



Seminar Advances in Deep Learning for Time Series (ADLTS)

Lecture 3: Deep Learning for Time Series

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Machine Learning and Data Analytics (MaD) Lab Friedrich-Alexander-Universität Erlangen-Nürnberg 17.10.2024



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Topics overview

Recorded Lectures

- I. Introduction
- II. The Tool Tracking dataset
- III. DL for Time Series
- IV. Time-aware models
- V. XAI for Time Series part 1
- VI. Active Learning for Time Series part 1
- VII. Semi-supervised Learning
- VIII.Domain-shifts, Ethics, and Bias
- IX. XAI for Time Series part 2
- X. Active Learning for Time Series part 2

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III. DL for Time Series

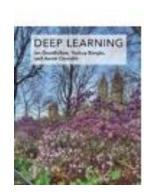
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References

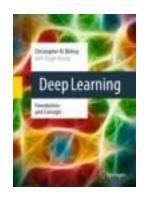
Deep Learning

by Ian Goodfellow, Yoshua Bengio, and Aaron Courville (2016)



Deep Learning: Foundations and Concepts

by C. Bishop, H. Bishop (2024)





Lecture outline

- 1. Introduction to Deep Learning
- 2. Convolutional Neural Networks (CNNs)
- 3. Recurrent models (RNNs and LSTMs)
- 4. Transformers







ADLTS \ DL for TS \ Introduction to Deep Learning





Why Deep Learning?

Previous method needed **handcrafted features**:

- MFCCs (speech processing) (1)
- I-Vector (speech processing) (2)
- Sift (scene alignment, videos) (3) → Needs expert knowledge about domain
- (1) Mermelstein, P. (1976). Distance measures for speech recognition, psychological and instrumental. *Pattern recognition and artificial intelligence*, *116*, 374-388.
- (2) V. Gupta, P. Kenny, P. Ouellet and T. Stafylakis, "I-vector-based speaker adaptation of deep neural networks for French broadcast audio transcription," *2014 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)*, 2014, pp. 6334-6338, doi: 10.1109/ICASSP.2014.6854823.
- (3) Lowe, D. G. (2004). Distinctive image features from scale-invariant keypoints. *International journal of computer vision*, 60(2), 91-110.



Why Deep Learning?

What if we can not define generally applicable features?

- High dimensional data
- Hard to come up with generally applicable features

- → With deep learning, we can find features in a data driven way
- → Can help capture complex non-linear relationships



The Human Brain

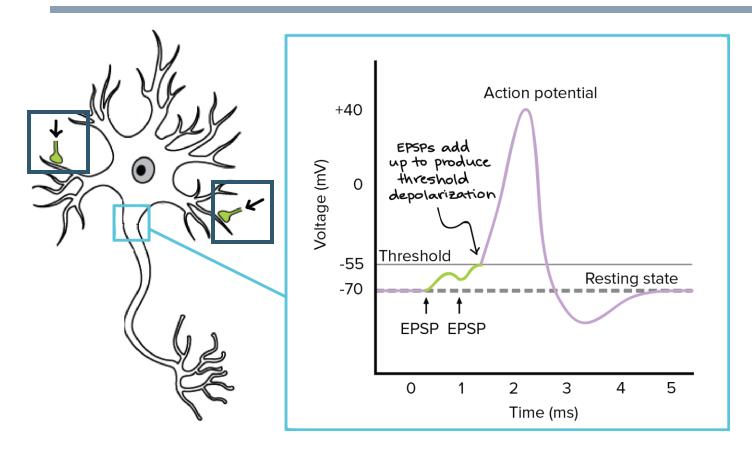
The human brain is our reference for an intelligent agent.

- It contains different areas specialized for some tasks (e.g., the visual cortex)
- It consists of neurons as the fundamental unit of "computation"





The Brain's Neuron



- Excitatory stimuli reach the neuron
- Threshold is reached
- Neuron fires and triggers action potential

EPSP = Excitatory postsynaptic potential



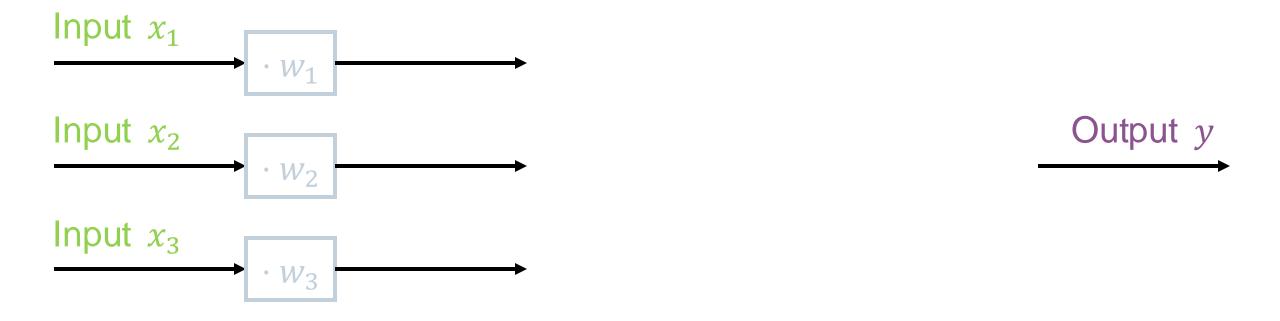
Let's build the computational model step by step:

1. Show the input and output of our neuron (which depends on the data and task)



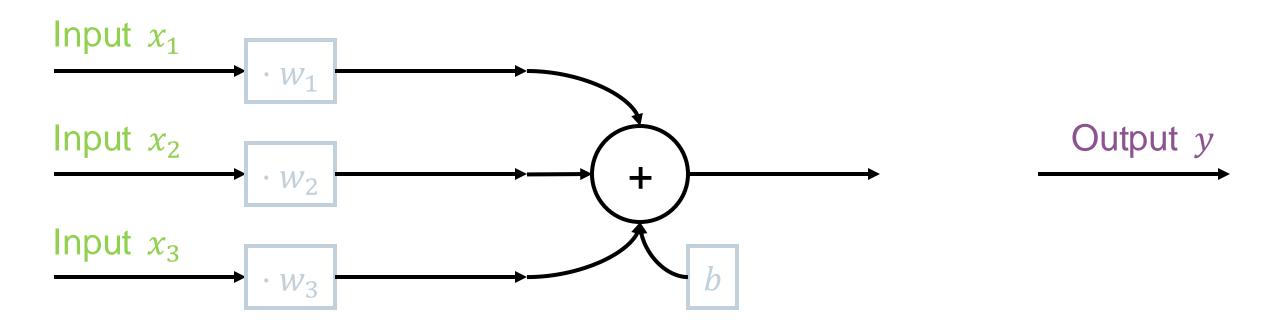


2. Weights can "select" or "deselect" input channels (not all are relevant for subsequent computations)



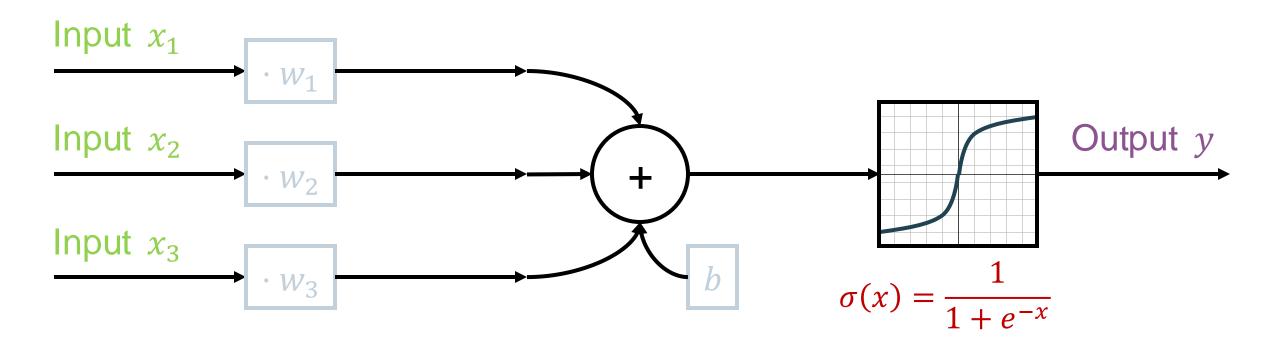


3. We add up all the excitatory signals and the **resting potential (or bias)** to determine the current potential.





4. A threshold function (or activation function) σ is applied to determine whether an action potential has to be sent in the output





Activation Functions

The goal of an **activation function** is to activate (or not) neuron.

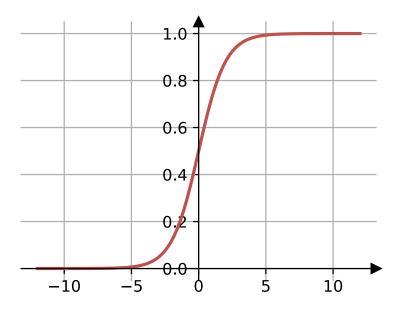
It brings **non-linearity** to a neural network:

→ The output is **not** a linear function of the input

Different activation functions can be considered, taking into account their:

- Output range
- Mean
- Gradient

Sigmoid

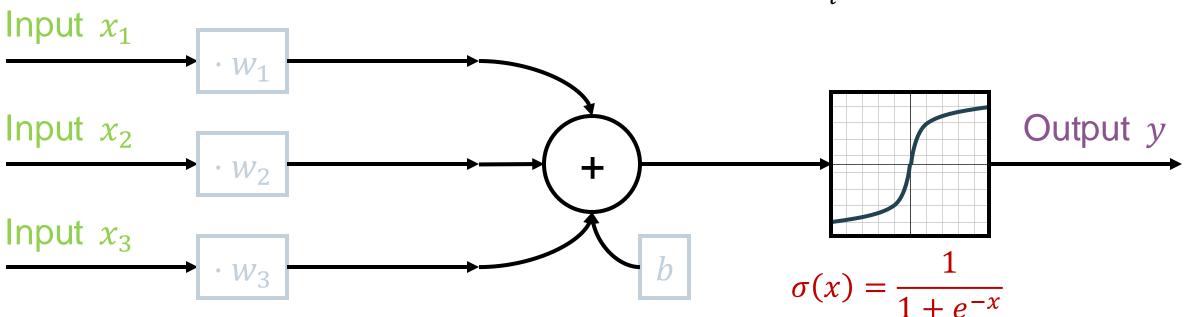


$$\sigma(x) = \frac{1}{1 + e^{-x}}$$



5. We can write the perceptron mathematical model to map inputs x_1, x_2, x_3 to the output y_1 using channel weights w_1, w_2, w_3, b :

$$y = \sigma(w_1 \cdot x_1 + w_2 \cdot x_2 + w_3 \cdot x_3 + b) = \sigma(\sum_{i} w_i \cdot x_i + b)$$



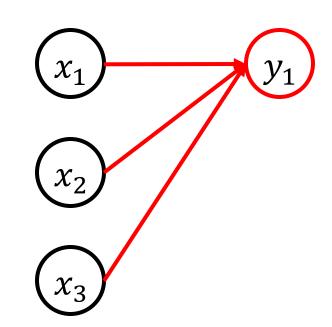


Single Layer Perceptron

We can combine multiple perceptrons to create a layer.

One perceptron has the following equation:

$$y_1 = \sigma(w_{11} \cdot x_1 + w_{12} \cdot x_2 + w_{13} \cdot x_3 + b)$$





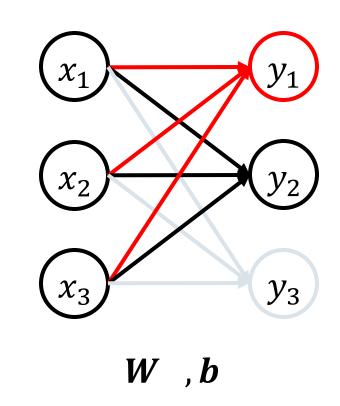
Single Layer Perceptron

We can combine multiple perceptrons to create a layer.

We can combine three perceptrons into one layer:

$$\begin{pmatrix} y_1 \\ y_2 \\ y_3 \end{pmatrix} = \begin{pmatrix} w_{11} & w_{12} & w_{13} \\ w_{21} & w_{22} & w_{23} \\ w_{31} & w_{32} & w_{33} \end{pmatrix} \begin{pmatrix} x_1 \\ x_2 \\ x_3 \end{pmatrix} + \begin{pmatrix} b_1 \\ b_2 \\ b_3 \end{pmatrix}$$

$$W$$



Or in a more simplified form: $y = \sigma(W \cdot x + b)$



Multi Layer Perceptron (MLP)

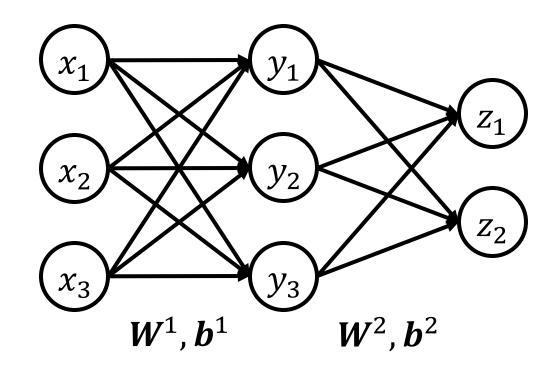
We can chain multiple layers

$$y = \sigma(W^1 \cdot x + b^1)$$

$$z = \sigma(W^2 \cdot y + b^2)$$

So:
$$z = \sigma(W^2 \cdot \sigma(W^1 \cdot x + b^1) + b^2)$$

- \rightarrow Each layer has its own set of parameters (weights W^i and bias b^i)
- → The activation function ensures that all layers do not collapse into one: it introduces nonlinearity





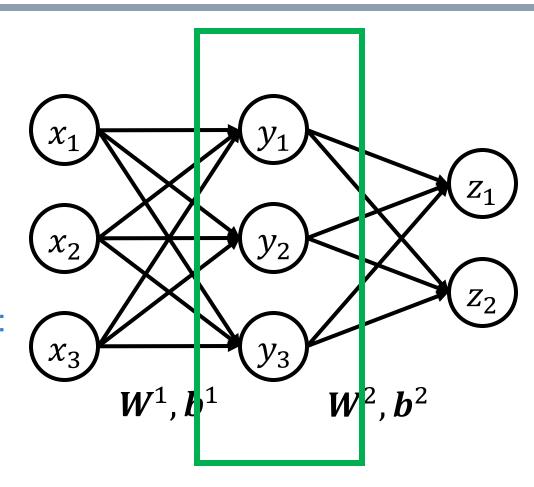
Multilayer perceptron (MLP)

We call "hidden layer" any layer in between the input and the output layers.

For example, this neural network (image on the right) is a Multi-Layer-Perceptron (MLP) with a **single hidden layer** (highlighted by the green box).

⇒ The underlying computation is described by: $y^{i+1} = \sigma(W^i \cdot y^i + b^i)$

Also referred to feedforward networks (no feedback connection)





How are Model Parameters Learned?

- 1) **Forward propagation:** This phase refers to the computation of the output using input and parameters.
- 2) **Loss calculation:** The output and expected output are then compare using a loss function.
- 3) **Backward propagation**: During this phase, the model computes the gradients of the loss with respect to each parameter (θ).

$\nabla \theta \leftarrow \text{Backward Pass}$

$$1.2 = x_1$$

$$y_1 = 2.9 \Leftrightarrow 3.2$$

$$1.4 = x_2$$

$$y_2 = 0.2 \Leftrightarrow 0.2$$

$$1.3 = x_3$$

Forward Pass $\rightarrow y = f(x, \theta)$



Backpropagation Algorithm: Overview

- 1. Calculate the forward pass and **store** results for \hat{y} , a_j^k , and z_j^k .
- 2. Calculate the backward pass and **store** results for $\frac{\partial L}{\partial w_{ij}}$, proceeding from the last layer:
 - a) **Evaluate** the error terms for the last layer δ_k
 - b) Backpropagate the error term for the computation of δ_i
 - c) Iterate to all previous layers
- 3. Combine the individual gradients (average)
- 4. Update the weights according to the **learning rate** λ







ADLTS \ DL for TS \ Convolutional Neural Networks (CNNs)

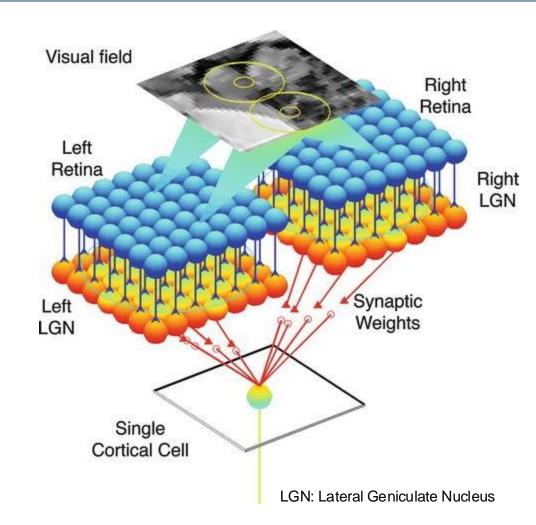




Motivation: The Human Visual Cortex

CNNs were inspired by a specialized brain area: the **visual cortex**.

- The visual cortex processes visual information collected by the retinae in a hierarchical manner.
- The **receptive field** of cortical neurons increases the later they are in this hierarchy.
 - Small receptive fields are stimulated by high spatial frequencies - or fine details
 - b) Large receptive fields are stimulated by low spatial frequencies or coarse details

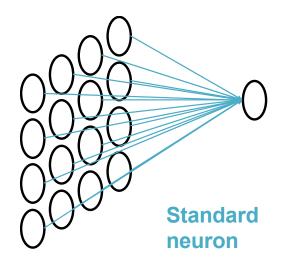


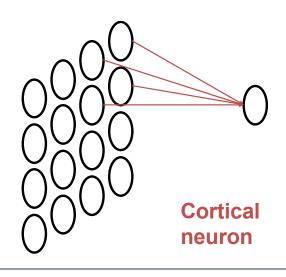


Modeling Receptive Fields

In standard multilayer perceptrons (MLP), layers are fully-connected, i.e. all units from one layer are connected to all units from the previous layer.

- Cortical neurons have small receptive fields: units from one layer have sparse and localized connection with units from the previous layer.
- → This is modeled by discrete convolutional operations.







Convolutional Operation

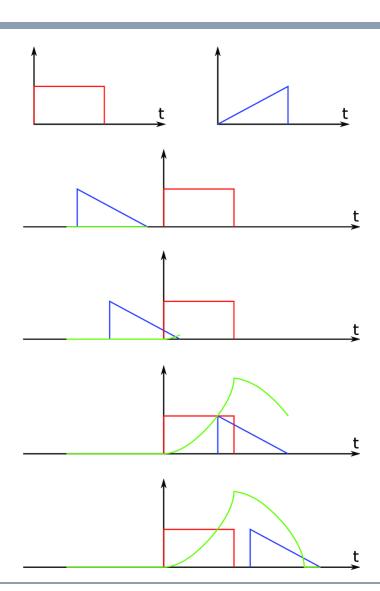
The **continuous convolution** is a operation on two functions f and g noted f * g that produces a third function.

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

It is the integral of the product of the two functions after one is reflected about the y-axis and shifted.

In deep learning, a **discrete** version of this operation is used.

Image from Pihlajamäki, Tapani. Multi-resolution Short-time Fourier Transform Implementation of Directional Audio Coding.





Discrete Convolution

The (discrete) **convolution** is a mathematical operation on two functions. On our case, the **input** and a smaller **filter** produce a third function (the feature map).

$$(input * filter)[n] = \sum_{m=-\infty}^{+\infty} input[m] \cdot filter[n-m]$$

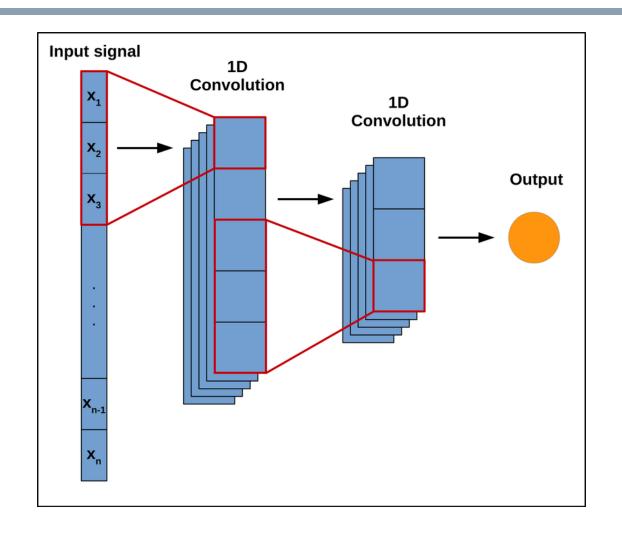
- → The sum is evaluated for all values of the shift.
- → The term convolution is used both to indicate the result function and process of computing it.



1-D Convolutional Neural Networks

CNNs can also be used to learn **temporal dependencies** on time series data.

- The convolution is applied along a single dimension, i.e., the temporal dimension.
- The resulting model is generally referred to as 1-D CNN.

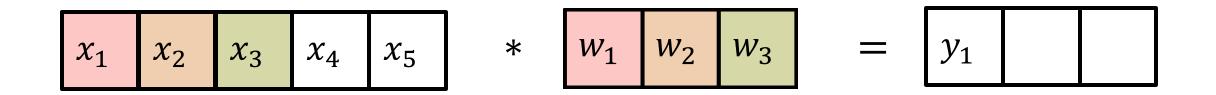




1D Convolutional Operation

Like the 2D operation, the output computation only depends on a **subset** of the time series:

$$y_1 = w_1 x_1 + w_2 x_2 + w_3 x_3$$

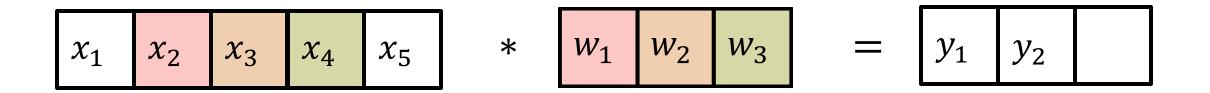




1D Convolutional Operation

Like the 2D operation, the output computation only depends on a **subset** of the time series:

$$y_2 = w_1 x_2 + w_2 x_3 + w_3 x_4$$

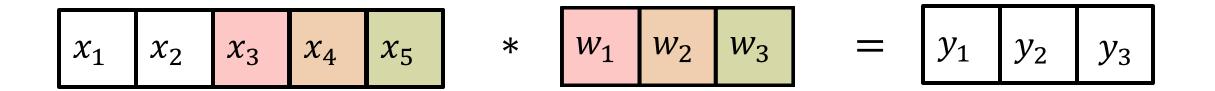




1D Convolutional Operation

Like the 2D operation, the output computation only depends on a **subset** of the time series:

$$y_3 = w_1 x_3 + w_2 x_4 + w_3 x_5$$





CNN Properties and Time Series

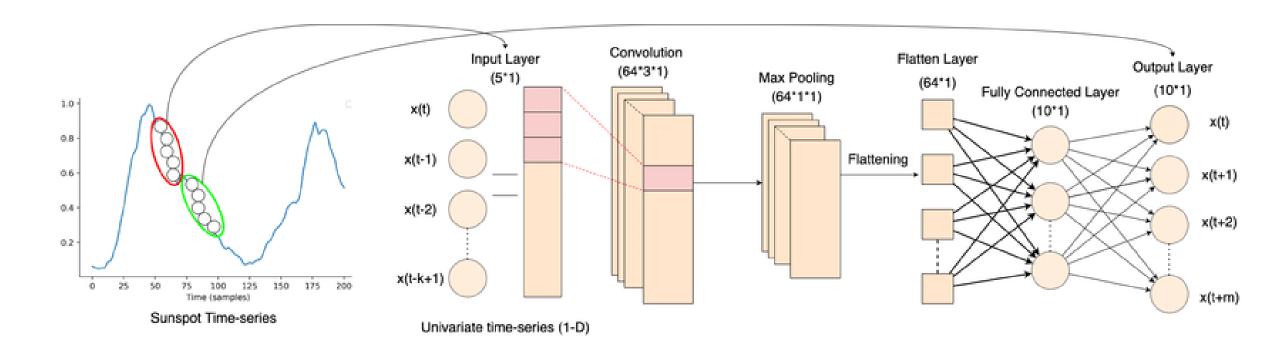
CNN properties make them very appropriate for **computer vision**. Those can also be useful for **time series**:

- Ability to extract local patterns: 1D CNNs can extract local patterns that occur in time series data (e.g. short term trends in stock prices, weather events)
- Translation Equivariance: 1D CNNs can identify those patterns regardless of their position, which is useful for tasks such as anomaly detection
- Dimensionality reduction of pooling layers: This can be useful when datasets are high dimensional.
- Hierarchical feature learning: Time series are composed of patterns at different time scales. (e.g. seasonal weather patterns vs sudden weather event)



CNN for Time Series Forecasting

Example of CNN for univariate time series forecasting [1].



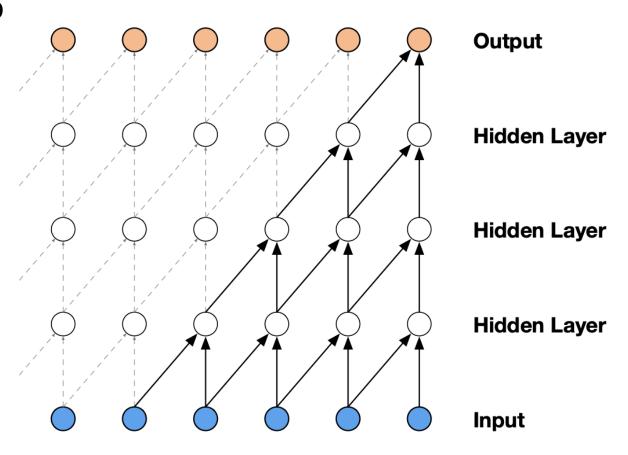
Chandra, Rohitash & Goyal, Shaurya & Gupta, Rishabh. (2021). Evaluation of deep learning models for multi-step ahead time series prediction. 10.48550/arXiv.2103.14250.



Temporal Convolutional Networks (TCNs)

The **causal convolution** is best suited to model causality in the data.

- For images it can be implemented by "masked convolutions", i.e., a tensor mask is applied before the actual convolution takes place.
- For 1D data, e.g., audio processing, it can be more easily implemented by shifting the output of a normal convolution by a few timesteps.









ADLTS \ DL for TS \ Recurrent models (RNNs and LSTMs)



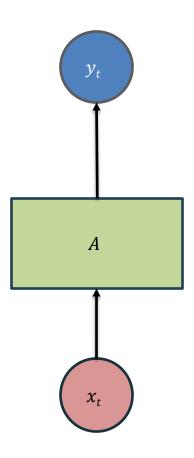


Limitations of NN for time series data

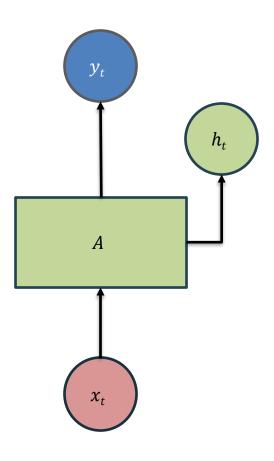
Feed-forward and Convolutional neural networks present some **disadvantages**, when applied to sequential data:

- 1. Cannot work online (sequence has to be fed all at once)
- 2. Consider only the current input
 - Canot memorize previous time steps
- 3. Cannot handle directly sequences of different lengths

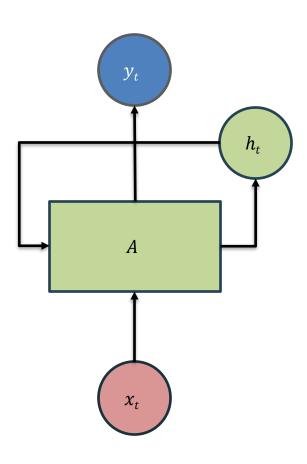




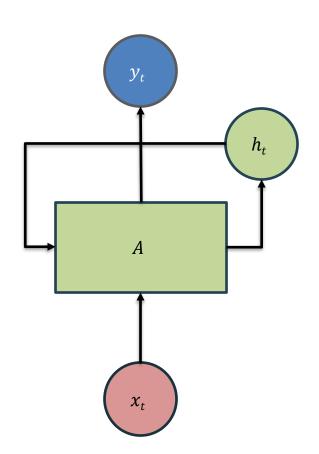












A recurrent neural network (RNN) is a neural network that contains feed-back connections.

- Activations can flow in a loop
- It allows for temporal processing

An RNN is composed by:

 x_t : input at time t

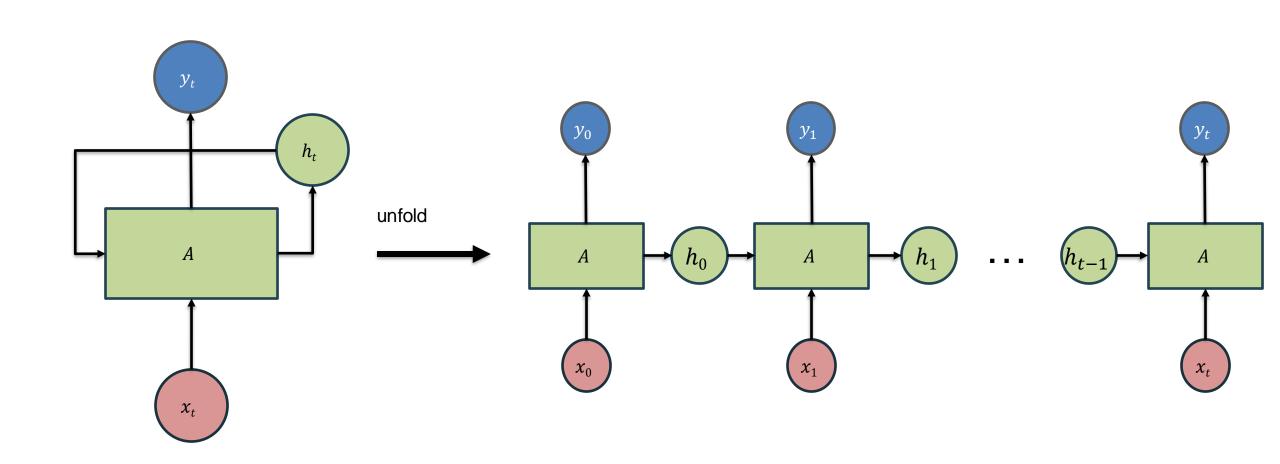
A: neural network

 h_t : hidden state

 y_t : output at time t

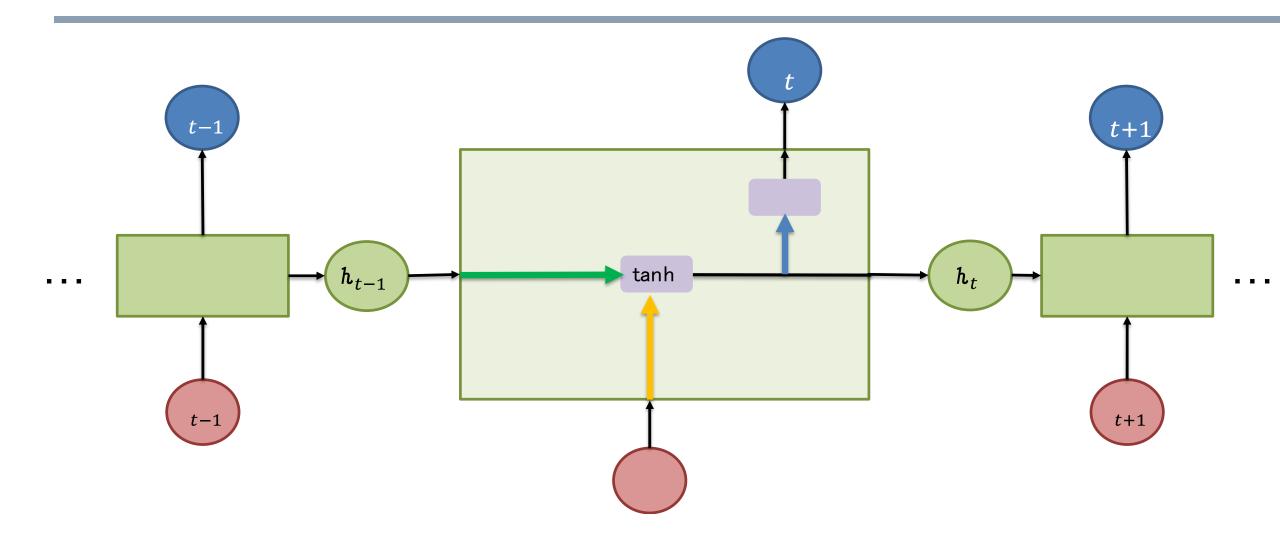


RNN Unfolding



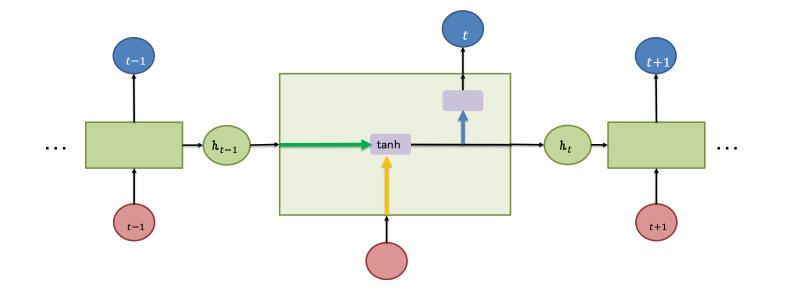


Mathematical formulation





Mathematical formulation



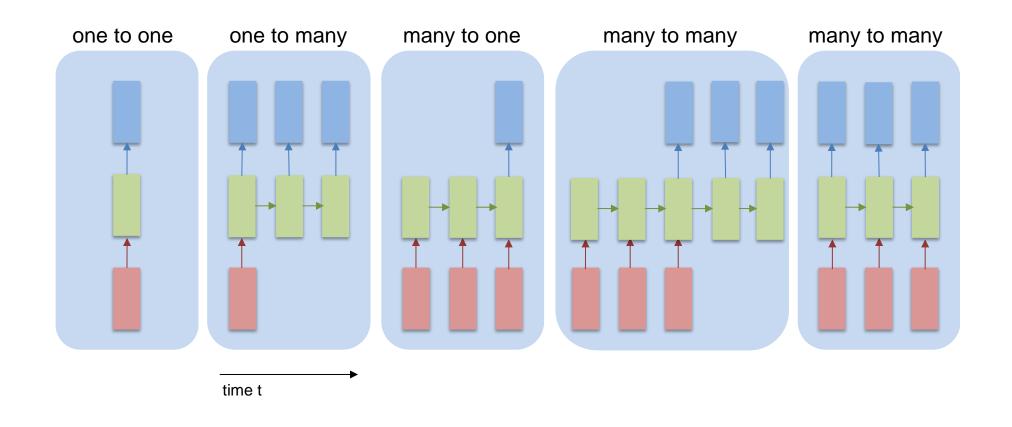
The behaviour of the RNN can be described as a **dynamical system** by the pair of non-linear matrix equations:

$$h_t = \tanh(\mathbf{W}_{hh}h_{t-1} + \mathbf{W}_{xh}x_t)$$
$$y_t = \sigma(\mathbf{W}_{hy}h_t)$$

The order of the dynamical system corresponds to the dimensionality of the state h_t .

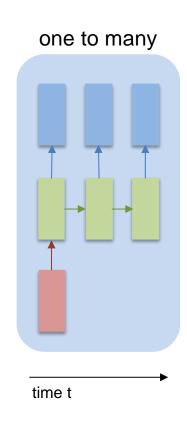


RNNs architecture





Example: one to many



A typical example of a one to many problem is that of **image captioning.**

Input:



Output:

4

cat

playing

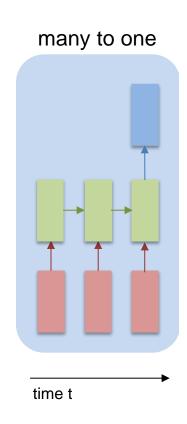
with

a

ball



Example: many to one



A typical example of a many to one problem is that of sentiment analysis.

Input:

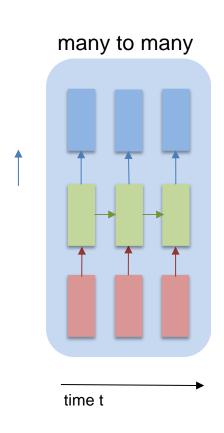
Horrible service the room was dirty

Output:





Example: many to many



A typical example of a many to many problem is that of name entity recognition.

Input:

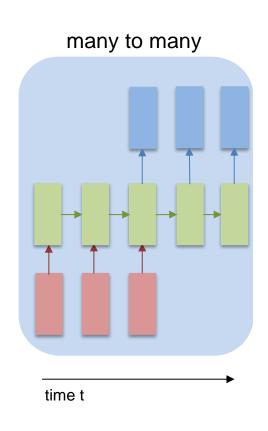
Harry Potter and Hermione invented a new spell

Output:

1 1 0 1 0 0 0



Example: many to many



Another example of a many to many problem is that of machine translation.

Input:

Horrible service the room was dirty

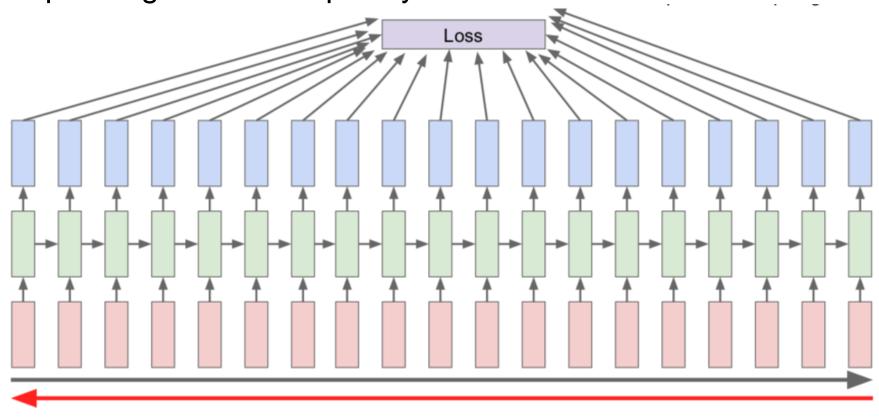
Output:

Un servizio orribile la camera era sporca



BPTT: Limitations

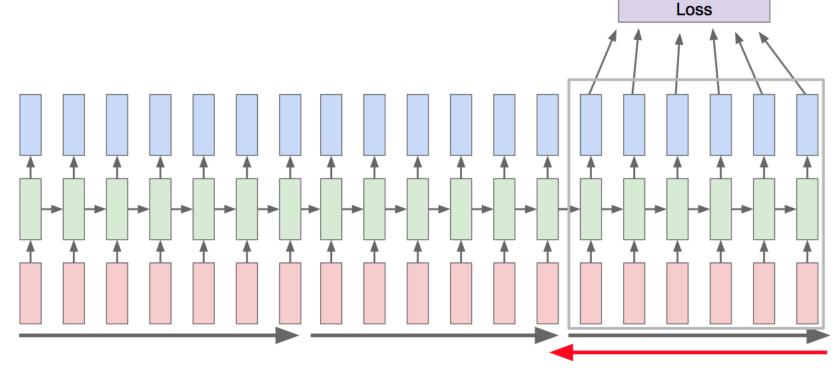
BPTT can be computationally very expensive as a lot of partial derivatives have to be computed, depending on the complexity of the network.



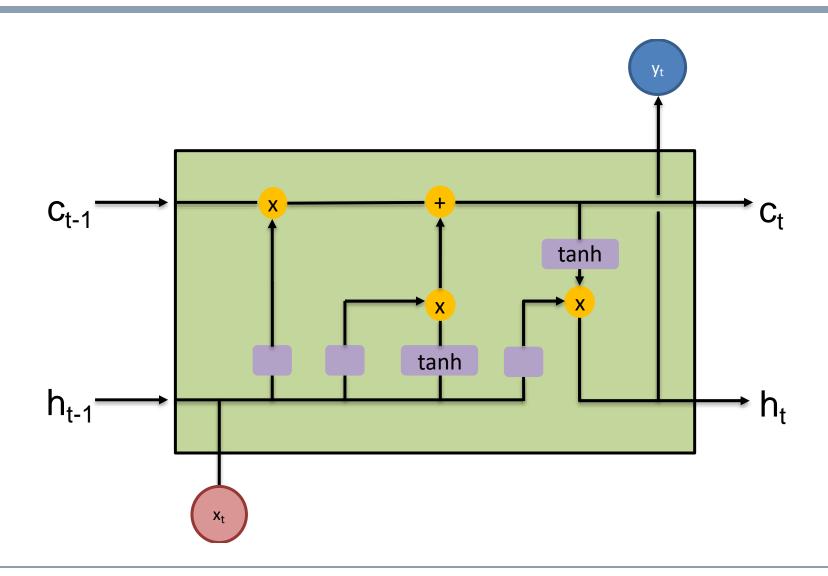


Truncated Backpropagation Through Time (Trunc-BPTT)

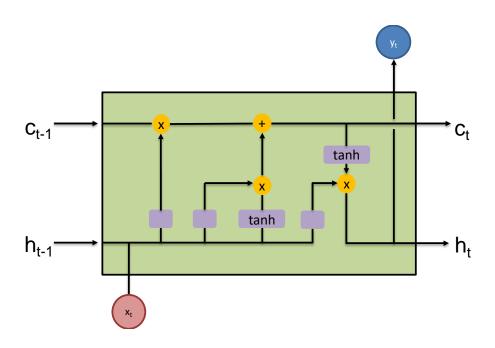
With the Truncated Backpropagation Through Time (**Trunc-BPTT**), instead of passing the whole sequence, we perform the forward and backpard pass on a subset.



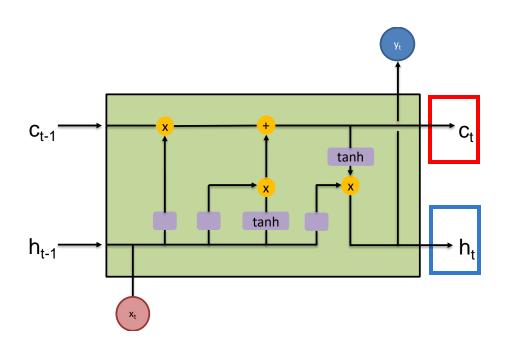








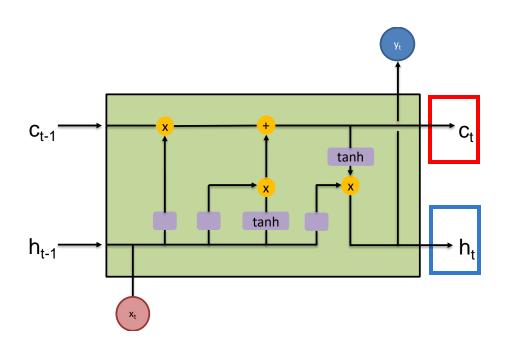




Observations:

• Compared to RNNs, additionally to the hidden state h_t , we have another state, c_t , said the cell state.

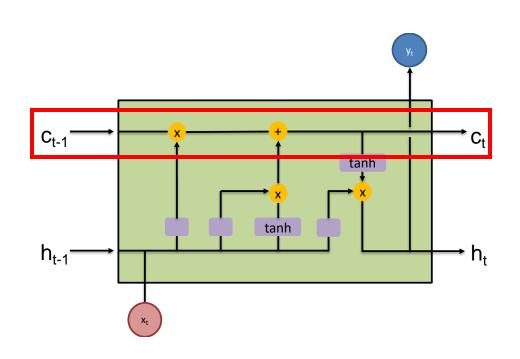




Observations:

- Compared to RNNs, additionally to the **hidden state** h_t , we have another state, c_t , said the **cell state**.
- The information flow is regulated by three gates: the forget gate, the input gate and the output gate.

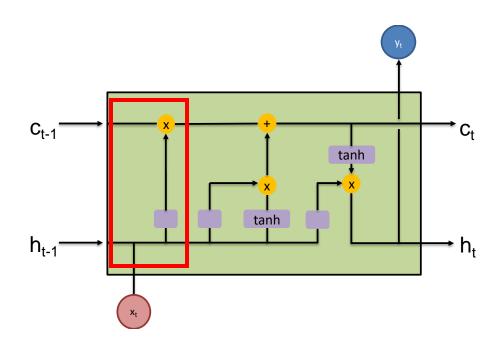




The cell state has:

- Only minor interactions
- Simple information flow
- Other gates regulates weather it is preserved/not-preserved or updated.

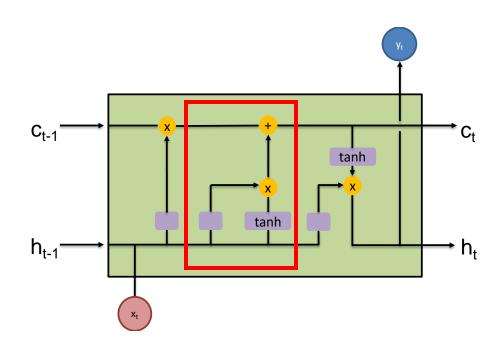




The **forget gate** decides how much information to retain from the previous cell state.

$$f_t = \sigma(W_f \cdot [h_{t-1}, x_t] + b_f)$$



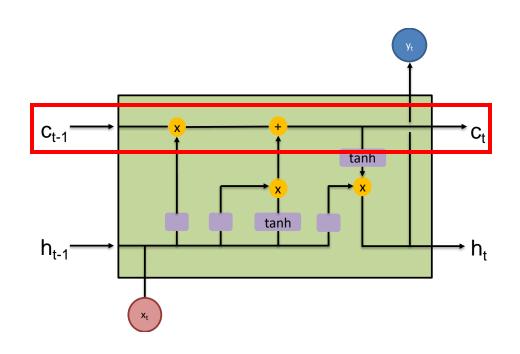


The **input gate** decides the information to be added to the cell state, based on the current input.

$$i_t = \sigma(W_i \cdot [h_{t-1}, x_t] + b_i)$$

$$g_t = tanh(W_c \cdot [h_{t-1}, x_t] + b_c)$$

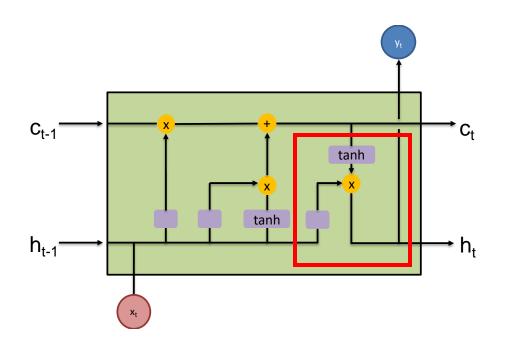




The values can be combined to update the cell state

$$c_t = f_t \odot c_{t-1} + i_t \odot g_t$$



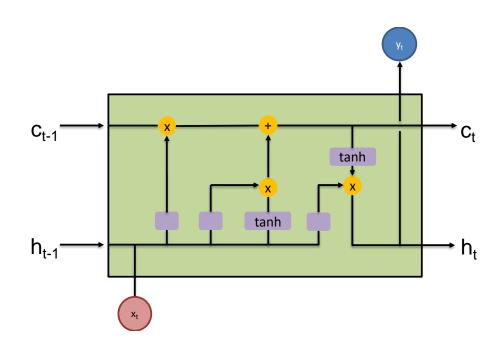


The **output gate** generates the output of the current LSTM cell, based on the current input, the previous output, and the updated cell state.

$$o_t = \sigma(W_o \cdot [h_{t-1}, x_t] + b_o)$$

$$h_t = o_t * \tanh(c_t)$$





To summarize, the LSTM cell is described by the following equations:

$$f_t = \sigma(W_f \cdot [h_{t-1}, x_t] + b_f)$$

$$i_t = \sigma(W_i \cdot [h_{t-1}, x_t] + b_i)$$

$$g_t = tanh(W_c \cdot [h_{t-1}, x_t] + b_c)$$

$$c_t = f_t \odot c_{t-1} + i_t \odot g_t$$

$$o_t = \sigma(W_o \cdot [h_{t-1}, x_t] + b_o)$$

$$h_t = o_t * tanh(c_t)$$







ADLTS \ DL for TS \ Transformers





Attention Is All You Need [1]

The **Transformer** [1] architecture was published in 2017. Models built on this architecture have become state-of-the-art in many domains, starting with **Natural Language Processing**.

- Completely built on the self-attention mechanism
- Does not use sequence any recurrent architecture:
 - More efficient: input sequences can be processed in parallel
 - Has no inherent understanding of sequence order
- Can be applied to various tasks, including in time series machine learning



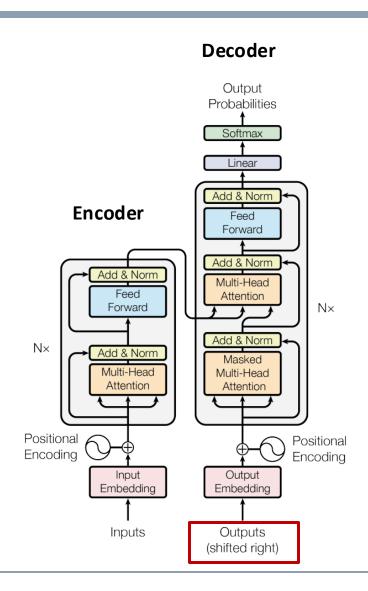


The Transformer Architecture

The fundamentals components of the Transformer architecture are:

- Positional encoding
- Multi-head self attention based scaled dot product attention
- An encoder-decoder architecture

The transformer was first proposed as a **machine translation** model.



[&]quot;Attention is all you need", Vaswani, et al.



Transformer: Positional Encoding

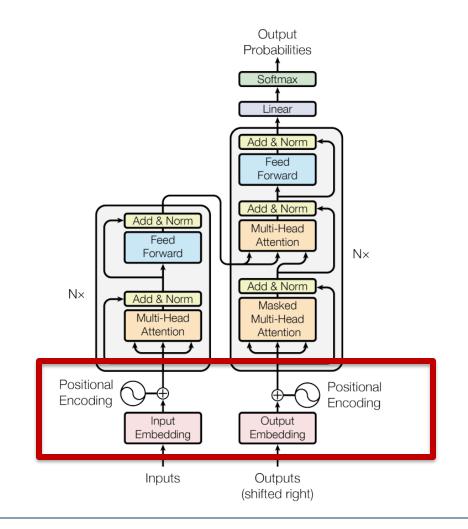
Positional encoding is added to the input and output sequence embeddings in the encoder and decoder. It is necessary to give information on **word order**.

Positional encoding is defined by using sine and cosine functions with different frequencies:

$$PE_{(pos, 2i)} = \sin\left(\frac{pos}{10000 \frac{2i}{d_{model}}}\right)$$

$$PE_{(pos, 2i+1)} = \cos\left(\frac{pos}{10000^{\frac{2i}{d_{model}}}}\right)$$

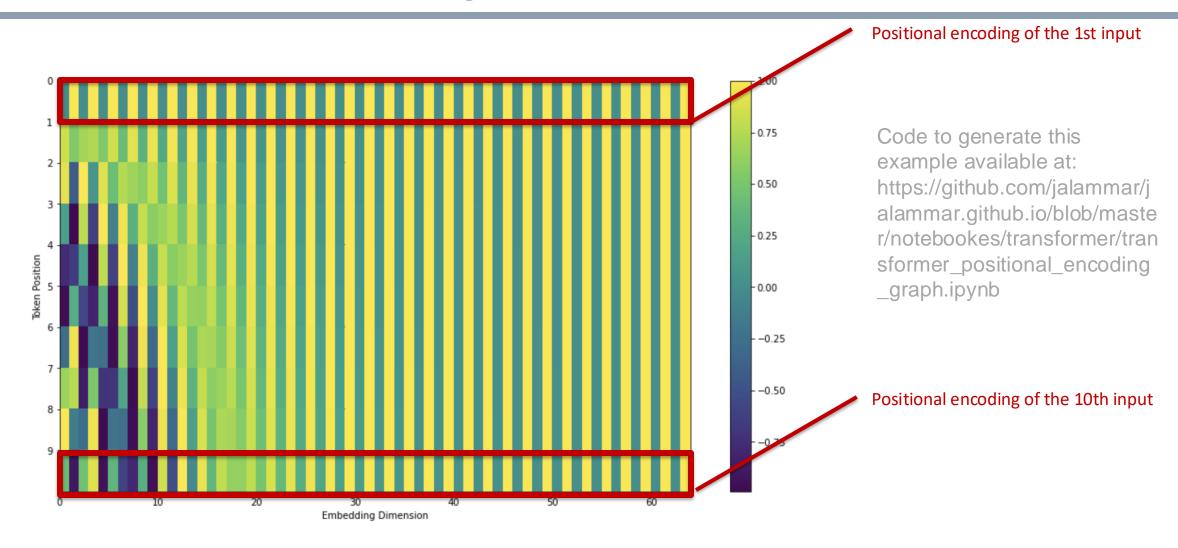
where pos is the position and i is the embedding dimension.



[&]quot;Attention is all you need", Vaswani, et al.

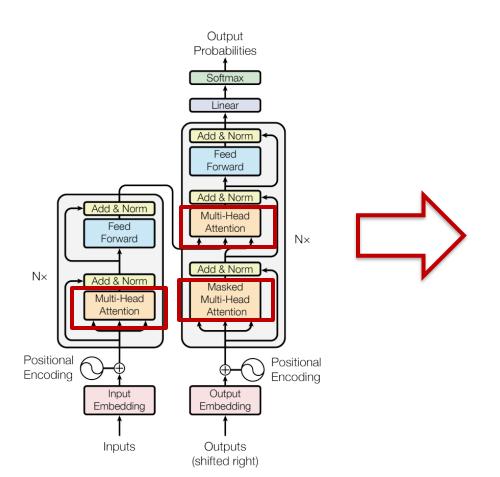


Transformer: Positional Encoding





Transformer: Multi-head Attention Mechanisms



Three multi-head attention mechanisms are used in the Transformer architecture:

- Encoder self-attention: captures relationships between input sequence tokens
- Decoder self-attention: captures relationships between output sequence tokens
- Encoder-decoder cross-attention: captures relationships between input and output sequences

[&]quot;Attention is all you need", Vaswani, et al.

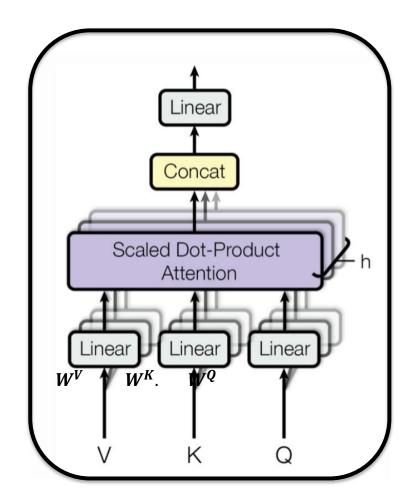


Transformer: Multi-Head Self-Attention

Multi-head attention is used to compute attention several times in **parallel**, using **independent** weight matrices $W^{V,i}$ $W^{K,i}$ and $W^{Q,i}$. These operations are called **attention heads**, and are then **concatenated**.

Basic scaled dot product attention is not sufficient to encode the **complexity** of language, as it might only focus on one aspect of relationships between tokens.

For instance, the multi-head attention mechanism can focus both on **long range and sort range relationships** through different attention heads.



Multi-head attention



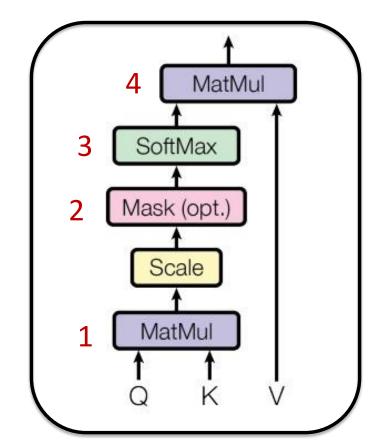
Transformer: Scaled Dot Product Attention

Then, the scaled dot-product attention is computed by:

$$Attention(Q, K, V) = softmax\left(\frac{QK^{T}}{\sqrt{d_{k}}}\right)V$$

where d_k is the dimension of \boldsymbol{Q} and \boldsymbol{K} .

- 1. How each token is related to the query (i.e., similarity score)
- 2. Converts into probabilities (sum to 1)
- 3. Causal masking is used to prevent information leakage
- 4. Weighted sum of values that highlights relevant vectors



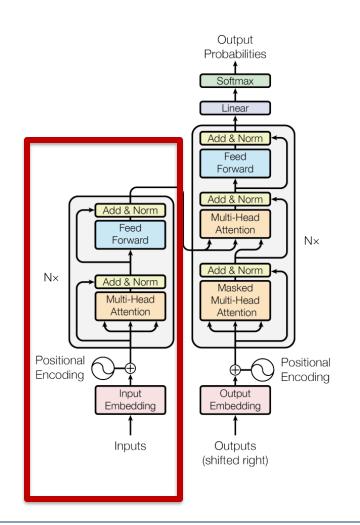
Scaled Dot-Product Attention



Transformer: Encoder

The encoder generates a **contextualized representation** of the **input sequence**. This representation has the same length as the input sequence, and is used to condition the decoder.

- It is made of N_x identical layers, each composed of
 - Multi-head self-attention
 - Feedforward neural network
 - Residual connection
 - Layer normalization



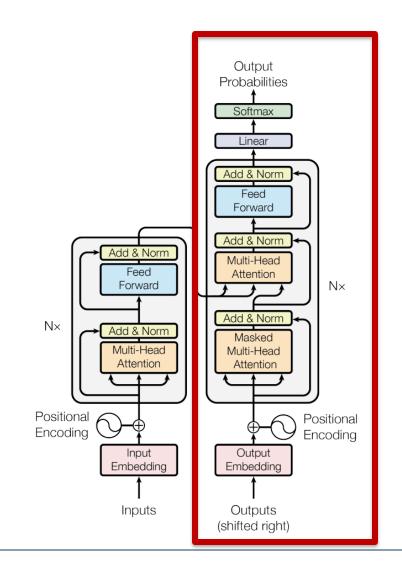
[&]quot;Attention is all you need", Vaswani, et al.



Transformer: Decoder

The **decoder** generates the output of the model (e.g., translated text). It generates tokens one at a time, using network information from the **encoded representation**.

- It is also made of N_x identical layers:
 - Masked multi-head self-attention
 - Multi-head cross-attention
 - Feedforward neural network
 - Residual connection
 - Layer normalization





Transformer Model: Pros and Cons

The main advantages of Transformer models are:

- Parallelization: Transformers can process all tokens in a sequence simultaneously.
- Long-Range Dependencies: Transformers can capturing dependencies between distant tokens, as long as they are all present in the same sequence

However:

Quadratic time complexity of attention mechanism







ADLTS \ DL for TS \ Recap





Lecture outline

- 1. Introduction to Deep Learning
- 2. Convolutional Neural Networks (CNNs)
- 3. Recurrent models (RNNs and LSTMs)
- 4. Transformers





