Low-latency Image Recognition with
GPU-accelerated Convolutional Networks
for Web-based Services

by

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Abstract

In this work, we describe an application of convolutional networks to object classification and detection in images. The task of image based object recognition is surveyed in the first chapter. Its application in internet advertisement is one of the main motivations of this work.

The architecture of the convolutional networks is described in details in the following chapter. Stochastic gradient descent is used to train the networks.

We then describe the data collection and labelling process. The set of training data labelled basically decides what kind of recognizer is being built. Four binary classifiers are trained for the object types of sailboat, car, motorbike, and dog.

GPU based massive parallel implementation of the convolutional networks is built. This enables us to run the convolution operations at close to 40 times faster than running on a traditional CPU. Details about how to implement the convolutional operation on NVIDIA GPUs using CUDA is discussed.

In order to apply the object recognizer in a production environment where millions of images are processed daily, we have built a platform with cloud computing. We describe how large scale and low latency image processing can be achieved with such a system.
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Chapter 1

Introduction

Images are literally everywhere. The web brought us the first wave of online images. Then came the photo-sharing websites such as ImageShack, Photobucket, Flickr that are dedicated to storing and sharing images. Social networking sites like Facebook and Google+ pushed image sharing to an even larger scale. Then came the mobile age, where smartphones equipped with cameras and apps such as Instagram and Twitpic have enabled users to create and share images anywhere at anytime.

Unlike texts, images are not amenable to indexing, since raw pixels have no meanings. Image understanding is sorely needed for any application that generates value based on the semantics of images. One example is the online advertisement. By placing ads that are relevant and compatible with the images on the webpage, the clicking rate on the ads can be significantly boosted.

1.1 Image recognition

Our work is motivated by real-world applications such as internet advertisement. We have built an image based object classification and detection system with
convolutional networks. Large data sets are labelled to train binary classifiers to identify specific object types against non-object background. We show that with bootstrapping, our classifiers can achieve good results on objects such as sailboats, cars, motorbikes, and dogs.

Both recognition accuracy and speed are critical to the success of the system. In the real-world environment, processing millions of images per day is quite common. The economy of the technology becomes viable only when large quantity of images can be processed efficiently. The user experiences also require processing low latency, which in turn demands a robust and high performance infrastructure.

In order to build a fast object classification and detection system, we have explored using the new technology of GPU programming. GPUs grew out of computer graphics applications, especially gaming, and have become the leading technology for supercomputing. GPUs are also very affordable, compared with the traditional supercomputing technologies. In this work, we show that with some modification, our system can be ported to Nvidia’s GPU, and achieve close to 40× speedup for convolution operations compared with running on a regular CPU.

The last part of this work is the constructing of an image processing production system. A fast web server has been built to respond to the online image processing requests, and a robust queuing system to dispatch workloads across a cluster of image processing instances. The system has been tested in real revenue-generating production environment.

1.2 Previous works

Image based object recognition, including classification and detection, has been an active and fruitful research field for decades. Many methods and approaches
have been proposed over the years. For example, color, texture, and contours have been advocated in [16], the use of global appearance templates in [19, 18, 25], silhouettes and edge information [22, 29, 16, 5, 25], pose-invariant feature histograms [17, 6, 1], distinctive local features [26, 35, 33, 12], and image fragments [30].

Several multi-layer feed-forward architectures, other than the convolutional nets [13, 31, 14, 15], have been proposed for object classification and detection. Like convolutional nets, they are also inspired by Hubel and Wiesel’s classic model of simple cell and complex cell for the early visual cortex. These include the Neocognitron [9], the hierarchies of features detectors based on image fragments [8], and variants of the HMAX architecture [28].

These models are all based on stacking the modules of convolutional filters, the non-linearities, and the spatial subsampling. In [9], the filters are learned with an unsupervised method that produces sparse features, and the non-linearities are piecewise linear rectifications. In [8], the filters are image fragments selected from training images using a mutual information criterion with the objects labels. Since the filters are relatively selective and class-specific, a large number of them is required. In [28], the first layer is simply a set of fixed Gabor filters, and the non-linearity/subsampling is a max over local filter outputs. A large number of Gabor filters are necessary to cover the orientation/scale space.

In Convolutional Networks [14, 36], all the filters are learned with a supervised gradient-based algorithm. Several recent works have shown advantages of pre-training each layer of a deep network in unsupervised mode, before tuning the whole system with a gradient-based algorithm [38, 39, 40].
Chapter 2

Object recognition with Conv-Nets

Convolution network [14] is a specific kind of neural networks inspired by biology. It has been used with great success in various image recognition applications, such as handwriting recognition for OCR.

The convolutional net architecture is designed for image processing, containing multiple alternated layers of trainable filters, point-wise non-linearities, and spatial subsampling. It is trained in supervised mode using a gradient-based algorithm that minimizes a loss function. Convolutional net have been shown empirically to automatically learn salient image features and yield good recognition accuracy [15].

2.1 Architecture

Convolutional nets have multi-layer architectures where the successive layers are designed to learn progressively higher-level features, until the last layer which
produces categories. All the layers are trained simultaneously to minimize an overall objective function. The feature extraction is therefore an integral part of the classification system, rather than a separate module, and is entirely trained from data, rather than designed.

A conv net consists of a stack of convolution and subsampling layers. A convolution (C-)layer computes the convolution between the input $\mathbf{x}_{in}$ with some small, trainable convolution kernels $k$:

$$\mathbf{x}_{out} = S(\sum_i \mathbf{x}_{in} \otimes k_i + b)$$

where $S$ is a non-linear function (a hyperbolic tangent sigmoid), and $b$ is a scalar bias. Multiple convolution kernels can be used on each C-layer. They will be configured differently by the training process.

The spatial subsampling (S-)layers take the average of a $n \times n$ pixel block, multiply it by a trainable scalar $\beta$, add a bias, and pass the result through another sigmoid:

$$\mathbf{x}_{out} = S(\beta \sum \mathbf{x}_{in}^{n \times n} + b)$$

The result is a feature map of lower resolution where some position information about features has been eliminated, thereby building some level of distortion invariance in the representation.

Alternated layers of convolution and subsampling can extract features from increasingly large receptive fields, with increasing robustness to irrelevant variabilities of the inputs. Different number of convolution and subsampling layers can be stacked together to create models with different characteristics.

The last layer of a convolutional network can be seen as a linear classifier operating on the feature representation extracted. It computes the product of the feature vector $v$ with a weight matrix $W$, adds a bias vector, and passes the
result through sigmoid functions. The Euclidean distance between the output vector and a target output vector $T^i$ is used as the loss function to be minimized:

$$\mathcal{L} = \| S(W.v + b) - T^i \|^2$$

where $W$ is a trainable weight matrix of the last layer. $T^i$ can be a traditional place code (one unit corresponding to the $i$-th element set to active, other units inactive), $i$ being the class label of the input $x$.

The network is trained by minimizing $\mathcal{L}$. Gradient descent based algorithms can be used for the optimization since all the layers are differentiable. The training process is described in more details in later sections.

### 2.2 Example model

In our work, a six-layer conv net is used, as seen in figure 2.1. The layers in the model are sequentially indexed, and named C1, S2, C3, S4, C5, and output.

In the figure, a testing image of size $96 \times 96$ is used. C1 uses 8 convolution kernels of size $5 \times 5$ to generate 8 feature maps. There are 208 (200 on $k_i$ and...
8 on b) trainable parameters in this layer. The output of C1 is a $8 \times 92 \times 92$ 3-dimensional array.

S2 is a $4 \times 4$ subsampling layer with 16 parameters, with output dimensions $8 \times 23 \times 23$.

C3 uses 96 convolution kernels of size $6 \times 6$ to output 24 feature maps. The C3 layer contains 3,480 trainable parameters, with output dimensions $24 \times 18 \times 18$. S4 is a $3 \times 3$ subsampling layer which outputs feature maps of size $24 \times 6 \times 6$.

C5 has 2400 kernels of size $6 \times 6$. A C5 layer outputting 100 feature maps has 86,500 trainable parameters, about 95% of the whole system’s parameter set. The output of the C5 layer is a 100-dimensional vector.

The output layer takes inputs from all C5 maps, transform them into a scalar. The binary decision of whether the $96 \times 96$ has the object is based on this scalar’s value.

### 2.3 Training with backpropagation

To minimize the loss function a stochastic version of the Levenberg-Marquardt algorithm with diagonal approximation of the Hessian was used.

The whole data set can be used for multiple passes to get progressively better results. The test error rate flattens out after about 10 passes. No significant over-training was observed, and no early stopping was performed.

One parameter controlling the training procedure must be heuristically chosen: the global step size of the stochastic gradient procedure. Best results are obtained by adopting a schedule in which this step size is progressively decreased from $2 \times 10^{-5}$ to $1 \times 10^{-6}$.

The convolutional nets are computationally very efficient. The training time scales sub-linearly with dataset size in practice.
Chapter 3

Training the object recognizer

In this chapter, we describe the application of the conv-nets to the image based object recognition. This includes the classification task, where objects are to be decided whether present in the image, and the detection task where the objects are localized within an image.

We describe how the training images are collected, labelled, and preprocessed. The architectural details of conv-nets are described. We evaluate the performance of the classification task with the recall-precision curve. We also show some examples of the detection results.

The system described in this chapter has been implemented and tested in a production environment. Daily processing of millions of images has been achieved with a fast implementation of the system, and the recognizer has shown impressive results in the real-world commercial environment.
3.1 data sets and preprocessing

Recognizers for four different object categories, namely sailboat, car, motorbike and dog, are built in this work. For each category, we search for images with such objects on the web with keywords.

We then label these images with bounding boxes around the desired objects. For the sailboat category, we pick images with side views of sailboats, containing a horizontal boat hull, vertical mast, and one or two sails attached to the mast. Figure 3.1 shows sample images for this category. The labelling choice decides what kind of recognizer we want to build.

We use about 600 sailboat images as the positive training set. The bounding boxes are randomly perturbed in orientations (-5 to +5 degrees) and scales (0.84× to 1.18×) to get 5 drawings for each box. Thus we have about 3,000 cropped sailboat images in total. These near-square images are then resized to 96×96 pixels to train the convolutional network.
The negative set are randomly cropped from a large number of images collected from imageshack.com, after removing sailboat images. About 3,000 negative images are used in our experiment. By randomly cropping 5 patches from each image, we have 15,000 negative samples for training. Figure 3.2 shows sample image for this set.

The image samples are preprocessed carefully to remove influence of variations of color, lighting condition, and contrast. The color images are first converted to gray scale. Then pixel intensity equalization is applied. Finally, a $19 \times 19$ high-pass filter is applied to the images. The operation makes the average pixel intensity to be near 0, hence removes lighting condition variations.

Special attention is needed to deal with the filtering at image edges, where for a pixel anchored with the center of the filter, not all surrounding pixels are available. The filter coefficients that correspond to the missing pixels need to be subtracted from the center coefficient of the filter.
3.2 training the recognizer

The convolutional network we used as the object recognizer has 5 layers: a convolution layer with 8 filters of sizes $5 \times 5$, followed by a $4 \times 4$ subsampling layer, then a second convolution layer 96 filters of sizes $6 \times 6$, and a $3 \times 3$ subsampling layer, and another convolution layer with 2400 filters of sizes $6 \times 6$.

With this convolution network, each image patch of size $96 \times 96$ is transformed into a $92 \times 92 \times 8$ sized stack of features, then subsampled to $23 \times 23 \times 8$. The 8 layers of features are then combined with certain configuration by the second batch of convolution filters to generate a $18 \times 18 \times 24$ features, subsampled to $6 \times 6 \times 24$, and finally combined by the last convolution layer into a 100 dimensional feature vector. This feature vector is linearly combined to generate a output value. During training process, we adjust the network parameter, through backpropagation such that for negative samples, the output value is as close to $-1.5$ as possible, and for positive samples, the output value should be pushed to near $+1.5$.

The stochastic gradient descent described in the previous chapter is used to learn the parameters of the whole system. We gradually decrease the learning step size from $2 \times 10^{-5}$ to $10^{-6}$ in the total 35 epochs used for training.

3.3 image classification, detection and evaluations

For images larger than $96 \times 96$ in size, we can use the trained template, and scan it through the image horizontally and vertically. Wherever a local patch correlates with the template highly, we have a positive detection of this object.

The convolutional net we use does this scanning automatically, except that it translates 12 pixels between two neighboring outputs, due to the two subsam-
pling (3×3 and 4×4) operations.

The process only detects objects of size 96×96. In reality, we wish to detect objects of different sizes. We can either use a set of templates with different sizes, or resize the image to different scales and use a fixed-size template which works equivalently.

In our experiment, we systematically resize the original image into a set of 13 scales that are $\sqrt{2}$ apart, from $1/32 \times$ to $2 \times$ scaling of the image. Such scaled images form a pyramid. Running the template with fixed 96×96 can detect objects of all sizes ranging from 48×48 to 3072×3072.

Out of the 13 scales, only those with object size of interest are searched. We use both an absolute size limit and a relative size bound. The choice of the limits are empirical: the object has to be larger than 48 pixels, and it must be larger than both $1/20$ of the longer side of the image and $1/3$ of the shorter side of the image, and the object must also be smaller than the shorter side of the image.

The performance of the system can be evaluated with two different tasks, classification and detection. The classification task is to decide whether the specific object is present in the image or not. This binary classification can be measured by the recall-precision curve.

The curve can be obtained as following. For each test sample, a numerical value is assigned by the classifier. This value is compared with pre-determined threshold $\theta$. If the value is higher than the threshold, then the system thinks at least one instance of the object is present.

A testing sample with a value above the threshold falls in 2 categories. It is a true positive ($TP$) if it comes from the positive training set. And it is a false positive ($FP$) if it comes from the negative training set, i.e., it does not actually contain the object.
On the other hand, a testing sample with a value lower than the threshold can be either a true negative (TN) if it is from the negative set, or a false negative (FN) if it is from the positive set and in fact does contain the object.

The above 4 elements of the confusion matrix contains all the information about the classifier’s performance. But in the image retrieval applications, the following two values are usually extracted from the confusion matrix to measure specific aspects of the system.

The recall, $TP/(TP + FN)$, is the number of true positive samples normalized by the number of samples in the positive training set. This describes the probability of detecting a positive testing sample. The recall value ranges from 0 to 1.

The precision, $TP/(TP + FP)$, is defined as among the samples with values above the threshold, what percentage is true positives.

For a given testing set, each threshold values gives us a recall-precision pair. And the recall-precision curve can be obtained by increasing the threshold from minimum value to maximum.

Ideally, we would like to have a system with both high recall and high precision. But these two measures usually work against each other. We can lower the threshold to get a higher recall rate, but then the precision will also go lower. A suitable threshold can be chosen according to the requirement of the application, depending whether a specific recall rate or a precision rate is more desirable.

Figure 3.3 shows the recall-precision curve for the sailboat classification. The testing set consists of 50 positive samples and 200 negative samples. The threshold is adjusted in the range [0, 100] to obtain the data points on the curve.

The detection task is similar to the classification except some post-processings are needed. For each scale, we threshold the array of the outputs. Those
points with values above threshold are then transformed into equivalent bounding boxes in the original image scale.

The bounding boxes collected from all scales are then compared with each other to remove redundant ones. For a pair of neighboring boxes with significant overlapping, the one with a smaller recognizer output value is removed. This is based on the observation that an image patch with an object usually produce a cluster of neighboring bounding boxes.

The detection performance is evaluated in a less rigorous way. We simply run the system on the 50 positive images. The system’s localization results are compared with the manual labelling. With a threshold of 60, there are 4 images where the boats are missed by the detector, and there is one false positive detection on a non-boat background area. The correct detection rate is therefore above 90%. Figure 3.4 shows some sample images with the detection.

Figure 3.3: The recall-precision curve of the boat classifier
One difficulty in building an object recognizer is the lack of good negative training samples. Conceptually the positive samples populate only a small neighborhood in the hyperspace due to their visual similarities, while the negative samples span the remaining vast space. We often need a very large negative set of samples to cover such space.

In the previous experiments, we collected the negative set by randomly cropping image patches from non-object images. Even though this set is much bigger than the positive set, these samples do not cover the non-object space well. This results in a lot of false positives yielded by the classifier.
In order to learn the decision boundary separating the positive and negative spaces, we only need the samples that are close to the boundary on both sides. That is, we need negative samples are easily confused with the positives. These samples lie near the decision boundary.

The detector built in the previous experiment can help us to pick such negative samples efficiently. We run the learned detector through the negative set and pick up the false positive patches, as shown in figure 3.5. These false positives are then merged into the negative training set to train a more refined system. In the literature, this process is called bootstrapping [43].

In the sailboat experiment, we collected 3,500 image patches from the negative set. These patches are then perturbed to generate 7,000 samples. The system is re-trained in the same way using this new training data. In our experiment, we observed that the training process converges with less number of epochs to a small training error. Figure 3.6 draws both the curve obtained in
the previous sections and the curve with the new system trained with the combined negative set. For the same recall rate, the precision rate is increased by 10% to 15%. The retraining with the newly collected data works very well.

3.5 why high precision for classification is hard

The recall-precision curve of the sailboat classifier in previous section is obtained on a test set with 50 positive samples and 200 negative samples. The ratio $\lambda$ between the number of negative samples $N_n$ versus the number of positive samples $N_p$, $\lambda = N_n/N_p$, can change when we chose a different test set. This ratio $\lambda$ influence directly the result of the precision.

This can be seen by transforming the definition of the precision:
Figure 3.7: Comparison of the recall-precision curves of test set with different negative to positive ratios

\[
\frac{TP}{TP + FP} = \frac{TP/N_p}{((TP/N_p) + (FP/N_n) \times (N_n/N_p))}
= \frac{TP_r}{TP_r + \lambda \times FP_r}
\]

where the true positive rate \(TP_r\) and the false positive rate \(FP_r\) are both only dependent on the threshold \(\theta\). Therefore, when we increase the ratio \(\lambda\), \(TP_r\) and \(FP_r\) remain the same, while the precision decreases.

The reason for precision to be dependent on \(\lambda\) is that precision combines results from both the positive set and negative set, and is therefore prior dependent. Recall, on the other hand, is only related to the positive set.

Figure 3.7 shows the two recall-precision curves with different ratio \(\lambda\) for the same sailboat classifier. The original curve is obtained with 200 negative samples, and ratio 4. The red curve is for a negative set of 1600 samples, and
Figure 3.8: Comparison of the recall-precision curves of the original classifier, and of the classifier trained with targeted negative samples.

In reality, this means that picking out sailboat images from a set that is known to have very few boat images is very hard. False positives tend to overwhelm the returned result. This is bad news. The prior knowledge about the distribution of the images is helpful in setting the expectations on the performance.

### 3.6 results with other objects

The process described in the previous sections can be used to train other object recognizers. We have built classifiers and detectors for 3 other categories: motorbike, car, and dog.
The motorbike category is very similar to the boat in that most images have side views of the objects. The car images have more view variations. The dog category is even more challenging since the objects are not rigid and can have different shapes depending on the configuration of the body parts.

The car training images contain mostly the between-front-and-side view, where at least one head light and its closest wheel are visible. For the dog recognizer, we choose to recognize only the head part, and with frontal or between-front-and-side views.

The same procedure of labeling, training and evaluation is used for these objects. Between 500 to 900 images are labeled for each category, and the conv-nets of the same architecture are trained. The trained detectors are then used to collect false positive samples from the negative training set. The results are combined with the randomly chosen negative samples to retrain the recognizer.
Figure 3.8 shows the recall-precision curve for the motorbike recognizer. The valuation set contains 50 positive samples and 200 negative samples. The lower curve is from the system trained with random negative samples, while the other curve is from the system trained with targeted negative samples. The retraining boosts the precision rate by 10% to 20%. Figure 3.9 and figure 3.10 shows the performances of the car and dog recognizers, respectively. Both these recognizers have a much higher precision rate when retrained with targeted negative samples.
Chapter 4

Fast Implementation of Convolutions

The convolutional network has shown a remarkable capability for the object recognition task, as we have shown in the last chapter. In order to apply this technology in the real world environment, an efficient implementation of the system is needed. In this chapter, we describe the work to implement the convolutional network on the GPU. The modern GPUs have thousands of cores capable of general computing. These cores can be programmed to do massive parallel computing on the desktop environment. We will show that the most computationally intensive part of the recognition system, the convolution operation, can be easily parallelized to achieve computing performance far above running a traditional CPU.
4.1 Convolution operation is computationally intensive

We first profile the convolutional net based object detection system to find the hot spot, the component that takes up most of the computing time. The hot spot is then optimized to speed up the system.

The system consists of three parts, the image preprocessing, convolution nets, and the post-processing. The image preprocessing includes the color to grayscale conversion, resizing, histogram equalization, filtering. These operations are well understood in terms of their computational complexities.

In the second step, the preprocessed image is fed through the convolution net to produce a two-dimensional confidence map, representing the confidences of a local image patch being of the target object. The post-processing marks the locations where objects are detected, based on the confidence map.

In this work, we first implement a reference system in C++. The the runtime complexity of the various steps can be measured with profilers. Figure 4.1 shows a call graph when running the system with a 216×120 test image on an Intel Q8200 CPU. Each node stands for one function, annotated with the accumulated running time.

We can see that the overall detection on the image takes 29.1 ms, while the convolutional operation, \texttt{op\_convolution}, takes 20.5 ms, which accounts for about 70\% of the whole process. It is obvious that optimizing the convolutional operation would speed up the entire process.

In the next section, we analyze the complexity of the convolution operation, and show that it can be easily parallelized.
4.2 Computational complexity of convolutions

Convolution operations can be applied to two arrays of the same dimensions. The convolutions between two 1-dimensional vectors are often used in audio signal processing. The convolutions between two 2-dimensional arrays are popular in image processing. The convolution net in this work uses a generalized version of the 3-dimensional convolution.

The 1-dimensional convolutions between two vectors $A$ and $B$, of size $N$ and $k$ respectively, where $N \gg k$, can be described as follows. From the longer vector $A$ of size $N$, we successively take sub-vectors of size $k$, with one element shifting each time, and take the inner product between it and the smaller vector $B$, the result is one element in the output vector $C$, as shown in 4.1. Obviously this process involves two nested loops, with the inner loop computing the inner product, and the computational complexity is $O(k(N - k + 1))$.

```
for (int i = 0; i < (N-k+1); ++i) {
    C[i] = 0;
    for (int j = 0; j < k; ++j)
        C[i] += A[i+j] * B[j];
}
```

Similarly, the complexity for the 2-d convolution between two arrays of size $N \times N$ and $k \times k$ is $O(k^2(N - k + 1)^2)$ with 4 nested loops. And 3-d convolutions
between arrays of size $N \times N \times N$ and $k \times k \times k$ has complexity of $O(k^3(N-k+1)^3)$ with 6 nested loops.

The convolution net uses a generalized version of the 3-d convolution described above. The first input array of the operation is of size $D \times N \times N$, and the second input, the kernel array, is of size $(d \times P) \times k \times k$. The first input is considered as a stack of 2-d features maps, where the stack dimension is treated differently from the other two dimensions of the feature maps.

For the operation, we take $d$ feature maps out of the $D$ maps, of size $d \times N \times N$, and apply the regular 3-d convolution with the $d \times k \times k$ subset of the kernel array. We get an output array of size $1 \times (N-k+1) \times (N-k+1)$, which can be degenerated into a 2-d $(N-k+1) \times (N-k+1)$ matrix. We repeat this process $P$ times, and stack the output matrices together to get a 3-d array of size $P \times (N-k+1) \times (N-k+1)$.

The choices of $d$ out of $D$ feature maps from the first input can be configured in a table of size $d \times P$ whose elements range between 0 and $D-1$.

The operation described above can be implemented with 6-level nested loop with $P$, $N-k+1$, $N-k+1$, $d$, $k$, $k$ iterations respectively. The first 3 loops iterate through each element of the $P \times (N-k+1) \times (N-k+1)$ sized output array, and each element is the inner product of two $d \times k \times k$ arrays. The following code snippet in 4.2 shows a detailed implement of the convolution operation.
Listing 4.2: Convolution operation

```c
void op_conv(const float *o1,
              int extent_1_0, int extent_1_1, int extent_1_2,
              int stride_1_0, int stride_1_1, int stride_1_2,
              const float *o2,
              int extent_2_0, int extent_2_1, int extent_2_2,
              int stride_2_0, int stride_2_1, int stride_2_2,
              float *o3,
              int extent_3_0, int extent_3_1, int extent_3_2,
              int stride_3_0, int stride_3_1, int stride_3_2,
              const int *o4,
              int extent_4_0, int extent_4_1,
              int stride_4_0, int stride_4_1,
              const float *o5,
              int stride_5_0)
{
    for(int h=0; h < extent_3_0; ++h) {
        for(int i=0; i < extent_3_1; ++i) {
            for(int j=0; j < extent_3_2; ++j) {
                const float *p3_1 = o1 + i * stride_1_1 + j * stride_1_2;
                const float *p3_2 = o2 + (h*extent_4_0) * stride_2_0;
                float f = 0;
                for(int m=0; m < extent_2_1; ++m) {
                    const float *p1_1 = p3_1 + m*stride_1_1;
                    const float *p1_2 = p3_2 + m*stride_2_1;
                    for(int n=0; n < extent_2_2; ++n) {
                        f += *p1_1++ * *p1_2++;
                    }
                }
                float *p0_3 = o3+h*stride_3_0+i*stride_3_1+j*stride_3_2;
                const float *p0_5 = o5 + h*stride_5_0;
                *p0_3 = stdsigmoid( f + *p0_5);
            }
        }
    }
}
```

The convolution operation described in 4.2 has a running cost of \(O(dP(N-k+1)^2k^2))\). In order to get a concrete idea of the run time complexity of the convolutional net, let us consider using a real image with size 120×216. The feed forward convolution net has 3 convolution layers and 2 subsampling layers. The first convolution layer has \(d=1, P=8, k=5\), and it generates an output array of size 8×116×212. The computational cost is \(1 \times 8 \times 116 \times 212 \times 5 \times 5 = 4.9 \times 10^6\)
configure & ops(million) & CPU(ms) \\ 
1x8x(116x212)x(5x5) & 9.8 & 10.4 \\ 
4x24x(24x48)x(6x6) & 7.8 & 5.3 \\ 
24x100x(3x11)x(6x6) & 5.6 & 3.8 \\ 
total & 23.2 & 19.5 \\ 

Table 4.1: The count of floating operations in convolution layers with an example image

*fused multiply-add* (FMA) operations, or about 10 millions floating operations, with each FMA counted as 2 floating operations. The following subsampling layer has \( k = 4 \), generating output of size \( 8 \times 29 \times 53 \). The next convolution layer has \( d = 4, P = 24, k = 6 \), costing \( 3.9 \times 10^6 \) FMAs, and generates an output of size \( 24 \times 24 \times 48 \). The following subsampling layer has \( k = 3 \), which brings the output to \( 24 \times 8 \times 16 \). The last convolution layer has \( d = 24, P = 100, k = 6 \), which takes \( 2.8 \times 10^6 \) FMAs to run and generates a confidence map of size \( 100 \times 3 \times 11 \).

With this example image, the total computational cost for convolutions adds up to about 23 million floating operations, as shown in the first column of the table 4.1. Profiled on an Intel Core2Quad 2.3 GHZ CPU Q8200, running in a single thread, the convolutions add up to about 20 ms, as shown in the second column in the same table. This yields about 1.2 GFLOPS performance. Note that this performance is not only related to the hardware, but also the code implementation and the compiling tools.

### 4.3 Parallelize the Convolution Operations

The previous sections have analyzed the computational complexity of the conv nets, and established that the most runtime intensive part is the convolution operations. To speed up the system, it is essential to implement the convolution operation efficiently.
The convolution operations turn out to be very amenable to parallelization. The key is to identify the parts that are independent of each other and can be run by a separate thread. As previously analyzed, the operation consists of 6 nested loops. The 3 inner-most loops compute the inner product between two \( d \times k \times k \) arrays. These inner product computations are inherently independent from each other. To make this fact more obvious, we can factor out the inner most 3 loops and put them in a standalone function, as shown in 4.3, where there are 3 extra arguments compared with the function \( op_{conv} \) described in 4.2, namely, \( h \), \( i \), \( j \). This tuple identifies which inner product the function is computing.

Listing 4.3: Function with the 3 inner loop in the convolution operation

```
void op_conv_kernel(..., int h, int i, int j)
{
    const float *p3_1 = o1 + i * stride_1_1 + j * stride_1_2;
    const float *p3_2 = o2 + (h*extent_4_0) * stride_2_0;
    float f = 0;
    for(int k=0; k < extent_4_0; k++) {
        const float *p2_1 = p3_1 + stride_1_0 + o4[k*stride_4_0+h];
        const float *p2_2 = p3_2 + stride_2_0 + k;
        for(int m=0; m < extent_1_2; ++m) {
            const float *p1_1 = p2_1 + m*stride_1_1;
            const float *p1_2 = p2_2 + m*stride_2_1;
            for(int n=0; n < extent_2_2; ++n) {
                f += *p1_1++ * *p1_2++;
            }
        }
    }
    float *p0_3 = o3 + h*stride_3_0+i*stride_3_1+j*stride_3_2;
    const float *p0_5 = o5 + h*stride_5_0;
    *p0_3 = std::sigmoid( f + *p0_5);
}
```

The outer 3 loops iterate through each element of the output array. Each iteration step invokes the above \( op_{conv\_kernel} \) to compute the inner product, as shown in 4.4.
void op_conv(...) {
    for(int h=0; h < extent_3.0; ++h) {
        for(int i=0; i < extent_3.1; ++i) {
            for(int j=0; j < extent_3.2; ++j) {
                op_conv_kernel(..., h, i, j);
            }
        }
    }
}

These steps, identified by the index variables \((h, i, j)\), are executed sequentially in the above implementation, running in a single thread. They can be executed in parallel by simply launching \(P \times (N-n+1) \times (N-n+1)\) threads, with each thread computing one element independently. The number of threads in this cube can be pretty large when the image size related \(N\) is big. When the available parallel computing units is limited, we need to break the cube into small batches, and run the batches in sequence. The advent of modern GPU provides a perfect fit for this need since GPUs tend to have lots of computing units, in the order of hundreds and thousands, and the batching process is supported by the hardware and its compiling tools, as exemplified in the CUDA (Compute Unified Device Architecture) technology developed at NVIDIA.

### 4.4 GPU Programming Model

The modern CPUs have achieved remarkable computational performance by increasing the clock frequency, the instruction-level parallelization, and the cache size. This trend has lasted two decades and brought GFLOPS performance to desktops. But it is now increasingly difficult to boost the performance within this traditional model.

Meanwhile, another type of hardware, the GPUs (graphical processing unit), have grown in popularity, and started to out-perform the CPUs in terms of
computational power. The GPUs have a different design philosophy. They usually have hundreds, even thousand of computing cores, compared with the CPUs with just dual cores or quad cores. Even though GPU cores usually have lower clock frequencies, and simplified control logic, their combined performance can easily top off the CPUs.

The GPU-enabled program runs in two modes. Most part of the program runs on the CPU, or the host. The part that needs to be parallelized runs on the GPU, or the device. The host and device exchange data by copying between the host memory and the device memory.

The CUDA framework extends the C++ syntax with a few key words that are recognized by its compiler nvcc. Functions running on the device are prefixed with _global_. Multiple copies of a device function can be launched simultaneously. Each one runs in its own thread, and is supplied with special variables blockIdx and threadIdx to identify itself.

The following code snippet in 4.5 is a rewrite of the function op\_conv\_kernel in 4.3. The compiler recognizes this function as a device function since it’s prefixed with _global_. It will be compiled separately into GPU machine code. Notice that its function arguments do not include the tuple \((h, i, j)\). Instead, they are computed from the special variables blockIdx and threadIdx.

This device function is remarkably similar to the regular CPU version. The only difference is that since threads are launched in batches, some redundant threads may be launched. We need to tell these threads to return without doing anything, as shown on line 9 and 10. The variable \(c\_table\) on line 7 is a global variable in the constant memory, a kind of memory specific to the GPU. We
store the configure in \textit{constant memory} for faster access.

Listing 4.5: Device function computing inner product

```c
__global__ void op_conv_kernel(...)
{
    int h = blockIdx.z * blockDim.z + threadIdx.z;
    int i = blockIdx.y * blockDim.y + threadIdx.y;
    int j = blockIdx.x * blockDim.x + threadIdx.x;

    const int *o4 = &c_table[table_offset];

    if( !(h < extent.3.0 && i < extent.3.1 && j < extent.3.2) )
    return;

    const float *p3_1 = o1 + i * stride.1_0 + j * stride.1_2;
    const float *p3_2 = o2 + (h*extent.4_0) * stride.2_0;

    float f = 0;
    for(int k=0; k < extent.4_0; ++k) {
        const float *p2_1 = p3_1 + stride.1_0 * o4[k * stride.4_0 + h];
        const float *p2_2 = p3_2 + h * stride.2_0 * k;
        for(int m=0; m < extent.2_1; ++m)
            f += p1_1++ * p1_2++;
    }

    float *p0_3 = o3 + h*stride.3_0 + i*stride.3_1 + j*stride.3_2;
    const float *p0_5 = o5 + h * stride.5_0;
    *p0_3 = stdsigmoid( f + *p0_5);
}
```

The function \textit{op\_conv} in 4.4 executes \textit{op\_conv\_kernel} in sequence with a single thread. This function also needs to be modified to instead launch a grid of P×(N-k+1)×(N-k+1) threads, with each thread running a copy of the \textit{device function}.

\textit{CUDA} provides a symbol ((⟨⟩)) between the \textit{device function} name and its parentheses to mark a thread-launching point. The compiler will fill in the routine instructions that create threads, copy the \textit{device function} machine code to GPU, and start the GPU threads. The code in 4.6 shows the new \textit{op\_conv} that launches parallel threads, instead of having 3 nested loops.

The arguments passed into the ((⟨⟩)) specify the cube of threads, P×(N-
k+1)\times(N-k+1) in our case. These threads are grouped into a 2-level hierarchy, since the number of threads may exceeds the number of available execution units. The threads in a block, whose size is specified by block size, are guaranteed to execute at the same time. The P\times(N-k+1)\times(N-k+1) threads are grouped into a grid of blocks, as specified by grid size. The grouping of threads is device dependent, and can have an impact on the execution speed.

The threads in a block, whose size is specified by block size, are guaranteed to execute at the same time. The P\times(N-k+1)\times(N-k+1) threads are grouped into a grid of blocks, as specified by grid size. The grouping of threads is device dependent, and can have an impact on the execution speed.

Listing 4.6: Host function for convolution operation

```c
#define BLOCK_WIDTH 16
#define BLOCK_HEIGHT 16

void op_conv(...) {
  int sz_0 = (int) ceil( extent_3_0 / (float) 1);
  int sz_1 = (int) ceil( extent_3_1 / (float) BLOCK_HEIGHT);
  int sz_2 = (int) ceil( extent_3_2 / (float) BLOCK_WIDTH);
  dim3 block_size( BLOCK_WIDTH, BLOCK_HEIGHT, 1);
  dim3 grid_size( sz_2, sz_1, sz_0);
  op_conv_kernel<<< grid_size, block_size >>>(...);
}
```

4.5 Performance improvement with GPU

In this section, we will profile the GPU implementation of convolutions as described in 4.5 and 4.6, tested on a NVIDIA GTX 560 Ti GPU. We show that with the small code changes, the convolution operation can be run on the GPU with 39× speedup, compared against running on an Intel CPU using a single core.

We use the same testing image of size 216×120, and the same convolution net with layers c-s-c-s-c, with the c-layer filter kernels of size 5×5, 6×6, 6×6, and the s-layer subsampling rates of 4×4 and 3×3. The configuration tables for the c-layers have parameters d×P as 1×8, 4×24, and 24×100.

The GTX 560 Ti GPU has 384 cores with clock frequency 1.7GHz. The GPU has 1GB global memory of clock rate 2GHz. It has CUDA capability.
<table>
<thead>
<tr>
<th>configure</th>
<th>CPU(ms)</th>
<th>GPU(ms)</th>
<th>speedup</th>
</tr>
</thead>
<tbody>
<tr>
<td>1x8x(212x116)x(5x5)</td>
<td>10.4</td>
<td>0.180</td>
<td>57</td>
</tr>
<tr>
<td>4x24x(48x24)x(6x6)</td>
<td>5.3</td>
<td>0.128</td>
<td>41</td>
</tr>
<tr>
<td>24x100x(11x3)x(6x6)</td>
<td>3.8</td>
<td>0.196</td>
<td>19</td>
</tr>
<tr>
<td><strong>total</strong></td>
<td><strong>19.5</strong></td>
<td><strong>0.5</strong></td>
<td><strong>39</strong></td>
</tr>
</tbody>
</table>

Table 4.2: Comparing the running time of CPU and GPU implementations

version 2.1. This capability mainly concerns the hardware’s thread scheduling, memory accessing mechanism etc.

We use the CUDA SDK toolkit version 4.0, including compiler, debugger, profiler, and runtime library. The NVIDIA profiler *compute prof* generates detailed information about the function running time, as well as a time trace of the entire program.

Table 4.2 shows the real running time of the 3 *op_conv* operations, on CPU and GPU respectively, and the speedups. The column for CPU is the same as reported in table 4.1. On average, the GPU implementation achieves 39× speedup against the one that runs on a CPU. This is about 46 GFLOPS performance.

The profiler *compute prof* also generates a time trace of the program, as shown in 4.2. This width plot shows the time stamp of various *device functions*, represented by dark colored bars. The blank area is when the code is running on the CPU. The unit of the plot is micro-second.

![Figure 4.2: The time trace of the object detection program](image)

In the time trace, the 3 *op_convolution* invocations and 2 *op_subsamples* add
up to about 0.8 ms, decreased from the 22.4 ms of the CPU version. They account for 30% of the running time of \textit{object\_detect}, unlike in 4.1 where they takes more than 70% of the running time.

The preprocessing part of the \textit{fprop}, where the image is resized, filtered and histogram equalized, uses the \textit{NPP} library routines for GPU. Using function from \textit{NPP} also gains some speedup against the \textit{OpenCV} library routines for CPU.

The speedup of the system’s overall running time is governed by the \textit{Amdahl’s law}. The law formulates the speedup as a function on the runtime percentage $p$ of the part of code that can be parallelized, and the level of parallelization $N$: $1/[(1 - p) + p/N]$. It tells the bad news about the parallelization. Assume $p$ is 50%, then when $N$ approximates infinity, the speedup saturates at 2. That is, we can at most get 2× speedup, no matter how much effort we put into parallelizing the 50% of the program that can be parallelized.

In our work, about 90% of the entire \textit{object\_detect} procedure can be parallelized and run on a GPU. Hence it can reach about 10× speedup in theory. The example testing image of size 216×120 takes about 2.6 ms to run a system with the afore mentioned GPU, down from 20 ms on an Intel CPU.

4.6 Modular Design of Convolution Net

To close the discussion about the implementation, we briefly describe the data structures and functions used. The functions described in the previous sections, such as the \textit{op\_conv} in 4.2, are low level functions which use arguments of plain pointers and integers. These functions need to be wrapped in higher level ones that are more descriptive and easier to understand and debug.

The basic data structure used is the \textit{multi\_array} type from the \textit{boost} library. It is a C++ template class parameterized with the element type and the dimen-
sionality of the array. The inputs and outputs of the convolution operation are 3-dim arrays with floating point elements, as \texttt{multi\_array\{}float, 3\}\texttt{.}

The convolution net is defined as a \textit{class} that includes all the interim results of the convolutions and sub-samplings, as well as the parameters of the convolution kernels, sigmoid biases, subsampling rates and scaling coefficients, as shown in 4.7.

Listing 4.7: Conv net class

```cpp
struct lenet7 {
  multi_array<float, 3> c_kernel_0;
  multi_array<int, 2> c_table_0;
  multi_array<float, 1> c_bias_0;
  multi_array<float, 1> s_coeff_0;
  size_t s_stridei_0;
  size_t s_stridej_0;
  multi_array<float, 1> s_bias_0;
  ...
  multi_array<float, 3> c0_state;
  multi_array<float, 3> s1_state;
  ...
  lenet7(const char *filename);
  fprop(const multi_array<float,3>&, multi_array<float,3>&);
};
```

The parameters are initialized when an object of this class is constructed.

The interim arrays are resized before the forward propagation, at which point the size of the input image is known, hence the interim array sizes can be calculated.

The forward propagation operation, \textit{fprop}, is defined as a method of the class, as shown in 4.7 on line 18. Its interface is very straightforward, with parameters for input array and output array, both of type \texttt{multi\_array\{}float, 3\}\texttt{.}

The output array is resized according to the input array and convolution kernel sizes. The function body, as shown in 4.8, consists of a sequence of alternated invocations of the functions \textit{op\_convolution} and \textit{op\_subsample}. 

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The convolution operation is implemented in the function `op_convolution`. The inputs to this function are the 3-dimensional input array `in`, the stack of convolution kernels `kernel`. The output `out` and the inputs are all of type `multi_array<float, 3>`. The function is also passed in the configure table `table` and the sigmoid bias as `bias`. This function invokes the `op_conv` in 4.2, by extracting the data pointer, dimension sizes, and strides.

There have been other independent research efforts on implementing fast convolutional networks with GPUs. The one in [44] implements 2D convolution with highly-optimized GPU code for training a large convolutional network.
Chapter 5

Object Recognition System

We have described an object recognizer based on convolution networks, and its GPU implementation in the previous chapters. To make the technology usable in the real production environment, a robust and scalable infrastructure is needed to support the recognition engine. More importantly, the system should be able to interact with the outside world.

In the internet advertisement industry, many players interact with each other to serve ads with targeted webpages, in the hope that the ad will get user clicks and hence generate revenues. Figure 5.1 shows the essential players.

Figure 5.1: The data-flow in the online advertisement system
In the process, the internet user browses a certain webpage by sending a request to the website. The requested webpage includes fixed contents (texts, images, and videos), as well as spaces reserved for ads that needs to be decided dynamically.

In our case, the ads are decided based on the prominent image on the current page. The website sends the image URL to the exchange server. The exchange server then offers to “sell” the ad space to the highest bidder. Our system acts as the ad network in this scenario. In this role, we have an inventory of ads to be served to webpages. Equipped with our object recognition system, we can find the optimal webpages by understanding the image. And we bid for the optimal ad spaces, under the assumption that serving ads on the right pages will garner higher clicking rates.

Therefore, we need to interact with the exchange server. Our system should be able to receive recognition request over the web, retrieve the requested image, give the image an unique ID, send the image into the recognition engines to be processed, and send back the recognition result once it is done.

As shown in 5.2, our system includes a high throughput webserver responding to web requests, a database that stores the recognition results, a queuing system that hold temporarily requests and results, and a recognition engine that takes requests from the queue and produce recognition metadata on the requested images.

In the following sections, we discuss each subsystem.

5.1 Webserver

At the front end, the system receives web requests from the ad exchange. We need a web server module for the task. Its functionality is simpler than a traditional web server, such as Apache, in that it does not need to serve static
contents. But the module needs to be efficient and robust to handle high volumes of requests. We built the module with the *pion-net* library. Our module is able to process more than 10,000 simultaneous connections.

The *pion-net* library has a HTTP specific layer which makes building a HTTP based web server easy. The library is based on the *boost::asio* library to achieve scalable performances. *asio* handles socket connections with non-blocking mechanisms, by using multiplexing system calls, such as *epoll* on the Linux platform. With its an asynchronous interface, programming is easy. The user just needs to write the callback code, which will be called when the I/O is finished.

### 5.2 Storing and Caching Results

When a new image is uploaded to the web server and made public, it usually will be viewed many times in a window of several days. Using a database to remember the recognition result from the same picture will save a lot computations, and also makes the recognition appears to have a very low latency from the second time the image is requested onward.

As shown in the figure 5.2, when the web server receives a request to recognize an image, it first check in a database to see if we already have the result for the image, possibly from a previous request with the same image.
If the image is seen for the first time, the front end puts the request and image ID in a request queue to be processed by the back end. When the back end finishes the processing, the result will be put in the result queue and then stored into the database. If this image is requested again later, we can get an immediate answer.

We can see that the web server only reads from the database, it never write directly into the database. Traditional SQL databases, including MySQL, tend to have slow reads. To bridge the fast web server module and the slower database, we need some caching mechanism.

We use the memcached to cache MySQL reads. It uses the image ID as the hashing key to retrieve the value that has the metadata about the image. This pairing usage of memcached and MySQL is a quite common practice in the industry.

5.3 Queuing System

The queuing system is the central link that connects the front end of web server and the back end of a set of worker instances, as shown in figure 5.2. When the web server finds that the request it received does not have an answer in the database, it puts the request in the request queue. The workers later pick up the requests and process them, and put the recognition results in the result queue when done. The contents in the result queue are periodically taken out and inserted into the database in the front end, so that they can be retrieved and sent back to the requesting clients.

With the queuing system, the front end and the back end loosely coupled. They are mutually unaware of the existence of each other. This makes the system more robust. Both the front end and the back end could fail without affecting the other. Queues should be tolerant to message overflow and underflow.
when only one end of the system stops working.

We surveyed and experimented many existing queuing systems. Many of them can not meet the requirements of heavy load and high throughput. We used a modified version of Q4M as our queuing system. Q4M is a plug-able storage engine for MySQL. Any table in MySQL using this storage engine works like a queue. One row in the table is like a message in the queue. The rows can not be sorted, only accessed in the order as they are inserted. It supports only *insert* and *delete*, not *update*.

5.4 Recognizers

The back end of the system is the set of recognizers, or workers as they are called in figure 5.2. They are configured to recognize and locate a specific type of objects in images. These workers retrieve images from the queue, producing results in uniform formats, and putting results into the result queue. They function independently.

The workers run on virtual computer instances in a cloud computing environment. Each worker is integrated into an instance image, which will automatically run after booting up, and knows which *request queue* to read from, and which *result queue* to write results to.

These instances can be dynamically scheduled. When the demand increases, the system only needs to start up more instances to have more workers. We used the cloud platform EC2 (Elastic Compute Cloud), provided by Amazon, to manage the instances. The system is scalable, and cost effective compared with using physical computing resources.
Chapter 6

Conclusions

This work is a natural extension of our previous works in [15, 36]. In this work, we have labelled a large set of images for objects of four categories, sailboat, car, motorbike, and dog. We have trained binary decision convolutional networks to differentiate images of a specific category against backgrounds. We have also used training processes similar to bootstrapping to improve recognition accuracies. The recall-precision curves are used to measure the accuracies of our classifiers. Object detectors are also built to find objects of different sizes up to 13 scales. The object classifiers and detectors have shown good performances.

Looking into the future, certainly detectors for objects of more categories are desired. Larger training data sets are very needed to improve the performances of the current classifiers. Some type of unsupervised learning for the early convolutional layers maybe beneficial to speed up the training process, as indicated in [38].

We built the detectors and classifiers for both CPU and GPU. The convolution operations show $40\times$ speedup when running on GPU. The GPU technology is still at the early stage of development, and is constantly evolving. We should
adapt our system to the new developments on both architecture and software fronts, to make the system even more efficient.

We have tested our work in the real world environment of online advertisement. Such applications provides solid validation to the research work. Infrastructures have been built to enable our system to process millions of images on daily basis. At the same time, we expect more engineering challenges to bring the research work into real applications.
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