→ Lab session 2: Language modeling

This lab covers classical and neural language models as seen in the theory lectures.

General instructions:

- Complete the code where needed
- · Provide answers to questions only in the cell where indicated
- Do not alter the evaluation cells (## evaluation) in any way as they are needed for the partly automated evaluation process

▼ How AI can write a paper!

We shall train our language model on a corpora of scientific articles and see if we can generate a new one!



```
# import necessary packages
from __future__ import division
from __future__ import unicode_literals
import random as rand
import numpy as np
import pandas as pd
from sklearn.model_selection import train_test_split
```

for reproducability

np.random.seed(SEED)

SEED = 42

→ Data exploration

Lets download and look into the data:

```
!wget "https://www.dropbox.com/s/99az9n1b57qkd9j/arxivData.json.tar.gz?dl=1" -0 arxivData.json.tar.gz
!tar -xvzf arxivData.json.tar.gz
data = pd.read_json("./arxivData.json")
data.sample(n=5)
```



--2020-03-17 18:40:00-- https://www.dropbox.com/s/99az9n1b57qkd9j/arxivData.json.tar.gz?dl=1

Resolving www.dropbox.com (www.dropbox.com (<

Connecting to www.dropbox.com (www.dropbox.com</a

HTTP request sent, awaiting response... 301 Moved Permanently

Location: /s/dl/99az9n1b57qkd9j/arxivData.json.tar.gz [following]

--2020-03-17 18:40:00-- https://www.dropbox.com/s/dl/99az9n1b57qkd9j/arxivData.json.tar.gz

Reusing existing connection to www.dropbox.com:443.

HTTP request sent, awaiting response... 302 Found

HTTP request sent, awaiting response... 200 OK

Length: 18933283 (18M) [application/binary]

Saving to: 'arxivData.json.tar.gz'

arxivData.json.tar. 100%[=========>] 18.06M 16.1MB/s in 1.1s

2020-03-17 18:40:03 (16.1 MB/s) - 'arxivData.json.tar.gz' saved [18933283/18933283]

arxivData.json

	author	day	id	link	month	summary	tag	
857	[{'name': 'A. N. Gorban'}, {'name': 'A. Y. Zin	2	0809.0490v2	[{'rel': 'related', 'href': 'http://dx.doi.org	9	In many physical, statistical, biological and	[{'term': 'cs.LG', 'scheme': 'http://arxiv.org	Principal Grap Mai
144	[{'name': 'Behnam Neyshabur'}, {'name': 'Ryota	27	1503.00036v2	[{'rel': 'alternate', 'href': 'http://arxiv.or	2	We investigate the capacity, convexity and cha	[{'term': 'cs.LG', 'scheme': 'http://arxiv.org	Norm-Based Ca Control in Ne
22961	[{'name': 'Maria De- Arteaga'}, {'name': 'Willi	27	1711.09522v2	[{'rel': 'alternate', 'href': 'http://arxiv.or	11	This is the Proceedings of NIPS 2017 Workshop	[{'term': 'stat.ML', 'scheme': 'http://arxiv.o	Proceedings o 2017 Worksl Mach
19448	[{'name': 'J. Keppens'}, {'name': 'Q. Shen'}]	30	1107.0035v1	[{'rel': 'related', 'href': 'http://dx.doi.org	6	The predominant knowledge-based approach to au	[{'term': 'cs.Al', 'scheme': 'http://arxiv.org	Compositional Repositor Dynan
2331	[{'name': 'Kevin T. Kelly'}, {'name': 'Conor M	15	1203.3488v1	[{'rel': 'alternate', 'href': 'http://arxiv.or	3	Over the past two decades, several consistent	[{'term': 'cs.LG', 'scheme': 'http://arxiv.org	Causal Conclusion Flip Repeatedly an

```
# Assemble lines: concatenate title and description
lines = data.apply(lambda row: row['title'] + ' ; ' + row['summary'], axis=1).tolist()
# Sample the first 3 lines...
sorted(lines, key=len)[:3]
```

['Differential Contrastive Divergence; This paper has been retracted.',
'What Does Artificial Life Tell Us About Death?; Short philosophical essay',
'P=NP; We claim to resolve the P=?NP problem via a formal argument for P=NP.']

```
# Convert lines into strings of space-separated tokens
from nltk.tokenize import WordPunctTokenizer
tknzr = WordPunctTokenizer()
lines = [tknzr.tokenize(sent.lower()) for sent in lines]
lines = [' '.join(sent) for sent in lines]
sorted(lines, key=len)[:3]
```

['differential contrastive divergence; this paper has been retracted .',
 'what does artificial life tell us about death ?; short philosophical essay',
 'p = np; we claim to resolve the p =? np problem via a formal argument for p = np .']

▼ N-Gram Language Model

A language model is a probabilistic model that estimates text probability: the joint probability of all tokens w_t in text X:

$$P(X) = P(w_1, \ldots, w_T).$$

It can do so by following the chain rule:

$$P(w_1,\ldots,w_T) = P(w_1)P(w_2 \mid w_1)\ldots P(w_T \mid w_1,\ldots,w_{T-1}).$$

The problem with such approach is that the final term $P(w_T \mid w_1, \dots, w_{T-1})$ depends on n-1 previous words. This probability is impractical to estimate for long texts, e.g. T=1000.

One popular approximation is to assume that the next word only depends on a finite amount of previous words:

$$P(w_t \mid w_1, \dots, w_{t-1}) = P(w_t \mid w_{t-n+1}, \dots, w_{t-1})$$

Such a model is called an **n-gram language model** where n is a parameter. For example, in 3-gram language model, each word only depends on the 2 previous words.

$$P(w_1,\ldots,w_n) = \prod_t P(w_t \mid w_{t-n+1},\ldots,w_{t-1}).$$

You can also sometimes see that approximation under the name of the *n-th order markov assumption*.

The first stage in building such a model is counting all word occurrences given the n-1 previous words:

```
from collections import defaultdict, Counter
# special tokens:
# - unk represents absent tokens,
# - eos is a special token after the end of sequence
UNK, EOS = " UNK ", " EOS "
def count ngrams(lines, n):
    Count how many times each word occured after (n - 1) previous words
    :param lines: an iterable of strings with space-separated tokens
    :returns: a dictionary { tuple(prefix tokens): {next token 1: count 1, next token 2: count 2}}
    When building counts, please consider the following two edge cases
    - if prefix is shorter than (n - 1) tokens, it should be padded with UNK. For n=3,
      empty prefix: "" -> (UNK, UNK)
      short prefix: "the" -> (UNK, the)
      long prefix: "the new approach" -> (new, approach)
    - you should add a special token, EOS, at the end of each sequence
      "... with deep neural networks ." -> (..., with, deep, neural, networks, ., EOS)
      count the probability of this token just like all others.
    11 11 11
```

```
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

dummy_lines = sorted(lines, key=len)[:100]
dummy_counts = count_ngrams(dummy_lines, n=3)
assert set(map(len, dummy_counts.keys())) == {2}, "please only count {n-1}-grams"
assert dummy_counts['author', '.']['_EOS_'] == 1
assert dummy_counts['p', '=']['np'] == 2
assert len(dummy_counts[('_UNK_', '_UNK_')]) == 78
assert dummy_counts['_UNK_', 'a']['note'] == 3

print('well done!')
```

well done!

Once we can count N-grams, we can build a probabilistic language model. The simplest way to compute probabilities is in proporiton to counts:

$$P(w| extit{prefix}) = rac{Count(extit{prefix},w)}{\sum_{w' \in extit{Vocab}} Count(extit{prefix},w')}$$

```
class NGramLanguageModel:
   def init (self, lines, n):
       Train a simple count-based language model:
       compute probabilities P(w | prefix) given n-gram counts
       :param n: computes probability of next token given (n - 1) previous words
       :param lines: an iterable of strings with space-separated tokens
       assert n >= 1
       self.n = n
       counts = count_ngrams(lines, self.n)
       self.probs = defaultdict(Counter)
       # compute token probabilities (self.probs), given the counts computed above
       # probs[(word1, word2)][word3] = P(word3 | word1, word2)
       for sent in lines:
        sent = sent.split() + [EOS]
        prefix = [UNK] * (n-1)
         for word in sent:
          sum = 0
          for count in counts[tuple(prefix)].values():
            sum += count
          self.probs[tuple(prefix)][word] = counts[tuple(prefix)][word] / sum
          if len(prefix) > 0:
            prefix.pop(0)
            prefix.append(word)
       def get possible next tokens(self, prefix):
       inanam profix: string with space-sonarated profix tokons
```

```
:returns: a dictionary {token : its probability} for all tokens with positive probabilities
"""

prefix = prefix.split()
prefix = prefix[max(0, len(prefix) - self.n + 1):]
prefix = [ UNK ] * (self.n - 1 - len(prefix)) + prefix
return self.probs[tuple(prefix)]

def get_next_token_prob(self, prefix, next_token):
"""

:param prefix: string with space-separated prefix tokens
:param next_token: the next token to predict probability for
:returns: P(next_token|prefix) a single number, 0 <= P <= 1
"""

return self.get_possible_next_tokens(prefix).get(next_token, 0)</pre>
```

Let's test it!

```
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

dummy_lm = NGramLanguageModel(dummy_lines, n=3)

p_initial = dummy_lm.get_possible_next_tokens('') # '' -> ['_UNK_', '_UNK_']
assert np.allclose(p_initial['learning'], 0.02)
assert np.allclose(p_initial[a], 0.13)
assert np.allclose(p_initial.get('meow', 0), 0)
assert np.allclose(sum(p_initial.values()), 1)

p_a = dummy_lm.get_possible_next_tokens('a') # '' -> ['_UNK_', 'a']
assert np.allclose(p_a['machine'], 0.15384615)
assert np.allclose(p_a['note'], 0.23076923)
assert np.allclose(p_a.get('the', 0), 0)
assert np.allclose(sum(p_a.values()), 1)

assert np.allclose(dummy_lm.get_possible_next_tokens('a note')['on'], 1)
assert dummy_lm.get_possible_next_tokens('a machine') == \
```

```
dummy_lm.get_possible_next_tokens("there have always been ghosts in a machine"), \
   "your 3-gram model should only depend on 2 previous words"

print("Good job!")
```



Now that you've got a working n-gram language model, let's see what sequences it can generate. But first, let's train it on the whole dataset.

```
lm = NGramLanguageModel(lines, n=3)
```

The process of generating sequences is... well, it's sequential. You maintain a list of tokens and iteratively add the next token by sampling from the probabilities over the vocabulary at each point in the sequence.

$$X = []$$

forever:

- $w_{next} \sim P(w_{next}|X)$
- $X = concat(X, w_{next})$

Instead of sampling with probabilities, one can also take the most likely token, sample from among the top-K most likely tokens, or sample with a certain *temperature*. In the latter case (temperature), one samples from

$$w_{next} \sim rac{P(w_{next}|X)^{1/ au}}{\sum_{w'} P(w'|X)^{1/ au}}$$

Where $\tau>0$ is the model temperature. If $\tau<<1$, more likely tokens will be sampled with even higher probability while less likely tokens will vanish. For sampling from a given probability distribution, the function <code>np.random.choice</code> can be used.

```
def get_next_token(lm, prefix, temperature=1.0):
    return next token after prefix;
    :param temperature: samples proportionally to lm probabilities ^ (1 / temperature)
        if temperature == 0, always takes most likely token. Break ties arbitrarily.
    """
```

```
if temperature == 0:
 inverse temperature = 1
else:
 inverse temperature = 1 / temperature
possible next tokens = lm.get possible next tokens(prefix)
probs dict = {}
sum = 0
for possible next token in possible next tokens:
 next token prob = lm.get next token prob(prefix, possible next token)
 next token prob with temperature = pow(next token prob, inverse temperature)
 probs dict[possible next token] = next token prob with temperature
 sum += next token prob with temperature
keys = list(probs dict.keys())
values = [value / sum for value in list(probs dict.values())]
if temperature == 0:
 max indices = np.where(values == np.amax(values))[0]
 token = keys[np.random.choice(max indices)]
else:
 token = np.random.choice(keys, p=values)
return token
```

```
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

from collections import Counter
test_freqs = Counter([get_next_token(lm, 'there have') for _ in range(10000)])
assert 250 < test_freqs['not'] < 450
assert 8500 < test_freqs['hoor'] < 9500</pre>
```

```
assert 1 < test_freqs['lately'] < 200
test_freqs = Counter([get_next_token(lm, 'deep', temperature=1.0) for _ in range(10000)])
assert 1500 < test_freqs['learning'] < 3000
test_freqs = Counter([get_next_token(lm, 'deep', temperature=0.0) for _ in range(10000)])
assert test_freqs['learning'] == 10000
test_freqs = Counter([get_next_token(lm, 'deep', temperature=0.5) for _ in range(10000)])
assert 8000 < test_freqs['learning'] < 9000

print("Looks nice!")</pre>
```



Let's have fun with this model:

```
prefix = 'artificial' # <- your ideas on the start of your AI generated scientific paper :)

for i in range(100):
    prefix += ' ' + get_next_token(lm, prefix)
    if prefix.endswith(EOS) or len(lm.get_possible_next_tokens(prefix)) == 0:
        break

print(prefix)</pre>
```

e artificial persuasion in abstract argumentation frameworks . this work , we use a technique to solve this problem has been succe

```
prefix = 'bridging the' # <- more of your ideas

for i in range(100):
    prefix += ' ' + get_next_token(lm, prefix, temperature=0.4)
    if prefix.endswith(EOS) or len(lm.get_possible_next_tokens(prefix)) == 0:
        break

print(prefix)</pre>
```

bridging the gap between the amount of data mining , and it is possible to train a deep neural networks (cnns) and the results

Question:

1. How does the temperature parameter affect the generated samples?

When the temperature is very small, more likely tokens will be sampled with an even higher probability. There will be a lower chance of getting 'exotic' words and sentences.

Evaluating language models: perplexity

Perplexity is a measure of how well does your model approximate true probability distribution behind data. **Smaller perplexity = better model**.

To compute perplexity on one sentence, use:

$$\mathbb{P}(w_1\dots w_N) = P(w_1,\dots,w_N)^{-rac{1}{N}} = \left(\prod_t P(w_t\mid w_{t-n},\dots,w_{t-1})
ight)^{-rac{1}{N}},$$

On the corpus level, perplexity is a product of probabilities of all tokens in all sentences to the power of 1, divided by **total length of all sentences** in corpora.

This number can quickly get too small for float32/float64 precision, so we recommend you to first compute log-perplexity (from log-probabilities) and then take the exponent.

```
def perplexity(lm, lines, min_logprob=np.log(10 ** -50.)):
    """
    :param lines: a list of strings with space-separated tokens
    :param min_logprob: if log(P(w | ...)) is smaller than min_logprob, set it equal to min_logrob
    :returns: corpora-level perplexity - a single scalar number from the formula above

Note: do not forget to compute P(w_first | empty) and P(eos | full_sequence)

PLEASE USE lm.get_next_token_prob and NOT lm.get_possible_next_tokens
```

```
N = 0 # number of tokens
   ppl = 0.0 # perplexity
   # https://stats.stackexchange.com/questions/129352/how-to-find-the-perplexity-of-a-corpus
   sum = 0
   for sent in lines:
     # calculate sentence probability
     sent log prob = 0
     sent = sent.split() + [EOS]
     prefix = [UNK] * (lm.n-1)
     for word in sent:
       N += 1
       word prob = lm.get next token prob(" ".join(prefix), word)
       if word prob == 0:
         sent_log_prob += min_logprob
       else:
         sent log prob += max(min logprob, np.log(word prob))
       if len(prefix) > 0:
         prefix.pop(0)
         prefix.append(word)
     # update sum with sentence log probability or
     sum += sent log prob
   # calculate perplexity
   ppl = np.exp(-1/N * sum)
   return ppl
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY
lm1 = NGramLanguageModel(dummy_lines, n=1)
```



Now let's measure the actual perplexity: we'll split the data into train and test and score model on test data only.

```
train_lines, test_lines = train_test_split(lines, test_size=0.25, random_state=SEED)

for n in (1, 2, 3):
    lm = NGramLanguageModel(n=n, lines=train_lines)
    ppx = perplexity(lm, test_lines)
    print("N = %i, Perplexity = %.5f" % (n, ppx))
```



Question:

2. Do you expect increasing/decreasing perplexities for language models with longer n-grams (i.e., higher values of n)? Does this correspond with the test output you observe above? If not: can you explain this?

The longer the n-gram, the lower the perplexity on the train set. When n becomes to big, too many sequences of words have nog been seen yet, so the results will get worse.

▼ LM Smoothing

The problem with our simple language model is that whenever it encounters an n-gram it has never seen before, it assigns it the probability of 0. Every time this happens, perplexity explodes. To battle this issue, there's a technique called **smoothing**.

The core idea is to modify counts in a way that prevents probabilities from getting too low. The simplest algorithm here is *additive* smoothing (aka <u>Lapace smoothing</u>):

$$P(w|\textit{prefix}) = rac{Count(\textit{prefix},w) + \pmb{\delta}}{\sum_{w' \in \textit{Vocab}}(\textit{Count}(\textit{prefix},w') + \pmb{\delta})}$$

If counts for a given prefix are low, additive smoothing will adjust probabilities to a more uniform distribution.

```
class LaplaceLanguageModel(NGramLanguageModel):
   def init (self, lines, n, delta=1.0):
       self.n = n
       counts = count ngrams(lines, self.n) #
       self.vocab = set(token for token counts in counts.values() for token in token counts) #
       self.probs = defaultdict(Counter)
       # compute token proabilities
       # probs[prefix][token] = ...
       for sent in lines:
         sent = sent.split() + [EOS]
         prefix = [UNK] * (n-1)
         for word in sent:
           sum = 0
           for count in counts[tuple(prefix)].values():
             sum += count
           self.probs[tuple(prefix)][word] = (counts[tuple(prefix)][word] + delta) / (sum + len(self.vocab)*delta)
           if len(prefix) > 0:
            prefix.pop(0)
            prefix.append(word)
       def get possible next tokens(self, prefix):
       token probs = super().get possible next tokens(prefix)
       missing prob total = 1.0 - sum(token probs.values())
       missing prob = missing prob total / max(1, len(self.vocab) - len(token probs))
       return {token: token probs.get(token, missing prob) for token in self.vocab}
   def get_next_token_prob(self, prefix, next_token):
       token_probs = super().get_possible_next_tokens(prefix)
       if next_token in token_probs:
           return token probs[next token]
```

```
else:
            missing prob total = 1.0 - sum(token probs.values())
            missing_prob_total = max(0, missing_prob_total) # prevent rounding errors
            return missing prob total / max(1, len(self.vocab) - len(token probs))
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY
# test that it's a valid probability model
for n in (1, 2, 3):
    dummy lm = LaplaceLanguageModel(dummy lines, n=n)
    assert np.allclose(sum([dummy lm.get next token prob('a', w i) for w i in dummy lm.vocab]), 1), "You broke something!:)"
print('Great!')
```



```
# calculate perplexity for LaplaceLanguageModel
from sklearn.model selection import train test split
train lines, test lines = train test split(lines, test size=0.25, random state=SEED)
for n in (1, 2, 3):
    lm = LaplaceLanguageModel(n=n, lines=train lines)
    ppx = perplexity(lm, test lines)
    print("N = %i, Perplexity = %.5f" % (n, ppx))
```



Question:

- 3. In a bigram language model (without smoothing), which of the following two phrases do you expect to have higher probablity? Why?
 - and and
 - this paper

The probability of "and and" is expected to be zero, as it is not part of the valid english grammar. The sequence "this paper" can appear in the corpus, so the probability is expected to be higher than 0, and thus also expected to higher than that of "and and".

4. If we add smoothing, how would the probability relation change for the above phrases?

The probability of "and and" will be higher than 0, but still lower than that of "this paper".

Train both language models (smoothing and non-smoothing version) on dummy_lines and report perplexity for the given phrases.

```
lm_names = ["without smoothing", "with smoothing"]
phrases = [["and and"], ["this paper"]]

############ for student ###############

ngram = NGramLanguageModel(n=2, lines=dummy_lines)
laplace = LaplaceLanguageModel(n=2, lines=dummy_lines)
models = [ngram, laplace]
for phrase in phrases:
```

without smoothing: phrase = '['and and']' --> pp = 1.00E+50 with smoothing: phrase = '['and and']' --> pp = 1.01E+03 without smoothing: phrase = '['this paper']' --> pp = 2.79E+33 with smoothing: phrase = '['this paper']' --> pp = 3.48E+02

Deep Learning Based Language Models

We've checked out statistical approaches to language models so far. Now let's go find out what deep learning has to offer. We're gonna use the same dataset as before.

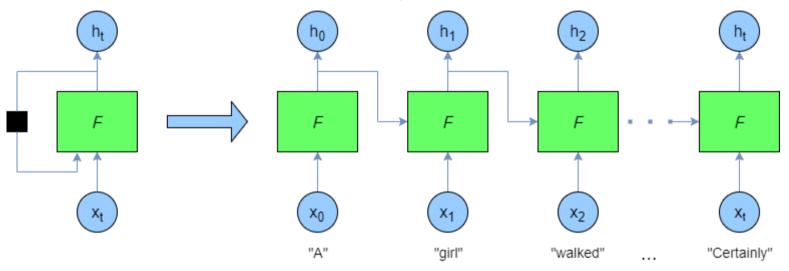


We are going to implement the simplest recurrent neural network (RNN) known as the Elman RNN. Its hidden state aims to encapsulate the information for all previous input elements in order to help the network to take into account *the past*. Since there is no hidden state during the first step, we feed the network with an initial state of zero values (or randomly initilized values). Next, we feed it the first token A together with the hidden state of the previous step, to predcit the next output (girl). We'll repeat this procudure until the end of sequence.

We can summarize the above explanation into a simple equation as:

$$h_t = F(x_t, h_{t-1}) = f(W_x x_t + W_h h_{t-1} + b),$$

where x_t is input, h_t is hidden state, b is bias term and f is the tanh non-linearity.



```
### If you don't have pytorch yet: install it in the current kernel first.
### Uncomment next 2 lines to do that.
import sys
!conda install --yes --prefix {sys.prefix} pytorch

# first import necessary packages
import torch
import torch.nn as nn
import matplotlib.pyplot as plt

%matplotlib inline
```

```
# for reproducibility
torch.manual_seed(SEED)
torch.cuda.manual_seed(SEED)

torch.backends.cudnn.enabled = False
torch.backends.cudnn.benchmark = False
torch.backends.cudnn.deterministic = True

device = torch.device('cuda' if torch.cuda.is_available() else 'cpu')
print(device)
```



▼ Tokenization

Before implementing the neural network itself, lets prepare the data. We need special tokens:

- Begin Of Sequence (**BOS**) this token is at the start of each sequence. We use it so that we always have non-empty input to our neural network. $P(x_t) = P(x_1 | BOS)$
- End Of Sequence (**EOS**) you guess it... this token is at the end of each sequence. The catch is that it should **not** occur anywhere else except at the very end. If our model produces this token, the sequence is over.

```
BOS, EOS = '<s>', '</s>'
text = [BOS + ' ' + line + ' ' + EOS for line in lines] # concatenate BOS and EOS to all sentences
text = [line.split() for line in text]

# let's print the first sentence
print(text[0])
```



Let us convert our raw text into sequences of ids. First, create two sorted dictionaries and map each token to an id. Second, create the target data sequences based on the input. Note that the target is the same as the input, except that it is one token ahead of the input.

```
from collections import Counter
def get data lm(data):
    word counts = Counter()
    for sent in data:
        word counts.update(sent)
    sorted token = sorted(word counts, key=word counts.get, reverse=True)
    # create two dictionaries to convert token to id or vice-verca
    id to token = {k: w for k, w in enumerate(sorted token)}
    token to id = {w: k for k, w in id to token.items()}
    n_token = len(id to token)
    tokenized text = [[token to id[w] for w in sent] for sent in data]
    # output is one token ahead
    inp text = [sent[:-1] for sent in tokenized text]
    out text = [sent[1: ] for sent in tokenized text]
    return id to token, token to id, n token, inp text, out text
id to token, token to id, n token, inp text, out text = get data lm(text[0:1])
print("-- raw text:", text[0:1])
print('-' * 100)
print("-- input text:", inp text)
print('-' * 100)
print("-- output text:", out_text)
```



For training, we won't put the entire sequence through the model at once. As explained in the theory lectures, we will limit the sequence length over which we apply back-propagation (which we'll call the *bptt*, or the back-propagation-through-time length), and arrange the bptt-long segments into mini-batches for parallel training (also see the slides, for considerations on choosing the mini-batch size).

Let us first convert the sequences into mini-batches

Suppose our batch size is 4 (for benefiting from a gpu, you'll need to scale this up), bptt is 3 (in practice it will be much longer, though), and our data consists of a 1-dimensional tensor containing 36 token id's. Each batch will contain a 4x3 input tensor and a 4x3 target tensor, except for the last batch (we can discard the last one during training). As you already know, the target batch is one token ahead (in terms of the original sequence) of the input batch since our task is language modeling! The input/output tensor for the first batch will be something like the following:

Input batch:

1	2	3	4	5	6	7	8	9	
10	11	12	13	14	15	16	17	18	
19	20	21	22	23	24	25	26	27	
28	29	30	31	32	33	34	35	36	

Target batch:

1	2	3	4	5	6	7	8	9
10	11	12	13	14	15	16	17	18
19	20	21	22	23	24	25	26	27
28	29	30	31	32	33	34	35	36

We'll convert the input (inp_text) and target (out_text) indices from the previous steps into mini-batches as follows:

```
from itertools import chain
from sklearn.utils import shuffle
def get batches(inp text, out text, batch size, seq size):
    # shuffle the sentences
    itext, otext = shuffle(inp text, out text)
    # flatten the data
    itext = list(chain(*itext))
    otext = list(chain(*otext))
    # work out how cleanly we can divide the dataset into batch size parts.
    num batches = int(len(itext) / (seq size * batch size))
    # trim off any extra elements that wouldn't cleanly fit
    itext = itext[:num batches * batch size * seq size]
    otext = otext[:num batches * batch size * seq size]
    itext = np.reshape(itext, (batch size, -1)) # batch size * tokens
    otext = np.reshape(otext, (batch size, -1)) # batch size * tokens
    for i in range(0, num batches * seq size, seq size):
        yield itext[:, i:i + seq size], otext[:, i:i + seq size]
```

▼ Question:

5. Does trimming, in above, reduce capabilites of the train model?

NO

```
bs = 4
bptt = 10
```

```
batches = get_batches(inp_text, out_text, bs, bptt)

for x,y in batches:
    print('input:', x.shape)
    print(x)
    print('-' * 35)
    print('target:', y.shape)
    print(y)
    break;
```



▼ Let's build the model

By extending the nn.Module you can easily develop your own recurrent cell in pytorch. In this part we shall implement **Elman** Recurrent Network. This module receives a sequence of feature vectors and returns two tensors. Please study the nodes as defined in __init__ and initialized using init_weights, and correctly fill in the forward method in line with the formula for the Elman RNN.

```
class RNN(nn.Module):
    def __init__(self, input_sz, hidden_sz):
        super(RNN, self).__init__()
        # necessary parameters in Elman RNN
        self.input_sz = input_sz
        self.hidden_sz = hidden_sz

        self.fc_x = nn.Linear(self.input_sz, self.hidden_sz, bias=False)
        self.fc_h = nn.Linear(self.hidden_sz, self.hidden_sz) #default: bias = True
```

```
self.init weights()
def init_weights(self):
   nn.init.xavier uniform (self.fc x.weight)
   nn.init.xavier uniform (self.fc h.weight)
   self.fc h.bias.data.fill (0.0)
def forward(self, x, state=None):
   :param x: batch of sequences of input symbols (represented as vectors)
             dimensions of x will be batch size * sequence length * input sz
   :param state: state vector, of size self.hidden sz
   :return: hidden seq, state; where state is final output state,
           hidden seq is list of hidden states h t (see fig. above)
   .....
   # things to do:
   # 1) if state is None, initialize it with zeros (use `torch.zeros`)
   # 2) iterate over time, each time applying the RNN formula, and
        concatenate the state tensors to hidden seg (with `torch.cat`)
   # 3) reshape hidden seq from (seq, batch, feature) to (batch, seq, feature), with `Tensor.transpose`
   hidden seq = []
   if state == None:
     state = torch.zeros((x.size(0),self.hidden sz)).to(device)
   x = x.permute(1,0,2)
   time steps = x.size(0)
   for t in range(time steps):
       state = torch.tanh(self.fc x(x[t]) + self.fc h(state))
       hidden_seq.append(state)
   out = torch.stack(hidden_seq)
   out = out nermute(1.0.2)
```

```
return out, state

## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

# Let's test the RNN module
arr = torch.rand([1, 2, 2]).to(device) # tensor dimension: batch size x bptt x features
my_rnn = RNN(2, 2).to(device)
out, state = my_rnn(arr)

assert out.shape == torch.Size([1, 2, 2]), out.shape
assert state.shape == torch.Size([1, 2]), state.shape

print("RNNCell completed!")
```



The recurrent module is only one part of the neural language model, we still need an embedding layer to convert our tokens into a feature vector and a decoding layer to predict subsequent tokens. To this end, we defined a wrapper and put everything in it. We provided the necessary modules you'll need, please complete the forward function.

```
# call the Embedding, RNN and linear decoder layer in the forward pass
       embeddings = self.encoder(x)
      out, state = self.rnn(embeddings)
      logits = self.decoder(out)
      return logits, state
       def zero state(self, batch size):
      return torch.zeros(batch size, self.hidden size)
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY
# Let's test the whole model
arr = torch.randint(n token, [64, 32], dtype=torch.long).to(device)
rnn language model = RNNLanguageModel(n token, 256).to(device)
state = rnn language model.zero state(64).to(device)
```



Sampling

out, state = rnn language model(arr, state)

print("Sounds good! The model is complete!")

assert type(out) != type(None), 'Do you return output?'
assert type(state) != type(None), 'Do you return state?'
assert out.shape == torch.Size([64, 32, n_token]), out.shape
assert state.shape == torch.Size([64, 256]), state.shape

You will need a function to generate text. For your convenience, we implemented it for you. The idea is to feed one token at a time to the model and concatenate the model's output token to previously predicted tokens.

```
def sample(preds, n token, temperature):
    if temperature == 0:
        choice = np.argmax(preds[0].tolist())
    else:
        preds = preds.squeeze() / temperature
        exp preds = np.exp(preds.tolist())
        preds = exp preds / np.sum(exp preds)
        choice = np.random.choice([*range(n token)], p=preds)
    return choice
def generate text(device, net, n token, token to id, id to token, temperature=1.0):
    # we are in evaluation mode
    net.eval()
    # initialize state
    state h = net.zero state(1).to(device)
    # manually feed some tokens
    initial words = ['recurrent', 'neural']
    for w in initial words:
        ix = torch.tensor([[token to id[w]]], dtype=torch.long).to(device)
        preds, state h = net(ix, state h)
    choice = sample(preds, n token, temperature)
    initial words.append(id to token[choice])
    # generate next tokens (50 tokens at most!)
    for in range(50):
        ix = torch.tensor([[choice]], dtype=torch.long).to(device)
        preds, state_h = net(ix, state_h)
        choice = sample(preds, n_token, temperature)
          # you can stop generation
```

```
# if id_to_token[choice] == EOS:
# break;

initial_words.append(id_to_token[choice])

return ' '.join(initial_words)
```

▼ Training loop

A typical set of steps for training in Pytorch is:

- set model in 'train' mode (note: it will only inform the inner mechanism that we are about to train, but not actually execute the training; we still need to do that ourselves)
- Reset all gradients
- Compute output, loss value, accuracy, etc
- · Perform back-propagation
- Update the network's parameters

```
seq_size = 32
batch_size = 16
hidden_size = 256
temperature = 1.0

dummy_text = text[0:100]
id_to_token, token_to_id, n_token, inp_text, out_text = get_data_lm(dummy_text)

net = RNNLanguageModel(n_token, hidden_size)
net = net.to(device)

criterion = nn.CrossEntropyLoss()
optimizer = torch.optim.Adam(net.parameters())

iteration = 0
total_posche = 100
```

```
rorat chorus - TAA
loss_history = []
for e in range(total epochs):
   batches = get batches(inp text, out text, batch size, seq size)
   state h = net.zero state(batch size)
   state h = state h.to(device)
   for x, y in batches:
       iteration += 1
       x = torch.tensor(x).to(device)
       y = torch.tensor(y).to(device)
       # Things to do:
       # - put model in `train` mode
       # - set gradients to zero
       # - model prediction for given input
       # - calculate loss for a mini-batch
       # - compute gradient
       # - detach state representation by `detach()` (If we did not detach the history of hidden states
         the back-propagated gradients would flow from the loss towards the beginning)
       # - clip gradients (optional)
       # - update parameters
       net.train()
       optimizer.zero grad()
       prediction, state_h = net(x, state_h)
       prediction = prediction.permute(0,2,1)
       loss = criterion(prediction, y)
       loss.backward()
       state_h.detach()
       optimizer.step()
```

```
if iteration % 50 == 0:
    print('epoch: {}/{} iteration: {} loss: {}'.format(e, total_epochs, iteration, loss.item()))

if iteration % 250 == 0:
    print('-' * 50)
    print(generate_text(device, net, n_token, token_to_id, id_to_token, temperature))
    print('-' * 50)

loss_history.append(loss.item())
```



```
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

assert sum(loss_history) / len(loss_history) < 2.0
assert loss_history[-1] < 1.0

print('Fantastico!')</pre>
```

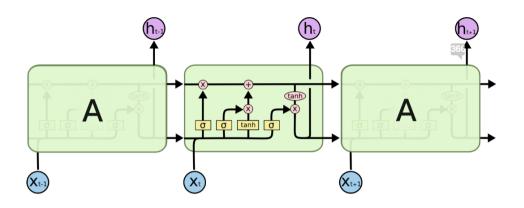
epoch_count = range(1, len(loss_history) + 1)

```
# Visualize loss history
plt.plot(epoch_count, loss_history, 'r--')
plt.legend(['Train Loss'])
plt.xlabel('Epoch')
plt.ylabel('Loss')
plt.show()
```



▼ More powerful cell!

In previous part, we implemented a simple Elman recurrent network. We can easily extend it to an **LSTM** by modifying our code slightly. Since we already learned how to implement the recurrent cell itself, in this part, we will simply use the existing pytorch nn.LSTM implementation. One major difference between these two networks is their **hidden state**. Unlike an Elman RNN which has just a single state, the hidden state of LSTM is made up of two parts. Please take a look at the online documentation of nn.LSTM as well as nn.LSTMCell



```
class LSTMLanguageModel(nn.Module):
   def __init__(self, n_tokens, hidden_sz):
       super(LSTMLanguageModel, self). init ()
       self.hidden size = hidden sz
       self.embedding = nn.Embedding(n tokens, hidden sz)
       self.rnn = nn.LSTM(hidden sz, hidden sz, batch first=True)
       self.decoder = nn.Linear(hidden sz, n tokens)
   def forward(self, x, prev state):
       :return: logits, state; where logits is output of the decoder,
               and state is the final rnn state.
       .....
       embeddings = self.embedding(x)
       out, (state_h, state_c) = self.rnn(embeddings, prev_state)
       logits = self.decoder(out)
       return logits, (state_h, state_c)
       def zero_state(self, batch_size, dev):
       # look up the dimensions of the nn.LSTM state (which is a tuple!)
       # write code to return a zero-initialized state (make use of torch.zeros(...))
       # already put on the correct device (dev)
```

8

LSTM is complete! Perfect!

assert isinstance(state, tuple)
print("LSTM is complete! Perfect!")

Evaluation method

As you may already notice, in previous part, each time we wanted to measure the loss we evaluate it only for a mini-batch which is not an elegant way to do it. Let's create an evaluation method to put everything inside it.

Complete the evaluate() method to return the overall (mean) cross-entropy loss.

assert out.shape == torch.Size([64, 32, n token]), out.shape

```
def evaluate(device, net, n_token, batch_size, seq_size, x_test, y_test):
    net.eval()
    total_loss = 0.

# intialize state
    (state h_state s) = net_zero state(batch_size_device)
```

```
(State_II) State_t/ - Het.ZetU_State(Dattil_StZe) Gevice/
# batchify data
batches = get batches(x test, y test, batch size, seg size)
# loop throgh the batchify data
# use 'criterion' to calculate loss
# detach() states
# return loss
n batches = 0
for x, y in batches:
   n batches += 1
   x = torch.tensor(x).to(device)
   y = torch.tensor(y).to(device)
   prediction, (state_h, state_c) = net(x, (state_h, state_c))
   prediction = prediction.permute(0,2,1)
   loss = criterion(prediction, y)
   state h.detach()
   state c.detach()
   total loss += loss.item()
return total_loss / n_batches
```

```
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY

id_to_token, token_to_id, n_token, inp_text, out_text = get_data_lm(dummy_text)
net = LSTMLanguageModel(n_token, 256).to(device)
dummy_loss = evaluate(device, net, n_token, 10, 64, inp_text, out_text)

assert np.exp(dummy_loss) < 1e4, 'your dummy ppl is too large!'

print("It sounds good")</pre>
```

▼ Train on whole dataset

Lets train the model on the whole train set this time:

```
id_to_token, token_to_id, n_token, inp_text, out_text = get_data_lm(dummy_text)
X_train, X_test, y_train, y_test = train_test_split(inp_text, out_text, test_size=0.25, random_state=SEED)
```

Modify the sampling and training parts from the elman RNN for use with the LSTM model:

```
net = LSTMLanguageModel(n token, hidden size)
net = net.to(device)
criterion = nn.CrossEntropyLoss()
optimizer = torch.optim.Adam(net.parameters())
iteration = 0
total epochs = 100
train loss = 0
train history = []
valid history = []
for e in range(total_epochs):
    batches = get batches(X train, y train, batch size, seq size)
    (state_h, state_c) = net.zero_state(batch_size, device)
   for x, y in batches:
       iteration += 1
```

```
x = torch.tensor(x).to(device)
   y = torch.tensor(y).to(device)
   net.train()
   optimizer.zero grad()
   prediction, (state h, state c) = net(x, (state h, state c))
   prediction = prediction.permute(0,2,1)
   loss = criterion(prediction, y)
   train loss += loss.item()
   loss.backward(retain graph=True)
   state h.detach()
   state c.detach()
   optimizer.step()
if iteration % 50 == 0:
       train loss = train loss / 50.0
       val loss = evaluate(device, net, n token, 10, seq size, X test, y test)
       train history.append(train loss)
       valid_history.append(val_loss)
       print('epoch: {}/{} iteration: {} train-Loss: {} val-loss: {}'.format(e, total epochs, iteration, train loss, val loss))
       train loss = 0
```

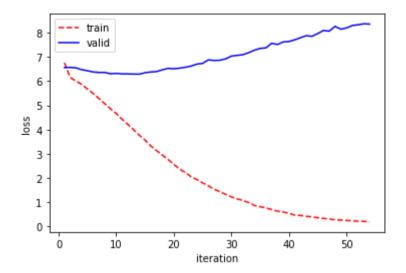
```
epoch: 1/100 iteration: 50 train-Loss: 6.755342435836792 val-loss: 6.559405531202044
epoch: 3/100 iteration: 100 train-Loss: 6.1463697719573975 val-loss: 6.564422539302281
epoch: 5/100 iteration: 150 train-Loss: 6.002133436203003 val-loss: 6.546098743166242
epoch: 7/100 iteration: 200 train-Loss: 5.861508817672729 val-loss: 6.470285211290632
epoch: 9/100 iteration: 250 train-Loss: 5.669991426467895 val-loss: 6.426055703844343
epoch: 11/100 iteration: 300 train-Loss: 5.484770631790161 val-loss: 6.3746671336037775
epoch: 12/100 iteration: 350 train-Loss: 5.2787089443206785 val-loss: 6.354996681213379
epoch: 14/100 iteration: 400 train-Loss: 5.066030349731445 val-loss: 6.356101546968732
epoch: 16/100 iteration: 450 train-Loss: 4.855803050994873 val-loss: 6.301157746996198
epoch: 18/100 iteration: 500 train-Loss: 4.656813659667969 val-loss: 6.317229134695871
epoch: 20/100 iteration: 550 train-Loss: 4.422775478363037 val-loss: 6.295559133802142
epoch: 22/100 iteration: 600 train-Loss: 4.213403840065002 val-loss: 6.2951182297297885
epoch: 24/100 iteration: 650 train-Loss: 3.996145668029785 val-loss: 6.288622685841152
epoch: 25/100 iteration: 700 train-Loss: 3.7648486948013304 val-loss: 6.285948548998151
epoch: 27/100 iteration: 750 train-Loss: 3.5639968967437743 val-loss: 6.345582519258771
epoch: 29/100 iteration: 800 train-Loss: 3.323235101699829 val-loss: 6.374503305980137
epoch: 31/100 iteration: 850 train-Loss: 3.126496920585632 val-loss: 6.393614053726196
epoch: 33/100 iteration: 900 train-Loss: 2.9470784950256346 val-loss: 6.4639672211238315
epoch: 35/100 iteration: 950 train-Loss: 2.76523015499115 val-loss: 6.52517260823931
epoch: 37/100 iteration: 1000 train-Loss: 2.5615936040878298 val-loss: 6.507562841687884
epoch: 38/100 iteration: 1050 train-Loss: 2.3678725242614744 val-loss: 6.532972301755633
epoch: 40/100 iteration: 1100 train-Loss: 2.226295516490936 val-loss: 6.5687135968889505
epoch: 42/100 iteration: 1150 train-Loss: 2.0489106440544127 val-loss: 6.620206219809396
epoch: 44/100 iteration: 1200 train-Loss: 1.93657053232193 val-loss: 6.699303524834769
epoch: 46/100 iteration: 1250 train-Loss: 1.7964438819885253 val-loss: 6.729496989931379
epoch: 48/100 iteration: 1300 train-Loss: 1.673508574962616 val-loss: 6.874975170407977
epoch: 49/100 iteration: 1350 train-Loss: 1.5340999484062194 val-loss: 6.849890300205776
epoch: 51/100 iteration: 1400 train-Loss: 1.4362503361701966 val-loss: 6.859814030783517
epoch: 53/100 iteration: 1450 train-Loss: 1.3218860363960265 val-loss: 6.917069605418614
epoch: 55/100 iteration: 1500 train-Loss: 1.2195318639278412 val-loss: 7.0314861706324985
epoch: 57/100 iteration: 1550 train-Loss: 1.1338822782039641 val-loss: 7.061163663864136
epoch: 59/100 iteration: 1600 train-Loss: 1.0726036953926086 val-loss: 7.087924923215594
epoch: 61/100 iteration: 1650 train-Loss: 0.9824717319011689 val-loss: 7.172810316085815
epoch: 62/100 iteration: 1700 train-Loss: 0.863843297958374 val-loss: 7.27536290032523
epoch: 64/100 iteration: 1750 train-Loss: 0.8114012372493744 val-loss: 7.347988162721906
epoch: 66/100 iteration: 1800 train-Loss: 0.7639252924919129 val-loss: 7.370683976582119
epoch: 68/100 iteration: 1850 train-Loss: 0.6960610103607178 val-loss: 7.561197485242571
epoch: 70/100 iteration: 1900 train-Loss: 0.6293037277460098 val-loss: 7.5093508788517545
epoch: 72/100 iteration: 1950 train-Loss: 0.5941761046648025 val-loss: 7.615212747028896
epoch: 74/100 iteration: 2000 train-Loss: 0.5411212784051895 val-loss: 7.633127587182181
epoch: 75/100 iteration: 2050 train-Loss: 0.4647019040584564 val-loss: 7.703417164938791
epoch: 77/100 iteration: 2100 train-loss: 0.45127547800540924 val-loss: 7.79028194291251
```

```
epoch: 79/100 iteration: 2150 train-Loss: 0.4182128345966339 val-loss: 7.872295958655221
     epoch: 81/100 iteration: 2200 train-Loss: 0.39653640270233154 val-loss: 7.848396914345877
     epoch: 83/100 iteration: 2250 train-Loss: 0.3563388305902481 val-loss: 7.961728743144444
     epoch: 85/100 iteration: 2300 train-Loss: 0.3317793944478035 val-loss: 8.088321685791016
     epoch: 87/100 iteration: 2350 train-Loss: 0.30687987685203555 val-loss: 8.06182759148734
    epoch: 88/100 iteration: 2400 train-Loss: 0.26878986418247225 val-loss: 8.260288204465594
     epoch: 90/100 iteration: 2450 train-Loss: 0.2660393610596657 val-loss: 8.140572377613612
     epoch: 92/100 iteration: 2500 train-Loss: 0.24605757534503936 val-loss: 8.19466849735805
     epoch: 94/100 iteration: 2550 train-Loss: 0.23134314984083176 val-loss: 8.291963849748884
    epoch: 96/100 iteration: 2600 train-Loss: 0.21438774347305298 val-loss: 8.324693100793022
     epoch: 98/100 iteration: 2650 train-Loss: 0.2102984294295311 val-loss: 8.363749299730573
     epoch: 99/100 iteration: 2700 train-Loss: 0.18215494185686112 val-loss: 8.353010518210274
## evaluation
## DON'T CHANGE THIS CELL IN ANY WAY
dummy_loss = evaluate(device, net, n_token, 10, 64, X_test, y_test)
assert np.exp(dummy loss) < 1e4, 'your dummy ppl is too large!'</pre>
print("Good job! It's almost done..")
    Good job! It's almost done..
epoch count = range(1, len(train history) + 1)
```

```
epoch_count = range(1, len(train_history) + 1)

# Visualize loss history
plt.plot(epoch_count, train_history, 'r--')
plt.plot(epoch_count, valid_history, 'b-')
plt.legend(['train', 'valid'])
plt.xlabel('iteration')
plt.ylabel('loss')
plt.show()
```





What you are experiencing is known as **overfitting** since the training loss is really small, but the validation error of the model is high. This is due to the model learning "too much" from the training dataset. You probably guess why this happened? We are still training the model on a subsample of the data (dummy_text)! If you train the model on the whole dataset, you will definitely get better validation scores.

Question

- Give at least 6 ideas on how you could make your neural language model better (short, bullet-style answers). (You can find inspiration online, for example here: https://arxiv.org/pdf/1708.02182.pdf)
- 1. Dropout (using same dropout mask over multiple time steps)
- 2. (Recurrent) Batch Normalization
- 3. Limiting updates to the RNN's hidden state
- 4. Restrictions on the recurrent matrices (capacity or elementwise)
- 5. Randomized-length backpropagation through time (BPTT)
- 6. DropConnect
- 7. Weight tying

If you had a lot more time (not within the scope of this lab!): you've learned the building blocks of neural language models, you can now build the ultimate monster:

- Weight tying: Two weight matrices have been used for input or output respectively (https://arxiv.org/abs/1608.05859)
- Make it char level or maybe use sub-word units like bpe;
- Use both char-level and word level features to train a word-level language model

Question

• Please give us a rough estimation of the hours you invested to complete this session? This will not affect your grade;) but it might help us with the design of our lab sessions for this new NLP course.

7 hours

Acknowledgment

If you received help or feedback from fellow students, please acknowledge that here. We count on your academic honesty:

I did not receive any help, I made the whole lab by myself and using the pytorch documentation or solutions from StackOverflow.