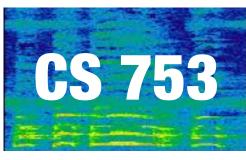
GANs +**Final practice questions**



Instructor: Preethi Jyothi

Lecture 23

Final Exam Syllabus

- 1. WFST algorithms/WFSTs used in ASR
- 2. HMM algorithms/EM/Tied state Triphone models
- 3. DNN-based acoustic models
- 4. N-gram/Smoothing/RNN language models
- 5. End-to-end ASR (CTC, LAS, RNN-T)
- 6. MFCC feature extraction
- 7. Search & Decoding
- 8. HMM-based speech synthesis models
- 9. Multilingual ASR
- 10. Speaker Adaptation
- 11. Discriminative training of HMMs

Questions can be asked on any of the 11 topics listed above. You will be allowed a single A-4 cheat sheet of **handwritten notes**; content on both sides permitted.



Final Project

Deliverables

- 4-5 page final report:
 - Analysis (if any), Summary
- Short talk summarizing the project: •
 - and ≈ 5 minutes for Q/A

✓ Task definition, Methodology, Prior work, Implementation Details, Experimental Setup, Experiments and Discussion, Error

✓ Each team will get 8-10 minutes for their presentation

Clearly demarcate which team member worked on what part

Final Project Grading

- Break-up of 20 points:
 - 6 points for the report
 - 4 points for the presentation
 - 6 points for Q/A
 - 4 points for overall evaluation of the project •

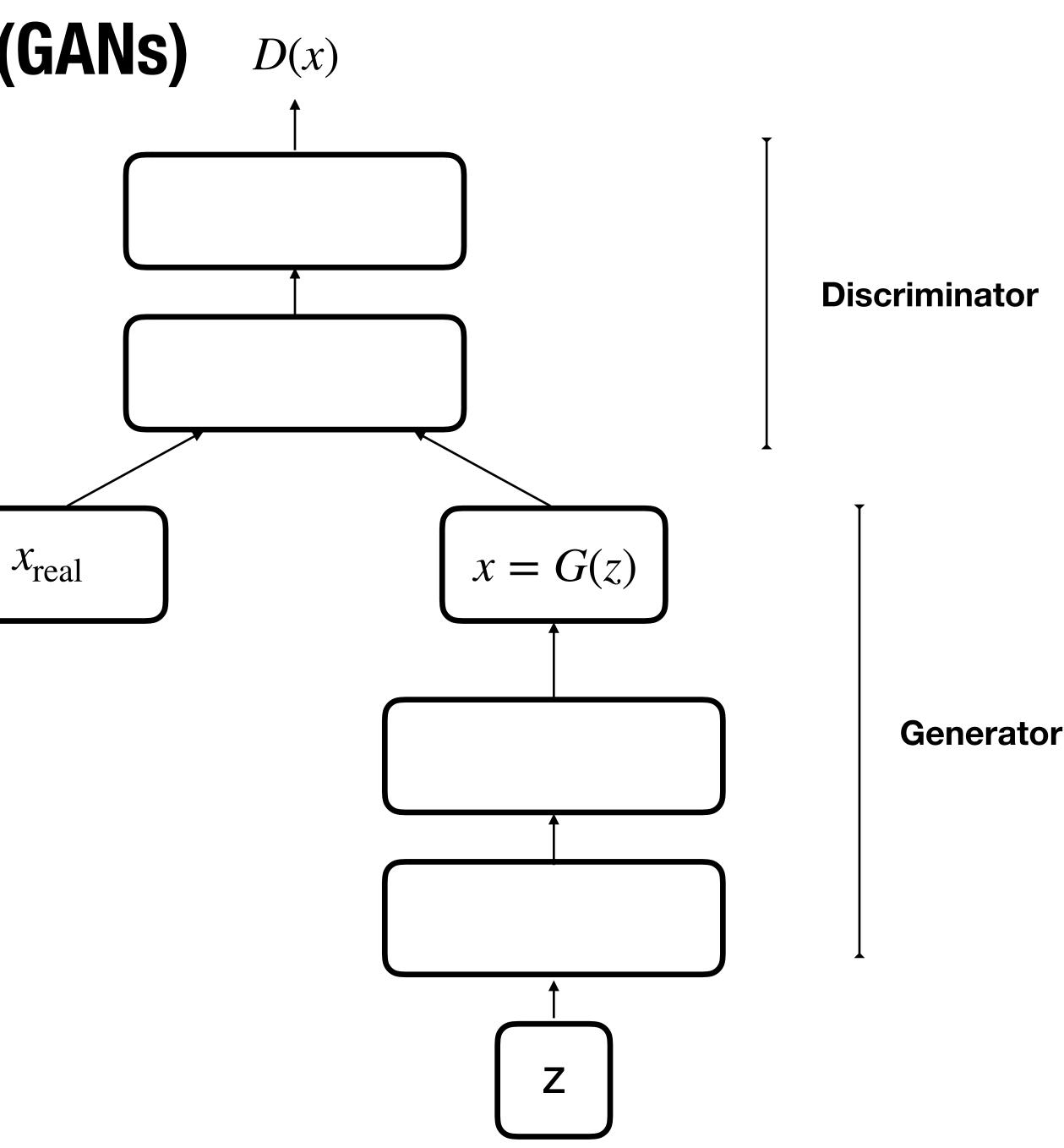
Final Project Schedule

- Presentations will be held on Nov 23rd and Nov 24th
- The final report in pdf format should be sent to pjyothi@cse.iitb.ac.in before Nov 24th
- and shared via Moodle before Nov 9th

The order of presentations will be decided on a lottery basis

Generative Adversarial Networks (GANs)

- Training process is formulated as a game between a generator network and a discriminative network
 - Objective of the generator: Create • samples that seem to be from the same distribution as the training data
 - Objective of the discriminator: • Examine a generated sample and distinguish between fake or real samples
- The generator tries to fool the discriminator network





Generative Adversarial Networks

- Cost function of the generator is the opposite of the discriminator's
- Minimax game: The generator and discriminator are playing a zero-sum game against each other

- $\max_{G} \min_{D} \mathcal{L}(G, D)$
- where $\mathcal{L}(G, D) = E_{x \in D}[-\log D(x)] + E_{z}[-\log(1 D(G(z)))]$

Training Generative Adversarial Networks

for number of training iterations do for k steps do

- $p_{\text{data}}(\boldsymbol{x}).$
- Update the discriminator by ascending its stochastic gradient:

$$\nabla_{\theta_d} \frac{1}{m} \sum_{i=1}^m \left[\log D\left(\boldsymbol{x}^{(i)} \right) + \log \left(1 - D\left(G\left(\boldsymbol{z}^{(i)} \right) \right) \right) \right].$$

end for

- Update the generator by descending its stochastic gradient:

$$\nabla_{\theta_g} \frac{1}{m} \sum_{i=1}^m \log\left(1 - D\left(G\left(\boldsymbol{z}^{(i)}\right)\right)\right).$$

end for

The gradient-based updates can use any standard gradient-based learning rule. We used momentum in our experiments.

• Sample minibatch of m noise samples $\{z^{(1)}, \ldots, z^{(m)}\}$ from noise prior $p_q(z)$. • Sample minibatch of m examples $\{x^{(1)}, \ldots, x^{(m)}\}$ from data generating distribution

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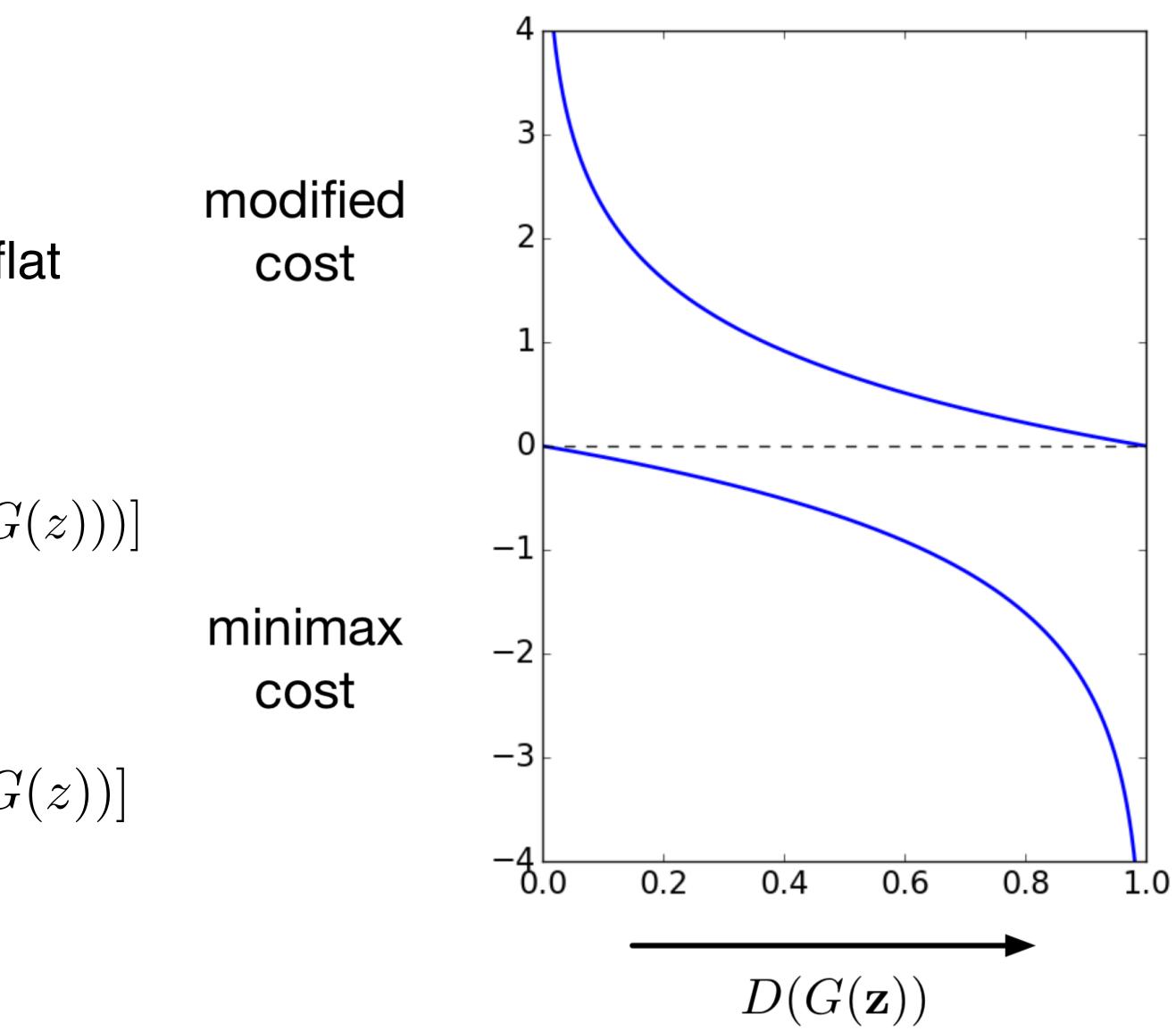
Better objective for the generator

- Problem of saturation: If the generated sample is really poor, the generator's cost is relatively flat
- Original cost

$$\mathcal{L}_{\text{GEN}}(G, D) = \mathcal{E}_z[\log(1 - D(G$$

Modified cost

 $\mathcal{L}_{\text{GEN}}(G, D) = \mathcal{E}_z[-\log D(G(z))]$



Large (& growing!) list of GANs

The GAN Zoo



- 3D-IWGAN Improved Adversarial Systems for 3D Object Generation and Reconstruction (github)
- 3D-PhysNet 3D-PhysNet: Learning the Intuitive Physics of Non-Rigid Object Deformations
- 3D-RecGAN 3D Object Reconstruction from a Single Depth View with Adversarial Learning (github)
- (github)
- ABC-GAN GANs for LIFE: Generative Adversarial Networks for Likelihood Free Inference
- AC-GAN Conditional Image Synthesis With Auxiliary Classifier GANs
- acGAN Face Aging With Conditional Generative Adversarial Networks
- ACGAN Coverless Information Hiding Based on Generative adversarial networks
- acGAN On-line Adaptative Curriculum Learning for GANs
- ACtuAL ACtuAL: Actor-Critic Under Adversarial Learning
- AdaGAN AdaGAN: Boosting Generative Models
- Adaptive GAN Customizing an Adversarial Example Generator with Class-Conditional GANs
- AdvGAN Generating adversarial examples with adversarial networks
- AE-GAN AE-GAN: adversarial eliminating with GAN

 3D-ED-GAN - Shape Inpainting using 3D Generative Adversarial Network and Recurrent Convolutional Networks • 3D-GAN - Learning a Probabilistic Latent Space of Object Shapes via 3D Generative-Adversarial Modeling (github)

• ABC-GAN - ABC-GAN: Adaptive Blur and Control for improved training stability of Generative Adversarial Networks

AdvEntuRe - AdvEntuRe: Adversarial Training for Textual Entailment with Knowledge-Guided Examples

Image from https://github.com/hindupuravinash/the-gan-zoo

 Generator and discriminator receive some additional conditioning information

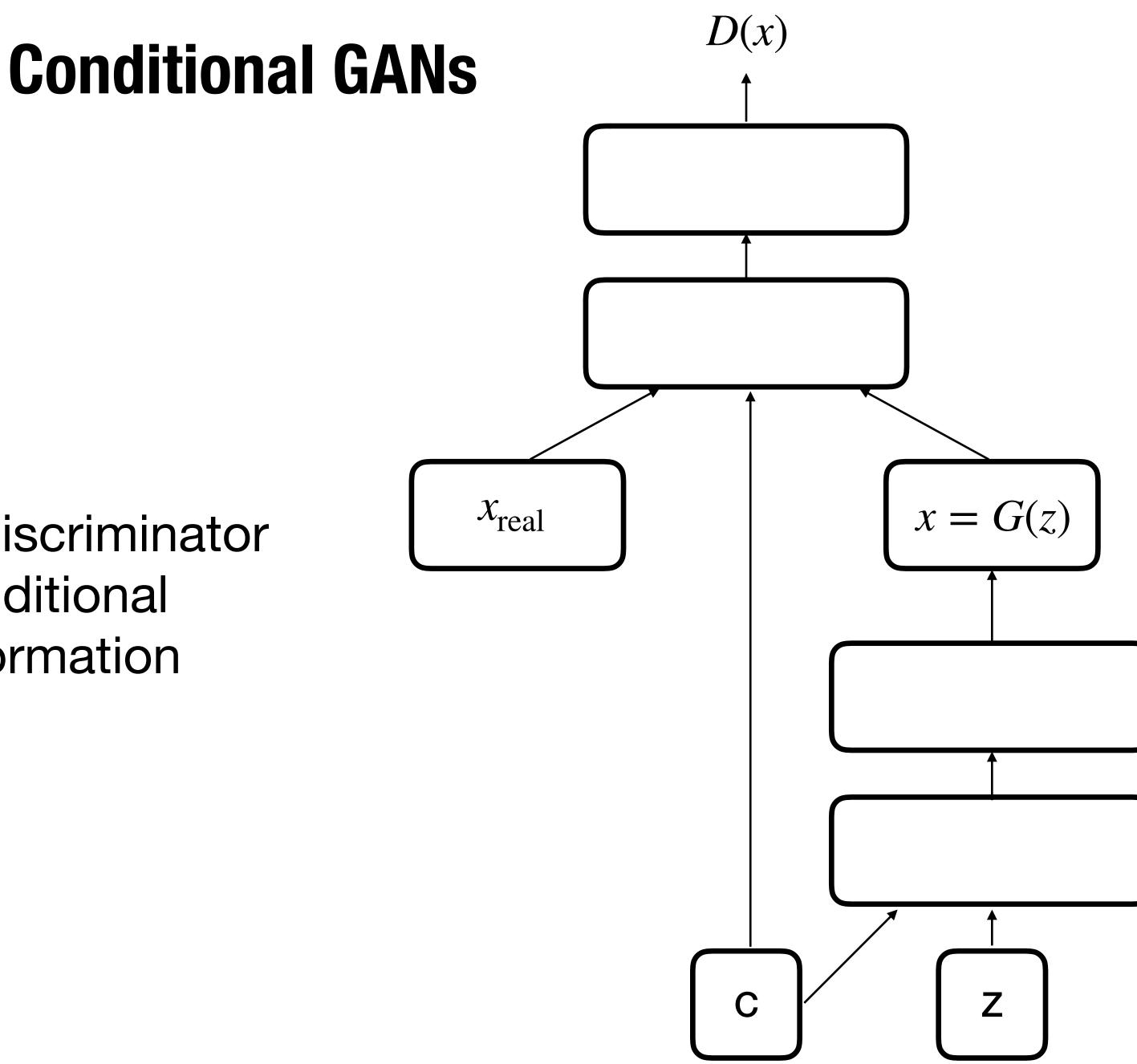
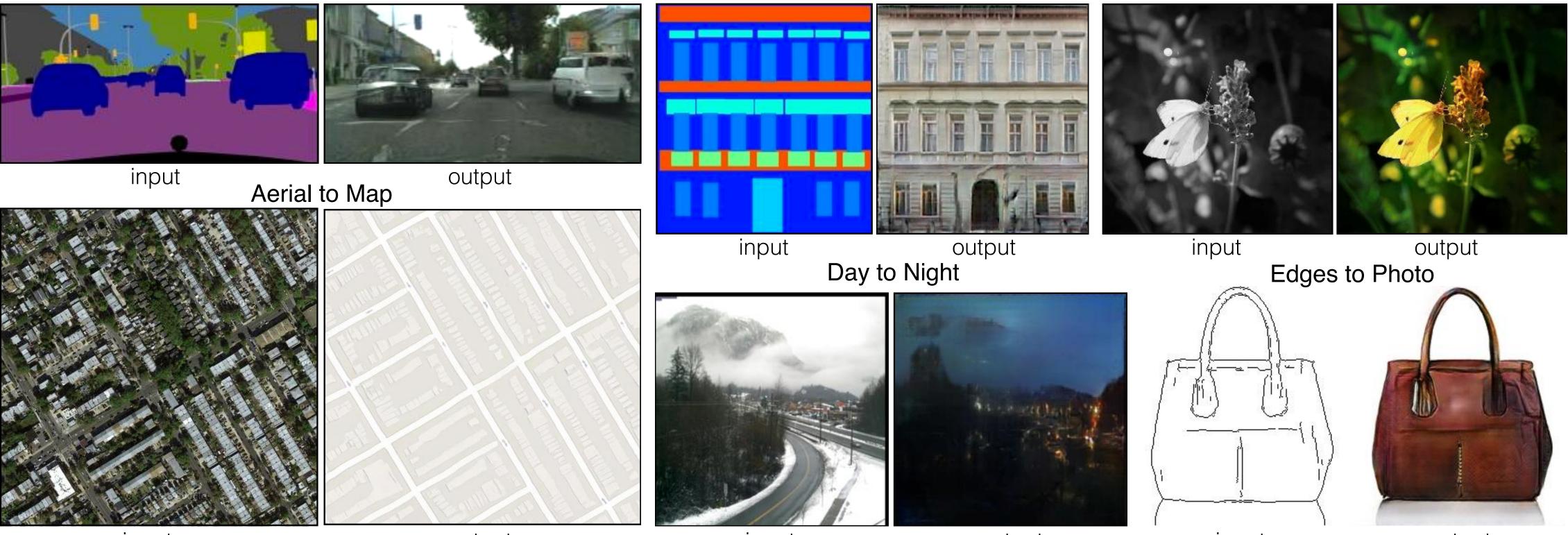




Image-to-image Translation using C-GANs

Labels to Street Scene



input

output

Labels to Facade

BW to Color

input

output

input

output

Text-to-Image Synthesis

this small bird has a pinkthis magnificent fellow isbreast and crown, and blackalmost all black with a redprimaries and secondaries.crest, and white cheek patch.



the flower has petals that are bright pinkish purple with white stigma





this white and yellow flower have thin white petals and a round yellow stamen



Image from Reed et al., ICML 2016, https://arxiv.org/pdf/1605.05396.pdf

Text-to-Image Synthesis

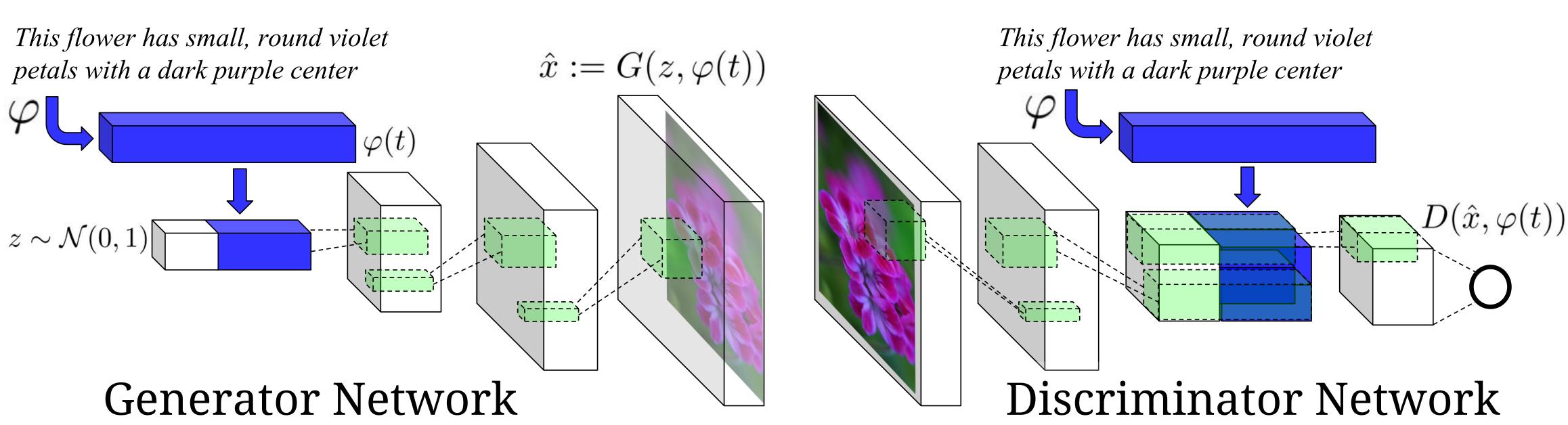


Image from Reed et al., ICML 2016, https://arxiv.org/pdf/1605.05396.pdf

Three Speech Applications of GANs

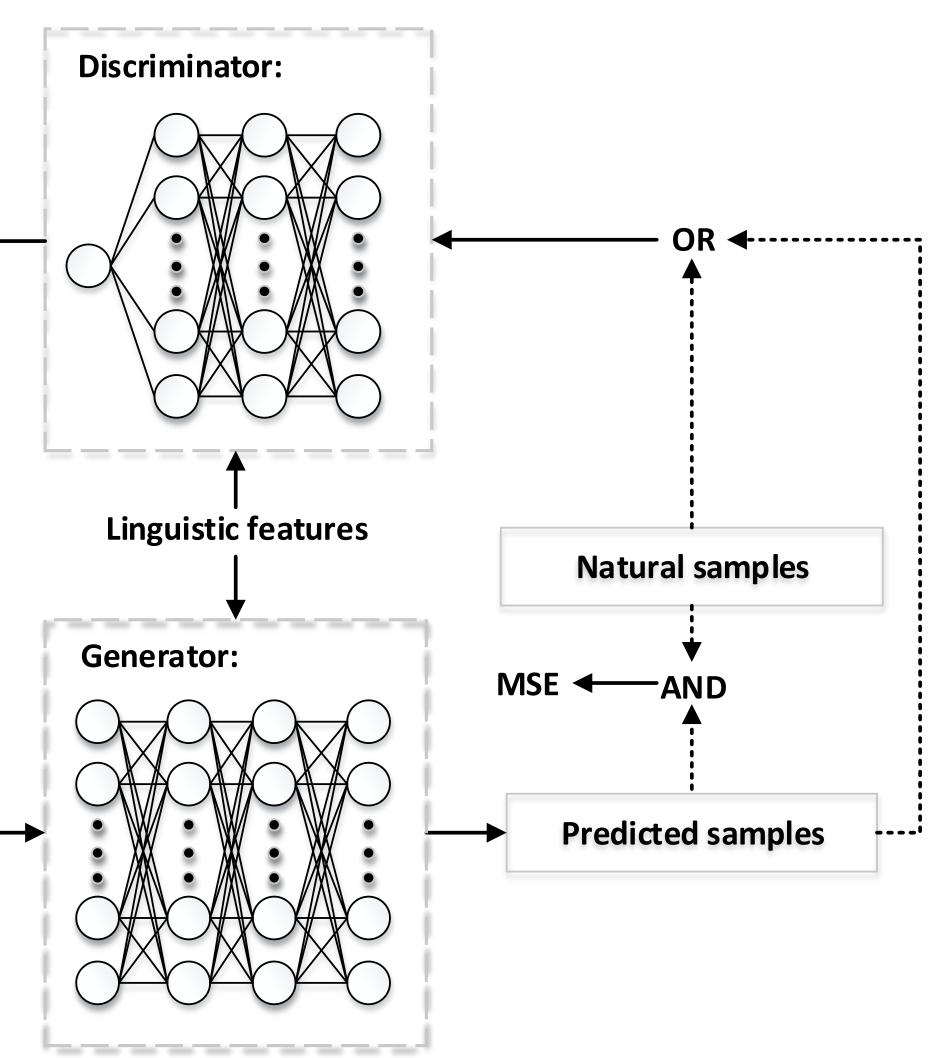
GANs for speech synthesis

 Generator produces synthesised speech which the Discriminator distinguishes from real speech



During synthesis, a random noise + linguistic features generates speech





SEGAN: GANs for speech enhancement

- Enhancement: Given an input noisy enhanced signal x
- inputs; G is fully convolutional

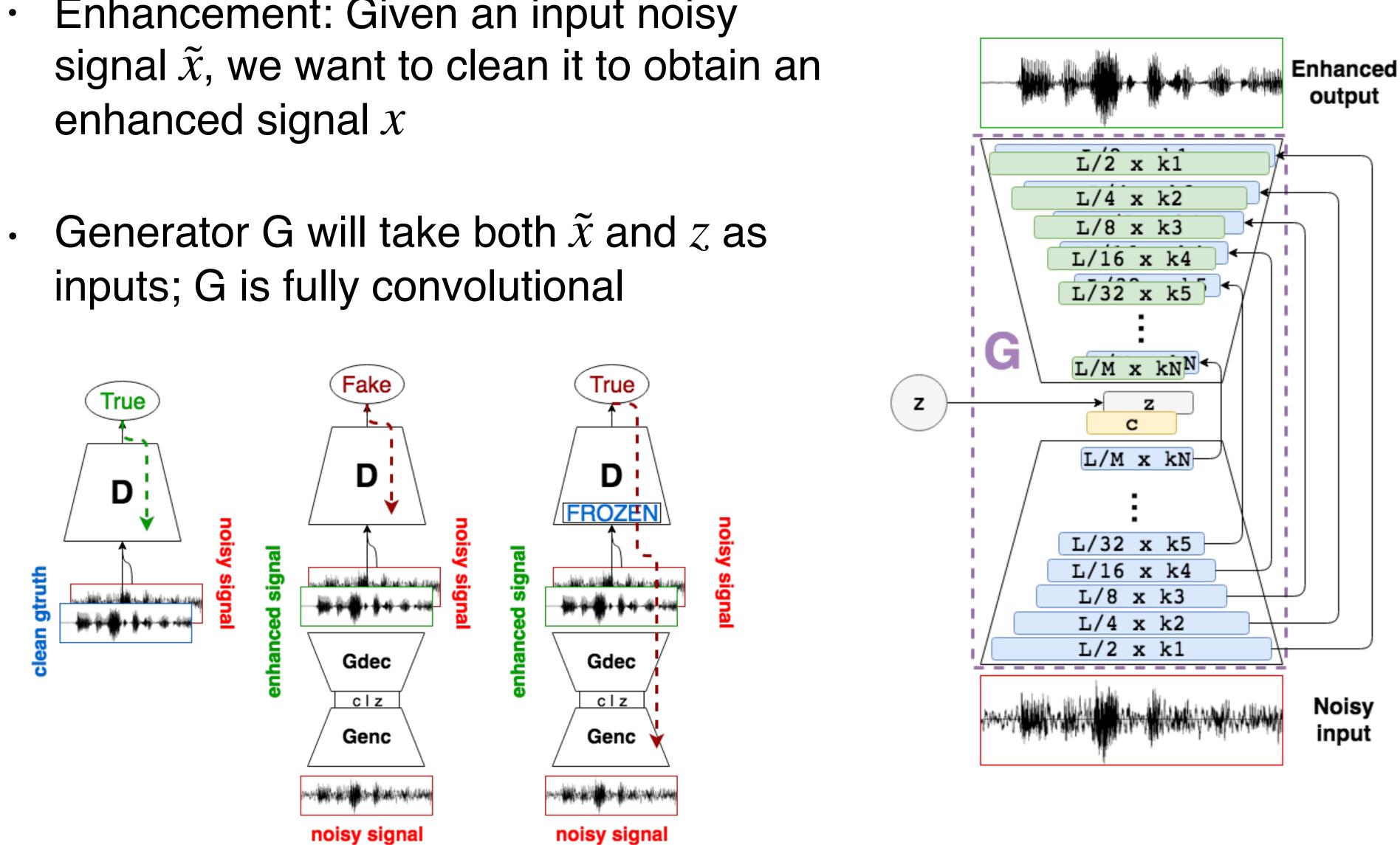


Image from https://arxiv.org/pdf/1703.09452.pdf

Voice Conversion Using Cycle-GANs

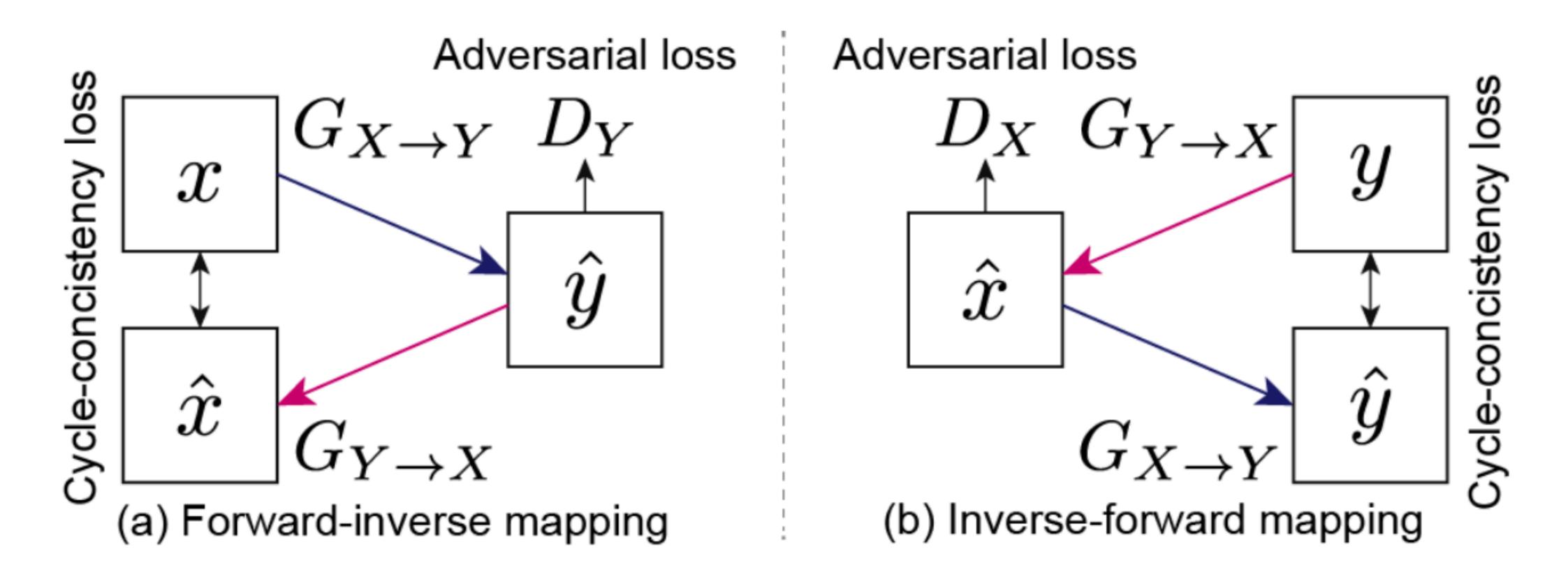
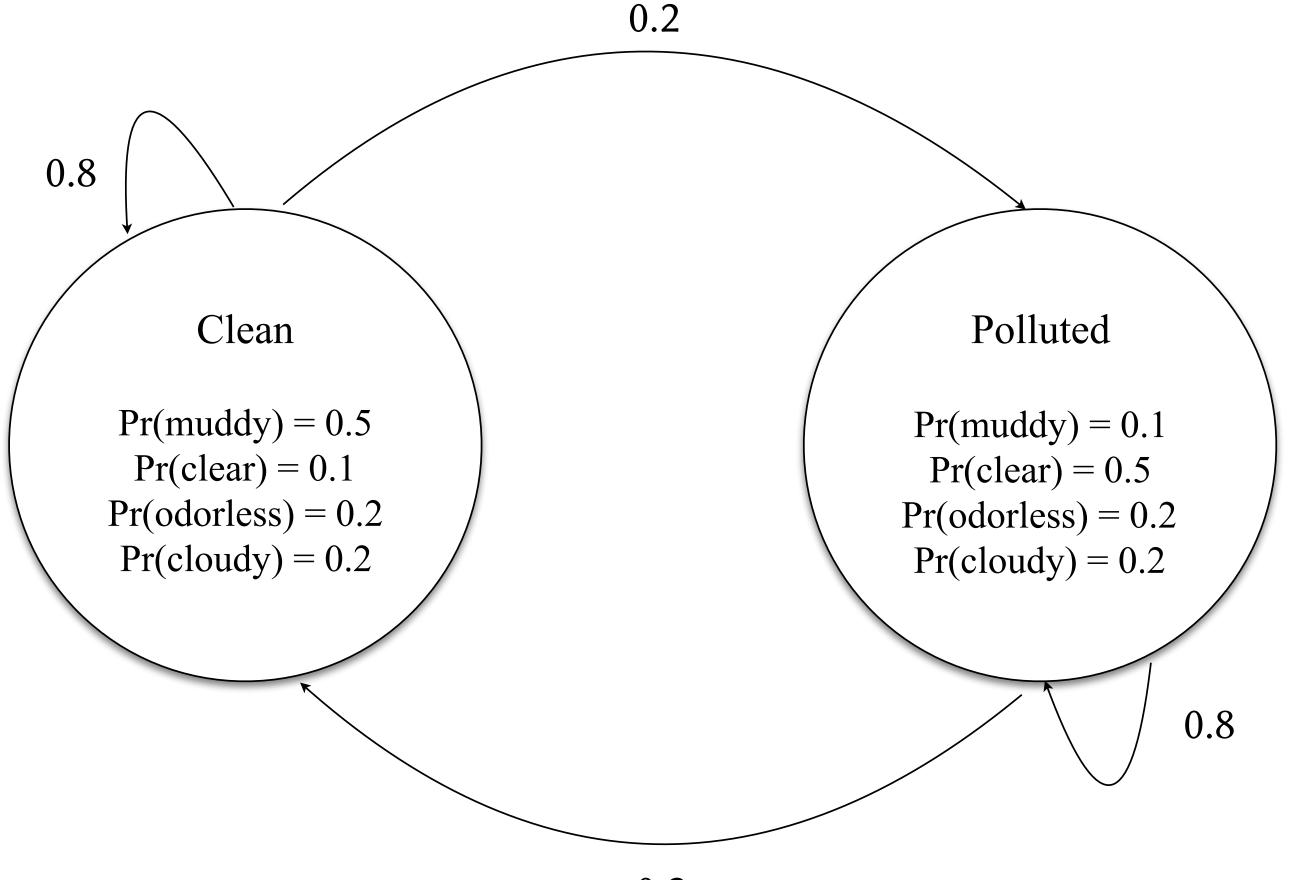


Image from https://arxiv.org/abs/1711.11293

Practice Questions

HMM 101

A water sample collected from Powai lake is either Clean or Polluted. However, this information is hidden from us and all we can observe is whether the water is muddy, clear, odorless or cloudy. We start at time step 1 in the Clean state. The HMM below models this problem. Let qt and Ot denote the state and observation at time step t, respectively.



a)What is
$$P(O_2 = clear)$$
?

b)What is $P(q_2 = Clean | O_2 = clear)$?

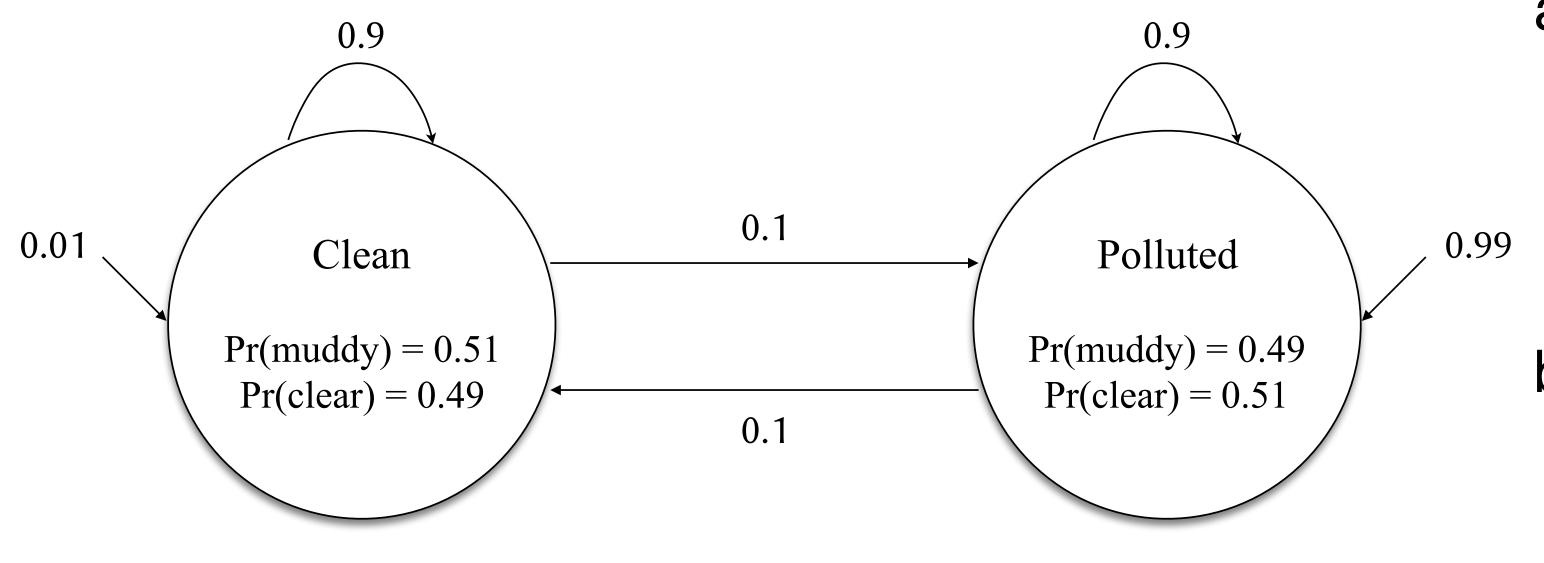
c)What is $P(O_{200} = cloudy)$?

d)What's the most likely sequence of states for the following observation sequence: $\{O_1 = \text{clear}, O_2 = \text{clear}, O_3 = \text{clear}, O_4 = \text{clear}, O_5 = \text{clear}\}$?

r)?

HMM 101

Say that we are now given a modified HMM for the water samples as shown below. Initial probabilities and transition probabilities are shown next to the arcs. (Note: You do not need to use the Viterbi algorithm to answer the next two questions.)



 a) What is the most likely sequence of states given a sequence of three observations: {muddy, muddy, muddy}?

b) Say we observe a very long sequence of "muddy" (e.g. 10 million "muddy" in a row). What happens to the most likely state sequence then?



Handling disfluencies in ASR

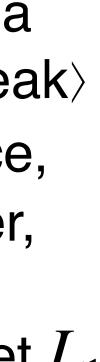
Recall that a pronunciation lexicon L maps a sequence of phones to a sequence of words. In this problem, we shall modify L in order to handle some limited forms of interruptions in speech (a.k.a. disfluencies). We will consider a dictionary of two words: W_1 with the phone sequence "a b c" and W_2 with the phone sequence "x y z".

a) Draw the state diagram of the finite-state machine L.

with breaks, and outputs a corresponding sequence of words. Draw the state diagram of L_1 .

b) We want to modify L such that it accounts for "breaks" when the speaker stops in the middle of a word and says the word all over again. For instance, the word W_1 may be pronounced as "a b (break) a b c," where \langle break \rangle is a special token produced by the acoustic model. In a valid phone sequence, breaks are allowed to appear only within a word, and not at the end or beginning of a word. Further, two consecutive (break) tokens are not allowed. But a word can be pronounced with an arbitrary number of breaks. E.g. W_1 can be pronounced also as "a b (break) a (break) a b (break) a b c". Let L_1 be an FST (obtained by modifying L from the previous part) that accepts all valid phone sequences







Handling disfluencies in ASR

Recall that a pronunciation lexicon L maps a sequence of phones to a sequence of words. In this problem, we shall modify L in order to handle some limited forms of interruptions in speech (a.k.a. disfluencies). We will consider a dictionary of two words: W_1 with the phone sequence "a b c" and W_2 with the phone sequence "x y z".

c) Next, we want to modify L_1 such that it can account for both "breaks" and "pauses." A pause corresponds to when the speaker briefly stops in the middle of a word and continues. For instance, the word W_1 may be pronounced as "a b (pause) c", "a (break) a (pause) b (break) a b c," etc. where (pause) is another special token produced by the acoustic model. In a valid phone sequence, these special tokens are allowed to appear only within a word, and two consecutive special tokens are not allowed. Let L_2 be an FST (obtained by modifying L_1 from the previous part) that accepts all valid phone sequences with breaks and pauses, and outputs a corresponding sequence of words. Draw the state diagram of L_2 .





2.Context-dependent HMMs are trained on these features 3. These HMMs are clustered using a decision tree 4. Durations of the HMM models are explicitly modeled

features (that are passed through a synthesis filter to produce speech). Say we (A)-(D) you would modify to add expressivity and briefly justify your choice.

Mixed Bag

- An HMM-based speech synthesis system can be described using the following steps:
 - 1.Spectral feature and excitation features are extracted from a speech database
- At synthesis time, for a given text sequence, the decision tree yields the appropriate HMM state sequence which in turn determines the output spectral and excitation want to add expressivity to the synthesized speech: i.e. we want to make the voice sound happy or sad, friendly or stern. Pick one of the above-mentioned steps from

Mixed Bag

Find the probability, Pr(drank|Mohan), given the following bigram counts:

Mol dra Mol drar Mo dra

 $\Pr(\text{drank}|\text{Mohan}) =$

Say you have an n-gram distribution which is smoothed using add- α smoothing for some $\alpha > 0$. The entropy of the smoothed distribution is

(B) less than (C) greater than (A) equal to the entropy of the original unsmoothed n-gram distribution. Pick one of (A), (B) or (C) and briefly justify

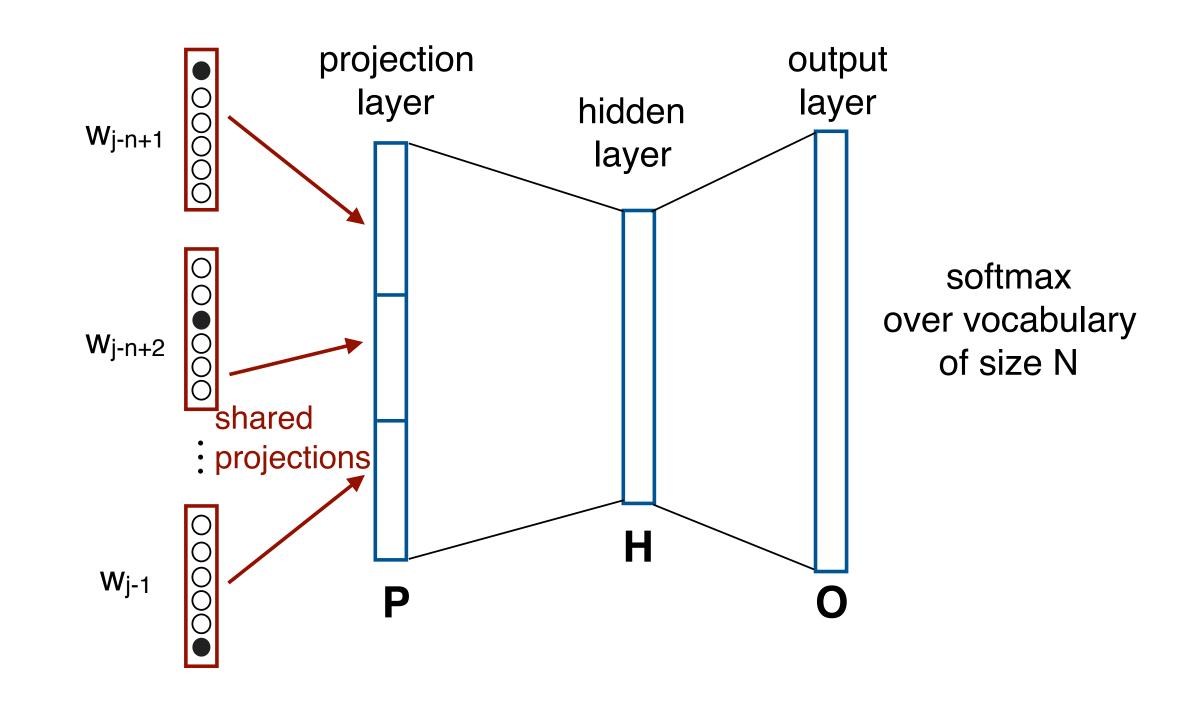
your choice.

han drank	10
ank coffee	1
han coffee	10
nk Mohan	5
ohan ate	10
ank water	20



Mixed Bag

Recall neural network language models (NNLMs) as shown in the schere 5c diagram below. For a given context of fixed length, each word in the context (drawn from a vocabulary of size N) is projected onto a P dimensional projection layer using a common $N \times P$ projection matrix, that is shared across the different word positions in the context. The value of the *i*th node in the output layer corresponds directly to the probability of a word *i* given its context.



The complexity to calculate probabilities using this NNLM is quite high. Describe one main reason why this evaluation is very costly in processing time.

CTC Alignments

of length N, the CTC objective function is given by:

 $P_{\rm CTC}(\mathbf{y}|\mathbf{x}) =$

where \mathcal{B} maps a per-frame output sequence $\mathbf{a} = (a_1, \ldots, a_T)$ to a final output sequence $\mathbf{y} = (y_1, \ldots, y_N)$

Consider a different definition of \mathcal{B} which first removes all occurrences of the blank symbol, and then compresses each run of an identical character to a run of length 1. Give an example of a sequence y such that there is no **a** with $\mathcal{B}(\mathbf{a}) = \mathbf{y}$, for this new \mathcal{B} . Briefly justify your answer.

Given an input sequence x of length T and an output character sequence y

$$\sum_{\mathbf{a}:\mathcal{B}(\mathbf{a})=\mathbf{y}} P(\mathbf{a}|\mathbf{x})$$



CTC Alignments

 $\ell_1 \text{ times} \qquad \ell_2 \text{ times}$ ℓ_M times $,\ldots,c_2,\ldots,c_M,\ldots,c_M)$ $_2$ times

Now suppose we would like to avoid the use of the blank symbol altogether. Towards this, we define a new \mathcal{B} which works as follows. Given $\mathbf{a} = (a_1, \ldots, a_T), \mathcal{B}$ defines the sequence $((c_1, \ell_1), (c_2, \ell_2), \ldots, (c_M, \ell_M))$ where $c_i \neq c_{i+1}$ and $\ell_i > 0$ for all i, and $\mathbf{a} = (\underbrace{c_1, \ldots, c_1}, \underbrace{c_2, \ldots, c_2}, \ldots, \underbrace{c_M, \ldots, c_M}).$ Then \mathcal{B} calculates the average run length $\overline{\ell} = \frac{1}{M} \sum_{i=1}^{M} \ell_i$, and outputs

$$\mathbf{y} = (\underbrace{c_1, \ldots, c_1}_{k_1 \text{ times}}, \underbrace{c_2, \ldots}_{k_2})$$

where $k_i = \max\{1, |\ell_i/\overline{\ell}|\}$. Here, k_i is an estimate of how many times c_i needs to be repeated, depending on how ℓ_i compares with the average run length $\overline{\ell}$. For example, $\mathcal{B}(a, a, b, b, b, b, b, b, b, c, c) = (a, b, b, c)$ because $\ell_1 = 2, \ell_2 = 8, \ell_3 = 2$ and therefore $k_1 = 2$ $1, k_2 = 2, k_3 = 1.$

Give an example of a sequence y such that there is no a with $\mathcal{B}(\mathbf{a}) = \mathbf{y}$, for this new \mathcal{B} .

