```
In [ ]: # Initialize Otter
import otter
grader = otter.Notebook()
```

CS187

Lab 2-6 - Language models with attention

In previous labs, you have explored two different language model architectures: the *n*-gram model and Recurrent Neural Network (RNN) model. Both of these models utilize context to predict the most likely next word in a sequence, but they do so in distinct ways.

The n-gram model bases its predictions solely on the previous n words. This method, though simple, is limited by its fixed context window. On the other hand, the RNN language model uses a hidden state that carries representations from previous time steps, theoretically enabling the model to capture long-range dependencies in text. However, in practice, RNNs often struggle with very long sequences due to issues like vanishing gradients, making them less effective for modeling extended texts.

In the upcoming lab, you will implement and explore the *attention mechanism*, a powerful technique for encoding and utilizing context from previous time steps. This will be your first step toward understanding the Transformer model, one of the most important language model innovations in recent years.

New bits of PyTorch used in the *solution set* for this lab, and which you may therefore find useful:

- torch.bmm
- torch.cumsum
- torch.masked fill
- torch.repeat
- torch.softmax

- torch.transpose
- torch.triu

Preparation – Loading packages and data

```
In [ ]: import json
        import math
        import os
        import random
        import wget
        import matplotlib.colors as mcolors
        import matplotlib.pyplot as plt
        import torch
        import torch.nn as nn
        from math import inf
        from tokenizers import Tokenizer
        from transformers import PreTrainedTokenizerFast
        from tqdm.notebook import tqdm
In [ ]: # Set random seeds for reproducibility
        SEED = 1234
        def reseed(seed=SEED):
            torch.manual seed(seed)
            random.seed(seed)
        reseed()
In [ ]: # GPU check, sets runtime type to "GPU" where available
        device = torch.device("cuda" if torch.cuda.is available() else "cpu")
        print(device)
```

We use the same data as in lab 2-3 - the Federalist papers.

```
In []: # Prepare to download needed data
def download_if_needed(source, dest, filename):
    if os.path.exists(f"./{dest}{filename}"):
        print (f"Skipping {filename}")
    else:
        print (f"Downloading {filename} from {source}")
        wget.download(source + filename, out=dest)

source_path = "https://raw.githubusercontent.com/nlp-course/data/refs/heads/dest_path = "./"
subdirectory_text = "Text/"
subdirectory_models = "Models/"
```

```
In [ ]: # Read in the raw data
dataset = json.load(open(dest_path + subdirectory_text + "federalist_data_pr

# Read in the pretrained tokenizer
hf_tokenizer = torch.load(dest_path + subdirectory_text + "tokenizer.pt", we
hf_tokenizer.split_special_tokens = False
```

Once again we use the dataset split into training, validation, and test sets. Since we have provided pretrained models, you won't be using the training set in this lab. For the validation set, we have separate ones for papers authored by Hamilton (validation_hamilton) and papers authored by Madison (validation_madison).

```
In []: ## Extract the test set
testing = dataset['test']

# Update the test set gold labels in-place (see lab 1-2 for the import of go
for example in testing:
    example['authors'] = 'Madison'
```

```
In []: ## Extract the validation sets
  validation = dataset['validation']

# one validation set for Madison...
  validation_madison = list(filter(lambda ex: ex['authors']=='Madison', valida

# ...and one for Hamilton
  validation_hamilton = list(filter(lambda ex: ex['authors']=='Hamilton', valida
```

```
In []: # We only consider the first 200 tokens of each document for speed
def truncate(document_set, k=200):
    for document in document_set:
        document['tokens'] = document['tokens'][:k]
```

Before diving into the *attention mechanism*, we'd like to offer some intuitions about its underlying idea. You'll begin by exploring simple models that will help you understand the basic concepts behind it.

A uniform attention model

Recall the forward step in RNN models, starting with converting words to an embedding:

$$\mathbf{x}_t = \mathbf{U}\mathbf{w}_t$$

followed by the RNN layer proper:

$$\mathbf{h}_t = \mathbf{W}\mathbf{h}_{t-1} + \mathbf{V}\mathbf{x}_i$$

and finally, a linear layer and softmax to convert to a distribution over the vocabulary.

$$\mathbf{y}_t = \operatorname{softmax}(\mathbf{X}\mathbf{h}_t)$$

Here,

- \mathbf{w}_t is the word (one-hot or vocabulary index) at time step t,
- \mathbf{x}_t is the corresponding embedding of size D,
- \mathbf{h}_t is the hidden value encapsulating the history of size H, and
- \mathbf{y}_t is the output.

The parameters of the model are just the matrices $\mathbf{U}, \mathbf{V}, \mathbf{W}, \mathbf{X}$.

Notice how \mathbf{h}_t has access to representations from previous time steps only through \mathbf{h}_{t-1} . In addition, calculating \mathbf{h}_t requires waiting for the computation of \mathbf{h}_{t-1} ; for this reason, RNNs are inherently serial in their computations. You will now test a new approach where the context at time step t directly incorporates word embeddings from all previous time steps.

You will first explore a simple and naive model called *The Uniform Attention Model*, so-called because at each time step the model assigns equal attention to all previous words when calculating the distribution for the next word.

At each time step t, \mathbf{h}_t is calculated simply based on the mean of all the previous word embeddings $\mathbf{x}_1, \mathbf{x}_2, \dots \mathbf{x}_t$:

$$\mathbf{h}_t = rac{1}{t} \sum_{i=1}^t \mathbf{x}_i$$

Now, complete the implementation of the forward function. (As usual, the final softmax is handled by the loss function, so you needn't handle that.)

Hint: You might find torch.cumsum and torch.arange helpful.

```
In [ ]: class UATTNLM(torch.nn.Module):
          def __init__(self, hf_tokenizer, embedding size, hidden size):
            super(). init ()
            self.hf tokenizer = hf_tokenizer
            self.pad index = hf tokenizer.pad token id
            self.hidden size = hidden size
            vocab size = len(hf tokenizer)
            # In this uniform attention model, the hidden size and the
            # embedding size must be the same, since the former is
            # just a sum over the latter
            assert hidden_size == embedding_size
            # Create modules
            self.embed = torch.nn.Embedding(vocab size, embedding size)
            self.hidden2output = torch.nn.Linear(hidden size, vocab size)
          def forward(self, context words):
            """Computes the distribution over the next word given
               context `history words`.
            Arguments:
              context_words: a list of word strings, could be an
                             empty list when generating the first word.
              logits, should be a tensor of size (bsz, vocab_size)
            self.eval()
            context = self.hf tokenizer(context words, is split into words=True, ret
                      .long()\
                      .to(device) # bsz, context len
            context_len = context.size(1)
            # For generating the first word, we feed in a special beginning-of-sente
            # which is also what we use for padding. In future labs we'll be using <
            # as training and evaluation use the same beginning-of-sentence symbol,
            # which particular symbol we use.
```

```
if context_len == 0:
    context = context.new(1, 1).fill_(self.pad_index)
    context_len = context.size(1)

hidden = None
# TODO: calculate output and set logits
# Logits should be a tensor of size (1, vocab_size)
# The structure of the network is
# embeddings -> output -> hidden2output -> logits

...
logits = logits[:, -1]
return logits
```

Question: For uniform attention to work as above, H and D, the hidden size and the embedding size, must be the same. Why?

Type your answer here, replacing this text.

Now, let's load the models for Hamilton and Madison. The model

u_attn_lm_madison was trained on documents authored by Madison, while

u attn lm hamilton was trained on documents authored by Hamilton.

```
In [ ]: # Create and load uniform attention LM for Madison
        u attn lm madison = UATTNLM(hf tokenizer,
                                    embedding size=128,
                                    hidden size=128,
                                   ).to(device)
        u attn lm madison.load state dict(
            torch.load(dest path + subdirectory models + 'ualm lm m.pt',
                       map location=device))
        u attn lm madison.eval()
        # Create and load uniform attention LM for Hamilton
        u attn lm hamilton = UATTNLM(hf tokenizer,
                                     embedding size=128,
                                     hidden size=128,
                                    ).to(device)
        u attn lm hamilton.load state dict(
            torch.load(dest path + subdirectory models + 'ualm lm h.pt',
                       map location=device))
        u_attn_lm_hamilton.eval()
```

Let's try to sample from the model. Don't expect a fluent text yet – more improvements are needed in the model.

```
In []: def sample(model, context):
    """Returns a token sampled from the `model` assuming the `context`"""
    logits = model(context) # calls internally to model.forward(context)

# Normalize to get probabilities
    probs = torch.softmax(logits, -1).view(-1) # vocab_size

# Match probabilities with actual word types
```

```
distribution = {}
    for i, prob in enumerate(probs):
      word = model.hf tokenizer.decode(i, clean up tokenization spaces=True)
      distribution[word] = prob.item()
   prob remaining = random.random()
    for token, prob in sorted(distribution.items()):
        if prob remaining < prob:</pre>
            return token
       else:
            prob remaining -= prob
    raise ValueError
def sample sequence(model, start context, count=100):
    """Returns a sequence of tokens of length `count` sampled successively
      from the `model` starting with the `start context`
    reseed() # for reproducibility
   context = list(start context)
    result = list(start context)
   for i in range(0, count):
       next = sample(model, tuple(context))
        result.append(next)
       context = context + [next]
    return result
```

In the previous model, we made a simple assumption – that all previous words are equally important. Of course, this simple assumption is wrong in general, as different time steps can have varying levels of importance and dependencies.

Next, you will build a model that differentiates between time steps.

Are you ready? You're about to learn about one of the most important and influential mechanisms invented in recent years.

The attention mechanism

The previous model was quite basic and suffered from an obvious problem:

At each time step, we assumed the dependencies on the previous time steps, the context, were static. Ideally, we want the model to determine dynamically at each stage which parts of the context are more relevant and which are less, that is, which parts of the context to *attend to*.

You'll now be introduced to the *attention mechanism* (proposed in this seminal paper), a powerful architecture that allows models to selectively focus on different parts of the input sequence. This mechanism has become a cornerstone in modern natural-language processing, significantly enhancing a model's ability to model language distributions.

Attention works by *querying* a (dynamically sized) sequence of *keys* associated with *values*. Each key-value pair is associated with the context elements of a particular token.

As usual, each query, key, and value is represented as a vector. Attention proceeds as follows: The query is compared to each of the keys, providing a score that specifies how much each key should be attended to. The attention can then be summarized by taking an average of the *values* weighted by the attention score of the corresponding *keys*. This *context vector* can then be used as another input to other processes you will learn about in the next lab.

More formally, let's suppose we have a query vector \mathbf{q} of size D, a set of S keys \mathbf{k}_i each of size D and a set of S values \mathbf{v}_i each of size D, where D is the embedding size. What we want to do through the attention mechanism is to use the query to attend to the keys, and summarize those values associated with the "relevant" keys into a fixed-size context vector \mathbf{c}_i also of size D. Note that this is different from directly compressing the key-value pairs into a fixed-size vector, since depending on the query, we might end up with different context vectors.

In general, not all of these need to be of the same size ${\cal D}$, but we restrict to that case here for simplicity.

We'll turn to implementing this attention mechanism shortly, but first, for concreteness, we show how the mechanism can be used to form a language model. Like the uniform attention mechanism above, we start with an embedding of the vocabulary.

$$\mathbf{x}_t = \mathbf{U}\mathbf{w}_t$$

Next, we construct a query, key, and value for each position with linear transforms for each:

$$\mathbf{q}_t = \mathbf{Q}\mathbf{w}_t \ \mathbf{k}_t = \mathbf{K}\mathbf{w}_t \ \mathbf{v}_t = \mathbf{V}\mathbf{w}_t$$

followed by the attention layer proper:

$$\mathbf{h}_t = \operatorname{attn}(\mathbf{q}_t, \mathbf{k}_{1:t}, \mathbf{v}_{1:t})$$

and finally, a linear layer and softmax to convert to a distribution over the vocabulary.

$$\mathbf{y}_t = \operatorname{softmax}(\mathbf{X}\mathbf{h}_t)$$

We write $\mathbf{k}_{1:t}$ for the sequence of elements $\langle \mathbf{k}_1, \mathbf{k}_2, \dots, \mathbf{k}_t
angle$.

Now back to how the attention calculation works.

To determine the score for a given query and key, it is standard to use a measure of similarity between the query and key. You've seen such similarity measures before, in labs 1-1 and 1-2. A good choice is simply the normalized dot product between query and key. We'll thus take the attention score for query ${\bf q}$ and key ${\bf k}_i$ to be

$$s_i = rac{\mathbf{q} \cdot \mathbf{k}_i}{\sqrt{D}}$$

and the attention vector to be the softmax of the score

$$\mathbf{a} = \operatorname{softmax}(\mathbf{s})$$

where \cdot denotes the dot product (inner product). (There are multiple ways of parameterizing the attention function, but the form we present here is the most popular one.).

The softmax guarantees that the attention scores ${\bf a}$ lie on a simplex (meaning $a_i \geq 0$ and $\sum_i a_i = 1$), so ${\bf a}$ can be viewed as a probability distribution over the key-value pairs. This also lends it some interpretability: the closer a_i is to 1, the more "relevant" a key k_i (and hence its value v_i) is to the given query.

Why divide through by the square root of the hidden size (the length of ${\bf q}$ and of ${\bf k}$)? You can think of this as a bit of neural network magic, one of many in the black art of machine learning. Without this scaling, the dot product values can get quite large, leading to numerical problems in performing the softmax.

But why divide by \sqrt{D} in particular. Here's a hand-waving argument. We'd like to keep all of the values over which we are taking the softmax to have similar means and standard deviations, let's say mean of 0 and standard deviation of 1. Suppose the q and k_i values do have mean of 0 and standard deviation of 1. Then it turns out that (under some simpifying assumptions) s_i will have mean of 0 and variance of D, and since variance grows as the square of the standard deviation, it will have standard deviation of

 $\sqrt{D}.$ Dividing through by that value thus returns us to a standard deviation of 1.

Finally, to compute the result of the attention query – the "context vector" ${\bf c}$ – we take the weighted sum of values using the corresponding attention scores as weights:

$$\mathbf{c} = \sum_{i=1}^S a_i \mathbf{v}_i$$

The closer a_i is to 1, the higher the weight \mathbf{v}_i receives, and the more the context vector "attends to" the value \mathbf{v}_i at position i.

In usage, the result of attention, the context vector for position t, plays the role of \mathbf{h}_t , as above.

Question: Suppose there exists i for which a_i is 1. What will the value of \mathbf{c} be?

Type your answer here, replacing this text.

In practice, instead of computing the context vector once for each query, we want to combine together computations for different queries (for different positions t in the input) to allow for parallel processing on GPUs. This will become especially useful for the transformer implementation in the next lab. We use a matrix $Q \in \mathbb{R}^{T \times D}$ to store T queries, a matrix $K \in \mathbb{R}^{S \times D}$ to store T0 keys, and a matrix T1 a matrix T2 to store the corresponding values. Then we can write down how we compute the attention scores T3 in a matrix form:

$$A = \operatorname{softmax}(QK^{ op}, \dim = -1)$$

and to get the context matrix $C \in \mathbb{R}^{T imes D}$, we have

$$C = AV$$

Question: What is the shape of A? What does A_{ij} represent?

Type your answer here, replacing this text.

Your first job is to implement this calculation by finishing the attention function below, which takes the Q, K, and V matrices and returns the A and C matrices. Note that for these matrices, there is one additional dimension for the batching, so instead of $Q \in \mathbb{R}^{T \times D}$, $K, V \in \mathbb{R}^{S \times D}$, $A \in \mathbb{R}^{T \times S}$, $C \in \mathbb{R}^{T \times D}$, we have $Q \in \mathbb{R}^{B \times T \times D}$, $K, V \in \mathbb{R}^{B \times S \times D}$, $A \in \mathbb{R}^{B \times T \times S}$, $C \in \mathbb{R}^{B \times T \times D}$, where B is the batch size. In addition, the function below also takes an argument mask of

size $\mathbb{R}^{B \times T \times S}$ to mark where attentions are disallowed. This is useful not only in disallowing attending to padding symbols, but also to efficiently train the model.

Hint: You might find torch.bmm helpful for batched matrix multiplications. You might need to transpose and reshape tensors to be able to use this function.

Hint: As mentioned in the beginning of the lab, you might also find torch.transpose, torch.reshape, torch.masked_fill, and torch.softmax useful.

Hint: A simple trick for masking an attention score is to set it to negative infinity before normalization.

Hint: Debugging the attention function can be tricky. Try defining a very small example of batched \mathbf{q} , \mathbf{k} , and \mathbf{v} values for which you can manually carry out the attention claculation and test the code on those, working your way up to more complex examples.

```
In [ ]: #TODO - finish implementing this function.
        def attention(batched_Q, batched_K, batched V, mask=None):
          Performs the attention operation and returns the attention matrix
          `batched A` and the context matrix `batched C` using queries
          `batched Q`, keys `batched K`, and values `batched V`.
          Arguments:
              batched Q: (bsz, q len, D)
              batched K: (bsz, k len, D)
              batched V: (bsz, k len, D)
              mask: (bsz, q len, k len). An optional boolean mask *disallowing*
                    attentions where the mask value is *`False`*.
          Returns:
              batched A: the normalized attention scores (bsz, q len, k len)
              batched C: a tensor of size (bsz, q len, D).
         # Check sizes
          D = batched Q.size(-1)
          bsz = batched Q.size(0)
          q len = batched Q.size(1)
          k len = batched K.size(1)
          assert batched K.size(-1) == D and batched V.size(-1) == D
          assert batched K.size(0) == bsz and batched V.size(0) == bsz
          assert batched V.size(1) == k len
          if mask is not None:
            assert mask.size() == torch.Size([bsz, q len, k len])
          batched A = \dots
          batched_C = ...
```

Causal attention mask

To efficiently train the model, we want to batch the attention operations together such that they can be fully parallelized along the sequence length dimension. (The non-attention operations are position-wise so they are trivally parallelizable.) Each word at position t attends to itself and the t-1 previous words $1,\ldots,t-1$, which means each word has access to a different set of key-value pairs. It might seem then that we can't compute attention for all of the positions at the same time. Is it possible to batch them together?

The solution is to use attention masks. For every position t, we give it all key-value pairs at positions $1,\ldots,T$, and we disallow attending to future words $t+1,t+2,\ldots,T$ through an attention mask. (Recall that the attention function takes a mask argument.) We usually call this attention mask a causal attention mask, as it prevents the leakage of information from the future into the present. Since every t has the same set of (key, value) pairs, we can batch them and compute the context vectors using a single call to the function attention.

What should such a mask be? Implement the causal_mask function below to generate this mask.

Hint: you might find torch.triu useful.

```
In []: #TODO - implement this function, which returns a causal attention mask
    def causal_mask(T):
        """
        Generate a causal mask.
        Arguments:
            T: the length of target sequence
        Returns:
            mask: a T x T tensor, where `mask[i, j]` should be `True`
            if position i can attend to position j, and `False` if i cannot
            attend to j
        """
        mask = ...
        return mask.to(device)
```

```
In [ ]: grader.check("causal_attention_mask")
```

We can visualize the attention mask and manually check if it's what we expected.

```
In [ ]: T = 7
        mask = causal mask(T)
        # Create a colormap where True is one color and False is another
        cmap = mcolors.ListedColormap(['gray', 'maroon'])
        bounds = [-0.5, 0.5, 1.5]
        norm = mcolors.BoundaryNorm(bounds, cmap.N)
        # Plot the matrix
        plt.imshow(mask, cmap=cmap, norm=norm)
        # Add colorbar with labels
        cbar = plt.colorbar(ticks=[0, 1])
        cbar.ax.set yticklabels(['False', 'True']) # Customize your labels here
        # Label the axes
        plt.ylabel('Position $i$')
        plt.xlabel('...can attend to $j$')
        plt.title('Mask');
        # Uncomment the line below if the plot does not show up
        # Make sure to comment that before submitting to gradescope
        # since there would be some autograder issues with `plt.show()`
        #plt.show()
```

By using causal masks, all positions (at the same layer) can be computed at once (if the lower layer has been computed). The parallelizability of the attention mechanism is the key to its success since it allows for training on vast amounts of data.

Now we are ready to complete the attention model.

Hint: The causal mask is a 2-D matrix, but we want to add a batch dimension, and expand it to be of the desired size. For this purpose, you can use torch.repeat.

Complete the forward function.

```
In []:
    def __init__(self, hf_tokenizer, embedding_size, hidden_size):
        super().__init__()
        self.hf_tokenizer = hf_tokenizer
        self.pad_index = hf_tokenizer.pad_token_id
        self.hidden_size = hidden_size
        vocab_size = len(hf_tokenizer)

# Create modules
    self.embed = torch.nn.Embedding(vocab_size, embedding_size)
        self.q_proj = nn.Linear(embedding_size, hidden_size)
        self.k_proj = nn.Linear(embedding_size, hidden_size)
        self.v_proj = nn.Linear(embedding_size, hidden_size)
        self.v_proj = nn.Linear(embedding_size, hidden_size)
        self.hidden2output = torch.nn.Linear(hidden_size, vocab_size)
```

```
def forward(self, context words):
  """Computes the distribution over the next word given context `history w
  Arguments:
    context words: batch a list of word strings, could be an empty list wh
                   the first word.
  Returns:
    logits, should be a tensor of size (batch , vocab size)
  self.eval()
  context = self.hf tokenizer(context words, is split into words=True, ret
            .long()\
            .to(device) # bsz, context len
  context len = context.size(1)
  context bsz = context.size(0)
  # For generating the first word, we feed in a special beginning-of-sente
  # which is also what we use for padding. In future labs we'll be using <
  # as training and evaluation use the same beginning-of-sentence symbol,
  # which particular symbol we use.
  if context len == 0:
    context = context.new(1, 1).fill (self.pad index)
    context len = context.size(1)
  hidden = None
  # TODO: finish feedforward and set logits
  # Logits should be a tensor of size (bsz, seq length, vocab size)
  # The structure of the network is
     embeddings -> output -> hidden2output -> logits
  output = attention(q, k, v, mask)[1]
  logits = self.hidden2output(output)
  logits = logits[:, -1]
  return logits
```

Now, let's load the pretrained models for Hamilton and Madison. The model attn_lm_madison was trained on documents authored by Madison, whereas attn_lm_hamilton was trained on documents authored by Hamilton.

Sampling from the attention model

Let's try to sample from the models.

Evaluating text according to the attention model

```
In [ ]: document_madison = validation_madison[0]['tokens']
    document_hamilton = validation_hamilton[0]['tokens']
```

Just like in Lab 2-3, we want to evaluate the models using perplexity.

```
In [ ]: def neglogprob(tokens, model):
            """Returns the negative log probability of a sequence of `tokens`
               according to a `model` returning an unnormalized score for each toker
            reseed()
            score = 0.0
            for i in tgdm(range(len(tokens))):
                context = tokens[:i]
                token = tokens[i]
                logits = model(context)
                probs = torch.softmax(logits, -1).view(-1) # vocab size
                distribution = {}
                for i, prob in enumerate(probs):
                    word = hf tokenizer.decode(i, clean up tokenization spaces=True)
                    distribution[word] = prob.item()
                    prob = distribution[token] if token in distribution else distrib
                # Handle underflow to prevent math domain errors
                if prob <= 0.0:
                    prob = 1e-20
                score += -math.log2(prob)
                #exit
            return score
```

```
In [ ]: def perplexity(tokens, model):
    """Returns the perplexity of a sequence of `tokens` according to a `mode
    """
    return 2 ** (neglogprob(tokens, model) / (len(tokens)))
```

Calculate the perplexity of each model on each document.

```
In []: # TODO
    attn_ppl_madison_model_madison_document = ...
    attn_ppl_hamilton_model_madison_document = ...
    attn_ppl_madison_model_hamilton_document = ...
    attn_ppl_hamilton_model_hamilton_document = ...
```

Now, let's compare those perplexity values. Do the results make sense to you?

Although attention is a powerful concept, it alone isn't enough to build a strong model. In the next lab, you will learn about the *Transformer language model*, which is built on the attention mechanism along with several other important components.

Lab debrief

Question: We're interested in any thoughts your group has about this lab so that we can improve this lab for later years, and to inform later labs for this year. Please list any issues that arose or comments you have to improve the lab. Useful things to comment on include the following:

- Was the lab too long or too short?
- Were the readings appropriate for the lab?
- Was it clear (at least after you completed the lab) what the points of the exercises were?
- Are there additions or changes you think would make the lab better?

Type your answer here, replacing this text.

End of Lab 2-6

To double-check your work, the cell below will rerun all of the autograder tests.

```
In [ ]: grader.check_all()
```