Convolutional models

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Local connectivity

• Sometimes, it might make sense to have local connectivity in a network



Why local connectivity?

Local patterns might be seen in multiple parts of the input

What if the same pattern might occur in different places?



• How can we learn to recognize a local pattern anywhere in the input?

• If we can learn a set of weights that are applied to a local area, this can serve as a feature extractor.

• We need tools to apply these weights across the input.



Convolution

Operation that combines two signals (1D definitions):
 – Continuous

$$x(t) * w(t) = \int_{-\infty}^{\infty} x(a)w(t-a)da$$

– Discrete

$$x[t] * w[t] = \sum_{-\infty}^{\infty} x[a]w[t-a]$$



Notes on notation: Engineers denote continuous signals with (), discrete with [] Goodfellow et al. use (x*w)(t) rather than x(t)*w(t)

Convolution

- Formula outputs sample at time t.
- By convention:

$$\underbrace{x[t]}_{input} * \underbrace{w[t]}_{kernel}_{(filter)} = \sum_{a=-\infty}^{\infty} x[a]w[t-a]$$

- The kernel shapes the response of the convolution and is time reversed.
- Kernel usually has finite support



Convolution in 1D





Demonstration courtesy of Karlsruhe Institute of Technology

https://phiresky.github.io/convolution-demo/

Convolution in 2D

• For audio, typically used on spectrograms or other time/frequency representations:

$$x[t,\omega] * w[t,\omega] = \sum_{m} \sum_{n} x[m,n]w[t-m,n-\omega]$$

• Many libraries implement cross-correlation instead of convolution (kernel not reversed, not important in practice):

$$\sum_{m} \sum_{n} x[t+m,\omega+n]w[m,n]$$

• Convolution is applied for each time-frequency cell $[t, \omega]$







Convolution in 2D



12.0	12.0	17.0
10.0	17.0	19.0
9.0	6.0	14.0

feature map

$$w = \begin{bmatrix} 0 & 1 & 2 \\ 2 & 2 & 0 \\ 0 & 1 & 2 \end{bmatrix}$$



Animation credit: Irhum Shafkat

Multiple kernels

- Suppose > 1 kernel, e.g., 10
- In the previous example, we would have 10 3x3 feature maps
- If the next layer convolved another 10 kernels across each of the 10 feature maps, we would have 100 feature maps.
- To prevent combinatorial explosion, we fuse the feature maps between layers, usually by addition, possibly after using an activation function



Key ideas

• Kernel is a learned feature extractor

- Desirable properties
 - Sparse interactions
 - Parameter sharing
 - Equivalence under translation



Sparse interaction

- Output points only depend on a local neighborhood.
- Similar to cutting connections (weights to zero) in a dense layer





Parameter sharing

- We slide the convolutional layer across the tensor.
- Each position produces a new output.
- We use the same kernel, thus the outputs from a single kernel have tied (shared) weights.



Receptive field

Subsequent convolutions increase the area that affects a layer output





Equivariance under translation

When something in the input shifts position, its representation (convolution with the kernel) will also shift position.



Image: John Tlumacki, Boston Globe





Convolutional layers learn a representation





North Atlantic right whale upcall

Key ideas

- For audio, convolutional layers frequently followed by
 - an optional RNN
 - flattening and a feed-forward network to perform the classification.



Common architecture

• Linear activation in first stages (not all networks use this)

• Nonlinear activations

• Pooling





Pooling layers

- Reduce dimensions
- Helps with translation invariance
- How it works:
 - specify mask size (2x2)
 - specify stride (2)
 - apply statistic to covered cells
 - Example is max-pooling





Pooling

- Can be seen as an alternative to resizing an input to a fixed size
- Example:
 - Learn a 9x9 feature vector to send to classification layer
 - Can adjust the pooling step sizes so that end result is 9x9 for many input sizes.



Convolutional layer connections

- Each convolutional layer *l* has l_f filters $f_l(i), 1 \le i \le l_f$.
- In layer l+1, most deepnet libraries will combine the l_f filter outputs with each of l+1's convolutional filters:

$$fout_{l+1}(i) = b_{l+1}(i) + \sum_{j=1}^{l_f} fout_l(j) * f_{(l+1)}(i)$$



How are CNNs used in audio

- Two main paradigms
 - Image processing paradigm
 - Spectrogram
 - Multiple convolutional and pooling layers
 - Framed speech
 - 1D convolution on time domain or spectrum
- Possible RNN network on extracted feature matrix
- Feed forward layer



Are CNNs appropriate for audio?

Not always:

• Position has meaning

• Harmonic structure changes across the spectrum



Harmonic structure













Shifting frequency changes meaning...





Harmonic structure

- Unless using constant-quality representations harmonics change differently
- A CNN filter may learn harmonic structure that is appropriate for one frequency, but not for a frequency that a little different:
 - $-100\ 200\ 300\ 400\ Hz\ vs$
 - $-\ 120\ 240\ 360\ 440\ Hz$



Keras convolutional layers

from tensorflow.keras.models import Sequential

from tensorflow.keras.layers import Dense, Input, Conv2D, MaxPooling2D
model = Sequential()

model.add(Input(shape=(1000,1000,1)) # 1000x1000 grayscale model.add(Conv2D(32, kernel_size=(5, 5), activation='relu'))) model.add(MaxPooling2D(pool_size=(2, 2), strides=(2, 2))) model.add(Conv2D(64, kernel_size=(5, 5), activation='relu')) model.add(MaxPooling2D(pool_size=(2, 2)))

Convert to vector, then feed forward network
model.add(Flatten())

```
model.add(Dense(1000, activation='relu'))
```

```
model.add(Dense(NumClasses, activation='softmax'))
```

San Diego State University

Use Conv1D for 1D convolution