Attention and transformers

Sequence modelling

- Text-to-text translation
- Speech-to-speech translation
- Code-to-code translation
- Time series

What it is: A set of symbols, i.e. embeddings (of words, sounds etc)

Sequence modelling - historical

- RNN
- LSTM

Problem

- Time series depend on past values OK
- Grammar also depends on future intention
- Example: I saw a/an (?) → man/elephant (aha!)

Historical solution

- 1) Run RNN/LSTM from left to right
- 2) Run RNN/LSTM from right to left
- 3) Provide both outputs as input to next layer

It works!

Attention

- Idea: Focus processing on the important parts no matter where they are.
- LSTM -> 1) Store important part 2) retrieve it where it is important
- Atttention → 1) Directly input parts that are most important for each new symbol

Classic database

- Query: «Star Wars»
- Key: <Title of each movie>
- Value: video.mpg

 1) Match query against each key. 2) Identify the best matching key 3) Return the corresponding video.mpg file/

Attention

Components: Key, query, value

- Query: Which symbol am I looking for?
- Key: What am I looking at in each symbol?
- Score: How to weight the value
- Value: The actual value of the symbol

Attention example: self-attention

Self-: Key, query, value are all based on X

- Query: $Q = W_qX$
- Key: $K = W_k X$
- Score: softmax(QK/n)
- Value: W_VX
- Output: W_VX softmax(QK)

Attention example explained

Components: Key, query, value

- Query: Q = W_qX looks for what is important to a symbol
- Key: K = W_kX looks for which other symbols match he query
- Score:S = softmax(QK/c) Checks which query is matching
- (c is some constant, usually $c = sqrt(d_K)$)
- Value: $V = W_V X$ The value (+ linear transform)
- Output: Y = V S Returns the value corresponding to the best matching key

Self-Attention

Attention (q, k, v) =
$$softmax \left(\frac{qk^T}{\sqrt{d_k}}\right)v$$

from to vector dimensionality of K, V

Each vector receives three representations ("roles")

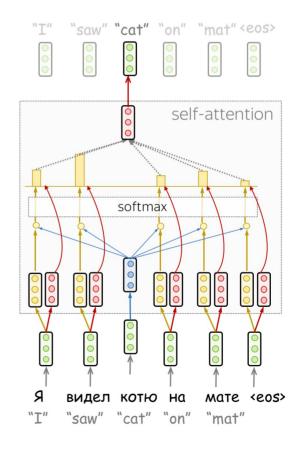
"Hey there, do you have this information?"

$$\begin{bmatrix} W_K \end{bmatrix} \times \begin{bmatrix} \circ \\ \circ \end{bmatrix} = \begin{bmatrix} \circ \\ \circ \\ \circ \end{bmatrix}$$
 Key: vector at which the query looks to compute weights

"Hi, I have this information – give me a large weight!"

$$\begin{bmatrix} W_V \end{bmatrix} \times \begin{bmatrix} \circ & \bullet \\ \circ & \bullet \end{bmatrix} = \begin{bmatrix} \bullet & \text{Value} \\ \bullet & \text{attention output} \end{bmatrix}$$

"Here's the information I have!"



11

Attention example explained

What's missing from Self-Attention?

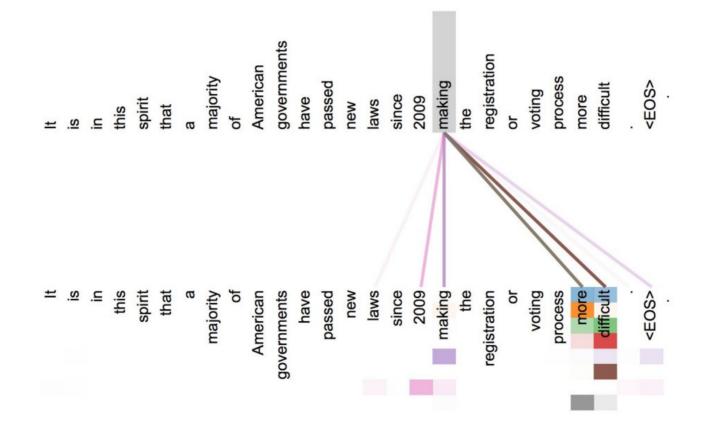
- Convolution: a different linear transformation for each relative position.
 Allows you to distinguish what information came from where.
- Self-Attention: a weighted average :(

Convolution

Self-Attention



The Fix: Multi-Head Attention



Multihead attention

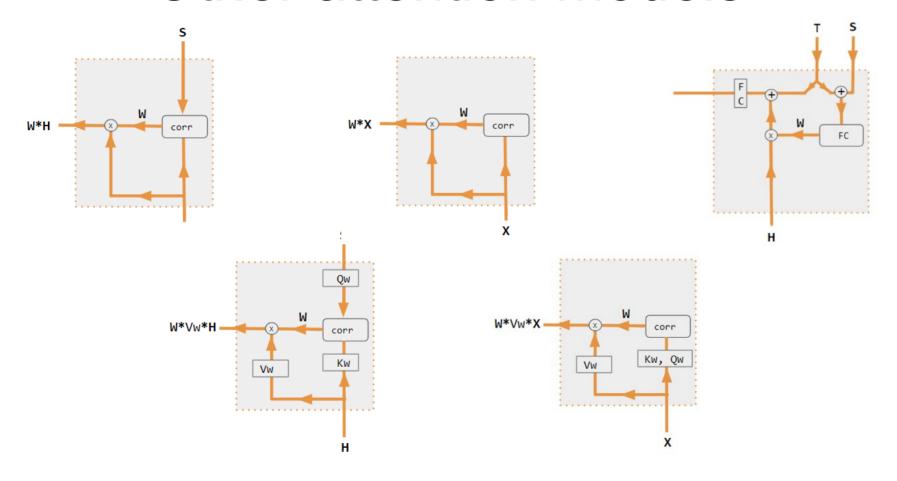
- We look for several values and concatenate
- MH(Q,K,V) = concat(head1(Q,K,V), head2(Q,K,V))
 W_O

- Example:
- MH(X) = concat(head₁(X), head₂(X))

Summary so far

- Y = Attention(K,Q,V) sequence of length L
- Multihead attention = (head1, head2 ...) W
- LxO where O is output dim
- There are different types of attention (inputs, score functions, non-linearities etc)
- Self-attention is K=Q=V (=X)

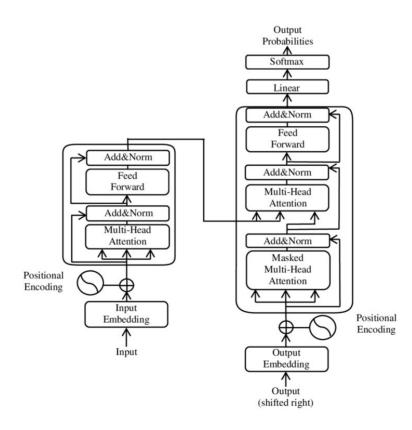
Other attention models



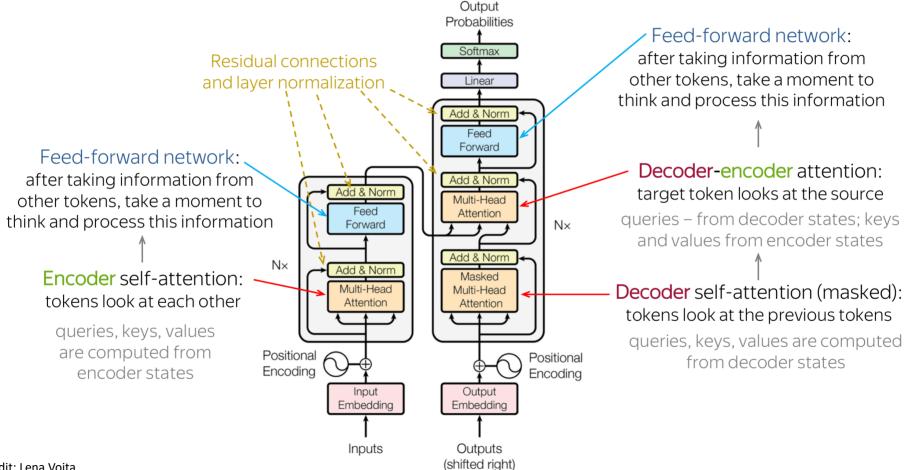
Transformer

- Attention is all you need (almost)
- Transformer: Layered self-attention
- An element in a sequence: a «symbol»
- Each layer:
 - N input symbols → N output symbols
 - Dimensions: $N^*d_x \rightarrow N^*d_y$

Transformer (original)



Transformer (annotated)



Decoder-encoder attention

- Idea: For each symbol in the decoder we perform one «search» for relevant information among the encoder symbols
- Query: Symbols from previous layer of decoder, D
- Key, value: Symbols from encoder, E
- Output: N_D symbols, i.e. same number as previous decoder layer
- Query: Q = W_qD
- Key: K = W_kE
- Score: softmax(QK/c)
- Value: W_VE
- Output: W_VX softmax(QK) (i.e. value*score)

New blocks: Add&Norm

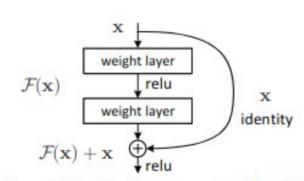
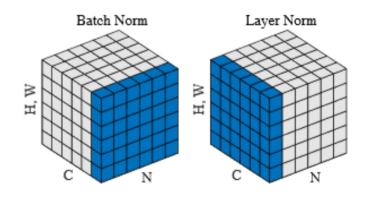


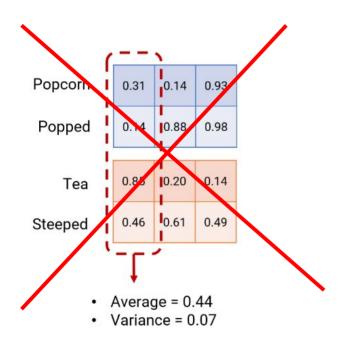
Figure 2. Residual learning: a building block.

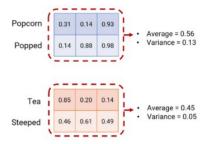


For each output: Sum of outputs for each sample sums up to 1 with unit variance.

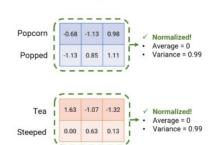
Note: This calculation is applied *once for each symbol*. N symbols create a N x d_o sized output for the layer, where d_o is the output size of the neural network.

Layer normalization





And again, after normalization, we'll have matrices with average of 0 and variance of 1:



New blocks: Embedding

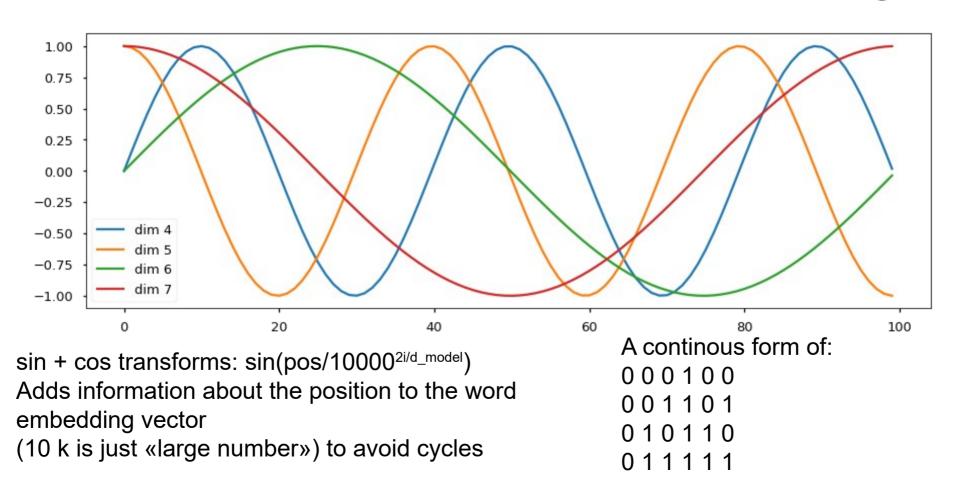
Some pretrained language model that converts each word to a vector of fixed size, e.g. [0.4 0.5 0.7]

New blocks: Masking

Prevent attention to unwanted words. In particular, future words.

$$a^K \quad b^K \quad c^K \quad D^K$$
 softmax(QK $^ op$ + M) $M = egin{array}{c} a^Q & \begin{bmatrix} 0 & -\infty & -\infty & -\infty \ 0 & 0 & -\infty & -\infty \ 0 & 0 & 0 & -\infty \ D^Q & 0 & 0 & 0 & 0 \end{bmatrix}$

New blocks: Positional encoding



New blocks: Positional encoding

$$\overrightarrow{p_t}^{(i)} = f(t)^{(i)} := egin{cases} \sin(\omega_k.\,t), & ext{if } i = 2k \ \cos(\omega_k.\,t), & ext{if } i = 2k+1 \end{cases}$$

Why d-dimensional?

Oddity: Add position encoding to word embedding (instead of concatenating)

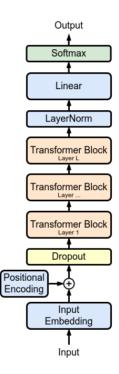
DeBERTa:

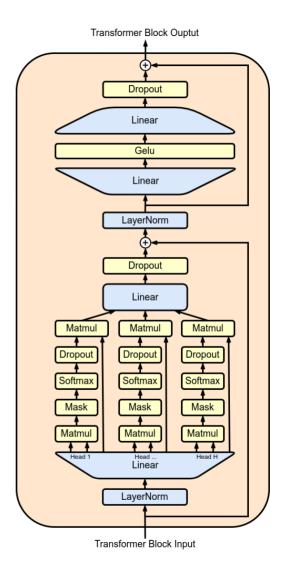
«the disentangled attention mechanism, where each word is represented using two vectors that encode its content and position...performs consistently better»

GPT2

and GPT3...

We use the same model and architecture as GPT-2 [RWC⁺19], including the modified initialization, pre-normalization, and reversible tokenization described therein, with the exception that we use alternating dense and locally banded sparse attention patterns in the layers of the transformer, similar to the Sparse Transformer [CGRS19]. To study the dependence of ML performance on model size, we train 8 different sizes of model, ranging over three orders of magnitude from 125 million parameters to 175 billion parameters, with the last being the model we call GPT-3. Previous work [KMH⁺20] suggests that with enough training data, scaling of validation loss should be approximately a smooth power law as a function of size; training models of many different sizes allows us to test this hypothesis both for validation loss and for downstream language tasks.





New blocks: Embedding

Some pretrained language model that converts each word to a vector of fixed size, e.g. [0.4 0.5 0.7]

RNN vs LSTM vs Self-attention

n: sequence length, d:encoding length

Multi-Head Attention with linear transformations. For each of the h heads, $d_q = d_k = d_v = d/h$	$n^2 \cdot d + n \cdot d^2$
Recurrent	$n \cdot d^2$
Convolutional	$n \cdot d^2$

Simplest: RNN

Most scalable with sequence length: LSTM/RNN Fastest on typical hardware if n<<d: Self-attention

Fastest on sparse bandwidth: Self-attention

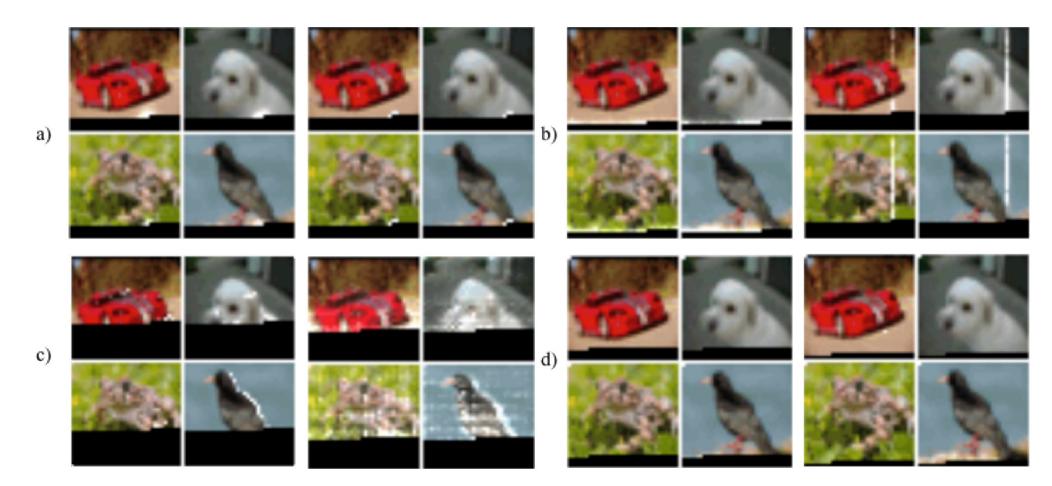
However, for training gradient quality and landscape might be more important

RNN vs LSTM vs Self-attention: Practical results

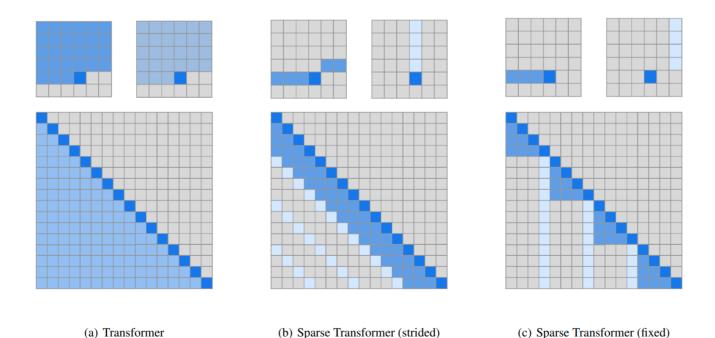
Machine Translation Results: WMT-14

Model	BL	EU	Training Co	Training Cost (FLOPs)		
Wodel	EN-DE	EN-FR	EN-DE	EN-FR		
ByteNet [17]	23.75					
Deep-Att + PosUnk [37]		39.2		$1.0 \cdot 10^{20}$		
GNMT + RL [36]	24.6	39.92	$2.3 \cdot 10^{19}$	$1.4 \cdot 10^{20}$		
ConvS2S [9]	25.16	40.46	$9.6 \cdot 10^{18}$	$1.5 \cdot 10^{20}$		
MoE [31]	26.03	40.56	$2.0\cdot 10^{19}$	$1.2\cdot 10^{20}$		
Deep-Att + PosUnk Ensemble [37]		40.4		$8.0 \cdot 10^{20}$		
GNMT + RL Ensemble [36]	26.30	41.16	$1.8 \cdot 10^{20}$	$1.1 \cdot 10^{21}$		
ConvS2S Ensemble [9]	26.36	41.29	$7.7\cdot 10^{19}$	$1.2\cdot 10^{21}$		
Transformer (base model)	27.3	38.1	3.3 ·	$3.3\cdot 10^{18}$		
Transformer (big)	28.4	41.0	2.3 ·	10^{19}		

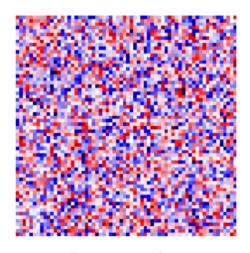
Sparse attention (white)



Sparse transformers



Block-sparse (on GPU)



Dense weights



Block-sparse weights

1 0 0 0 0 0 1 1 1 1 1 1 0 0 1 1 1 0 1 1 1 1 1 1								
1 0 0 0 0 0 1 1 1 1 1 1 0 0 1 1 1 0 1 1 1 1 1 1	0	0	1	1	1	1	1	1
1 1 1 1 0 0 1 1 1 0 1 1 1 1 1 1	1	1	1	0	0	1	1	0
1 0 1 1 1 1 1	1	0	0	0	0	0	1	1
	1	1	1	1	0	0	1	1
	1	0	1	1	1	1	1	1
0 1 0 0 0 0 0 0	0	1	0	0	0	0	0	0
1 1 1 0 1 1 0 0	1	1	1	0	1	1	0	0
1 0 0 0 0 1 1 1	1	0	0	0	0	1	1	1

Corresponding sparsity pattern

GPT2 code

```
def model(hparams, X, past=None, scope='model', reuse=False):
    with tf.variable scope(scope, reuse=reuse):
        results = {}
        batch, sequence = shape list(X)
        wpe = tf.get variable('wpe', [hparams.n ctx, hparams.n embd],
                            initializer=tf.random normal initializer(stddev=0.01))
        wte = tf.get variable('wte', [hparams.n vocab, hparams.n embd],
                             initializer=tf.random normal initializer(stddev=0.02))
        past length = 0 if past is None else tf.shape(past)[-2]
        h = tf.gather(wte, X) + tf.gather(wpe, positions for(X, past length))
        # Transformer
        presents = []
        pasts = tf.unstack(past, axis=1) if past is not None else [None] * hparams.n layer
        assert len(pasts) == hparams.n layer
        for layer, past in enumerate(pasts):
           h, present = block(h, 'h%d' % layer, past=past, hparams=hparams)
            presents.append(present)
        results['present'] = tf.stack(presents, axis=1)
       h = norm(h, 'ln f')
        # Language model loss. Do tokens <n predict token n?
       h_flat = tf.reshape(h, [batch*sequence, hparams.n_embd])
        logits = tf.matmul(h flat, wte, transpose b=True)
        logits = tf.reshape(logits, [batch, sequence, hparams.n vocab])
        results['logits'] = logits
        return results
```