

Физтех-Школа Прикладной математики и информатики (ФПМИ) МФТИ

Some parts of the notebook are almost the exact copy of https://github.com/yandexdataschool/nlp_course

Attention

Attention layer can take in the previous hidden state of the decoder s_{t-1} , and all of the stacked forward and backward hidden states H from the encoder. The layer will output an attention vector a_t , that is the length of the source sentence, each element is between 0 and 1 and the

entire vector sums to 1.

Intuitively, this layer takes what we have decoded so far s_{t-1} , and all of what we have encoded H, to produce a vector a_t , that represents which words in the source sentence we should pay the most attention to in order to correctly predict the next word to decode \hat{y}_{t+1} . The decoder input word that has been embedded y_t .

You can use any type of the attention scores between previous hidden state of the encoder s_{t-1} and hidden state of the decoder $h \in H$, you prefer. We have met at least three of them:

$$ext{score}(m{h}, m{s}_{t-1}) = egin{cases} m{h}^ op m{s}_{t-1} & ext{dot} \ m{h}^ op m{W}_{m{a}} m{s}_{t-1} & ext{general} \ m{v}_a^ op anh(m{W}_{m{a}} \left[m{h}; m{s}_{t-1}
ight]) & ext{concat} \end{cases}$$

We wil use "concat attention":

First, we calculate the *energy* between the previous decoder hidden state s_{t-1} and the encoder hidden states H. As our encoder hidden states H are a sequence of T tensors, and our previous decoder hidden state s_{t-1} is a single tensor, the first thing we do is repeat the previous decoder hidden state T times. \Rightarrow

We have:

$$H = egin{bmatrix} m{h}_0, \dots, m{h}_{T-1} \end{bmatrix} \ egin{bmatrix} m{s}_{t-1}, \dots, m{s}_{t-1} \end{bmatrix}$$

The encoder hidden dim and the decoder hidden dim should be equal: **dec hid dim = enc hid dim**.

We then calculate the energy, E_t , between them by concatenating them together:

$$ig[[oldsymbol{h}_0,oldsymbol{s}_{t-1}],\ldots,[oldsymbol{h}_{T-1},oldsymbol{s}_{t-1}]ig]$$

And passing them through a linear layer (attn = W_a) and a anh activation function:

$$E_t = \tanh(\operatorname{attn}(H, s_{t-1}))$$

This can be thought of as calculating how well each encoder hidden state "matches" the previous decoder hidden state.

We currently have a **[enc hid dim, src sent len]** tensor for each example in the batch. We want this to be **[src sent len]** for each example in the batch as the attention should be over the length of the source sentence. This is achieved by multiplying the energy by a **[1, enc hid dim]** tensor, v.

$$\hat{a}_t = v E_t$$

We can think of this as calculating a weighted sum of the "match" over all enc_hid_dem elements for each encoder hidden state, where the weights are learned (as we learn the parameters of v).

Finally, we ensure the attention vector fits the constraints of having all elements between 0 and 1 and the vector summing to 1 by passing it through a softmax layer.

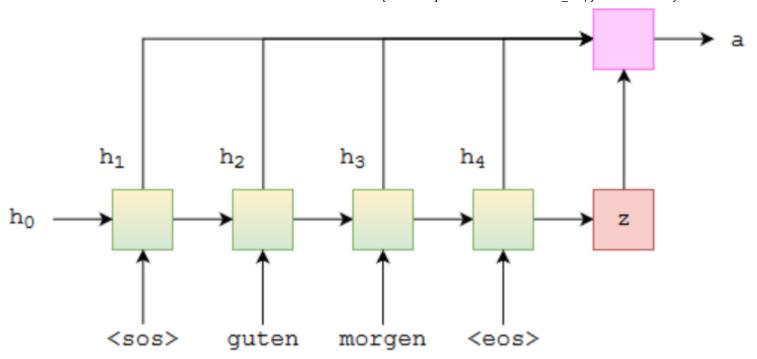
$$a_t = \operatorname{softmax}(\hat{a_t})$$

▼ Temperature SoftMax

$$\operatorname{softmax}(x)_i = \frac{e^{\frac{y_i}{T}}}{\sum_{j}^{N} e^{\frac{y_j}{T}}}$$

This gives us the attention over the source sentence!

Graphically, this looks something like below. $z=s_{t-1}$. The green/yellow blocks represent the hidden states from both the forward and backward RNNs, and the attention computation is all done within the pink block.



Neural Machine Translation

Write down some summary on your experiments and illustrate it with convergence plots/metrics and your thoughts. Just like you would approach a real problem.

```
! wget https://drive.google.com/uc?id=1NWYqJgeG_4883LINdEjKUr6nLQPY6Yb_ -0 data.txt
# Thanks to YSDA NLP course team for the data
# (who thanks tilda and deephack teams for the data in their turn)

--2021-04-08 07:56:29-- https://drive.google.com/uc?id=1NWYqJgeG_4883LINdEjKUr6nLQPY6Yb_
Resolving drive.google.com (drive.google.com)... 74.125.20.102, 74.125.20.100, 74.125.20.113, ...
Connecting to drive.google.com (drive.google.com)|74.125.20.102|:443... connected.
HTTP request sent, awaiting response... 302 Moved Temporarily
```

```
Location: https://doc-14-00-docs.googleusercontent.com/docs/securesc/ha0ro937gcuc717deffksulhg5h7mbp1/gnp8ilp3gfk0odo98n00mt686
     Warning: wildcards not supported in HTTP.
     --2021-04-08 07:56:30-- https://doc-14-00-docs.googleusercontent.com/docs/securesc/ha0ro937gcuc717deffksulhg5h7mbp1/gnp8ilp3gf
     Resolving doc-14-00-docs.googleusercontent.com (doc-14-00-docs.googleusercontent.com)... 74.125.195.132, 2607:f8b0:400e:c09::84
     Connecting to doc-14-00-docs.googleusercontent.com (doc-14-00-docs.googleusercontent.com) 74.125.195.132 443... connected.
     HTTP request sent, awaiting response... 200 OK
     Length: unspecified [text/plain]
     Saving to: 'data.txt'
     data.txt
                             <=>
                                                  1 12.31M 35.7MB/s
                                                                         in 0.3s
     2021-04-08 07:56:31 (35.7 MB/s) - 'data.txt' saved [12905334]
import torch
import torch.nn as nn
import torch.optim as optim
import torchtext
from torchtext.legacy.data import Field, BucketIterator
import spacy
import random
import math
import time
import numpy as np
import matplotlib
matplotlib.rcParams.update({'figure.figsize': (16, 12), 'font.size': 14})
import matplotlib.pyplot as plt
%matplotlib inline
from IPython.display import clear output
from nltk.tokenize import WordPunctTokenizer
```

We'll set the random seeds for deterministic results.

```
SEED = 1234

random.seed(SEED)
np.random.seed(SEED)
torch.manual_seed(SEED)
torch.cuda.manual_seed(SEED)
torch.backends.cudnn.deterministic = True
```

▼ Preparing Data

Here comes the preprocessing

```
tokenizer W = WordPunctTokenizer()
def tokenize ru(x, tokenizer=tokenizer W):
    return tokenizer.tokenize(x.lower())[::-1]
def tokenize_en(x, tokenizer=tokenizer_W):
    return tokenizer.tokenize(x.lower())
SRC = Field(tokenize=tokenize ru,
            init token = '<sos>',
            eos_token = '<eos>',
            lower = True)
TRG = Field(tokenize=tokenize en,
            init_token = '<sos>',
            eos_token = '<eos>',
            lower = True)
dataset = torchtext.legacy.data.TabularDataset(
```

```
patii= uata.txt ,
    format='tsv',
    fields=[('trg', TRG), ('src', SRC)]
print(len(dataset.examples))
print(dataset.examples[0].src)
print(dataset.examples[0].trg)
     50000
     ['.', 'собора', 'троицкого', '-', 'свято', 'от', 'ходьбы', 'минутах', '3', 'в', ',', 'тбилиси', 'в', 'расположен', 'cordelia',
     ['cordelia', 'hotel', 'is', 'situated', 'in', 'tbilisi', ',', 'a', '3', '-', 'minute', 'walk', 'away', 'from', 'saint', 'trinit
train data, valid data, test data = dataset.split(split ratio=[0.8, 0.15, 0.05])
print(f"Number of training examples: {len(train data.examples)}")
print(f"Number of validation examples: {len(valid data.examples)}")
print(f"Number of testing examples: {len(test data.examples)}")
     Number of training examples: 40000
     Number of validation examples: 2500
     Number of testing examples: 7500
SRC.build vocab(train data, min freq = 2)
TRG.build vocab(train data, min freq = 2)
print(f"Unique tokens in source (ru) vocabulary: {len(SRC.vocab)}")
print(f"Unique tokens in target (en) vocabulary: {len(TRG.vocab)}")
     Unique tokens in source (ru) vocabulary: 14129
     Unique tokens in target (en) vocabulary: 10104
```

And here is example from train dataset:

When we get a batch of examples using an iterator we need to make sure that all of the source sentences are padded to the same length, the same with the target sentences. Luckily, TorchText iterators handle this for us!

We use a BucketIterator instead of the standard Iterator as it creates batches in such a way that it minimizes the amount of padding in both the source and target sentences.

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

    device(type='cuda')

def _len_sort_key(x):
    return len(x.src)

BATCH_SIZE = 128

train_iterator, valid_iterator, test_iterator = BucketIterator.splits(
        (train_data, valid_data, test_data),
        batch_size = BATCH_SIZE,
        device = device,
        sort_key=_len_sort_key
)
```

▼ Let's use modules.py

```
# from google.colab import drive
# drive.mount('/content/drive')
```

- # !ls /content/drive/MyDrive/Colab_Notebooks/DLSchool_mipt_2sem
- # %cd /content/drive/MyDrive/Colab_Notebooks/DLSchool_mipt_2sem

▼ Encoder

For a multi-layer RNN, the input sentence, X, goes into the first (bottom) layer of the RNN and hidden states, $H = \{h_1, h_2, \dots, h_T\}$, output by this layer are used as inputs to the RNN in the layer above. Thus, representing each layer with a superscript, the hidden states in the first layer are given by:

$$h_t^1 = \operatorname{EncoderRNN}^1(x_t, h_{t-1}^1)$$

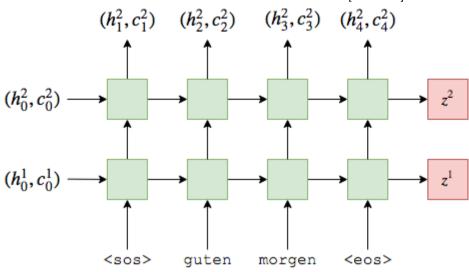
The hidden states in the second layer are given by:

$$h_t^2 = \text{EncoderRNN}^2(h_t^1, h_{t-1}^2)$$

Extending our multi-layer equations to LSTMs, we get:

$$(h_t^1,c_t^1) = \operatorname{EncoderLSTM}^1(x_t,(h_{t-1}^1,c_{t-1}^1))$$

$$(h_t^2, c_t^2) = \mathrm{EncoderLSTM}^2(h_t^1, (h_{t-1}^2, c_{t-1}^2))$$



```
import random
import torch
from torch import nn
from torch.nn import functional as F
# you can paste code of encoder from modules.py
# the encoder can be like seminar encoder but you have to return outputs
# and if you use bidirectional you won't make the same operation like with hidden
# because outputs = [src sent len, batch size, hid dim * n directions]
class Encoder(nn.Module):
    def __init__(self, input_dim, emb_dim, hid_dim, n_layers, dropout, bidirectional):
        super(). init ()
        self.input dim = input dim
        self.emb_dim = emb_dim
        self.hid_dim = hid_dim
        self.n_layers = n_layers
        self.dropout = dropout
        self.bidirectional = bidirectional
```

```
self.embedding = nn.Embedding(input dim, emb dim)
   self.rnn = nn.LSTM(emb dim, hid dim, num layers=n layers, dropout=dropout, bidirectional=bidirectional)
    self.dropout = nn.Dropout(p=dropout)
def forward(self, src):
    #src = [src sent len, batch size]
    # Compute an embedding from the src data and apply dropout to it
   embedded = self.dropout(self.embedding(src))
   #embedded = [src sent len, batch size, emb dim]
   # Compute the RNN output values of the encoder RNN.
    # outputs, hidden and cell should be initialized here. Refer to nn.LSTM docs ;)
   outputs, (hidden, cell) = self.rnn(embedded)
    hidden = torch.tanh(hidden) # !!
   #outputs = [src sent len, batch size, hid dim * n directions]
   #hidden = [n layers * n directions, batch size, hid dim]
   #cell = [n layers * n directions, batch size, hid dim]
   # #outputs are always from the top hidden layer
   if self.bidirectional:
       hidden = hidden.reshape(self.n layers, 2, -1, self.hid dim)
       hidden = hidden.transpose(1, 2).reshape(self.n layers, -1, 2 * self.hid dim)
        cell = cell.reshape(self.n layers, 2, -1, self.hid dim)
        cell = cell.transpose(1, 2).reshape(self.n layers, -1, 2 * self.hid dim)
    return outputs, hidden, cell
```

Attention

```
\operatorname{score}(\boldsymbol{h}, \boldsymbol{s}_{t-1}) = \boldsymbol{v}_{-1}^{\top} \tanh(\boldsymbol{W}_{-1}[\boldsymbol{h}, \boldsymbol{s}_{t-1}]) - concat attention
# you can paste code of attention from modules.py
def softmax(x, temperature=10): # use your temperature
    e x = torch.exp(x / temperature)
    return e x / torch.sum(e x, dim=0)
class Attention(nn.Module):
    def init (self, enc hid dim, dec hid dim, temperature=10):
         super(). init ()
         self.enc hid dim = enc hid dim
         self.dec hid dim = dec hid dim
         self.temperature = temperature
         self.attn = nn.Linear(enc hid dim + dec hid dim, enc hid dim)
         self.v = nn.Linear(enc hid dim, 1, bias=False)
    def forward(self, hidden, encoder outputs):
         # encoder outputs = [src sent len, batch size, enc hid dim]
         # hidden = [1, batch size, dec hid dim]
         # repeat hidden and concatenate it with encoder outputs
         '''your code'''
         hidden = hidden[-1].repeat(encoder outputs.shape[0], 1, 1)
         # print("hidden.shape", hidden.shape)
         # print("encoder outputs.shape", encoder outputs.shape)
         catted = torch.cat((hidden, encoder outputs), dim=2)
         # calculate energy
         '''your code'''
         # print("hidden.shape", hidden.shape)
         # print("encoder_outputs.shape", encoder_outputs.shape)
         # print("catted.shape", catted.shape)
         # print(self.attn.in features)
```

```
energy = torch.tanh(self.attn(catted)) # [src sent len, batch size, enc_hid_dim]

# get attention, use softmax function which is defined, can change temperature
'''your code'''

attention = self.v(energy) # [src sent len, batch size, 1]

attention = softmax(attention, self.temperature).permute(1, 0, 2) # [batch size, src sent len, 1]

return attention # '''your code'''
```

Decoder with Attention

To make it really work you should also change the Decoder class from the classwork in order to make it to use Attention. You may just copypaste Decoder class and add several lines of code to it.

The decoder contains the attention layer attention, which takes the previous hidden state s_{t-1} , all of the encoder hidden states H, and returns the attention vector a_t .

We then use this attention vector to create a weighted source vector, w_t , denoted by weighted, which is a weighted sum of the encoder hidden states, H, using a_t as the weights.

$$w_t = a_t H$$

The input word that has been embedded y_t , the weighted source vector w_t , and the previous decoder hidden state s_{t-1} , are then all passed into the decoder RNN, with y_t and w_t being concatenated together.

$$s_t = \operatorname{DecoderGRU}([y_t, w_t], s_{t-1})$$

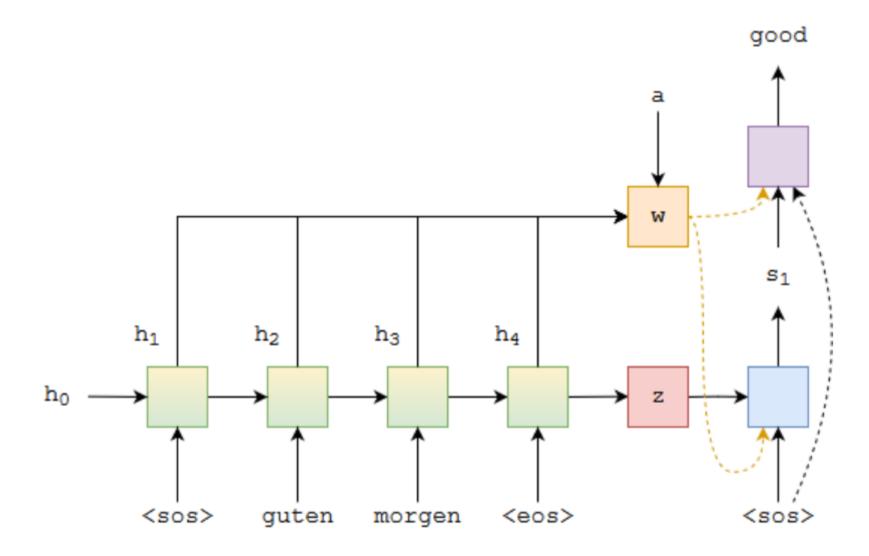
We then pass y_t , w_t and s_t through the linear layer, f, to make a prediction of the next word in the target sentence, \hat{y}_{t+1} . This is done by concatenating them all together.

$$\hat{y}_{t+1} = f(y_t, w_t, s_t)$$

The image below shows decoding the **first** word in an example translation.

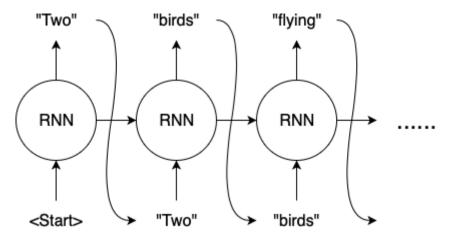
The green/yellow blocks show the forward/backward encoder RNNs which output H, the red block is $z=s_{t-1}=s_0$, the blue block shows the decoder RNN which outputs \hat{y}_{t+1} and the orange block shows the

calculation of the weighted sum over H by a_t and outputs $w_t.$ Not shown is the calculation of $a_t.$

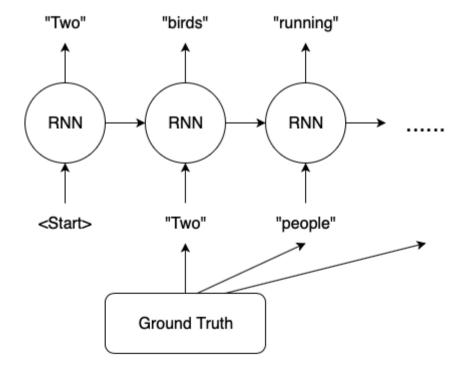


▼ Teacher forcing

Teacher forcing is a method for quickly and efficiently training recurrent neural network models that use the ground truth from a prior time step



Without Teacher Forcing



With Teacher Forcing

When training/testing our model, we always know how many words are in our target sentence, so we stop generating words once we hit that many. During inference (i.e. real world usage) it is common to keep generating words until the model outputs an <eos> token or after a certain amount of words have been generated.

Once we have our predicted target sentence, $\hat{Y} = \{\hat{y}_1, \hat{y}_2, \dots, \hat{y}_T\}$, we compare it against our actual target sentence, $Y = \{y_1, y_2, \dots, y_T\}$, to calculate our loss. We then use this loss to update all of the parameters in our model.

```
# you can paste code of decoder from modules.py
class DecoderWithAttention(nn.Module):
    def init (self, output dim, emb dim, enc hid dim, dec hid dim, dropout, attention):
        super().__init__()
        self.emb dim = emb dim
        self.enc hid dim = enc hid dim
        self.dec hid dim = dec hid dim
        self.output dim = output dim
        self.attention = attention
        self.embedding = nn.Embedding(output dim, emb dim) # '''your code'''
        self.rnn = nn.GRU(emb dim + enc hid dim, dec hid dim, dropout=dropout) #'''your code''' # use GRU
        self.out = nn.Linear(dec_hid_dim + emb_dim + dec_hid_dim, output_dim) # '''your code''' # linear layer to get next word
        self.dropout = nn.Dropout(dropout)
    def forward(self, input, hidden, encoder_outputs):
        #input = [batch size]
        #hidden = [n layers * n directions, batch size, hid dim]
        #n directions in the decoder will both always be 1, therefore:
        #hidden = [n layers, batch size, hid dim]
```

```
input = input.unsqueeze(0) # because only one word, no words sequence
#input = [1, batch size]
embedded = self.dropout(self.embedding(input))
#embedded = [1, batch size, emb dim]
# get weighted sum of encoder outputs
#outputs = [src sent len, batch size, hid dim * n directions]
#attention = [batch size, src sent len, 1]
'''vour code'''
# print("self.attention(hidden, encoder outputs)", self.attention(hidden, encoder outputs).shape)
# print("encoder outputs.permute(1, 2, 0)", encoder_outputs.permute(1, 2, 0).shape)
hidden = hidden[-1].unsqueeze(0)
weights = self.attention(hidden, encoder outputs)
context = torch.bmm(weights.permute(0, 2, 1), encoder outputs.permute(1, 0, 2)).permute(1, 0, 2) # [1, batch size, hid dim '
# concatenate weighted sum and embedded, break through the GRU
'''vour code'''
rnn_input = torch.cat([embedded, context], dim=2) # [1, B, E+H]
# print("rnn input.shape", rnn input.shape)
# print("hidden.shape", hidden.shape)
output, hidden = self.rnn(rnn input, hidden)
# get predictions
'''your code'''
prediction = self.out(torch.cat((output.squeeze(0), context.squeeze(0), embedded.squeeze(0)), dim=1))
#prediction = [batch size, output dim]
return prediction, hidden # '''your code'''
```

▼ Seq2Seq

Main idea:

- $w_t = a_t H$
- $s_t = \text{DecoderGRU}([y_t, w_t], s_{t-1})$
- $\bullet \ \ \hat{y}_{t+1} = f(y_t, w_t, s_t)$

Note: our decoder loop starts at 1, not 0. This means the 0th element of our outputs tensor remains all zeros. So our trg and outputs look something like:

$$ext{trg} = [< sos >, y_1, y_2, y_3, < eos >] \ ext{outputs} = [0, \hat{y}_1, \hat{y}_2, \hat{y}_3, < eos >]$$

Later on when we calculate the loss, we cut off the first element of each tensor to get:

$$\operatorname{trg} = [y_1, y_2, y_3, < eos >]$$

```
# you can paste code of seq2seq from modules.py

class Seq2Seq(nn.Module):
    def __init__(self, encoder, decoder, device):
        super().__init__()

    self.encoder = encoder
        self.decoder = decoder
        self.device = device

    assert encoder.hid_dim * 2 == decoder.dec_hid_dim, \
         "Hidden dimensions of encoder and decoder must be equal!"

def forward(self, src, trg, teacher_forcing_ratio = 0.5):

# src = [src sent len, batch size]
    # trg = [trg sent len, batch size]
    # teacher forcing ratio is probability to use teacher forcing
```

```
# e.g. if teacher forcing ratio is 0.75 we use ground-truth inputs 75% of the time
# Again, now batch is the first dimention instead of zero
batch size = trg.shape[1]
trg len = trg.shape[0]
trg vocab size = self.decoder.output dim
#tensor to store decoder outputs
outputs = torch.zeros(trg len, batch_size, trg_vocab_size).to(self.device)
#last hidden state of the encoder is used as the initial hidden state of the decoder
enc states, hidden, cell = self.encoder(src)
#first input to the decoder is the <sos> tokens
input = trg[0, :]
for t in range(1, trg len):
    '''vour code'''
   output, hidden = self.decoder(input, hidden, enc states) # !!
   outputs[t] = output
   #decide if we are going to use teacher forcing or not
   teacher force = random.random() < teacher forcing ratio</pre>
   #get the highest predicted token from our predictions
   top1 = output.argmax(-1)
   #if teacher forcing, use actual next token as next input
   #if not, use predicted token
   input = trg[t] if teacher force else top1
return outputs
```

Training

For reloading

```
# import modules
# import imp
# imp.reload(modules)
# Encoder = modules.Encoder
# Attention = modules.Attention
# Decoder = modules.DecoderWithAttention
# Seq2Seq = modules.Seq2Seq
INPUT DIM = len(SRC.vocab)
OUTPUT DIM = len(TRG.vocab)
ENC EMB DIM = 256
DEC EMB DIM = 256
HID DIM = 512
N LAYERS = 1 # simple model: n layers=1
ENC DROPOUT = 0.5
DEC DROPOUT = 0.5
BIDIRECTIONAL = True
TEMPERATURE = 10
enc = Encoder(INPUT_DIM, ENC_EMB_DIM, HID_DIM // 2, N_LAYERS, ENC_DROPOUT, BIDIRECTIONAL)
attention = Attention(HID DIM, HID DIM, TEMPERATURE)
dec = DecoderWithAttention(OUTPUT DIM, DEC EMB DIM, HID DIM, HID DIM, DEC DROPOUT, attention)
# dont forget to put the model to the right device
model = Seq2Seq(enc, dec, device).to(device)
     /usr/local/lib/python3.7/dist-packages/torch/nn/modules/rnn.py:63: UserWarning: dropout option adds dropout after all but last
       "num layers={}".format(dropout, num layers))
def init_weights(m):
    for name, param in m.named_parameters():
        nn.init.uniform (param, -0.08, 0.08)
```

```
model.apply(init weights)
     Seq2Seq(
       (encoder): Encoder(
         (embedding): Embedding(14129, 256)
         (rnn): LSTM(256, 256, dropout=0.5, bidirectional=True)
         (dropout): Dropout(p=0.5, inplace=False)
       (decoder): DecoderWithAttention(
         (attention): Attention(
           (attn): Linear(in features=1024, out features=512, bias=True)
           (v): Linear(in features=512, out features=1, bias=False)
         (embedding): Embedding(10104, 256)
         (rnn): GRU(768, 512, dropout=0.5)
         (out): Linear(in features=1280, out_features=10104, bias=True)
         (dropout): Dropout(p=0.5, inplace=False)
def count parameters(model):
    return sum(p.numel() for p in model.parameters() if p.requires grad)
print(f'The model has {count parameters(model):,} trainable parameters')
     The model has 22,694,008 trainable parameters
from torch.optim.lr scheduler import StepLR
PAD IDX = TRG.vocab.stoi['<pad>']
optimizer = optim.Adam(model.parameters())
scheduler = StepLR(optimizer, step size=5, gamma=0.5)
criterion = nn.CrossEntropyLoss(ignore index = PAD IDX)
def train(model, iterator, optimizer, criterion, clip, train_history=None, valid_history=None):
    model.train()
    epoch_loss = 0
    history = []
```

```
HERE L
for i, batch in enumerate(iterator):
    src = batch.src
    trg = batch.trg
    optimizer.zero grad()
    output = model(src, trg)
    #trg = [trg sent len, batch size]
    #output = [trg sent len, batch size, output dim]
    output = output[1:].view(-1, OUTPUT DIM)
    trg = trg[1:].view(-1)
    #trg = [(trg sent len - 1) * batch size]
    #output = [(trg sent len - 1) * batch size, output dim]
    loss = criterion(output, trg)
    loss.backward()
    # Let's clip the gradient
    torch.nn.utils.clip grad norm (model.parameters(), clip)
    optimizer.step()
    epoch loss += loss.item()
    history.append(loss.cpu().data.numpy())
    if (i+1)\%10==0:
        fig, ax = plt.subplots(nrows=1, ncols=2, figsize=(12, 8))
        clear_output(True)
        ax[0].plot(history, label='train loss')
        ax[0].set_xlabel('Batch')
        ax[0].set_title('Train loss')
```

```
it train history is not None:
                ax[1].plot(train_history, label='general train history')
                ax[1].set xlabel('Epoch')
            if valid history is not None:
                ax[1].plot(valid history, label='general valid history')
            plt.legend()
            plt.show()
    return epoch loss / len(iterator)
def evaluate(model, iterator, criterion):
    model.eval()
    epoch loss = 0
   history = []
    with torch.no grad():
        for i, batch in enumerate(iterator):
            src = batch.src
            trg = batch.trg
            output = model(src, trg, 0) #turn off teacher forcing
            #trg = [trg sent len, batch size]
            #output = [trg sent len, batch size, output dim]
            output = output[1:].view(-1, OUTPUT DIM)
            trg = trg[1:].view(-1)
            #trg = [(trg sent len - 1) * batch size]
            #output = [(trg sent len - 1) * batch size, output dim]
```

```
loss = criterion(output, trg)
            epoch loss += loss.item()
    return epoch loss / len(iterator)
def epoch_time(start_time, end_time):
    elapsed time = end time - start time
    elapsed mins = int(elapsed time / 60)
    elapsed secs = int(elapsed time - (elapsed mins * 60))
    return elapsed mins, elapsed secs
import matplotlib
matplotlib.rcParams.update({'figure.figsize': (16, 12), 'font.size': 14})
import matplotlib.pyplot as plt
%matplotlib inline
from IPython.display import clear output
train history = []
valid history = []
N EPOCHS = 12
CLIP = 5
best_valid_loss = float('inf')
for epoch in range(N EPOCHS):
    start_time = time.time()
    train loss = train(model, train iterator, optimizer, criterion, CLIP, train history, valid history)
    valid loss = evaluate(model, valid iterator, criterion)
    # scheduler.step()
    end_time = time.time()
```

```
epoch_mins, epoch_secs = epoch_time(start_time, end_time)

if valid_loss < best_valid_loss:
    best_valid_loss = valid_loss
    torch.save(model.state_dict(), 'best-val-model.pt')

train_history.append(train_loss)
valid_history.append(valid_loss)
print(f'Epoch: {epoch+1:02} | Time: {epoch_mins}m {epoch_secs}s')
print(f'\tTrain Loss: {train_loss:.3f} | Train PPL: {math.exp(train_loss):7.3f}')
print(f'\t Val. Loss: {valid_loss:.3f} | Val. PPL: {math.exp(valid_loss):7.3f}')</pre>
```

```
Train loss
                                                                                  general train history
                                                                                  general valid history
      2.6
Let's take a look at our network quality:
                            def cut on eos(tokens iter):
    for token in tokens iter:
        if token == '<eos>':
            break
        vield token
def remove tech tokens(tokens iter, tokens to remove=['<sos>', '<unk>', '<pad>']):
    return [x for x in tokens iter if x not in tokens to remove]
def generate translation(src, trg, model, TRG vocab):
    model.eval()
    output = model(src, trg, 0) #turn off teacher forcing
    output = output[1:].argmax(-1)
    original = remove tech tokens(cut on eos([TRG vocab.itos[x] for x in list(trg[:,0].cpu().numpy())]))
    generated = remove tech tokens(cut on eos([TRG vocab.itos[x] for x in list(output[:, 0].cpu().numpy())]))
    print('Original: {}'.format(' '.join(original)))
    print('Generated: {}'.format(' '.join(generated)))
    print()
def get text(x, TRG vocab):
     generated = remove tech tokens(cut on eos([TRG vocab.itos[elem] for elem in list(x)]))
     return generated
#model.load_state_dict(torch.load('best-val-model.pt'))
batch = next(iter(test_iterator))
```

```
for idx in range(10):
    src = batch.src[:, idx:idx+1]
    trg = batch.trg[:, idx:idx+1]
    generate translation(src, trg, model, TRG.vocab)
     Original: there is a 24 - hour front desk at the property .
     Generated: you will find a 24 - hour front desk at the property .
     Original: you will find a 24 - hour front desk at the property.
     Generated: you will find a 24 - hour front desk at the property .
     Original: there is a 24 - hour front desk at the property .
     Generated: you will find a 24 - hour front desk at the property .
     Original: free private parking is available.
     Generated: free private parking is available on site .
     Original: there are several restaurants in the surrounding area .
     Generated: restaurants can be found nearby .
     Original: the property also offers free parking .
     Generated: free parking is available .
     Original: the unit is fitted with a kitchen .
     Generated: the unit is equipped with a kitchen .
     Original: the bathroom has a shower .
     Generated: the bathroom comes with a shower .
     Original: there is also a fireplace in the living room .
     Generated: there is a living fireplace .
     Original: you will find a coffee machine in the room .
     Generated: you will find a coffee machine in the room .
```

▼ Bleu

link

$$BP = \begin{cases} 1 & \text{if } c > r \\ e^{(1-r/c)} & \text{if } c \le r \end{cases}.$$

Then,

BLEU= BP · exp
$$\left(\sum_{n=1}^{N} w_n \log p_n\right)$$
.

The ranking behavior is more immediately apparent in the log domain,

log BLEU = min
$$(1 - \frac{r}{c}, 0) + \sum_{n=1}^{N} w_n \log p_n$$
.

In our baseline, we use N = 4 and uniform weights $w_n = 1/N$.

from nltk.translate.bleu_score import corpus_bleu

```
# """ Estimates corpora-level BLEU score of model's translations given inp and reference out """
# translations, _ = model.translate_lines(inp_lines, **flags)
# Note: if you experience out-of-memory error, split input lines into batches and translate separately
# return corpus_bleu([[ref] for ref in out_lines], translations) * 100

import tqdm
original_text = []
generated_text = []
model.eval()
with torch.no_grad():
```

```
for i, batch in tqdm.tqdm(enumerate(test iterator)):
        src = batch.src
        trg = batch.trg
        output = model(src, trg, 0) #turn off teacher forcing
        #trg = [trg sent len, batch size]
        #output = [trg sent len, batch size, output dim]
        output = output[1:].argmax(-1)
        original text.extend([get text(x, TRG.vocab) for x in trg.cpu().numpy().T])
        generated text.extend([get text(x, TRG.vocab) for x in output.detach().cpu().numpy().T])
# original text = flatten(original text)
# generated text = flatten(generated text)
     59it [00:06, 8.87it/s]
corpus bleu([[text] for text in original text], generated text) * 100
     29.565204851035727
```

Recommendations:

- · use bidirectional RNN
- change learning rate from epoch to epoch
- · when classifying the word don't forget about embedding and summa of encoders state
- you can use more than one layer

You will get:

- 2 points if 21 < bleu score < 23
- 4 points if 23 < bleu score < 25
- 7 points if 25 < bleu score < 27
- 9 points if 27 < bleu score < 29
- 10 points if bleu score > 29

When your result is checked, your 10 translations will be checked too

▼ Your Conclusion

- · information about your the results obtained
- difference between seminar and homework model

✓ 0 сек. выполнено в 11:28