constant across the layers. Preserving the information about the input would require keeping the total number of activations (number of feature maps times number of pixel positions) to be non-decreasing from one layer to the next. The number of feature maps directly controls capacity and depends on the number of available examples and on the complexity of the task.

FILTER SHAPE AND SIZE Common filter shapes vary greatly in the literature, and are usually chosen based on the dataset. The challenge is, thus, to find the right level of granularity so as to create abstractions at the proper scale, given a particular dataset.

MAX POOLING SHAPE Typical values are 2x2. Very large input volumes may warrant 4x4 pooling in the first layers. However, choosing larger shapes will dramatically reduce the dimension of the signal, and may result in discarding too much information. Of-

ten, non-overlapping pooling windows perform best, as shown in [Scherer, Müller, and Behnke 2010].

AMOUNT OF EXAMPLES For many applications, only a small amount of training data is available. Convolutional neural networks usually require a large amount of training data in order to avoid overfitting. A common technique is to train the network on a larger data set from a related domain. Once the network parameters have converged an additional training step is performed using the in-domain data to fine-tune the network weights. This allows convolutional networks to be successfully applied to problems with small training sets.

2.5 APPLICATIONS

We conclude this chapter with some examples of application on CNNs in real-world domains. These kind of networks can be used with success to solve many different task, but, as said before, they work well with data with a grid-like topology.

2.5.1 *Image recognition*

Convolutional neural networks are often used in image recognition systems. They have achieved an error rate of 0.23 percent on the MNIST database, which as of February 2012 is the lowest achieved on the database [Ciregan, Meier, and Schmidhuber 2012]. When applied to facial recognition, they were able to contribute to a large decrease in error rate [Lawrence et al. 1997].

The ImageNet Large Scale Visual Recognition Challenge is a benchmark in object classification and detection, with millions of images and hundreds of object classes. In the ILSVRC 2014, almost every highly ranked team used a CNN as basic framework. The winner GoogLeNet [Szegedy et al. 2014] (the foundation of Google DeepDream) increased the mean average precision of object detection to 0.439329, and reduced classification error to 0.06656, the best result to date. Its network applied more than 30 layers.

The performance of convolutional neural networks on the ImageNet tests is now close to that of humans. In 2015 a many-layered CNN demonstrated the ability to spot faces from a wide range of angles, including upside down, even when partially occluded with competitive performance [Farfade, Saberian, and Li 2015].

2.5.2 *Video analysis*

Compared to image data domains, there is relatively little work on applying CNNs to video classification. Video is more complex than

images since it has another (temporal) dimension. However, some extensions of convolutional neural networks into the video domain have been explored. One approach is to treat space and time as equivalent dimensions of the input and perform convolutions in both time and space [Ji et al. 2013].

2.5.3 *Natural language processing*

Convolutional neural networks have also seen use in the field of natural language processing. CNN models have subsequently been shown to be effective for various NLP problems and have achieved excellent results in semantic parsing, search query retrieval, sentence modeling, classification, prediction, and other traditional NLP tasks.

2.5.4 *Drug discovery*

Convolutional neural networks have been used in drug discovery. Predicting the interaction between molecules and biological proteins can be used to identify potential treatments that are more likely to be effective and safe.

In 2015, Atomwise introduced AtomNet, the first deep learning neural network for structure-based rational drug design. [Wallach, Dzamba, and Heifets 2015] Subsequently, AtomNet was used to predict novel candidate biomolecules for several disease targets, most notably treatments for the Ebola virus and multiple sclerosis.

2.5.5 *Playing Go*

Convolutional neural networks have been used in computer Go. In December 2014, Christopher Clark and Amos Storkey published a paper ([Clark and Storkey 2014]) showing a convolutional network trained by supervised learning from a database of human professional games could outperform GNU Go and win some games against Monte Carlo tree search Fuego 1.1 in a fraction of the time it took Fuego to play.

Shortly after it was announced that a large 12-layer convolutional neural network had correctly predicted the professional move in 55% of positions, equalling the accuracy of a 6 dan human player.

Finally, a couple of CNNs for choosing moves and evaluating positions were used by AlphaGo, Google Deepmind's program that was the first to beat a professional human player.

3 TENSORFLOW BASICS

TensorFlow (TF) is an open source software library for machine learning in various kinds of perceptual and language understanding tasks. It is currently used for both research and production by different teams in dozens of commercial Google products [Abadi et al. 2016], such as speech recognition, Gmail, Google Photos, and Google Search, many of which had previously used its predecessor DistBelief [Dean et al. 2012].

TensorFlow was originally developed by the Google Brain team for Google's research and production purposes and later released under the Apache 2.0 open source license on November 9, 2015. Many teams nowadays at Google have migrated from DistBelief to TensorFlow for research and production uses.

This library of algorithms originated from Google's need to instruct neural networks, to learn and reason similarly to how humans do, so that new applications can be derived which are able to assume roles and functions previously reserved only for capable humans; the name TensorFlow itself derives from the operations which such neural networks perform on multidimensional data arrays. These multidimensional arrays are referred to as "tensors" but this concept is not identical to the mathematical concept of tensors. Its purpose is to train neural networks to detect and decipher patterns and correlations.

The aim of this chapter is to present the main data structures and operations provided by TF as well as to explain their basic usage. In particular the notions of computation graph, constant, variable and feed will be explained. Then, TensorBoard, a powerful tool for visualizing learning, will be presented. Finally, the last section includes a very simple example model built using TensorFlow.

The material provided in this chapter is taken form [*Tensorflow, An open-source software library for Machine Intelligence* 2017].

3.1 OVERVIEW

TensorFlow is Google Brain's second generation machine learning system. While the reference implementation runs on single devices, TF can run on multiple CPUs and GPUs (with optional CUDA extensions for general-purpose computing on graphics processing units). It runs on 64-bit Linux or Mac OS X desktop or server systems, as well as on mobile computing platforms, including Android and Apple's iOS.

It provides a Python API, as well as a less documented C++ API. In particular, the TensorFlow Python API supports Python 2.7 and

Python 3.3+. It can be easily installed as a common Python package through `pip` or in Anaconda with a `conda` installation.

TensorFlow computations are expressed as stateful dataflow **graphs**. Nodes in the graph are called *ops* (i. e. operations). An op takes zero or more `Tensor` objects, performs some computation, and produces zero or more `Tensors`. In TensorFlow terminology, a Tensor is simply a typed multi-dimensional array. For instance, a batch of images can be represented as a 4D tensor `Tensor` of floating point numbers with dimensions `[batch, height, width, channels]`.

In other words, a TF graph is a description of some computations. In order to actually compute anything, a graph must be launched in a `Session`. A `Session` object places the graph ops onto `Devices`, i. e. CPUs or GPUs and provides methods to execute them. In Python, these methods return tensors produced by ops as numpy `ndarray` objects.

3.2 THE COMPUTATION GRAPH

TensorFlow programs are usually structured into a construction phase, that assembles a graph, and an execution phase that uses a session to execute ops in the graph.

For example, it is common to create a graph to represent a neural network in the construction phase, and then repeatedly execute a set of training ops in the graph in the execution phase.

3.2.1 *Building the graph*

To build a graph it's useful start with ops that do not need any input (source ops), such as `Constant`, and pass their output to other ops that do computation. The ops constructors in the Python library return objects that stand for the output of the constructed ops. You can pass these to other ops constructors to use as inputs.

The TensorFlow Python library has a default graph to which ops constructors add nodes that is sufficient for many applications. However, it is possible to explicitly manage multiple graphs.

The following code builds a graph composed by three nodes. The two `constant()` operations produce two matrices (a 1x2 and a 2x1 matrix), while the `matmul()` operation takes as input the two previously created matrices and outputs the result of the matrix multiplication.

```
1 import tensorflow as tf
2
3 matrix1 = tf.constant([[3., 3.]])
4 matrix2 = tf.constant([[2.],[2.]])
5
6 product = tf.matmul(matrix1, matrix2)
```

In order to actually multiply the matrices, and get the result of the multiplication, the graph must be launched in a `Session`.

3.2.2 *Launching the graph*

Launching follows construction. To launch a graph, a `Session` object. Without arguments the session constructor launches the default graph.

The following code creates a session, launches the previously created graph and prints he result of the matrix multiplication.

```
# Launch the default graph
sess = tf.Session()

result = sess.run(product)
print(result)
# ==> [[ 12.]]

# Close the Session when we're done
sess.close()
```

To run the matmul op we call the session `run()` method, passing `product` which represents the output of the `matmul` op. This indicates to the call that we want to get the output of the `matmul` op back. All inputs needed by the op are run automatically by the session, typically in parallel. The call `run(product)` thus causes the execution of three ops in the graph: the two `constants` and `matmul`. The output of the matrix multiplication is returned in `result`.

Sessions should always be closed to release resources. They can be also surrounded with a `with` block. In this way, the `Session` closes automatically at the end of the block. The code then becomes:

```
with tf.Session() as sess:
    result = sess.run([product])
    print(result)
```

3.3 VARIABLES

`Variable` objects maintain state across executions of the graph. When a model is trained, `Variables` are used to hold and update parameters. They are in-memory buffers containing tensors. They must be explicitly initialized and can be saved to disk during and after training. Saved values can be later restored to exercise or analyze the model.

3.3.1 *Creation*

When a `Variable` a `Tensor` is passed to the `Variable()` constructor as initial value. TF provides a collection of ops that produce tensors that can be used for initialization from constants or random values. All these ops require to specify the shape of the tensors. That shape automatically becomes the shape of the variable. Variables generally have

a fixed shape, but TensorFlow provides advanced mechanisms to reshape them.

The following code creates two variables, one for weights and one for biases.

```
weights = tf.Variable(tf.random_normal([784, 200], stddev=0.35),
                      name="weights")
biases = tf.Variable(tf.zeros([200]), name="biases")
```

Calling `tf.Variable()` adds several ops to the graph:

- A `Variable` op the holds the variable value;
- An initializer op that sets the variable to its initial value;
- The ops for the initial value (e. g. the `zeros` op for `biases`).

The output is an instance of the Python class `tf.Variable`.

3.3.2 *Initialization*

Variable initializers must be run explicitly before other ops in the model can be run. The easiest way to do that is to add an op that runs all the variable initializers, and run that op before using the model.

The op `tf.global_variables_initializer()` can be used to add an op to run variable initializers. The op must be run after the model has been fully constructed.

The following code initializes the two variables created before:

```
init_op = tf.global_variables_initializer()

with tf.Session() as sess:
    # Run the init operation.
    sess.run(init_op)
    ...
```

3.3.3 *Saving and restoring*

The easiest way to save and restore a model is to use a `tf.train.Saver` object. The constructor adds `save` and `restore` ops to the graph for all, or a specified list, of the variables in the graph. The saver object provides methods to run these ops, specifying paths for the checkpoint files to write to or read from.

Checkpoints are binary files that, roughly, contain a map from variable names to tensor values.

The same `tf.train.Saver()` object can be used to manage all the variables in the model. Restored variables do not have to be initialized beforehand.

The following two pieces of code save and restore two variables:

```
v1 = tf.Variable(..., name="v1")
v2 = tf.Variable(..., name="v2")

init_op = tf.global_variables_initializer()

saver = tf.train.Saver()

with tf.Session() as sess:
    sess.run(init_op)
    ...
    save_path = saver.save(sess, "/tmp/model.ckpt")
    print("Model saved in file: %s" % save_path)
```

```
v1 = tf.Variable(..., name="v1")
v2 = tf.Variable(..., name="v2")

saver = tf.train.Saver()

with tf.Session() as sess:
    saver.restore(sess, "/tmp/model.ckpt")
    print("Model restored.")
    ...
```

If no argument is passed to `tf.train.Saver()`, the saver handles all variables in the graph. Each one of them is saved under the name that was passed when the variable was created.

It is sometimes useful to explicitly specify names for variables in the checkpoint files. For example, there might be a trained model with a variable named "weights" whose value have to be to restored in a new variable named "params".

It is also sometimes useful to only save or restore a subset of the variables used by a model. For example, a neural net with 5 layers could have been trained. In order to train a new model with 6 layers, the parameters from the 5 layers of the previously trained model can be restored into the first 5 layers of the new model.

The names and variables to save can be easily specified by passing to the `tf.train.Saver()` constructor a Python dictionary: keys are the names to use, values are the variables to manage.

3.4 FEEDS

The examples above introduced tensors into the computation graph by storing them in constants and variables. TF also provides a feed mechanism for patching a tensor directly into any operation in the graph.

A feed temporarily replaces the output of an operation with a tensor value. Deed data are supplied as an argument to a `run()` call. The feed is only used for the run call to which it is passed.

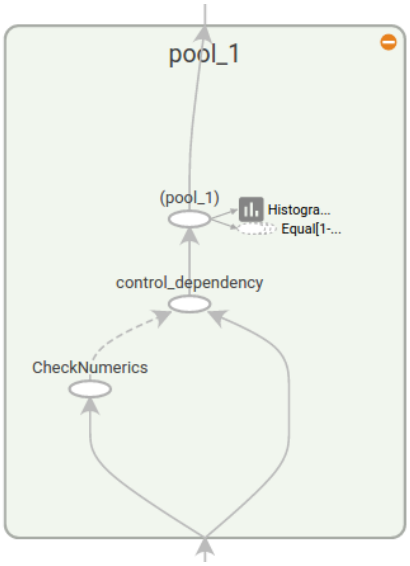

Figure 11: Graph visualization using TensorBoard

The most common use case involves designating specific operations to be "feed" operations by using `tf.placeholder()` to create them:

```
input1 = tf.placeholder(tf.float32)
input2 = tf.placeholder(tf.float32)
output = tf.mul(input1, input2)

with tf.Session() as sess:
    sess.run([output], feed_dict={input1:[7.], input2:[2.]})
```

A `placeholder()` operation generates an error if a feed for it isn't supplied.

3.5 TENSORBOARD

The computations performed with TensorFlow - like training a massive deep neural network - can be complex and confusing. To make them easier to understand, debug, and optimize TensorFlow programs, the developers included a suite of visualization tools called TensorBoard. It can be used to visualize a TensorFlow graph, plot quantitative metrics about the execution, and show additional data like images that pass through the graph.

TensorBoard operates by reading TensorFlow events files, which contain summary data that can be generated when running a TensorFlow graph.

Figure 11 shows an example of graph visualization using this tool. Every node can be expanded or minimized in order to analyze the graph at different granularities.

Further information about the usage of TensorBoard are are available in the dedicated page on the TensorFlow website.

3.6 A VERY SIMPLE MODEL

We conclude this chapter showing a very simple linear model built with TF. The program makes up some data in two dimensions, and then fits a line to them.

```
import tensorflow as tf
import numpy as np

# Create 100 phony x, y data points in NumPy, y = x * 0.1 + 0.3
x_data = np.random.rand(100).astype(np.float32)
y_data = x_data * 0.1 + 0.3

# Try to find values for W and b so that y_data = W * x_data + b
# (We know that W should be 0.1 and b 0.3, but TensorFlow will
# figure that out for us.)
W = tf.Variable(tf.random_uniform([1], -1.0, 1.0))
b = tf.Variable(tf.zeros([1]))
y = W * x_data + b

# Minimize the mean squared errors.
loss = tf.reduce_mean(tf.square(y - y_data))
optimizer = tf.train.GradientDescentOptimizer(0.5)
train = optimizer.minimize(loss)

# Before starting, initialize the variables.
init = tf.global_variables_initializer()

# Launch the graph.
sess = tf.Session()
sess.run(init)

# Fit the line.
for step in range(201):
    sess.run(train)
    if step % 20 == 0:
    print(step, sess.run(W), sess.run(b))

# Learns best fit is W: [0.1], b: [0.3]
```

The first part of the code builds the data flow graph. TensorFlow does not actually run any computation until the session is created and the run function is called, in the second part of the code.

4

A BASIC TENSORFLOW MODEL

In this chapter we will discover a basic Tensorflow model developed in order to classify images from the MNIST dataset. Our scope is to illustrate the structure of a CNNs and how that basic model can be readjust as needed, gathering data about training accuracy during the epochs and test accuracy on known and unknown images from the dataset.

After an introduction to the MNIST dataset, we will study in deep the structure of the Convolutional Neural Network used in our experiments, trying to understand which parameters are relevant to the improvement of the results, and how the structure can influence most of all on the accuracy of the network during the training.

5 ANALYSIS OF THE COLLECTED DATA

In order to reach satisfying conclusions, we will analyze carefully the collected data every time the structure of the network has been changed, trying to understand how it performs, what kind of adjustment we can do to improve the results, and comparing the different structured networks performances on the same classifying task.

6 CONCLUSIONS AND FUTURE WORK

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