Transformers-Original

November 11, 2024

Before you turn this problem in, make sure everything runs as expected. First, **restart the kernel** (in the menubar, select Kernel \rightarrow Restart) and then **run all cells** (in the menubar, select Cell \rightarrow Run All).

Make sure you fill in any place that says YOUR CODE HERE or "YOUR ANSWER HERE", as well as your name and collaborators below:

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Make sure you fill in any place that says YOUR CODE HERE or "YOUR ANSWER HERE", as well as your name and collaborators below:

```
[3]: NAME = ""
COLLABORATORS = ""
```

```
[4]: # Importing libraries
     # PyTorch
     import torch
     import torch.nn as nn
     from torch.utils.data import Dataset, DataLoader, random_split
     # from torch.utils.tensorboard import SummaryWriter
     # Math
     import math
     # Miscellaneous
     # import pynuml
     import random
     import time
     # HuggingFace libraries
     from datasets import load_dataset
     from tokenizers import Tokenizer
     from tokenizers.models import WordLevel
     from tokenizers.trainers import WordLevelTrainer
```

```
from tokenizers.pre_tokenizers import Whitespace

# Pathlib
from pathlib import Path

# typing
from typing import Any

# Library for progress bars in loops
from tqdm import tqdm

# Importing library of warnings
import warnings

# torch
import torch.nn.functional as F

#dgx package
# from gpu_manager.acquire_release import get_gpu, release_gpu_lock,u
prelease_gpu_memory
```

Before coding, let's take a look at the Transformer architecture.

Source: Attention Is All You Need

The Transformer architecture has two main blocks: the encoder and the decoder. Let's take a look at them further.

Encoder: It has a Multi-Head Attention mechanism and a fully connected Feed-Forward network. There are also residual connections around the two sub-layers, plus layer normalization for the output of each sub-layer. All sub-layers in the model and the embedding layers produce outputs of dimension $d_{model} = 512$.

Decoder: The decoder follows a similar structure, but it inserts a third sub-layer that performs multi-head attention over the output of the encoder block. There is also a modification of the self-attention sub-layer in the decoder block to avoid positions from attending to subsequent positions. This masking ensures that the predictions for position i depend solely on the known outputs at positions less than i.

Both the encoder and decode blocks are repeated N times. In the original paper, they defined N=6, and we will define a similar value in this notebook.

When we observe the Transformer architecture image above, we can see that the Embeddings

represent the first step of both blocks.

The InputEmbedding class below is responsible for converting the input text into numerical vectors of d_model dimensions. To prevent that our input embeddings become extremely small, we normalize them by multiplying them by the $\sqrt{d_{model}}$.

In the image below, we can see how the embeddings are created. First, we have a sentence that gets split into tokens—we will explore what tokens are later on—. Then, the token IDs—identification numbers—are transformed into the embeddings, which are high-dimensional vectors.

Source: vaclavkosar.com

```
[5]: # Creating Input Embeddings
class InputEmbeddings(nn.Module):

    def __init__(self, d_model: int, vocab_size: int):
        super().__init__()
        self.d_model = d_model # Dimension of vectors (512)
        self.vocab_size = vocab_size # Size of the vocabulary
        self.embedding = nn.Embedding(vocab_size, d_model) # PyTorch layer that_u
        -converts integer indices to dense embeddings

    def forward(self, x):
        return self.embedding(x) * math.sqrt(self.d_model) # Normalizing the_u
        -variance of the embeddings
```

In the original paper, the authors add the positional encodings to the input embeddings at the bottom of both the encoder and decoder blocks so the model can have some information about the relative or absolute position of the tokens in the sequence. The positional encodings have the same dimension d_{model} as the embeddings, so that the two vectors can be summed and we can combine the semantic content from the word embeddings and positional information from the positional encodings.

In the PositionalEncoding class below, we will create a matrix of positional encodings pe with dimensions (seq_len, d_model). We will start by filling it with 0s.We will then apply the sine function to even indices of the positional encoding matrix while the cosine function is applied to the odd ones.

Even Indices
$$(2i)$$
: $PE(pos, 2i) = \sin\left(\frac{pos}{10000^{2i/d_{model}}}\right)$ (1)

Odd Indices
$$(2i+1)$$
: PE(pos, $2i+1$) = $\cos\left(\frac{\text{pos}}{10000^{2i/d_{model}}}\right)$ (2)

We apply the sine and cosine functions because it allows the model to determine the position of a word based on the position of other words in the sequence, since for any fixed offset k, PE_{nos+k}

can be represented as a linear function of PE_{pos} . This happens due to the properties of sine and cosine functions, where a shift in the input results in a predictable change in the output.

```
[6]: # Creating the Positional Encoding
     class PositionalEncoding(nn.Module):
         def __init__(self, d_model: int, seq_len: int, dropout: float) -> None:
             super().__init__()
             self.d model = d model # Dimensionality of the model
             self.seq_len = seq_len # Maximum sequence length
             self.dropout = nn.Dropout(dropout) # Dropout layer to prevent
      →overfitting
             # Creating a positional encoding matrix of shape (seq_len, d_model)_{\sqcup}
      ⇔filled with zeros
             pe = torch.zeros(seq_len, d_model)
             # Creating a tensor representing positions (0 to seq_len - 1)
             position = torch.arange(0, seq_len, dtype = torch.float).unsqueeze(1) #_
      →Transforming 'position' into a 2D tensor['seq_len, 1']
             # Creating the division term for the positional encoding formula
             div_term = torch.exp(torch.arange(0, d_model, 2).float() * (-math.
      →log(10000.0) / d_model))
             # Apply sine to even indices in pe
             pe[:, 0::2] = torch.sin(position * div_term)
             # Apply cosine to odd indices in pe
             pe[:, 1::2] = torch.cos(position * div_term)
             # Adding an extra dimension at the beginning of pe matrix for batch
      \hookrightarrow handling
             pe = pe.unsqueeze(0)
             # Registering 'pe' as buffer. Buffer is a tensor not considered as au
      ⇔model parameter
             self.register_buffer('pe', pe)
         def forward(self,x):
             \# Addind positional encoding to the input tensor X
             x = x + (self.pe[:, :x.shape[1], :]).requires_grad_(False)
             return self.dropout(x) # Dropout for regularization
```

When we look at the encoder and decoder blocks, we see several normalization layers called Add

& Norm.

The LayerNormalization class below performs layer normalization on the input data. During its forward pass, we compute the mean and standard deviation of the input data. We then normalize the input data by subtracting the mean and dividing by the standard deviation plus a small number called epsilon to avoid any divisions by zero. This process results in a normalized output with a mean 0 and a standard deviation 1.

We will then scale the normalized output by a learnable parameter alpha and add a learnable parameter called bias. The training process is responsible for adjusting these parameters. The final result is a layer-normalized tensor, which ensures that the scale of the inputs to layers in the network is consistent.

```
[7]: # Creating Layer Normalization
     class LayerNormalization(nn.Module):
         def __init__(self, eps: float = 10**-6) -> None: # We define epsilon as 0.
      →000001 to avoid division by zero
             super().__init__()
             self.eps = eps
             # We define alpha as a trainable parameter and initialize it with ones
             self.alpha = nn.Parameter(torch.ones(1)) # One-dimensional tensor that_{\sqcup}
      ⇔will be used to scale the input data
             # We define bias as a trainable parameter and initialize it with zeros
             self.bias = nn.Parameter(torch.zeros(1)) # One-dimensional tenso that
      will be added to the input data
         def forward(self, x):
             mean = x.mean(dim = -1, keepdim = True) # Computing the mean of the
      →input data. Keeping the number of dimensions unchanged
             std = x.std(dim = -1, keepdim = True) # Computing the standard
      -deviation of the input data. Keeping the number of dimensions unchanged
             # Returning the normalized input
             return self.alpha * (x-mean) / (std + self.eps) + self.bias
```

In the fully connected feed-forward network, we apply two linear transformations with a ReLU activation in between. We can mathematically represent this operation as:

$$FFN(x) = \max(0, xW_1 + b_1)W_2 + b_2 \tag{3}$$

 W_1 and W_2 are the weights, while b_1 and b_2 are the biases of the two linear transformations.

In the FeedForwardBlock below, we will define the two linear transformations—self.linear_1 and self.linear_2—and the inner-layer d_ff. The input data will first pass through the self.linear_1 transformation, which increases its dimensionality from d_model to d_ff. The output of this operation passes through the ReLU activation function, which introduces non-linearity so the network can learn more complex patterns, and the self.dropout layer is applied to mitigate overfitting. The final operation is the self.linear_2 transformation to the dropout-modified tensor, which transforms it back to the original d_model dimension.

```
[8]: # Creating Feed Forward Layers
class FeedForwardBlock(nn.Module):

def __init__(self, d_model: int, d_ff: int, dropout: float) -> None:
    super().__init__()
    # First linear transformation
    self.linear_1 = nn.Linear(d_model, d_ff) # W1 & b1
    self.dropout = nn.Dropout(dropout) # Dropout to prevent overfitting
    # Second linear transformation
    self.linear_2 = nn.Linear(d_ff, d_model) # W2 & b2

def forward(self, x):
    # (Batch, seq_len, d_model) --> (batch, seq_len, d_ff) --> (batch, u)
    seq_len, d_model)
    return self.linear_2(self.dropout(torch.relu(self.linear_1(x))))
```

The Multi-Head Attention is the most crucial component of the Transformer. It is responsible for helping the model to understand complex relationships and patterns in the data.

The image below displays how the Multi-Head Attention works. It doesn't include batch dimension because it only illustrates the process for one single sentence.

Source: YouTube: Coding a Transformer from scratch on PyTorch, with full explanation, training and inference by Umar Jamil.

The Multi-Head Attention block receives the input data split into queries, keys, and values organized into matrices Q, K, and V. Each matrix contains different facets of the input, and they have the same dimensions as the input.

We then linearly transform each matrix by their respective weight matrices W^Q , W^K , and W^V . These transformations will result in new matrices Q', K', and V', which will be split into smaller matrices corresponding to different heads h, allowing the model to attend to information from different representation subspaces in parallel. This split creates multiple sets of queries, keys, and values for each head.

Finally, we concatenate every head into an H matrix, which is then transformed by another weight matrix W^o to produce the multi-head attention output, a matrix MH - A that retains the input dimensionality.

```
[9]: # Creating the Multi-Head Attention block
     class MultiHeadAttentionBlock(nn.Module):
         def __init__(self, d_model: int, h: int, dropout: float) -> None: # h =_
      →number of heads
             super().__init__()
             self.d_model = d_model
             self.h = h
             # We ensure that the dimensions of the model is divisible by the number ...
      ⇔of heads
             assert d_model % h == 0, 'd_model is not divisible by h'
             # d_k is the dimension of each attention head's key, query, and value_
             self.d k = d model // h # d k formula, like in the original "Attention"
      → Is All You Need" paper
             # Defining the weight matrices
             self.w_q = nn.Linear(d_model, d_model) # W_q
             self.w_k = nn.Linear(d_model, d_model) # W_k
             self.w_v = nn.Linear(d_model, d_model) # W_v
             self.w_o = nn.Linear(d_model, d_model) # W_o
             self.dropout = nn.Dropout(dropout) # Dropout layer to avoid overfitting
         Ostaticmethod
         def attention(query, key, value, mask, dropout: nn.Dropout): # mask => When_
      we want certain words to NOT interact with others, we "hide" them
             d_k = query.shape[-1] # The last dimension of query, key, and value
             # We calculate the Attention(Q,K,V) as in the formula in the image
      \rightarrowabove
             attention scores = (query @ key.transpose(-2,-1)) / math.sqrt(d_k) # @_
      →= Matrix multiplication sign in PyTorch
             # Before applying the softmax, we apply the mask to hide some_
      →interactions between words
             if mask is not None: # If a mask IS defined...
                 attention_scores.masked_fill_(mask == 0, -1e9) # Replace each value_
      \hookrightarrowwhere mask is equal to 0 by -1e9
             attention_scores = attention_scores.softmax(dim = -1) # Applying softmax
             if dropout is not None: # If a dropout IS defined...
                 attention_scores = dropout(attention_scores) # We apply dropout to_{\square}
      ⇔prevent overfitting
```

```
return (attention_scores @ value), attention_scores # Multiply the_
→output matrix by the V matrix, as in the formula
  def forward(self, q, k, v, mask):
      query = self.w_q(q) # Q' matrix
      key = self.w_k(k) # K' matrix
      value = self.w_v(v) # V' matrix
       # Splitting results into smaller matrices for the different heads
       # Splitting embeddings (third dimension) into h parts
       query = query.view(query.shape[0], query.shape[1], self.h, self.d_k).
→transpose(1,2) # Transpose => bring the head to the second dimension
      key = key.view(key.shape[0], key.shape[1], self.h, self.d_k).
stranspose(1,2) # Transpose => bring the head to the second dimension
      value = value.view(value.shape[0], value.shape[1], self.h, self.d k).
stranspose(1,2) # Transpose => bring the head to the second dimension
       # Obtaining the output and the attention scores
      x, self.attention scores = MultiHeadAttentionBlock.attention(query,
⇒key, value, mask, self.dropout)
       # Obtaining the H matrix
      x = x.transpose(1, 2).contiguous().view(x.shape[0], -1, self.h * self.
\rightarrowd_k)
      return self.w_o(x) # Multiply the H matrix by the weight matrix W_o,_
⇔resulting in the MH-A matrix
```

When we look at the architecture of the Transformer, we see that each sub-layer, including the self-attention and Feed Forward blocks, adds its output to its input before passing it to the Add & Norm layer. This approach integrates the output with the original input in the Add & Norm layer. This process is known as the skip connection, which allows the Transformer to train deep networks more effectively by providing a shortcut for the gradient to flow through during backpropagation.

The ResidualConnection class below is responsible for this process.

```
[10]: # Building Residual Connection
class ResidualConnection(nn.Module):
    def __init__(self, dropout: float) -> None:
        super().__init__()
```

```
self.dropout = nn.Dropout(dropout) # We use a dropout layer to prevent

overfitting

self.norm = LayerNormalization() # We use a normalization layer

def forward(self, x, sublayer):

# We normalize the input and add it to the original input 'x'. This

ocreates the residual connection process.

return x + self.dropout(sublayer(self.norm(x)))
```

We will now build the encoder. We create the EncoderBlock class, consisting of the Multi-Head Attention and Feed Forward layers, plus the residual connections.

Encoder block. Source: researchgate.net.

In the original paper, the Encoder Block repeats six times. We create the Encoder class as an assembly of multiple EncoderBlocks. We also add layer normalization as a final step after processing the input through all its blocks.

```
[11]: # Building Encoder Block
      class EncoderBlock(nn.Module):
          # This block takes in the MultiHeadAttentionBlock and FeedForwardBlock, as \square
       well as the dropout rate for the residual connections
          def __init__(self, self_attention_block: MultiHeadAttentionBlock,__
       feed_forward_block: FeedForwardBlock, dropout: float) -> None:
              super(). init ()
              # Storing the self-attention block and feed-forward block
              self.self_attention_block = self_attention_block
              self.feed_forward_block = feed_forward_block
              self.residual connections = nn.ModuleList([ResidualConnection(dropout)])
       →for _ in range(2)]) # 2 Residual Connections with dropout
          def forward(self, x, src_mask):
              # Applying the first residual connection with the self-attention block
              x = self.residual_connections[0](x, lambda x: self.
       →self_attention_block(x, x, x, src_mask)) # Three 'x's corresponding to_
       ⇔query, key, and value inputs plus source mask
              # Applying the second residual connection with the feed-forward block
              x = self.residual_connections[1](x, self.feed_forward_block)
              return x # Output tensor after applying self-attention and feed-forward \Box
       → layers with residual connections.
```

Similarly, the Decoder also consists of several DecoderBlocks that repeat six times in the original paper. The main difference is that it has an additional sub-layer that performs multi-head attention with a cross-attention component that uses the output of the Encoder as its keys and values while using the Decoder's input as queries.

Decoder block. Source: edlitera.com.

For the Output Embedding, we can use the same InputEmbeddings class we use for the Encoder. You can also notice that the self-attention sub-layer is masked, which restricts the model from accessing future elements in the sequence.

We will start by building the DecoderBlock class, and then we will build the Decoder class, which will assemble multiple DecoderBlocks.

```
[14]: # Building Decoder
      # A Decoder can have several Decoder Blocks
      class Decoder(nn.Module):
          # The Decoder takes in instances of 'DecoderBlock'
          def __init__(self, layers: nn.ModuleList) -> None:
              super().__init__()
              # Storing the 'DecoderBlock's
              self.layers = layers
              self.norm = LayerNormalization() # Layer to normalize the output
          def forward(self, x, encoder_output, src_mask, tgt_mask):
              # Iterating over each DecoderBlock stored in self.layers
              for layer in self.layers:
                  # Applies each DecoderBlock to the input 'x' plus the encoder_{\sqcup}
       →output and source and target masks
                  x = layer(x, encoder_output, src_mask, tgt_mask)
              return self.norm(x) # Returns normalized output
```

You can see in the Decoder image that after running a stack of DecoderBlocks, we have a Linear Layer and a Softmax function to the output of probabilities. The ProjectionLayer class below is responsible for converting the output of the model into a probability distribution over the vocabulary, where we select each output token from a vocabulary of possible tokens.

```
[15]: # Buiding Linear Layer class ProjectionLayer(nn.Module):
```

We finally have every component of the Transformer architecture ready. We may now construct the Transformer by putting it all together.

In the Transformer class below, we will bring together all the components of the model's architecture.

```
[16]: # Creating the Transformer Architecture
      class Transformer(nn.Module):
          # This takes in the encoder and decoder, as well the embeddings for the
       ⇒source and target language.
          \# It also takes in the Positional Encoding for the source and target \sqcup
       →language, as well as the projection layer
          def __init__(self, encoder: Encoder, decoder: Decoder, src_embed:_
       →InputEmbeddings, tgt_embed: InputEmbeddings, src_pos: PositionalEncoding,
       utgt_pos: PositionalEncoding, projection_layer: ProjectionLayer) -> None:
              super().__init__()
              self.encoder = encoder
              self.decoder = decoder
              self.src_embed = src_embed
              self.tgt_embed = tgt_embed
              self.src_pos = src_pos
              self.tgt pos = tgt pos
              self.projection_layer = projection_layer
          # Encoder
          def encode(self, src, src_mask):
              src = self.src_embed(src) # Applying source embeddings to the input_
       ⇒source language
              src = self.src_pos(src) # Applying source positional encoding to the
       ⇒source embeddings
              return self.encoder(src, src_mask) # Returning the source embeddings_
       ⇒plus a source mask to prevent attention to certain elements
          # Decoder
```

```
def decode(self, encoder_output, src_mask, tgt, tgt_mask):
    tgt = self.tgt_embed(tgt) # Applying target embeddings to the input_
    target language (tgt)
    tgt = self.tgt_pos(tgt) # Applying target positional encoding to the
    target embeddings

# Returning the target embeddings, the output of the encoder, and both_
    source and target masks
    # The target mask ensures that the model won't 'see' future elements of
    the sequence
    return self.decoder(tgt, encoder_output, src_mask, tgt_mask)

# Applying Projection Layer with the Softmax function to the Decoder output def project(self, x):
    return self.projection_layer(x)
```

The architecture is finally ready. We now define a function called build_transformer, in which we define the parameters and everything we need to have a fully operational Transformer model for the task of machine translation.

We will set the same parameters as in the original paper, Attention Is All You Need, where $d_{model} = 512$, N = 6, h = 8, dropout rate $P_{drop} = 0.1$, and $d_{ff} = 2048$.

```
[17]: # Building & Initializing Transformer
      # Definin function and its parameter, including model dimension, number of \Box
       ⇔encoder and decoder stacks, heads, etc.
      def build_transformer(src_vocab_size: int, tgt_vocab_size: int, src_seq_len:u
       →int, tgt_seq_len: int, d_model: int = 512, N: int = 6, h: int = 8, dropout:
       ⇔float = 0.1, d_ff: int = 2048) → Transformer:
          # Creating Embedding layers
          src_embed = InputEmbeddings(d_model, src_vocab_size) # Source language_
       → (Source Vocabulary to 512-dimensional vectors)
          tgt_embed = InputEmbeddings(d_model, tgt_vocab_size) # Target language_u
       → (Target Vocabulary to 512-dimensional vectors)
          # Creating Positional Encoding layers
          src_pos = PositionalEncoding(d_model, src_seq_len, dropout) # Positional_
       →encoding for the source language embeddings
          tgt pos = PositionalEncoding(d model, tgt seq len, dropout) # Positional
       →encoding for the target language embeddings
          # Creating EncoderBlocks
          encoder_blocks = [] # Initial list of empty EncoderBlocks
          for _ in range(N): # Iterating 'N' times to create 'N' EncoderBlocks (N = 6)
```

```
encoder_self_attention_block = MultiHeadAttentionBlock(d_model, h,_
⇔dropout) # Self-Attention
       feed_forward_block = FeedForwardBlock(d_model, d_ff, dropout) #__
\rightarrow FeedForward
       # Combine layers into an EncoderBlock
       encoder_block = EncoderBlock(encoder_self_attention_block,_
→feed_forward_block, dropout)
       encoder_blocks.append(encoder_block) # Appending EncoderBlock to the
⇔list of EncoderBlocks
  # Creating DecoderBlocks
  decoder_blocks = [] # Initial list of empty DecoderBlocks
  for _ in range(N): # Iterating 'N' times to create 'N' DecoderBlocks (N = 6)
       decoder_self_attention_block = MultiHeadAttentionBlock(d_model, h, u
⇔dropout) # Self-Attention
       decoder_cross_attention_block = MultiHeadAttentionBlock(d_model, h,_
⇔dropout) # Cross-Attention
       feed_forward_block = FeedForwardBlock(d_model, d_ff, dropout) #__
\hookrightarrow FeedForward
       # Combining layers into a DecoderBlock
       decoder block = DecoderBlock(decoder self attention block,
decoder_cross_attention_block, feed_forward_block, dropout)
       decoder_blocks.append(decoder_block) # Appending DecoderBlock to the
⇔list of DecoderBlocks
   # Creating the Encoder and Decoder by using the EncoderBlocks and
→DecoderBlocks lists
  encoder = Encoder(nn.ModuleList(encoder_blocks))
  decoder = Decoder(nn.ModuleList(decoder_blocks))
  # Creating projection layer
  projection_layer = ProjectionLayer(d_model, tgt_vocab_size) # Map the_
→output of Decoder to the Target Vocabulary Space
  # Creating the transformer by combining everything above
  transformer = Transformer(encoder, decoder, src_embed, tgt_embed, src_pos, __
→tgt_pos, projection_layer)
   # Initialize the parameters
  for p in transformer.parameters():
      if p.dim() > 1:
           nn.init.xavier_uniform_(p)
```

```
return transformer # Assembled and initialized Transformer. Ready to be _{\!\!\!\perp} _{\!\!\!\!\perp} trained and validated!
```

The model is now ready to be trained!

Tokenization is a crucial preprocessing step for our Transformer model. In this step, we convert raw text into a number format that the model can process.

There are several Tokenization strategies. We will use the word-level tokenization to transform each word in a sentence into a token.

Different tokenization strategies. Source: shaankhosla.substack.com.

After tokenizing a sentence, we map each token to an unique integer ID based on the created vocabulary present in the training corpus during the training of the tokenizer. Each integer number represents a specific word in the vocabulary.

Besides the words in the training corpus, Transformers use special tokens for specific purposes. These are some that we will define right away:

- [UNK]: This token is used to identify an unknown word in the sequence.
- [PAD]: Padding token to ensure that all sequences in a batch have the same length, so we pad shorter sentences with this token. We use attention masks to "tell" the model to ignore the padded tokens during training since they don't have any real meaning to the task.
- [SOS]: This is a token used to signal the Start of Sentence.
- [EOS]: This is a token used to signal the End of Sentence.

In the build_tokenizer function below, we ensure a tokenizer is ready to train the model. It checks if there is an existing tokenizer, and if that is not the case, it trains a new tokenizer.

For this task, we will use the OpusBooks dataset, available on Hugging Face. This dataset consists of two features, id and translation. The translation feature contains pairs of sentences in different languages, such as Spanish and Portuguese, English and French, and so forth.

I first tried translating sentences from English to Portuguese—my native tongue — but there are only 1.4k examples for this pair, so the results were not satisfying in the current configurations for this model. I then tried to use the English-French pair due to its higher number of examples—127k—but it would take too long to train with the current configurations. I then opted to train the model on the English-Italian pair, the same one used in the Coding a Transformer from scratch on PyTorch, with full explanation, training and inference video, as that was a good balance between performance and time of training.

We start by defining the get_all_sentences function to iterate over the dataset and extract the sentences according to the language pair defined—we will do that later.

```
[19]: # Iterating through dataset to extract the original sentence and its

stranslation

def get_all_sentences(ds, lang):
   for pair in ds:
        yield pair['translation'][lang]
```

The get_ds function is defined to load and prepare the dataset for training and validation. In this function, we build or load the tokenizer, split the dataset, and create DataLoaders, so the model can successfully iterate over the dataset in batches. The result of these functions is tokenizers for the source and target languages plus the DataLoader objects.

```
[20]: def get_ds(config):

# Loading the train portion of the OpusBooks dataset.
```

```
# The Language pairs will be defined in the 'config' dictionary we will
⇒build later
  ds_raw = load_dataset('opus_books',__

-f'{config["lang_src"]}-{config["lang_tgt"]}', split = 'train[:1%]')

   # Building or loading tokenizer for both the source and target languages
  tokenizer_src = build_tokenizer(config, ds_raw, config['lang_src'])
  tokenizer_tgt = build_tokenizer(config, ds_raw, config['lang_tgt'])
  # Splitting the dataset for training and validation
  train_ds_size = int(0.9 * len(ds_raw)) # 90% for training
  val_ds_size = len(ds_raw) - train_ds_size # 10% for validation
  train_ds_raw, val_ds_raw = random_split(ds_raw, [train_ds_size,_
→val_ds_size]) # Randomly splitting the dataset
   # Processing data with the Bilingual Dataset class, which we will define
⇒below
  train_ds = BilingualDataset(train_ds_raw, tokenizer_src, tokenizer_tgt,_

config['lang_src'], config['lang_tgt'], config['seq_len'])

  val_ds = BilingualDataset(val_ds_raw, tokenizer_src, tokenizer_tgt,_

→config['lang_src'], config['lang_tgt'], config['seq_len'])
  # Iterating over the entire dataset and printing the maximum length found
→in the sentences of both the source and target languages
  max_len_src = 0
  max_len_tgt = 0
  for pair in ds_raw:
      src_ids = tokenizer_src.encode(pair['translation'][config['lang_src']]).
⊶ids
      tgt_ids = tokenizer_src.encode(pair['translation'][config['lang_tgt']]).
⊶ids
      max_len_src = max(max_len_src, len(src_ids))
      max_len_tgt = max(max_len_tgt, len(tgt_ids))
  print(f'Max length of source sentence: {max_len_src}')
  print(f'Max length of target sentence: {max_len_tgt}')
  # Creating dataloaders for the training and validadion sets
   # Dataloaders are used to iterate over the dataset in batches during_
⇔training and validation
  train_dataloader = DataLoader(train_ds, batch_size = config['batch_size'],
shuffle = True) # Batch size will be defined in the confiq dictionary
  val_dataloader = DataLoader(val_ds, batch_size = 1, shuffle = True)
  return train_dataloader, val_dataloader, tokenizer_src, tokenizer_tgt #_
→Returning the DataLoader objects and tokenizers
```

We define the casual_mask function to create a mask for the attention mechanism of the decoder. This mask prevents the model from having information about future elements in the sequence.

We start by making a square grid filled with ones. We determine the grid size with the size parameter. Then, we change all the numbers above the main diagonal line to zeros. Every number on one side becomes a zero, while the rest remain ones. The function then flips all these values, turning ones into zeros and zeros into ones. This process is crucial for models that predict future tokens in a sequence.

```
[21]: def casual_mask(size):
    # Creating a square matrix of dimensions 'size x size' filled with ones
    mask = torch.triu(torch.ones(1, size, size), diagonal = 1).type(torch.
    int)
    return mask == 0
```

The BilingualDataset class processes the texts of the target and source languages in the dataset by tokenizing them and adding all the necessary special tokens. This class also certifies that the sentences are within a maximum sequence length for both languages and pads all necessary sentences.

```
[22]: class BilingualDataset(Dataset):
          # This takes in the dataset containing sentence pairs, the tokenizers for
       starget and source languages, and the strings of source and target languages
          # 'seg len' defines the sequence length for both languages
          def __init__(self, ds, tokenizer_src, tokenizer_tgt, src_lang, tgt_lang,__
       ⇒seq_len) -> None:
              super().__init__()
              self.seq_len = seq_len
              self.ds = ds
              self.tokenizer_src = tokenizer_src
              self.tokenizer_tgt = tokenizer_tgt
              self.src_lang = src_lang
              self.tgt_lang = tgt_lang
              # Defining special tokens by using the target language tokenizer
              self.sos_token = torch.tensor([tokenizer_tgt.token_to_id("[SOS]")],_
       →dtype=torch.int64)
              self.eos_token = torch.tensor([tokenizer_tgt.token_to_id("[EOS]")],_
       ⇒dtype=torch.int64)
              self.pad_token = torch.tensor([tokenizer_tgt.token_to_id("[PAD]")],__
       ⇒dtype=torch.int64)
          # Total number of instances in the dataset (some pairs are larger than \sqcup
       ⇔others)
          def __len__(self):
```

```
return len(self.ds)
   # Using the index to retrive source and target texts
  def __getitem__(self, index: Any) -> Any:
      src_target_pair = self.ds[index]
      src_text = src_target_pair['translation'][self.src_lang]
       tgt_text = src_target_pair['translation'][self.tgt_lang]
       # Tokenizing source and target texts
       enc_input_tokens = self.tokenizer_src.encode(src_text).ids
       dec input tokens = self.tokenizer tgt.encode(tgt text).ids
       # Computing how many padding tokens need to be added to the tokenized
\rightarrow texts
       # Source tokens
       enc_num_padding_tokens = self.seq_len - len(enc_input_tokens) - 2 # u
→Subtracting the two '[EOS]' and '[SOS]' special tokens
       # Target tokens
       dec_num_padding_tokens = self.seq_len - len(dec_input_tokens) - 1 # # dec_num_padding_tokens
→Subtracting the '[SOS]' special token
       # If the texts exceed the 'seq_len' allowed, it will raise an error.
This means that one of the sentences in the pair is too long to be processed
       # given the current sequence length limit (this will be defined in the
⇔config dictionary below)
       if enc_num_padding_tokens < 0 or dec_num_padding_tokens < 0:</pre>
           raise ValueError('Sentence is too long')
       # Building the encoder input tensor by combining several elements
       encoder_input = torch.cat(
           self.sos_token, # inserting the '[SOS]' token
           torch.tensor(enc_input_tokens, dtype = torch.int64), # Inserting_
⇔the tokenized source text
           self.eos_token, # Inserting the '[EOS]' token
           torch.tensor([self.pad_token] * enc_num_padding_tokens, dtype =__
→torch.int64) # Addind padding tokens
          ]
       # Building the decoder input tensor by combining several elements
       decoder_input = torch.cat(
           Γ
               self.sos_token, # inserting the '[SOS]' token
               torch.tensor(dec_input_tokens, dtype = torch.int64), #__
→ Inserting the tokenized target text
```

```
torch.tensor([self.pad_token] * dec_num_padding_tokens, dtype =__
→torch.int64) # Addind padding tokens
      )
      # Creating a label tensor, the expected output for training the model
      label = torch.cat(
           Γ
              torch.tensor(dec_input_tokens, dtype = torch.int64), #__
→ Inserting the tokenized target text
              self.eos token, # Inserting the '[EOS]' token
              torch.tensor([self.pad_token] * dec_num_padding_tokens, dtype =_
→torch.int64) # Adding padding tokens
      )
      # Ensuring that the length of each tensor above is equal to the defined

  'seq_len'

      assert encoder_input.size(0) == self.seq_len
      assert decoder input.size(0) == self.seg len
      assert label.size(0) == self.seg len
      return {
           'encoder_input': encoder_input,
           'decoder_input': decoder_input,
           'encoder_mask': (encoder_input != self.pad_token).unsqueeze(0).

unsqueeze(0).int(),
           'decoder_mask': (decoder_input != self.pad_token).unsqueeze(0).
unsqueeze(0).int() & casual_mask(decoder_input.size(0)),
           'label': label,
           'src_text': src_text,
           'tgt_text': tgt_text
      }
```

We will now create two functions for the validation loop. The validation loop is crucial to evaluate model performance in translating sentences from data it has not seen during training.

We will define two functions. The first function, greedy_decode, gives us the model's output by obtaining the most probable next token. The second function, run_validation, is responsible for running the validation process in which we decode the model's output and compare it with the reference text for the target sentence.

```
[23]: # Define function to obtain the most probable next token
      def greedy_decode(model, source, source_mask, tokenizer_src, tokenizer_tgt,__

→max_len, device):
          # Retrieving the indices from the start and end of sequences of the target,
       →tokens
          sos_idx = tokenizer_tgt.token_to_id('[SOS]')
          eos_idx = tokenizer_tgt.token_to_id('[EOS]')
          # Computing the output of the encoder for the source sequence
          encoder_output = model.encode(source, source_mask)
          # Initializing the decoder input with the Start of Sentence token
          decoder_input = torch.empty(1,1).fill_(sos_idx).type_as(source).to(device)
          # Looping until the 'max_len', maximum length, is reached
              if decoder_input.size(1) == max_len:
                  break
              # Building a mask for the decoder input
              decoder_mask = casual_mask(decoder_input.size(1)).type as(source_mask).
       →to(device)
              # Calculating the output of the decoder
              out = model.decode(encoder_output, source_mask, decoder_input,_
       →decoder mask)
              # Applying the projection layer to get the probabilities for the next_{\sqcup}
       \rightarrow token
              prob = model.project(out[:, -1])
              # Selecting token with the highest probability
              _, next_word = torch.max(prob, dim=1)
              decoder_input = torch.cat([decoder_input, torch.empty(1,1).__
       stype_as(source).fill_(next_word.item()).to(device)], dim=1)
              # If the next token is an End of Sentence token, we finish the loop
              if next_word == eos_idx:
                  break
          return decoder_input.squeeze(0) # Sequence of tokens generated by the
       \rightarrow decoder
```

```
[24]: # Defining function to evaluate the model on the validation dataset
# num_examples = 2, two examples per run
def run_validation(model, validation_ds, tokenizer_src, tokenizer_tgt, max_len, u
→device, print_msg, global_state, writer, num_examples=2):
```

```
model.eval() # Setting model to evaluation mode
  count = 0 # Initializing counter to keep track of how many examples have
⇒been processed
  console_width = 80 # Fixed witdh for printed messages
  # Creating evaluation loop
  with torch.no_grad(): # Ensuring that no gradients are computed during this_
⇔process
      for batch in validation_ds:
          count += 1
           encoder input = batch['encoder input'].to(device)
          encoder_mask = batch['encoder_mask'].to(device)
           # Ensuring that the batch_size of the validation set is 1
          assert encoder_input.size(0) == 1, 'Batch size must be 1 for_
⇔validation.'
           # Applying the 'greedy_decode' function to get the model's output_{\square}
⇔for the source text of the input batch
          model_out = greedy_decode(model, encoder_input, encoder_mask,__
⇒tokenizer src, tokenizer tgt, max len, device)
           # Retrieving source and target texts from the batch
           source_text = batch['src_text'][0]
          target_text = batch['tgt_text'][0] # True translation
          model_out_text = tokenizer_tgt.decode(model_out.detach().cpu().
onumpy()) # Decoded, human-readable model output
           # Printing results
          print_msg('-'*console_width)
          print_msg(f'SOURCE: {source_text}')
          print_msg(f'TARGET: {target_text}')
          print_msg(f'PREDICTED: {model_out_text}')
           # After two examples, we break the loop
           if count == num_examples:
              break
```

We are ready to train our Transformer model on the OpusBook dataset for the English to Italian translation task.

We first start by defining the get_model function to load the model by calling the build_transformer function we have previously defined. This function uses the config dictionary to set a few parameters.

I have mentioned the config dictionary several times throughout this notebook. Now, it is time to create it.

In the following cell, we will define two functions to configure our model and the training process.

In the get_config function, we define crucial parameters for the training process. batch_size for the number of training examples used in one iteration, num_epochs as the number of times the entire dataset is passed forward and backward through the Transformer, lr as the learning rate for the optimizer, etc. We will also finally define the pairs from the OpusBook dataset, 'lang_src': 'en' for selecting English as the source language and 'lang_tgt': 'it' for selecting Italian as the target language.

The get_weights_file_path function constructs the file path for saving or loading model weights for any specific epoch.

```
[26]: # Define settings for building and training the transformer model
      def get_config():
          return{
              'batch_size': 2,
              'num_epochs': 1,
              'lr': 10**-4,
              'seq_len': 350,
              'd_model': 512, # Dimensions of the embeddings in the Transformer. 512
       →like in the "Attention Is All You Need" paper.
              'lang_src': 'en',
              'lang_tgt': 'it',
              'model_folder': 'weights',
              'model_basename': 'tmodel_',
              'preload': None,
              'tokenizer_file': 'tokenizer_{0}.json',
              'experiment_name': 'runs/tmodel'
          }
      # Function to construct the path for saving and retrieving model weights
      def get weights file path(config, epoch: str):
```

```
model_folder = config['model_folder'] # Extracting model folder from the config

model_basename = config['model_basename'] # Extracting the base name for model files

model_files

model_filename = f"{model_basename}{epoch}.pt" # Building filename

return str(Path('.')/ model_folder/ model_filename) # Combining current currectory, the model folder, and the model filename
```

We finally define our last function, train_model, which takes the config arguments as input.

In this function, we will set everything up for the training. We will load the model and its necessary components onto the GPU for faster training, set the Adam optimizer, and configure the CrossEntropyLoss function to compute the differences between the translations output by the model and the reference translations from the dataset.

Every loop necessary for iterating over the training batches, performing backpropagation, and computing the gradients is in this function. We will also use it to run the validation function and save the current state of the model.

```
[27]: def train model(config, device):
         print(f"Using device {device} in training")
         # Creating model directory to store weights
         Path(config['model_folder']).mkdir(parents=True, exist_ok=True)
         \# Retrieving dataloaders and tokenizers for source and target languages \sqcup
       ⇔using the 'get_ds' function
         train_dataloader, val_dataloader, tokenizer_src, tokenizer_tgt = __

get_ds(config)
         # Initializing model on the GPU using the 'get_model' function
         model = get model(config,tokenizer_src.get_vocab_size(), tokenizer_tgt.
       →get_vocab_size()).to(device)
         # Tensorboard
         #writer = SummaryWriter(config['experiment name'])
         # Setting up the Adam optimizer with the specified learning rate from the '
         # config' dictionary plus an epsilon value
         optimizer = torch.optim.Adam(model.parameters(), lr=config['lr'], eps =__
       →1e-9)
         # Initializing epoch and global step variables
         initial_epoch = 0
         global_step = 0
```

```
# Checking if there is a pre-trained model to load
  # If true, loads it
  if config['preload']:
      model_filename = get_weights_file_path(config, config['preload'])
      print(f'Preloading model {model_filename}')
      state = torch.load(model_filename) # Loading model
      # Sets epoch to the saved in the state plus one, to resume from where u
⇔it stopped
      initial_epoch = state['epoch'] + 1
      # Loading the optimizer state from the saved model
      optimizer.load_state_dict(state['optimizer_state_dict'])
      # Loading the global step state from the saved model
      global_step = state['global_step']
  # Initializing CrossEntropyLoss function for training
  # We ignore padding tokens when computing loss, as they are not relevant \sqcup
→ for the learning process
  # We also apply label_smoothing to prevent overfitting
  loss_fn = nn.CrossEntropyLoss(ignore_index = tokenizer_src.
# Initializing training loop
  # Iterating over each epoch from the 'initial epoch' variable up to
  # the number of epochs informed in the config
  for epoch in range(initial_epoch, config['num_epochs']):
      # Initializing an iterator over the training dataloader
      # We also use tqdm to display a progress bar
      batch_iterator = tqdm(train_dataloader, desc = f'Processing epoch_
\hookrightarrow {epoch: 02d}')
      # For each batch...
      for batch in batch iterator:
          model.train() # Train the model
          # Loading input data and masks onto the GPU
          encoder_input = batch['encoder_input'].to(device)
          decoder_input = batch['decoder_input'].to(device)
          encoder_mask = batch['encoder_mask'].to(device)
          decoder_mask = batch['decoder_mask'].to(device)
          # Running tensors through the Transformer
          encoder_output = model.encode(encoder_input, encoder_mask)
          decoder_output = model.decode(encoder_output, encoder_mask,__

→decoder_input, decoder_mask)
```

```
proj_output = model.project(decoder_output)
           # Loading the target labels onto the GPU
           label = batch['label'].to(device)
           # Computing loss between model's output and true labels
           loss = loss_fn(proj_output.view(-1, tokenizer_tgt.

    get_vocab_size()), label.view(-1))
           # Updating progress bar
           batch_iterator.set_postfix({f"loss": f"{loss.item():6.3f}"})
           #writer.add_scalar('train loss', loss.item(), global_step)
           #writer.flush()
           # Performing backpropagation
           loss.backward()
           # Updating parameters based on the gradients
           optimizer.step()
           # Clearing the gradients to prepare for the next batch
           optimizer.zero_grad()
           global_step += 1 # Updating global step count
       # We run the 'run_validation' function at the end of each epoch
       # to evaluate model performance
      run_validation(model, val_dataloader, tokenizer_src, tokenizer_tgt,_
→config['seq_len'], device, lambda msg: batch_iterator.write(msg),
⇒global_step, None)
       # Saving model
      model filename = get weights file path(config, f'{epoch:02d}')
       # Writting current model state to the 'model filename'
```

We can now train the model!

```
selected_qpu, utilized_memory, acquire_qpu_index =qet_qpu()
    torch.cuda.set_device(selected_gpu)
  device = torch.device('cuda' if torch.cuda.is_available() else 'cpu')
  #device = torch.cuda.current_device()
  try:
      train model(config,device)
     CheckExecution = True
  except Exception as e:
     CheckExecution=False
      # Handle out of memory error here
      # release_gpu_lock(selected_gpu, acquire_gpu_index)
     print("The GPU memory is currently unavailable, please run after all
→minute.")
      # release_qpu_memory()
  # release_gpu_lock(selected_gpu, acquire_gpu_index)
  print("Completed!")
  # release_gpu_memory()# Training model with the config arguments
```

Using device cuda in training

[NVSHARE] [INFO]: Successfully initialized nvshare GPU

[NVSHARE] [INFO]: Client ID = ea1091ab3cd40160

Else block: Tokenizer json files are used Else block: Tokenizer json files are used

Max length of source sentence: 190
Max length of target sentence: 159

Processing epoch 00: 100% | 145/145 [00:04<00:00, 34.29it/s,

loss=5.2061

SOURCE: At intervals, while turning over the leaves of my book, I studied the aspect of that winter afternoon. Afar, it offered a pale blank of mist and cloud; near a scene of wet lawn and storm-beat shrub, with ceaseless rain sweeping away wildly before a long and lamentable blast.

TARGET: Di tanto in tanto, sfogliando il libro, gettavo un'occhiata al difuori e studiavo l'aspetto di quella serata d'inverno; in lontananza si scorgeva una pallida striscia di nebbia con nuvole, più vicino alberi bagnati, piante sradicate dal temporale e, infine, una pioggia incessante, che lunghe e lamentevoli ventate respingevano sibilando.

SOURCE: No; moonlight was still, and this stirred; while I gazed, it glided up to the ceiling and quivered over my head.

TARGET: No, la luna è immobile e quella luce vacillava, e mentre io la fissava scorse sul soffitto e si fermò sulla mia testa.

PREDICTED: e e e e e e .

GPU GI CI PID

Completed!

As you can see below, we trained for 20 epochs, and the model has been slowly improving. The last epoch had the best performance, at 2.094. Training for more epochs, as well as fine-tuning some parameters, could lead to more promising results.

```
[29]: assert CheckExecution==True
[30]: !nvidia-smi
   Mon Nov 11 15:48:08 2024
   +-----
   ----+
   | NVIDIA-SMI 535.183.01
                        Driver Version: 535.183.01 CUDA Version:
   |-----
   | GPU Name
                    Persistence-M | Bus-Id
                                      Disp.A | Volatile
   Uncorr. ECC |
   | Fan Temp Perf
                   Pwr:Usage/Cap | Memory-Usage | GPU-Util
   Compute M. |
   MIG M. |
   O NVIDIA GeForce RTX 4090 Off | 00000000:01:00.0 Off |
   Off |
                    150W / 450W |
   I 0% 44C
                               2017MiB / 24564MiB | 17%
   Default |
                             N/A |
   | Processes:
```

Type Process name

GPU

[31]: exit(0)