

Deep Learning Project: Follow Me!

Robotics Nanodegree, Udacity

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Abstract

This report describes my solution of the deep learning project. First I describe the convolutional network as applied to the given semantic segmentation task. Then I describe the data collection process, the network architecture and training process. In the last section I present the results and discuss the performance of the tested models. The code for this project can be found in [this repository](#).

Contents

1	Fully Convolutional Network for Semantic Segmentation	3
1.1	Overview	3
1.2	Implementation	4
2	Data Collection	5
3	Network Architecture and Training	6
3.1	Architecture Parameters	6
3.2	Hyperparameter Tuning	7
3.3	Training with additional data	9
4	Results	11
4.1	Future Enhancements	11

List of Tables

1	FCN model properties	7
2	Architecture experiment	8
3	Hyperparameter tuning experiment	9
4	Collected data experiment	11

List of Figures

1	Paths for data collection run 1	5
2	Paths for data collection run 2	6
3	Diagram of FCN model	8
4	Training curves for tuning run 6	10
5	Demo run for detecting object types	13

1 Fully Convolutional Network for Semantic Segmentation

For writing this section I used what I learned in the classroom and references [1, 2, 3].

1.1 Overview

The goal of this project is to build and train a neural network that enables a drone to follow a person in a simulated environment.

The network is supposed to do semantic segmentation of the given image and detect the location of the hero person in the scene. To accomplish this goal, each pixel is classified as either belonging to the target person or being part of another person or not being part of any person. If the model were to distinguish more target types (dog, cat, car, etc.), the number of output classes would change. The depth of the final layer would have to be adjusted accordingly, and the network trained with additional labeled image data containing these target types.

The image segmentation task can be accomplished with a fully convolutional network (FCN). Such architectures consist of convolution layers exclusively.

Encoding and Decoding

The architecture used in this project is comprised of an encoding and a decoding stage. The encoder gradually reduces the spatial size of the image with pooling layers. The decoder scales the feature maps back up to the original size.

The convolution layers in the encoder extract feature maps from the image data. So they learn details about the image which will be used for segmentation. The complexity of the encoded features increases with the number of layers.

The pooling layers in the encoder perform a downsampling operation along the width and height of the image. This enables the model to generalize to unseen input, and it reduces the risk of overfitting to the training data. The drawback of pooling is that the original image gets coarse-grained. The resolution is lowered and some of the corresponding spatial information is lost.

The purpose of the decoder block is to scale the low resolution feature maps up to the original image resolution, so that the network can classify all of the pixels in the input image.

The upsampling operation fills the gaps between missing pixel values by interpolation. The information which was lost during pooling can't be fully restored by the upsampling

operation, and the decoder can't recover the spatial resolution completely.

To help the decoder recover the object details, skip connections are introduced. They connect the high resolution encoder layers directly to the decoder, so that some of the information from those layers can be restored.

Depthwise Separable Convolution

For writing this section I read section 3 in [3] and section 1 in [2].

The encoder architecture is built as a depthwise separable convolution. The standard convolution layer is factorized into a depthwise convolution looking at spatial correlations and a 1×1 pointwise convolution that detects correlations in the color channels. The underlying assumption is that the information encoded in the spatial dimensions is independent of the information in the depth channel.

So the last layer in the encoder is a regular convolution with kernel size 1×1 and stride of 1. So each neuron has a receptive field of only one pixel, therefore the input and output volumes have equal height and width. The output tensor is of size $H \times W \times M$ with the height and width and depth of the input volume. In comparison, a fully connected final layer would reduce the output into a single vector, where the class scores are arranged along the depth dimension. So the output tensor of the final layer would have size $1 \times 1 \times D$ where D is the number of classes.

This depthwise factorization has the advantage of drastically reducing the model size and the computational cost. The efficiency of the encoder network improves, and the number of parameters is reduced. The drawback is that the assumption that the color channel information is decoupled from the spatial information might not hold for the actual image data we are using in our model.

1.2 Implementation

The encoding stage consists of two separable convolutions followed by a regular 1×1 convolution with a stride of one and same padding.

The purpose of the decoder block is to scale the encoder output up to the size of the input image and decode the data stored in the previous layers. It is comprised of three parts.

- A non-trainable bilinear upsampling layer. The value of an output pixel is determined by interpolation of its nearest neighbors in the input.
- A non-trainable concatenation layer. It implements skip connections and thereby retains some of the information from the previous higher resolution layers.



Figure 1: Drone and hero paths for data collection run 1.

- One or more separable convolutions which learn from the data stored in the concatenated layer.

The final layer of the model is a regular convolution with softmax activation. It assigns class probabilities to each pixel in the image.

2 Data Collection

I collected image data in the `QuadRotor` simulator. The goal was to collect sequences that the drone will encounter in application, with the target person in it. I ran two different path layouts.

1. Drone patrols directly over the hero. The paths are laid out such that the drone follows a straight line and the hero zigzags around that line, as shown in figure 1.
2. The drone circles around the hero and many distracting people. See figure 2.

I collected 270 data sets in the first run and 186 data sets in the second run. One such data set consists of four camera images. I ran the image processing script `preprocess_imgs.py` which returned 456 image/mask combinations. I used these data points as additional training data, so I added them to the training data set and left the data sets for validation and evaluation as they were.

The resulting new training data set `train_combined.zip` (file size 56.7 MB) consists of 4587 image/mask combinations and can be downloaded from the [repository](#). I used this

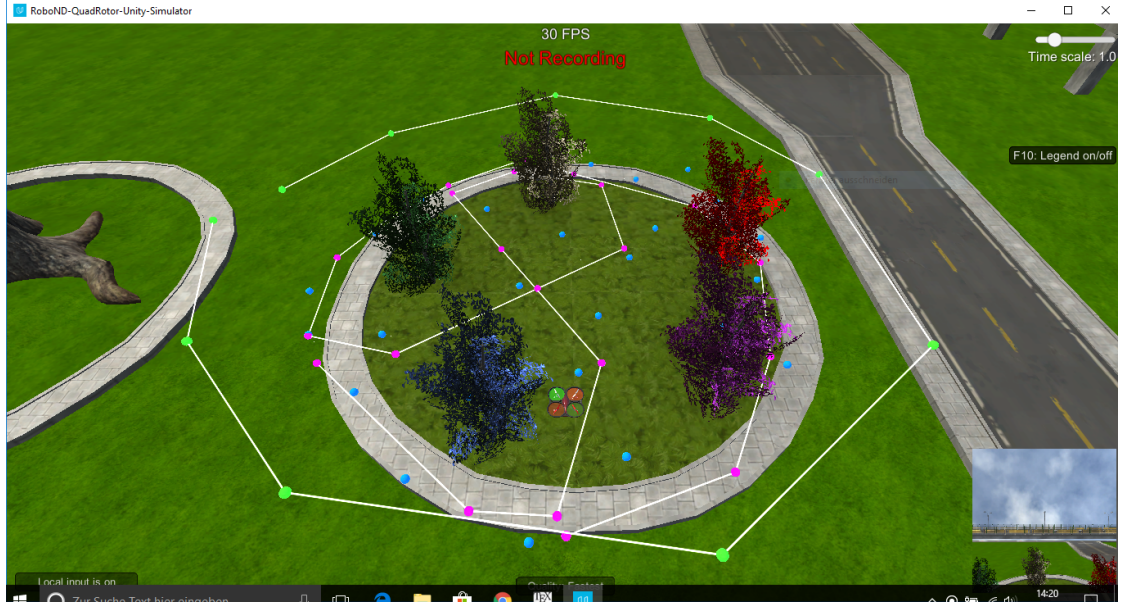


Figure 2: Drone and hero paths for data collection run 2.

data set to train the network as described in section 3.3. The results of this experiment can be found in table 4.

3 Network Architecture and Training

I trained the network on a AWS p2.xlarge instance with the Udacity Robotics Laboratory Community AMI installed. The training code can be found in the notebook [model_training.ipynb](#).

The evaluation is done using the intersection over union metric as the final score.

I conducted three different experiments in order to find the combination of network architecture and hyperparameter values that achieves the highest possible final score. I used the data sets [train.zip](#), [validation.zip](#) and [sample_evaluation_data.zip](#) as a reference, so the results of the training runs are comparable.

For all training runs described here, the number of validation steps is 50 and the number of workers is 4.

3.1 Architecture Parameters

I tuned and tested the following parameters of the FCN architecture described in section 1.

layer	purpose	volume	implementation
input	input	$160 \times 160 \times 3$	InputLayer
layer_1	encoder	$80 \times 80 \times 32$	SeparableConv2D
layer_2	encoder	$40 \times 40 \times 64$	SeparableConv2D
layer_3	1×1 conv	$40 \times 40 \times 128$	Conv2D
layer_4	decoder	several sizes	decoder_block
layer_5	decoder	several sizes	decoder_block
output	output	$160 \times 160 \times 3$	Conv2D

Table 1: Overview of FCN model used for semantic segmentation of drone images, as shown in figure 3. For implementation and detailed model summary see the notebook [model_training.ipynb](#).

- The filter sizes of the encoder blocks. I started with a depth of 32 for the first layer, 64 for the second layer and 128 for the 1×1 convolution.
- The number of separable convolution layers in the decoder block. I started with a two layer configuration.

For setting the initial values of these parameters I used knowledge from the classroom material, the segmentation lab and communication with other students on Slack. One such advice is to keep the network simple at the outset and add complexity later.

To test these assumptions I ran the architecture experiment as shown in table 2. I compared the filter size sequence $32 \rightarrow 64 \rightarrow 128$ with the sequence $32 \rightarrow 64 \rightarrow 128$. I tested the usage of one and two separable convolution layers in the decoder block.

The network with two separable convolutions in the decoder and a filter size sequence of $32 \rightarrow 64 \rightarrow 128$ achieved the highest final scores. Therefore I used this configuration in the subsequent experiments. This network is described in table 1 and depicted in figure 3.

3.2 Hyperparameter Tuning

I tested the reference architecture described in section 3.1 with a variety of values for the learning rate, the batch size and number of epochs. I used the reference data sets for training, validation and evaluation. The results of this experiment are given in table

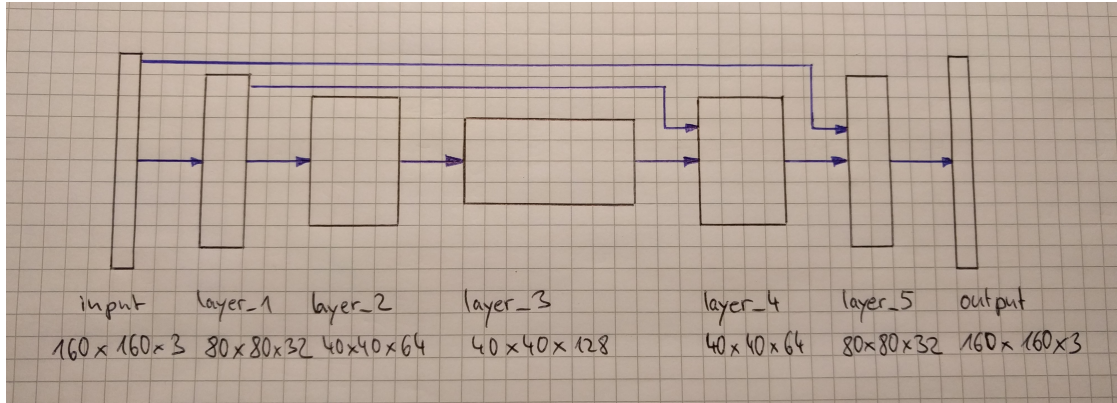


Figure 3: Diagram of FCN model. Rectangles represent input/output layers and encoder/decoder blocks. Blue arrows represent connections. Layer names follow the notebook [model_training.ipynb](#). For further model properties see table 1. The internal structure of the blocks is not shown in this diagram.

Run	1	2	3	4
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Architecture

Decoder convolutions	2	1	2	1
Maximum depth	128	128	256	256

Evaluation: Final Score

Run a	0.3908	0.3213	0.3834	0.3338
Run b	0.4277	0.3576	0.3955	0.4030
Average \pm deviation	0.41 ± 0.02	0.34 ± 0.02	0.39 ± 0.01	0.37 ± 0.04

Table 2: Results of the architecture experiment. Maximum depth refers to the filter size of the 1×1 convolution layer. Learning rate is 0.005. Batch size is 32. Network configurations are trained over 10 epochs with 200 steps. The final scores are rounded to four significant digits for the individual runs and two significant digits for the average and deviation over the runs.

Run	1	2	3	4	5	6
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Hyperparameter Values

learning rate	0.008	0.008	0.008	0.008	0.005	0.005
batch size	20	32	32	32	32	32
number of epochs	4	4	10	16	16	50
steps per epoch	200	200	200	200	200	200

Evaluation

final score	0.3198	0.4009	0.4260	0.4054	0.4387	0.4285
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Table 3: Results of the hyperparameter tuning experiment on the reference architecture (see figure 3) using the reference data sets for training, validation and evaluation.

3. The corresponding model files can be found in the `weights` directory named as `model_weights_tune_n` where `n` is the number of the run.

I started with default values from the lab and trained 4 epochs. In the subsequent runs I increased the number of epochs and observed an improving final score. At 16 epochs I noticed the final score dropping. So I decreased the learning rate in order to prevent overfitting. I ran experiment 5 with a learning rate of 0.005 and the final score improved.

In run 6 I increased the number of epochs further to 50. The training curve of this run is shown in figure 4. The final score did not improve compared to the previous run.

From the results of the tuning experiment I deduced that the current best learning rate to work with is 0.005.

I left the number of steps per epoch at 200, in future work this can also be tuned. The number of epochs can be increased and the number of steps decreased.

3.3 Training with additional data

When I had explored several configurations of architecture and hyperparameter values, I started to use the data collected in the simulator. I took the reference network as shown in figure 3 and the hyperparameter values of tuning run 6 (see table 3). I trained this configuration on the combined data (see section 2 for the data collection process). The

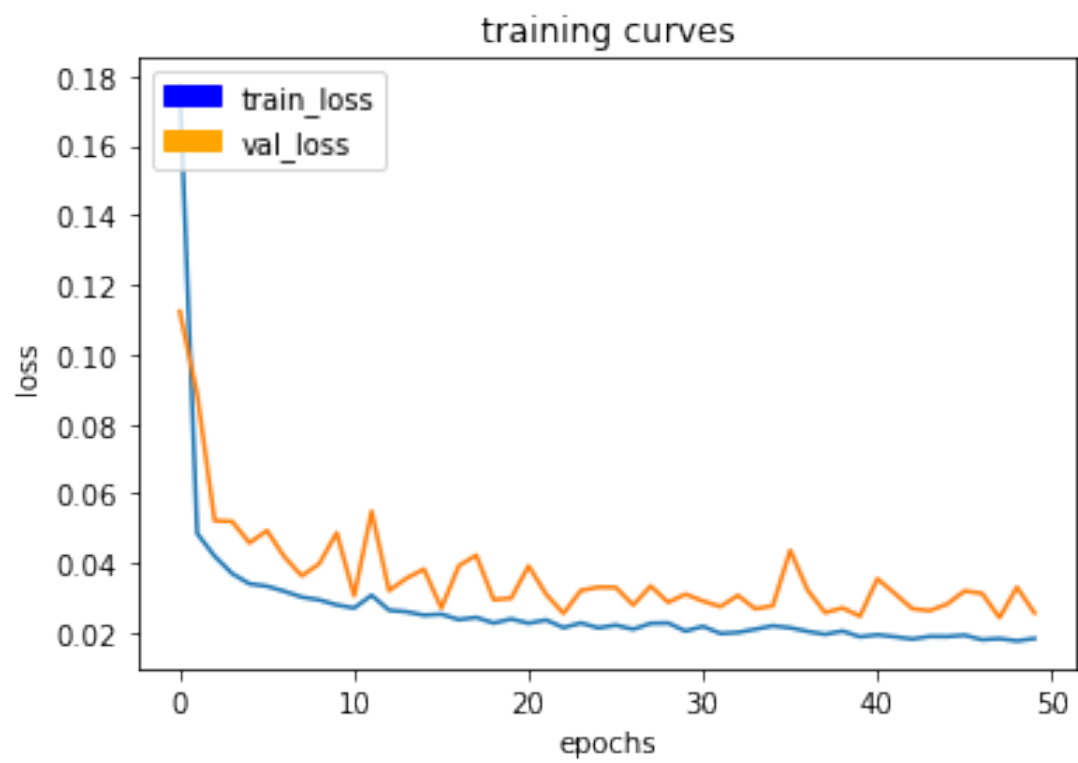


Figure 4: Training curves for tuning run 6 with 50 epochs at a learning rate of 0.005.

Run	1	2
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Hyperparameter Values

learning rate	0.005	0.005
batch size	32	32
number of epochs	16	50
steps per epoch	200	200

Evaluation

final score	0.3734	0.4044
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Table 4: Results of the collected data experiment using the collected data set for training and the reference data sets for validation and evaluation. The experiment is performed on the network configuration described in 1 and figure 3.

results are given in table 4.

4 Results

The model trained in tuning run 6 (last column in table 3) shows the best performance. It reaches final scores of approximately 0.43. Therefore I submit the corresponding weight file `model_weights_tune_6`.

Against expectations, gathering additional data did not have a significant effect on the network performance. I assume that this can be improved by collecting much more image sequences, where the hero is present in the camera image, and with more variation in the background scenery.

4.1 Future Enhancements

The following improvements of the network architecture, the hyperparameter tuning and the data collection process seem to be promising next steps to increase the performance of the model.

- Tune the number of steps per epoch. Increasing the number of epochs and decreasing the number of steps per epoch might increase the model performance

while taking the same training time.

- Experiment with learning rate decay and early stopping. Take full advantage of all the options provided by the [Keras optimizers package](#).
- Use `sklearn.grid_search.GridSearchCV` to automate the hyperparameter tuning experiments.
- Collect more image data, possibly with the target being present in a higher fraction of the camera images.
- In the decoder block, use transpose convolutions (see [Keras](#)) instead of the less accurate bilinear upsampling technique.

Regarding the task of detecting more types of objects (dog, cat, car, etc.) I did some research and found the [Mask R-CNN](#) project. As a first step I let that model run object detection on the image `5_run6cam1.00197.jpeg` from the validation data set. The result is encouraging, since besides the hero person a potted plant is detected as well, see figure 5.

I'm curious to compare both projects, their architecture and data sets, for example train the Follow Me! network on the multiple-object images provided there and vice versa.

References

- [1] [Stanford CS231n](#) *Convolutional Neural Networks for Visual Recognition*, [Class Notes on Convolutional Neural Networks](#)
- [2] François Chollet, *Xception: Deep Learning with Depthwise Separable Convolutions*, CoRR, abs/1610.02357, 2016, <http://arxiv.org/abs/1610.02357>
- [3] Andrew G. Howard and Menglong Zhu and Bo Chen and Dmitry Kalenichenko and Weijun Wang and Tobias Weyand and Marco Andreetto and Hartwig Adam, *MobileNets: Efficient Convolutional Neural Networks for Mobile Vision Applications*, CoRR, abs/1704.04861, 2017, <http://arxiv.org/abs/1704.04861>



Figure 5: Result of demo run for recognizing multiple objects using the Mask R-CNN model.