

Machine Learning Course - CS-433

Neural Nets – Convolutional Nets

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ÉCOLE POLYTECHNIQUE
FÉDÉRALE DE LAUSANNE

Basic Structure of Convolutional Nets

Recall that for the standard neural network structure we introduced, every node at level l is connected to every node at level $l - 1$. The advantage of this structure is that it is very general and powerful. The disadvantage is that such a network has a large number of parameters and so generically lots of data is needed to train it.

In some scenarios it is intuitive that “local” processing of data should suffice. E.g., consider an audio stream given by samples $x^{(0)}[n]$. It is natural to process such a stream by running it through a linear time-invariant *filter*, whose output, call it $x^{(1)}[n]$, is given by the *convolution* of the input $x^{(0)}[n]$ and the filter $f[n]$,

$$x^{(1)}[n] = \sum_k f[k]x^{(0)}[n - k].$$

The filter $f[n]$ is often “local”, i.e., $f[k] = 0$ for $|k| \geq K$. Much of signal processing is based on this type of operation. By choosing an appropriate type of filter we can bring out various aspects of the underlying signal. E.g., we can “smooth features” by averaging or we can enhance differences between neighboring elements by taking a so-called “high-pass” filter.

We have essentially the same scenario if we think of a picture. Now the signal $x^{(0)}[n, m]$ is two-dimensional and the convolution takes the form

$$x^{(1)}[n, m] = \sum_{k,l} f[k, l]x^{(0)}[n - k, m - l].$$

As before, we can take filters $f[n, m]$ that are “local” i.e., only have non-zero coefficients for small values of $|n|$ and $|m|$.

There are two important aspects about this structure. First, the output $x^{(1)}$ at position $[n, m]$ only depends on the value of the input $x^{(0)}$ at positions close to $[n, m]$. If we use this structure in a nn then we no longer get a fully connected network but the structure is much more sparse and local. This implies that we have significantly fewer parameters to deal with.

Second, this structure implies that we should use the *same* filter (e.g., not only the same connection-pattern but also the same weights) at every position! This is called *weight sharing*. Weight sharing drastically reduces the number parameters further.

Figure 1 shows two layers of a very small nn where we see the difference between a fully connected network and one where connections are sparse and only local.

Why is it meaningful to use the same filter at every position in the network? Consider e.g. a photo. Photos are inherently “shift invariant.” We can imagine that there is an essentially infinite-sized 2D photo in reality and we are seeing a small portion of it. The portion we are seeing is more or less random and so the exact position of any object in this photo is more or less random as well. It therefore makes sense that we treat each position essentially equally.

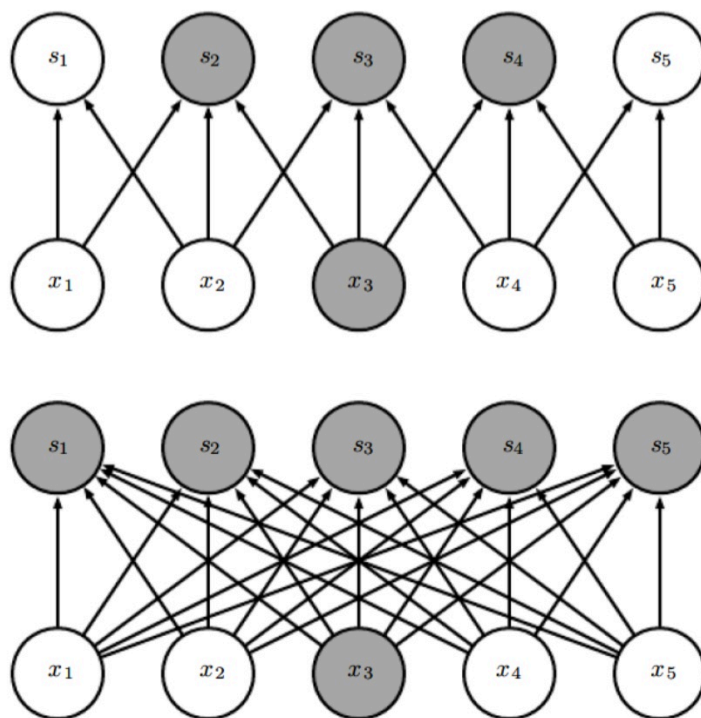


Figure 1: Fully connected versus sparse connections.

Layout

It is common to lay out the data in a convolutional network according to its “natural” layout. E.g., if the input is a photo then it is natural to use a 2D layout, whereas if the data is naturally one-dimensional then it makes sense to use a 1D layout.

Handling of Borders

Consider a 2D case. Assume that we have an input of dimension $n_1 \times n_2$ as shown in Figure ?? and a $k_1 \times k_2$ kernel. There are several natural ways of dealing with the boundary. The first is to *pad* the original input data with zeros at the boundary. More precisely, instead of considering the input of size $n_1 \times n_2$, consider the input to be of size $(n_1 + 2(k_1 -$

$1)) \times (n_2 + 2(k_2 - 1))$ where the “center piece” is the original data and there is an additional boundary of width $k_1 - 1$ and $k_2 - 1$ respectively which is set to 0. If we now perform the convolution then the output will again be of size $n_1 \times n_2$. For obvious reasons this is called *zero padding*.

The second option is to compute an output which is only of size $(n_1 - k_1 + 1) \times (n_2 - k_2 + 1)$, i.e., to perform the convolution only for positions so that the whole filter lies inside the original data. This is sometimes called *valid padding*.

Training

As we discussed, there are two aspects that make CNN special. First, only some of the edges are present. This means that the weight matrices are sparse and this does not require any change in the learning steps when we use SGD and backpropagation.

Second, weight sharing is used, i.e., many edges use weights that are the same. As we already mentioned in the previous lecture, with a small modification the back propagation algorithm can still be used to train a cnn with weight sharing: run backpropagation ignoring that some weights are shared, considering each weight on each edge to be an independent variable. Once the gradient has been computed for this network with independent weights, just sum up the gradients of all edges that share the same weight. This gives us the gradient for the network with weight sharing.

To see this, consider a simple example. Let $f(x, y, z)$ be a function from $\mathbb{R}^3 \rightarrow \mathbb{R}$. Let $g(x, y) = f(x, y, x)$. In words, z is no longer an independent variable but we set $z = x$. If

we now want to compute the gradient

$$\left(\frac{\partial g(x, y)}{\partial x}, \frac{\partial g(x, y)}{\partial y}\right)$$

then we can compute this by first computing

$$\left(\frac{\partial f(x, y, z)}{\partial x}, \frac{\partial f(x, y, z)}{\partial y}, \frac{\partial f(x, y, z)}{\partial z}\right)$$

and then realizing that

$$\left(\frac{\partial g(x, y)}{\partial x}, \frac{\partial g(x, y)}{\partial y}\right) = \left(\frac{\partial f(x, y, z)}{\partial x} + \frac{\partial f(x, y, z)}{\partial z}, \frac{\partial f(x, y, z)}{\partial y}\right).$$

FFT

The convolutional structure can be used in order to compute the output of a CNN very efficiently. To keep things simple, assume that our data layout is one-dimensional and that each layer is connected via a convolution where we use zero-padding at the boundaries. Let us consider the first layer to be specific. We then have

$$x^{(1)}[n] = \sum_k f[k]x^{(0)}[n - k].$$

zero padding cyclic convolution Fourier transform: convolution goes into multiplication FFT