### Introduction to Sequence Models

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### Shameless Advertising

This is a technical lecture that lays out mathematical foundations you need for a deep understanding of language models (of any size).

If you or someone you know wants a broad tutorial for thinking about language models, without math, take a look at "Language Models: A Guide for the Perplexed" (Serrano et al., 2023). There's also a talk video on my website. You're in the middle of writing an email or text message, and the system predicts your next ...

The heart of the language modeling task: what is the next word likely to be, given the preceding ones?

# Motivation II: Conversation

You ask a question, and the system responds with an answer, like in a dialogue with another person.

# Motivation III: Other Text-Output Applications

Other tasks that have text (or speech) as output:

- transcription of speech-audio to text
- translation from one language to another
- conversational systems
- document summarization
- image captioning
- optical character recognition
- spelling and grammar correction

If we're mapping inputs i to word sequences w, then:

$$m{w}^* = \operatorname*{argmax}_{m{w}} \operatorname{Faithfulness}(m{w};m{i}) + \operatorname{Fluency}(m{w})$$

Language models can provide a "fluency" score.

## Motivation IV: Science

If we have two theories about language,  $\boldsymbol{A}$  and  $\boldsymbol{B},$  and

 $\operatorname{Surprise}(A; \operatorname{Data}) < \operatorname{Surprise}(B; \operatorname{Data}),$ 

then A is the preferred theory.

Language models can give us a notion of "surprise."

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Always true:  

$$p(X = x, Y = y) = p(X = x \mid Y = y) \cdot p(Y = y)$$

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▶ Sometimes true:  $p(X = x, Y = y) = p(X = x) \cdot p(Y = y)$ 

The difference between true and estimated probability distributions

## Notation and Definitions

- V is a finite set of (discrete) symbols (words or characters);
  V = |V|
- $\blacktriangleright \ \mathcal{V}^*$  is the (infinite) set of sequences of symbols from  $\mathcal{V}$
- ► In language modeling, we imagine a sequence of random variables X<sub>1</sub>, X<sub>2</sub>,... that continues until some X<sub>n</sub> takes the value "○" (a special end-of-sequence symbol).
- V<sup>†</sup> is the (infinite) set of sequences of V symbols, with a single ○, which is at the end.

## The Language Modeling Problem

Input: training data  $oldsymbol{x}=\langle x_1,\ldots,x_N
angle$  in  $\mathcal{V}^\dagger$ 

Sometimes it's useful to consider a collection of observations, each in V<sup>†</sup>, but it complicates notation.

Output: a function  $p: \mathcal{V}^{\dagger} \to \mathbb{R}$ 

Think of p as a measure of plausibility.

### Questions to Answer

- 1. How do we quantitatively evaluate language models?
- 2. How do we build language models?
- 3. How do we use language models?

# Probabilistic Language Model

We let p be a probability distribution, which means that

$$orall oldsymbol{x} \in \mathcal{V}^{\dagger}, p(oldsymbol{x}) \geq 0$$
 $\sum_{oldsymbol{x} \in \mathcal{V}^{\dagger}} p(oldsymbol{x}) = 1$ 

Advantages:

- Interpretability
- We can apply the maximum likelihood principle to build a language model from data

# Maximum Likelihood Principle/Estimation

Let x be your observations (data).

If  $\mathcal{P}$  is the set of probability distributions that are consistent with your assumptions about the data, then the distribution you should choose is:

 $p_{\text{MLE}} = \operatorname*{argmax}_{p \in \mathcal{P}} p(\boldsymbol{x})$ 

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In practice, we usually let  ${\cal P}$  be a family of probabilistic models with parameters  ${\pmb \theta}$  and choose:

$$\boldsymbol{\theta}_{\mathrm{MLE}} = \operatorname*{argmax}_{\boldsymbol{\theta}} p(\boldsymbol{x}; \boldsymbol{\theta})$$

Let  ${\pmb x}$  be a sequence of N observed coin flips, i.e., drawn from  $\{h,t\}^+.$ 

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$$p(\boldsymbol{x}; \theta) = \prod_{i=1}^{N} \theta^{\mathbf{1}\{x_i=h\}} \cdot (1-\theta)^{\mathbf{1}\{x_i=t\}}$$
$$\theta_{\text{MLE}} = \underset{\theta \in [0,1]}{\operatorname{argmax}} p(\boldsymbol{x}; \theta)$$
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For binomial (and more generally, multinomial) event-based probabilistic models, the MLE equates to "count and normalize."

We should prefer a language model that is less "surprised" by new data that wasn't used to build it. This could be data that's drawn from the same distribution as the training data, or it could be from a different distribution.

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- 2. That value will be tiny, because  $\mathcal{V}^{\dagger}$  is infinitely large, and p will decrease exponentially in the length of  $\bar{x}$ . So we transform it:

$$\operatorname{Perplexity}(\bar{\boldsymbol{x}}; p(\cdot; \boldsymbol{\theta})) = \sqrt[\bar{N}]{\frac{1}{p(\bar{\boldsymbol{x}}; \boldsymbol{\theta})}}$$

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#### Special cases:

- lf the model were to put *all* of its probability on  $\bar{x}$ , perplexity would be 1 (minimal possible value).
- ► If the model assigns zero probability to x
  , perplexity is +∞. So it's important to make sure that p assigns strictly positive probability to *every* sequence of words.

You can interpret perplexity as "effective size of the vocabulary."

# Perplexity

- Warning: you can only directly compare perplexity of models that use exactly the same V.
- Perplexity on conventionally accepted test sets is often reported in papers.
- I won't discuss perplexity numbers, because:
  - Perplexity is only an intermediate measure of performance.
  - Understanding the models is more important than remembering how well they perform on specific train/test sets; your data will always be different!
- If you're curious, look up numbers in the literature; always take them with a grain of salt.
- New perplexity benchmark: Paloma (Magnusson et al., 2023)

## Reflection

We can also measure perplexity on the training data. Do you expect training perplexity to be lower (i.e., better) than test perplexity, or higher (i.e., worse)? Why?

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Vocabulary will be huge, because there are a *lot* of words, they all have many variants with different capitalization, punctuation, and mis)spellings. Also, some languages don't use whitespace!

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  - This might work, but it means your text will be heavily fragmented and your sequences will be extremely long. This isn't yet a practical solution for current models.

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Conventional solution: use a sample of data to learn a series of rules that start from bytes or characters and iteratively merge common sequences into larger units, up to some maximum vocabulary size (Sennrich et al., 2016; Wu et al., 2016). The most common technique is known as "byte pair encoding."

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This approach is "safe" in the sense that you can always fall back to byte-level tokenization.

#### Our Universe, For Now

We will focus on *probabilistic* language models with a fixed, finite vocabulary  $\mathcal{V}$ .

Training will start from the maximum likelihood principle.

Training data is  $\boldsymbol{x} = \langle x_1, \dots, x_N \rangle$  and we evaluate perplexity on test data  $\bar{\boldsymbol{x}} = \langle \bar{x}_1, \dots, \bar{x}_{\bar{N}} \rangle$ .

# A First Language Model

 $p(\boldsymbol{x}) \propto \operatorname{count}(\boldsymbol{x})$ 

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What if  $\bar{x}$  is not (in) the training data?

# A First Language Model

#### $p(\boldsymbol{x}) \propto \operatorname{count}(\boldsymbol{x})$

If we think of the training data as *multiple* sequences, the issue remains.

#### Using the Chain Rule

$$p(\mathbf{X} = \mathbf{x}) = \begin{pmatrix} p(X_1 = x_1) \\ \cdot p(X_2 = x_2 \mid X_1 = x_1) \\ \cdot p(X_3 = x_3 \mid \mathbf{X}_{1:2} = \mathbf{x}_{1:2}) \\ \vdots \\ \cdot p(X_N = \bigcirc \mid \mathbf{X}_{1:N-1} = \mathbf{x}_{1:N-1}) \end{pmatrix}$$
$$= \prod_{i=1}^N p(X_i = x_i \mid \mathbf{X}_{1:i-1} = \mathbf{x}_{1:i-1})$$

The game is to "summarize" the history well enough to predict each word in turn.

# Unigram Model: Empty History

$$p(\boldsymbol{X} = \boldsymbol{x}) = \prod_{i=1}^{N} p(X_i = x_i \mid \boldsymbol{X}_{1:i-1} = \boldsymbol{x}_{1:i-1})$$

$$\overset{\text{assumption}}{=} \prod_{i=1}^{N} p(X_i = x_i; \boldsymbol{\theta}) = \prod_{i=1}^{N} \theta_{x_i}$$

Maximum likelihood estimate: for every  $v \in \mathcal{V}$ ,

$$\theta_v^* = \frac{\sum_{i=1}^N \mathbf{1} \{x_i = v\}}{N}$$
$$= \frac{\operatorname{count}_{\boldsymbol{x}}(v)}{N}$$

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## Example

The probability of

Presidents tell lies .

is:

 $p(X_1 = \mathsf{Presidents}) \cdot p(X_2 = \mathsf{tell}) \cdot p(X_3 = \mathsf{lies}) \cdot p(X_4 = .) \cdot p(X_5 = \bigcirc)$ 

In unigram model notation:

$$\theta_{\mathsf{Presidents}} \cdot \theta_{\mathsf{tell}} \cdot \theta_{\mathsf{lies}} \cdot \theta_{\cdot} \cdot \theta_{\bigcirc}$$

Using the maximum likelihood estimate for  $\theta$ , we could calculate:

$$\frac{\operatorname{count}_{\boldsymbol{x}}(\mathsf{Presidents})}{N} \cdot \frac{\operatorname{count}_{\boldsymbol{x}}(\mathsf{tell})}{N} \cdots \frac{\operatorname{count}_{\boldsymbol{x}}(\bigcirc)}{N}$$

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#### Reflection

Consider a unigram model that is completely agnostic; it assigns  $\theta_v = \frac{1}{V}$  for all  $v \in \mathcal{V}$ .

What will its perplexity be? Hint: as long as the test data is restricted to words in  $\mathcal{V}$ , the test data doesn't matter!

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# Unigram Models: Assessment

#### Pros:

- Easy to understand
- Cheap
- Good enough for information retrieval (maybe)

#### Cons:

- Fixed, known vocabulary assumption
- "Bag of words" assumption is linguistically inaccurate
  - $\label{eq:product} \blacktriangleright \ p(\text{the the the the}) \gg \\ p(\text{I want ice cream})$

## Markov Models $\equiv$ n-gram Models

$$p(\boldsymbol{X} = \boldsymbol{x}) = \prod_{i=1}^{N} p(X_i = x_i \mid \boldsymbol{X}_{1:i-1} = \boldsymbol{x}_{1:i-1})$$

$$\stackrel{\text{assumption}}{=} \prod_{i=1}^{N} p(X_i = x_i \mid X_{i-n+1:i-1} = \boldsymbol{x}_{i-n+1:i-1}; \boldsymbol{\theta})$$

$$= \prod_{i=1}^{N} \theta_{x_i \mid \boldsymbol{x}_{i-n+1:i-1}}$$

(n-1)th-order Markov assumption  $\equiv$  n-gram model

- Unigram model is the n = 1 case
- In speech and translation systems, trigram models (n = 3) were widely used, then later 5-grams, ...

#### Reflection

# What is the maximum likelihood estimate for the n-gram model's probability of v given a (n - 1)-length history h?

### Solution

$$\theta_{v|h} = p(X_i = v \mid X_{i-n+1:i-1} = h)$$

$$= \frac{p(X_i = v, X_{i-n+1:i-1} = h)}{p(X_{i-n+1:i-1} = h)}$$

$$= \frac{\operatorname{count}_{\boldsymbol{x}}(\boldsymbol{h}v)}{N} / \frac{\operatorname{count}_{\boldsymbol{x}}(\boldsymbol{h})}{N}$$

$$= \frac{\operatorname{count}_{\boldsymbol{x}}(\boldsymbol{h}v)}{\operatorname{count}_{\boldsymbol{x}}(\boldsymbol{h})}$$

A common mistake is to forget that  $\theta_{v|h}$  is a *conditional* probability and estimate the joint probability p(hv) instead.

#### Reflection

Given a sequence of words, what procedure would you use to calculate its n-gram probability? To make this procedure as fast as possible, what properties would you want for the data structure that stores  $\theta$ ?

# Choosing n is a Balancing Act

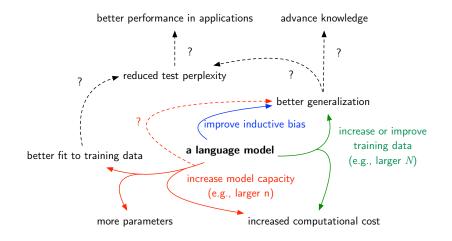
If n is too small, your model can't learn very much about language.

As n gets larger:

- The number of parameters grows with  $O(V^n)$ .
  - What's a parameter?
- Most n-grams will never be observed, so you'll have lots of zero probability n-grams. This is an example of data sparsity.
- Your model depends increasingly on the training data; you need (lots) more data to learn to generalize well.

This is a beautiful illustration of the bias-variance tradeoff.

## Language Modeling Research in a Nutshell



#### Smoothing: Attempts to Improve Inductive Bias

The game: prevent  $\theta_{v|h} = 0$  for any v and h, while keeping  $\sum_{x} p(x) = 1$  so that perplexity stays meaningful.

- ► Simple method: add λ > 0 to every count (including counts of zero) before normalizing; Eisenstein (2019) calls this "Lidstone" smoothing
- Longstanding champion: modified Kneser-Ney smoothing (Chen and Goodman, 1998)
- Reasonable, easy solution when you don't care about perplexity: stupid backoff (Brants et al., 2007)

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Empirica solution: try different values, and choose one using a **validation** dataset.

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- Never the test set, because that's cheating!

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Better solution: tune them using a systematic and replicable search procedure; report this procedure. See Dodge et al. (2019).

# n-gram Models: Assessment

#### Pros:

- Easy to understand
- Cheap (Lin and Dyer, 2010)
- Fine in some applications and when training data is scarce

#### Cons:

- Fixed, known vocabulary assumption
- Markov assumption is linguistically inaccurate
  - (But not as bad as unigram models!)
- Data sparseness problem

Instead of a lookup for a word and fixed-length history  $(\theta_{v|h})$ , define a vector function:

$$p(X_i \mid \boldsymbol{X}_{1:i-1} = \boldsymbol{x}_{1:i-1}) = \mathbf{NN}(\mathbf{enc}(\boldsymbol{x}_{1:i-1}); \boldsymbol{\theta})$$

where  $\theta$  do the work of *encoding* the history and *transforming* it into a distribution over the next word.

The transformation is described as a composed series of simple transformations or "layers."

Like many things from machine learning, the name invites confusion.

Formally, it's a function NN from  $\theta$  (learned parameters) and inputs to outputs, all of which are real-valued vectors (or matrices, or tensors, or collections of them).

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 "Nonlinear" means there does not exist a matrix A such that NN(v; θ) = Av, for all v.

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For a neural language model:

We need an encoder that maps word histories h to vectors/matrices.

• We interpret the output as  $p(X_i | X_{1:i-1} = h)$ .

If you let the label set be  $\mathcal{V}$ , then you can reduce language modeling to training a supervised classification trained on N instances (one per word). Input is a history, output is a prediction of the next word (or distribution over the possibilities).

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Note that the instances will not be independent, so it's a bit different from the usual classification setup.

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E.g., the multinomial logistic regression probability function is differentiable with respect to  $\theta$  (its weights).:

$$p(X_t = v \mid \boldsymbol{X}_{i-\mathsf{n}+1:i-1} = \boldsymbol{h}) = \frac{\exp \boldsymbol{\theta}^\top \mathbf{f}(\boldsymbol{h}, v)}{\sum_{v' \in \mathcal{V}} \exp \boldsymbol{\theta}^\top \mathbf{f}(\boldsymbol{h}, v')}$$

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# NLM v. 0: Classification

Lau et al. (1993), among others

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E.g., the multinomial logistic regression probability function is differentiable with respect to  $\theta$  (its weights).:

$$p(X_t = v \mid \boldsymbol{X}_{i-n+1:i-1} = \boldsymbol{h}) = \frac{\exp \boldsymbol{\theta}^\top \mathbf{f}(\boldsymbol{h}, v)}{\sum_{v' \in \mathcal{V}} \exp \boldsymbol{\theta}^\top \mathbf{f}(\boldsymbol{h}, v')}$$

Remember, though, that to do this, you need to decide what **features** of h and each candidate next word to use.

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#### Reflection

What do you think made these models impractical for realistic language modeling?

## Multinomial Logistic Regression



If you understand the principles, it's easier to learn the models to come.

## Why So Many Models?

We're going to see a lot of neural network approaches to language modeling.

The general ideas used in the series of models shown here have been used across NLP. This happens a lot in AI in general; understanding similarities and differences across approaches makes it easier to make sense of new developments!

## Two Key Developments

- 1. "Embedding" words as vectors.
- 2. Layering to increase capacity (i.e., the set of distributions that can be represented).

Same as before: we run stochastic (sub)gradient descent algorithms to maximize likelihood.

Different from before: likelihood is not necessarily convex in  $\theta$ .

#### "One Hot" Vectors

Let  $\mathbf{e}_i \in \mathbb{R}^V$  be the *i*th column of the identity matrix  $\mathbf{I}$ .

$$\mathbf{e}_{1} = \begin{bmatrix} 1\\0\\\vdots\\0\\0 \end{bmatrix}; \quad \mathbf{e}_{2} = \begin{bmatrix} 0\\1\\\vdots\\0\\0 \end{bmatrix}; \quad \dots; \quad \mathbf{e}_{V} = \begin{bmatrix} 0\\0\\\vdots\\0\\1 \end{bmatrix}$$

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A neural language model starts by "looking up" each history-word by multiplying its one hot vector by a matrix  $\mathbf{M}$ ;  $\mathbf{e}_v^\top \mathbf{M} = \mathbf{m}_v$ , the "embedding" of v.

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**M** becomes part of the parameters  $(\theta)$ .

The dense embeddings in  ${\bf M}$  lead us to an interesting idea: words can be closer or farther in different dimensions.

Many have attempted to connect this notion to word meanings.

## Brief Tangent: Word Vectors

This lecture focuses on language models; neural language models require that we represent vocabulary words as vectors.

That idea—"word vectors"—also came about in a separate thread of research (Schütze, 1992), independent of language modeling. We could easily spend an hour on that topic!

Indeed, we could have started from word vectors and worked our way up to the language models you're learning here, an approach I take in Smith (2020).

#### Standalone Word Vector Methods

- Most methods start from cooccurrence statistics between words: how often does v appear just after (or before) v'? More often than we'd expect by chance under a unigram model?
- Relatedly: guess a word at position i given a word in a nearby position.
- Popular methods include continuous bag of words and skip-gram (Mikolov et al., 2013a,b) in the "word2vec" package; these are trained much like neural classifiers, and have a close relationship to matrix factorization of a coocurrence-statistic matrix (Levy and Goldberg, 2014).

Three Approaches to Word Vectors in Language Models

- 1. Most common today: treat  $\mathbf{M}$  as "just more parameters," initialized randomly and learned during language model training.
- "Pretrain" M using a different algorithm (like skip-gram), then plug them in as fixed ("frozen") values. Train the other LM parameters.
- 3. Use pretrained word embeddings as initial values and "finetune" **M** during NLM training.

The appeal of options 2 and 3: if pretraining is cheap, we could get  $\mathbf{M}$  from lots more data and spend less computation learning the other LM parameters.

#### Sequences of Word Vectors

Given a word sequence  $\langle v_1, v_2, \ldots, v_k \rangle$ , we transform it into a sequence of word vectors,

 $\mathbf{m}_{v_1}, \mathbf{m}_{v_2}, \dots, \mathbf{m}_{v_k}$ 

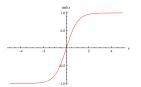
Using neural networks in NLP requires decisions about how to deal with *variable-length* input, because the lengths of the histories vary.

# Adding Layers

Neural networks are built by composing functions, a mix of

- affine, v' = Wv + b (note that the dimensionality of v and v' might be different)
- nonlinearity, including softmax, elementwise hyperbolic tangent

$$v'_i = \tanh(v_i) = \frac{e^{v_i} - e^{-v_i}}{e^{v_i} + e^{-v_i}},$$



and rectified linear ("relu") units,  $v'_i = \max(0, v_i)$ .

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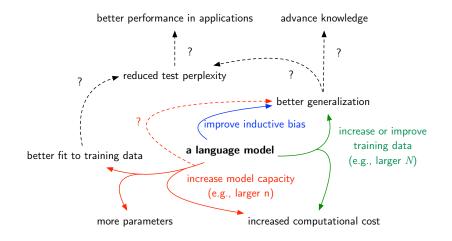
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The typical pattern is affine, nonlinear, affine, nonlinear, ...

More layers  $\Rightarrow$  increased capacity (more parameters, more computational cost, better training data fit)

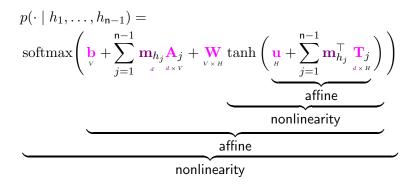
#### Language Modeling Research in a Nutshell



# NLM v. 1: Feedforward

(Bengio et al., 2003)

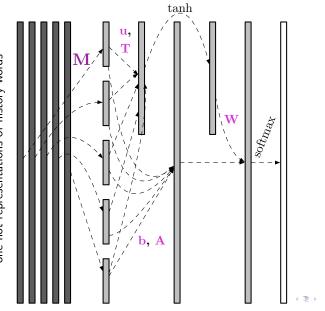
Define the n-gram probability as follows:



Parameters  $\boldsymbol{\theta}$  include  $\mathbf{M}$  and everything in pink.

Hyperparameters: dimensionalities  $d \mbox{ and } H$ 

## Feedforward NLM Computation Graph



one-hot representations of history words

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#### Interpretation?

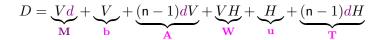
It's a bit like a multinomial logistic regression classifier language model with two kinds of "features":

- Concatenation of context-word embeddings vectors m<sub>hj</sub> (but these "word feature" vectors are themselves learned, not fixed in advance)
- tanh-affine transformation of the above

New parameters arise from (i) embeddings and (ii) affine transformations.

No single parameter will have any intuitive meaning.

#### Number of Parameters



For Bengio et al. (2003),  $V \approx 18000$  (after OOV processing);  $d \in \{30, 60\}$ ;  $H \in \{50, 100\}$ ; n - 1 = 5. So D = 461V + 30100 parameters, compared to  $O(V^n)$  for classical n-gram models.

- Forcing A = 0 eliminated 300V parameters and performed a bit better, but training was slower to converge.
- ► If we averaged m<sub>hj</sub> instead of concatenating, we'd get to 221V + 6100 (this is a variant of "continuous bag of words," Mikolov et al., 2013a).

 Historical answer: multiple layers and nonlinearities allow feature *combinations* a linear model can't get.

## Why does it work?

- Historical answer: multiple layers and nonlinearities allow feature *combinations* a linear model can't get.
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## Why does it work?

- Historical answer: multiple layers and nonlinearities allow feature *combinations* a linear model can't get.
  - Neural models seem to smoothly explore lots of approximately-conjunctive features.
- Modern answer: representations of words and histories are tuned, simultaneously, to the next-word prediction task.
- Word embeddings: a powerful idea!

## Reminders about Training

Good news: apply maximum likelihood principle and SGD as with v. 0. Lots more details in Eisenstein (2019) section 3.3 and Goldberg (2015).

Bad news:

- Log-likelihood function is not convex.
  - So any perplexity experiment is evaluating the model, the initial value of θ (usually random), and an algorithm for estimating it.
- Calculating log-likelihood and its gradient is very expensive (5 epochs took 3 weeks on 40 CPUs).

## Observations about NLMs (So Far)

- ► There's no knowledge built in that the most recent word h<sub>n-1</sub> is "closer" than earlier ones; it must be learned (probably learnable?).
- ► Hyperparameters: in addition to choosing n, also have to choose dimensionalities *d* and *H*.
- ▶ Parameters of these models are mostly hard to interpret.
- Architectures are not especially intuitive.
- Impressive perplexity reduction got people's interest.

#### Feedforward Networks



Like v. 0, but more layers and harder to understand.

## Neural Networks for Sequences

A feedforward network is fine if our input is bounded in length and we believe each position comprises its own features.

- That's not really how language works, though; there's nothing special about (for example) "the word four positions back."
- It also doesn't scale to longer sequences well (consider parameters specifically tied to the 974th word of a document).
- It also doesn't capture the way words tend to combine locally (e.g., with their neighbors) to form bigger meanings (compositionality).

What follows are three families or styles of networks that reuse parameters to **encode** sequences of arbitrary length.

#### NLM v. 2: Convolutional Networks (Sliding Windows)

Consider the entire history for word t,  $h = \langle x_1, x_2, \dots, x_{t-1} \rangle$  (no Markov assumption).

Start with  $\mathbf{X}^{(0)} = [\mathbf{m}_{x_1}; \mathbf{m}_{x_2}; \dots; \mathbf{m}_{x_{t-1}}].$ 

We will define a new matrix,  $\mathbf{X}^{(\ell)}$ , at each layer of the network, by applying a *convolution* function to the matrix  $\mathbf{X}^{(\ell-1)}$ . The vector  $\mathbf{X}^{(\ell)}[*,m]$  can be considered a "hidden state" representation of history word m at layer  $\ell$ .

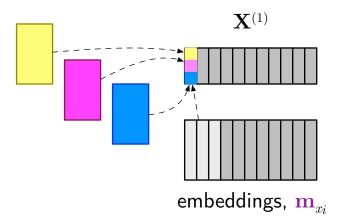
#### Convolution Layers

A convolution layer applies a feedforward-like "affine + nonlinear" sliding window function across the input matrix, at each position.

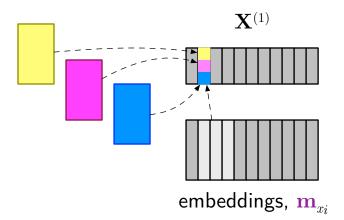
$$\mathbf{X}^{(1)}[k,m] = f\left(b_k + \sum_{i=1}^d \sum_{j=1}^w \mathbf{C}^{(k)}[i,j] \cdot \mathbf{X}^{(0)}[i,m+j-1]\right)$$

f is a nonlinearity (like tanh). w is the width of the sliding window. Each k is a different "filter" and each m is a word position.

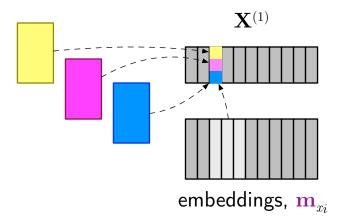
Hyperparameters: number of layers, and, at every layer,  $f, \ w,$  number of filters



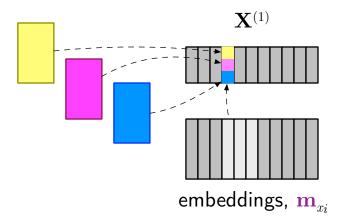
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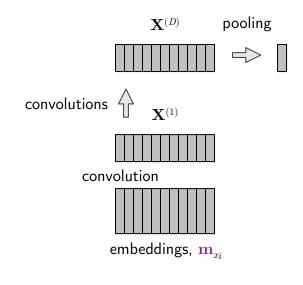
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#### Convolutional Network: Pooling

Let the dimensionality of the last (Dth) layer be  $d_{out}$ .

Pooling takes  $\mathbf{X}^{(D)} \in \mathbb{R}^{d_{out} \times (t-1)}$  and maps it into  $\mathbb{R}^{d_{out}}$ .

Two standard options (with no additional parameters) are max pooling,

$$z_k = \max_j \mathbf{X}^{(D)}[k, j];$$

and average pooling,

$$z_k = \frac{1}{t-1} \sum_{j=1}^{t-1} \mathbf{X}^{(D)}[k, j].$$

Finally,  $\operatorname{softmax}(\mathbf{z})$  gives a probability distribution over outputs.

### Reflection

Consider the computations required for encoding the history of word  $x_t$  and the history of word  $x_{t+1}$ . Do you see a way to make training efficient that wouldn't have been available for the feedforward NLM?

## Historical and Practical Notes

Convolutional neural networks originated in computer vision; similar ideas emerged in speech recognition.

Seminal use of convolutional networks for text classification: Kim (2014). Example use in language modeling: Dauphin et al. (2017).

Dilated convolutional networks use longer "strides" at deeper levels, skipping over increasingly more of the words, allowing effectively longer windows; see Yu and Koltun (2015).

# Convolutional Networks



An import from computer vision, often touted for their speed.

#### NLM v. 3: Recurrent Neural Network Mikolov et al. (2010)

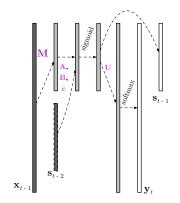
- Again, no Markov assumption; the history for word t is  $h = \langle x_1, x_2, \dots, x_{t-1} \rangle$ , mapped to  $\langle \mathbf{m}_{x_1}, \mathbf{m}_{x_2}, \dots, \mathbf{m}_{x_{t-1}} \rangle$ .
- ▶ The history is encoded as a fixed-length "state" vector,  $s_{t-1}$ .

$$p(\cdot \mid \boldsymbol{x}_{1:(t-1)}) = \mathbf{y}_t = \operatorname{softmax} \left( \mathbf{s}_{t-1}^\top \mathbf{U} \right)$$
$$\mathbf{s}_i = \operatorname{sigmoid} \left( \mathbf{m}_{x_i}^\top \mathbf{A} + \mathbf{s}_{i-1}^\top \mathbf{B} + \mathbf{c} \right)$$
$$\mathbf{s}_0 = \mathbf{0}$$

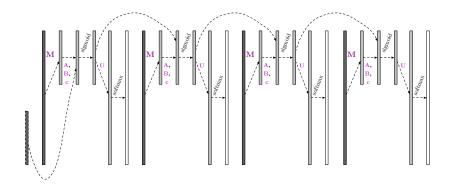
Note the recurrence.

The "depth" of the network corresponds to the position in the sequence (here, t).

# Computation Graph: RNN



# Visualization



### Improvements to RNN Language Models

The simple RNN is known to suffer from two related problems:

- "Vanishing gradients" during learning make it hard to propagate error into the distant past.
- State tends to change a lot on each iteration; the model "forgets" too much.

Some variants:

- "Stacking" the functions to make deeper networks, feeding the output of one in as the input to the next.
- Sundermeyer et al. (2012) use "long short-term memories" (LSTMs, Hochreiter and Schmidhuber, 1997; see Olah, 2015) and Cho et al. (2014) use "gated recurrent units" (GRUs) to define the recurrence.

## **Recurrent Networks**



Established the dominance of neural models in NLP, strongest option for many settings for several years.

Four NLMs so far:





- v. architecture
- 0 multinomial logistic regression
- 1 feedforward neural network



- 2 convolutional neural network
- 3 recurrent neural network

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None of these were designed specifically for language modeling, though arguably they are increasingly "language savvy" in their handling of sequences.

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Also increasingly expensive.

#### Four NLMs so far:







- v. architecture
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2 convolutional neural network



3 recurrent neural network

The last model, v. 4, is called the "transformer" (Vaswani et al., 2017). I won't have time to teach that!

### Sequence-to-Sequence

So far, we've focused on modeling  $p(X_i | X_{1:i-1})$ , which can be composed to give p(X).

That makes sense for "autocomplete" (the first motivation for this lecture), but what if we have an *input*?

Sequence-to-sequence ("seq2seq") models, also called "encoder-decoder" models (as opposed to "decoder-only") are about  $p(\boldsymbol{Y} \mid \boldsymbol{X})$ , where both random variables are sequences.

The driving application motivating seq2seq models is automatic translation between natural languages, known as "machine translation" (MT).

The seq2seq family of approaches was developed for MT, and we'll focus on that use case. Today, it's applied to many problems in NLP; often, it's an easy starting point.

# MT Evaluation

Intuition: good translations are **fluent** in the target language and **faithful** to the original meaning.

Bleu score (Papineni et al., 2002):

- Compare to a human-generated reference translation
- Or, better: multiple references
- Weighted average of n-gram precision (across different n)

There are many more recent alternatives; they often use models specifically trained for this problem.

Better: human evaluations that compare output to reference.

### Warren Weaver to Norbert Wiener, 1947

One naturally wonders if the problem of translation could be conceivably treated as a problem in cryptography. When I look at an article in Russian, I say: 'This is really written in English, but it has been coded in some strange symbols. I will now proceed to decode.'

A pattern for modeling a pair of random variables, X and Y:

$$\boxed{\texttt{source}} \longrightarrow Y \longrightarrow \boxed{\texttt{channel}} \longrightarrow X$$

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• Decoding: select y given X = x.

$$y^{*} = \underset{y}{\operatorname{argmax}} p(y \mid x)$$
  
= 
$$\underset{y}{\operatorname{argmax}} \frac{p(x \mid y) \cdot p(y)}{p(x)}$$
  
= 
$$\underset{y}{\operatorname{argmax}} \underbrace{p(x \mid y)}_{y} \cdot \underbrace{p(y)}_{\text{channel model source model}}$$

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# Translation

Successful translation requires generating a word sequence that is:

Faithful to the input

#### Fluent

If we're mapping a French word sequence f to an English word sequence e, then:

$$e^* = \operatorname*{argmax}_{e} \operatorname{Faithfulness}(e; f) + \operatorname{Fluency}(e)$$

Language models can provide a "fluency" score.

# Translation

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$$= \underset{e}{\operatorname{argmax}} \underbrace{\log p(f \mid e)}_{\text{channel model}} + \underbrace{\log p(e)}_{\text{source model}}$$

Language models can provide a "fluency" score.

Let f and e be two sequences in French and English, respectively.

If we have enough such examples, we could estimate a conditional distribution  $p(\pmb{F}\mid\pmb{E}),$  known as the translation model.

In a noisy channel machine translation system, we could use this together with source/language model  $p({\bm E})$  to "decode"  ${\bm f}$  into an English translation.

### Reflection

Where might we find parallel data?

# History of Pre-Neural MT in One Slide

- Many approaches based on formal automata and expert-crafted rules, going back to the 1950s. Often these were based on linguistic theories.
- Brown et al. (1993) introduced the noisy channel approach and channel models models based on bitext. This was complicated stuff!
- Open source implementation (Al-Onaizan et al., 1999) and automatic evaluation (Papineni et al., 2002) followed.

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- Open source implementation (Al-Onaizan et al., 1999) and automatic evaluation (Papineni et al., 2002) followed.
- By the early 2000s, it was becoming clear that modeling translation "word-by-word" was missing out on powerful contextual cues. There were two solutions in friendly competition:
  - Phrase-based translation: work with chunks of words instead of words (Koehn et al., 2003), sometimes organized hierarchically (Chiang, 2007).
  - Syntax-based translation: use syntactic parse trees (from linguistics) of input, output, or both (Galley et al., 2004).

Some good overviews: Lopez (2008); Koehn (2009)

Original idea proposed by Forcada and Ñeco (1997); resurgence in interest starting around 2013. Strong starting points for current work: Sutskever et al. (2014); Bahdanau et al. (2014).

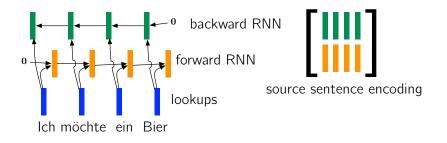
Take care: here, the terminology "encoder" and "decoder" are used differently than in the noisy-channel pattern.

### High-Level Model

$$\begin{split} p(\boldsymbol{E} = \boldsymbol{e} \mid \boldsymbol{f}) &= p(\boldsymbol{E} = \boldsymbol{e} \mid \mathsf{encode}(\boldsymbol{f})) \\ &= \prod_{j=1}^{\ell} p(e_j \mid e_0, \dots, e_{j-1}, \mathsf{encode}(\boldsymbol{f})) \end{split}$$

The encoding of the source sentence is a *deterministic* function of the words in that sentence.

# Neural MT Source-Sentence Encoder



**F** is a  $d \times m$  matrix encoding the source sentence f (length m). Originally, RNNs (depicted here) were used; now transformers dominate (Vaswani et al., 2017).

## Decoder: Contextual Language Model

Two inputs, the previous word and the source sentence context.

$$\begin{split} \mathbf{s}_t &= g_{\text{recurrent}}(\mathbf{e}_{e_{t-1}}, \underbrace{\texttt{access}(\texttt{encode}(\boldsymbol{f}))}_{\texttt{``context''}}, \mathbf{s}_{t-1}) \\ & \\ \mathbf{y}_t &= g_{\text{output}}(\mathbf{s}_t) \\ p(E_t = v \mid e_1, \dots, e_{t-1}, \boldsymbol{f}) &= [\mathbf{y}_t]_v \end{split}$$

(The forms of the two component gs are suppressed; just remember that they (i) have parameters and (ii) are differentiable with respect to those parameters.) The "access" function is a topic for later.

The neural language model we discussed earlier (Mikolov et al., 2010) didn't have the context as an input to  $g_{\text{recurrent}}$ .

### Learning and Decoding

$$\log p(\boldsymbol{e} \mid \mathsf{encode}(\boldsymbol{f})) = \sum_{i=1}^{m} \log p(e_i \mid \boldsymbol{e}_{0:i-1}, \mathsf{encode}(\boldsymbol{f}))$$

is differentiable with respect to all parameters of the neural network, allowing "end-to-end" training.

Decoding typically uses beam search.

# Beam Search for Seq2Seq Models

Input: beam size k, maximum length M, scoring function (typically  $\log p(Y_i \mid \pmb{y}_{0:i-1}, \texttt{encode}(\pmb{x}))$ 

 $B_0 \leftarrow \{\langle 0, \bigcirc \rangle\} // B_t$  is the beam of *t*-length prefixes  $F_0 \leftarrow \varnothing // F_t$  is the set of finished hypotheses of length  $\leq t$  for  $t \in \{1, \dots, M-1\}$ :

 $\blacktriangleright \ H \leftarrow \varnothing \ // \text{ hypotheses of length } t$ 

$$\blacktriangleright F_t \leftarrow F_{t-1}, B_t \leftarrow \varnothing$$

• for 
$$\langle s, \boldsymbol{y} \rangle \in B_{t-1}$$
 and  $v \in \mathcal{V}$ :

• add  $\langle s + \mathsf{score}(v \mid \boldsymbol{y}), \boldsymbol{y}v \rangle$  to H

- while  $|B_t| < k$ :
  - $\langle s, y \rangle \leftarrow \mathsf{pop}(H) \; // \; \mathsf{the \; max-scoring \; hypothesis}$
  - if  $\boldsymbol{y}$  doesn't end in  $\bigcirc$ , then add  $\langle s, \boldsymbol{y} \rangle$  to  $B_t$ ; else:
    - $\blacktriangleright \ \mathsf{add} \ \langle s, \boldsymbol{y} \rangle \ \mathsf{to} \ F_t$
    - If |F<sub>t</sub>| ≥ k, then return the max scoring item from F<sub>t</sub> // stop as soon as we have k finished hypotheses

### Notes on Beam Search

- Runtime depends on beam width, vocabulary size, maximum output length.
- Special cases:
  - k = 1 is greedy left-to-right decoding.
  - As  $k, M \to \infty$ , you're doing brute force, exhaustive search.
- ► Generally: no guarantee.
- Lots of variations; some add randomness, pruning, patience, and more!

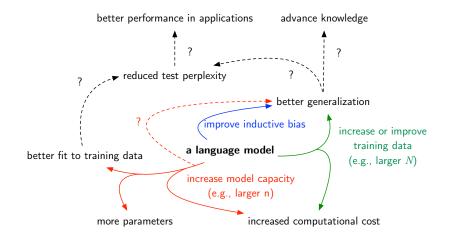
# On Data

The pervasive attitude for many years: more data is better (Church and Mercer, 1993).

The growth of the web makes easier to get more, and more diverse data. Today's datasets are too large to inspect manually, and often treated as trade secrets. Recent work documenting the construction of an open language modeling dataset: Soldaini et al. (2024).

The emergence of NLMs for generation (motivation III on slide 5) and the sheer quantity of data used have opened up new concerns about data quality, fairness, privacy, and cultural biases that NLMs can learn (and then repeat); see for example Gehman et al. (2020).

## Language Modeling Research in a Nutshell



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