Game-Theoretic Models for Generative Learning

PhD Thesis Proposal

Abstract

Machine learning has achieved great success in classification tasks with the booming of powerful deep neural networks. However, there's still a big gap between computer and human intelligence. I identify three levels of artificial intelligence: memorization, recognition and creativity. Discriminative learning attempts to infer knowledge from data by modeling the conditional distribution of outputs given inputs, while generative learning tries to estimate the entire distribution of data and produces new samples. My thesis will focus on the latter problem and investigate generative learning from a game-theoretic perspective. We first formulate the problem as a distributionally robust game with payoff uncertainty, and then develop a robust optimization algorithm to solve the Nash equilibrium. Meanwhile, we will study the distance metric to measure similarity between distributions, which is vital to build the objective functions. Last, we plan to test our approach in simulations and apply it to train several generative models for images and audio.

1 Introduction

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- In the last decade, machine learning approaches have achieved great success with the boom of deep neural networks. There are three levels of human intelligence: memorization, recognition and generation. Computers are approaching humans on the first two levels, and now going to the third.
- Apart from recognition and classification tasks, people hope to learn the entire distribution of data
- 19 and generate new samples. In mathematics, it means to learn a function describing the structure of
- 20 unlabeled data, for example, density estimation is a direct application. It requires inferring the real
- data distribution given a set of observations. There's no straightforward way to evaluate the accuracy
- of generative models, because the goal is to produce lifelike artificial examples and the evaluation is
- 23 subjective in most cases.
- 24 In generative learning, new samples produced by the learned model should be indistinguishable
- 25 from the original data and have enough diversity. Generative models are powerful tools for many
- 26 tasks such as image and audio generation, video prediction, style transfer, voice conversion, and
- 27 semi-supervised learning.
- 28 If we work on low-dimension data, or just want some statistics instead of producing new samples,
- 29 there's no need to model the data distribution. A bunch of unsupervised learning algorithms work
- 30 well, e.g., k-means, Gaussian mixture model, EM, PCA, etc. But for complex tasks such as generating
- 31 images, audios and videos, neural networks demonstrate more power. There are three popular
- mainstream methods in deep generative learning, generative adversarial network (GAN), variational
- auto-encoders (VAE) and autoregressive models (e.g., WaveNet, PixelRNN).
- 34 In general, training deep generative models is hard and time consuming. For example, it requires
- 35 8 GPUs training 6 days to learn the WaveNet autoencoder. The high dimensional training data and
- 36 complex objective structures lead to many problems in optimization, such as algorithm instability,
- saturation, and mode collapse. Moreover, the model should have strong generalization power to
- produce diverse artificial examples instead of just memorizing the training set.
- In this research work, we plan to develop a new game-theoretic framework for generative learning.
- We formulate the problem as a distributionally robust game with payoff uncertainty, and develop

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- 41 a learning algorithm to solve the robust Nash equilibrium. In this game there are several groups
- 42 of players that are competitive, noncooperative and have different objectives. Each player works
- in a continuous action space to optimize its expected worst-case performance. The players are
- 44 implemented with neural network models to perform prediction, classification or data generation
- tasks. Agents with similar objectives form a group and work together against the others in order to
- optimize their expected payoffs. The distributionally robust Nash equilibrium is achieved by solving
- a minimax optimization problem. We consider using stochastic optimization techniques to deal with
- 48 large-scale learning tasks.
- 49 Iterative optimization algorithms travel to the equilibria by minimizing a loss function, which is in our
- 50 case a distance metric defined to measure the similarity between two distributions. As an important
- part of this research, We will study the pros and cons of several kinds of distance metrics, and compare
- 52 Wasserstein distance with the most prevalent information-based metrics such as Kullback-Leibler and
- 53 Jensen-Shannon divergence. We plan to develop a practical method to approximately calculate the
- distance between distributions and give theoretical analysis and numerical evaluations.
- 55 We will first verify the theoretical results through simulations, and then apply our approach on real
- datasets for image and audio generation. We plan to work on several tasks such as object detection,
- 57 vehicle tracking, image generation and audio synthesis. This research contributes to the areas of
- 58 distributionally robust game, deep generative learning, stochastic optimization and time-series data
- 59 analysis.

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2 Distributionally Robust Games

- 61 2.1 Introduction
- 62 2.1.1 Distribution Uncertainty Set
- 63 2.1.2 Related Work
- 64 2.2 Problem Formulation
- 65 2.2.1 From unsupervised learning to Generative Model
- 66 2.2.2 Game Theoretic Framework for Learning
- 67 2.2.3 Definition
- 68 2.2.4 The Existence of Distributionally Robust Nash Equilibria
- 69 2.3 Minimax Robust Game
- 70 2.3.1 From Duality to Triality Theory
- 71 2.3.2 Dimension Reduction
- 72 2.3.3 Evaluation
- 73 2.4 Case Study: Learning a Generative Adversarial Model

74 3 Wasserstein Metric

- All headings should be lower case (except for first word and proper nouns), flush left, and bold.
- First-level headings should be in 12-point type.
- 77 3.1 Introduction
- 78 3.1.1 Optimal Transportation Problem
- 79 **3.1.2 Definition**
- Second-level headings should be in 10-point type.

81 3.2 From KL divergence to Wasserstein Metric

Third-level headings should be in 10-point type.

83 3.3 Other Metrics: L1, L2, Maximum Mean Discrepancy

84 3.4 Dynamic Optimal Transport

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- with the text, with the heading followed by 1 em of space.

87 3.5 Case Study: A Toy Example

88 4 Learning Algorithms

89 These instructions apply to everyone.

90 4.1 Bregman Learning under f-divergence

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- long as you maintain internal consistency. As to the format of the references themselves, any style is
- acceptable as long as it is used consistently.
- 94 The documentation for natbib may be found at
- 95 http://mirrors.ctan.org/macros/latex/contrib/natbib/natnotes.pdf
- 96 Of note is the command \citet, which produces citations appropriate for use in inline text. For example,
- % \citet{hasselmo} investigated\dots
- 99 produces

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- Hasselmo, et al. (1995) investigated...
- If you wish to load the natbib package with options, you may add the following before loading the nips_2018 package:
- 103 \PassOptionsToPackage{options}{natbib}
- 104 If natbib clashes with another package you load, you can add the optional argument nonatbib when loading the style file:
- 106 \usepackage[nonatbib] {nips_2018}

4.2 Distributionally Robust Optimization

- Footnotes should be used sparingly. If you do require a footnote, indicate footnotes with a number 108
- in the text. Place the footnotes at the bottom of the page on which they appear. Precede the footnote
- with a horizontal rule of 2 inches (12 picas).
- Note that footnotes are properly typeset *after* punctuation marks.²

112 4.3 Train a Deep Generative Model

- All artwork must be neat, clean, and legible. Lines should be dark enough for purposes of reproduction.
- The figure number and caption always appear after the figure. Place one line space before the figure

¹Sample of the first footnote.

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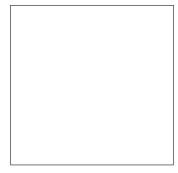


Figure 1: Sample figure caption.

Table 1: Sample table title

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caption and one line space after the figure. The figure caption should be lower case (except for first word and proper nouns); figures are numbered consecutively.

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4.4 Case Study: Unsupervised Learning for Clustering

All tables must be centered, neat, clean and legible. The table number and title always appear before the table. See Table 1.

Place one line space before the table title, one line space after the table title, and one line space after the table. The table title must be lower case (except for first word and proper nouns); tables are numbered consecutively.

Note that publication-quality tables *do not contain vertical rules*. We strongly suggest the use of the booktabs package, which allows for typesetting high-quality, professional tables:

128 This package was used to typeset Table 1.

29 4.5 Case Study: Generative Modeling for Image Synthesis

130 5 Domain Transfer as a Minimax Game

131 5.1 Introduction

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Many problems in machine learning involve translating data from one domain to another. For example, transfer photograph to artistic painting, convert one person's voice to another, translate music to imitate different musical instruments, etc. In supervised learning, it assumes that the test samples have the same distribution as the training set. However, it is not valid in many practical cases, so the knowledge needs to be transferred across domains to capture the domain shift. The general problem is to learn a mapping from one domain to another, on condition that it changes the appearance style while keeps the underlying content.

Let $x^s \in \mathbb{X}^S$ be samples in the source domain and $x^t \in \mathbb{X}^T$ be samples in the target domain. The goal of domain transfer is to learn a mapping: $G: \mathbb{X}^S \to \mathbb{X}^T$ such that the generated output $\hat{x}^t = G(x^s)$

is indistinguishable from the real samples drawn from the target domain. The optimal mapping G transports \mathbb{X}^S to $\hat{\mathbb{X}}^T$, which should have the same distribution as \mathbb{X}^T .

We propose a game-theoretic approach to learn the mapping. The approach is based on neural network representations to capture the high-level and low-level features, and distributionally robust optimization to find the Nash equilibrium. In the game, there are several groups of players with different objectives. The intergroup competition and intragroup collaboration enable the players to learn from others and optimize their worst-case performance.

This work has a wide range of use cases. In classification, since labelled data is scarce and expensive, 149 people want to learn from unlabelled target domain by exploring the knowledge learnt from a 150 well-labeled source domain. Domain transfer algorithms help to translate source data samples to target domain as well as the corresponding labels. In visual and performing arts, it's inspiring to 151 automatically generate artificial paintings with user-specified style or play synthetic music with 152 desired timbre and musical instrument. In informatics, it's useful to transform speaker identity by 153 modifying his voice to sound like another person. It is also possible to learn and mimic animal's 154 vocalization and study the feedback on the artificially generated sound. For case study, we apply our 155 approach in two typical domain transfer tasks: image style transfer and emotional voice conversion. 156

5.2 Motivation

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In machine learning, discriminative models predict labels from data by learning a conditional distribution p(y|x), while generative models produce new data with desired labels by drawing samples from distribution p(x|y). From a probabilistic modeling perspective, domain transfer models traslate samples from source domain to target domain by learning a joint distribution $p(x^s, x^t)$. If we do not have paired training data $\{x_i^s, x_i^t\}_{i=1}^N$, the problem is estimating joint distribution by its marginal distributions $p(x^s), p(x^t)$. According to coupling theory [1], the choice of valid joint distribution is generally not unique. Therefore, we need to make a choice so that the marginal distributions are related in a desirable way.

There are bunch of assumptions and constrains proposed to deal with this ill-posed problem. Some suggest to partially preserve the content of source domain data, such as pixel intensity, gradient and object boundary [2][3]; others propose to keep certain properties unchanged during the transfer, such as semantic features and class labels [4].

Gatys et al. [7] separate and recombine the content and style of images under the Convolutional Neural Network (CNN) representation. In the hierarchical network, high-level content information is stored in higher layers of the network, and low-level information such as texture and artistic style is captured by the correlations between filter responses in different CNN layers. They assume the representations of content and style in the Convolutional Neural Network are well separable. However, the style comes from only one image, not the entire training set. This limination prevents it from capturing the general theme of the target domain.

Thu et al. [10] proposed a very straightforward constraint called cycle-consistency. It assumes if we transfer a sample from the source domain to the target domain and then translate back, we should get exactly the same sample. Choi et al. [11] generalized it to perform multiple-domain translation using a single generative model. However, domain transfer is not a one-to-one mapping, but many-to-many. That's why the cycle-consistency constraint is too strong and leads to the lack of diversity in the translated outputs.

Based on the similar idea, Liu et al. [8] developed the UNIT framework by making a fully shared latent space assumption: corresponding images across domains can be mapped to the same latent code in a shared-latent space. This assumption implies the cycle-consistency constraint. Xun et al. [9] extended this idea to a partially shared latent space assumption, where each data sample is generated from a shared latent code for content and a domain-specific latent code for style. Images are translated across domains by reconstructing through encoder and decoder networks.

Another idea is to make assumptions over data distributions. Covariate shift [5] assumes unchanged conditional distributions $p(y^s|x^s)$, $p(y^t|x^t)$ and the only difference across domains exist in the input distributions. If the distributions share a common support, then importance weighting [6] can help to estimate the target density $\hat{p}(x^t) = \hat{w}(x)p(x^s)$, where $w(x) = p(x^s)/p(x^t)$ is estimated by minimizing the Kullback-Leibler divergence $KL(p(x^t), \hat{p}(x^t))$. The weighting parameters

correspond to the last layer of the decoder network and the first layer of the encoder network, which are low-level features representing the style.

Other approaches [11][12][13] assume there exist a transformation $\mathcal T$ so that the source and target 196 distributions can be matched with the new representations, $p(\mathcal{T}(x^s)) = p(\mathcal{T}(x^t))$. It requires to 197 minimize the distance between distributions, which has been discussed in chapter 3. This assumption 198 is equivalent to finding a transporation plan \mathcal{T} such that $p(\mathcal{T}(x^s)) = p(x^t)$. A particular solution is the 199 optimal plan with the minimum transportation cost, i.e., the Wasserstein distance between the source 200 and target distributions. As a by-product of this optimization problem, minimizing the transporation 201 cost is equivalent to matching the samples with common representations and labels, yielding better 202 knowledge transfer across domains. Under this idea, Damodaran et al. [14] propose to minimize 203 the Wasserstein distance between joint distributions of data-label pair $p(x^s, y^s), p(x^t, f(x^t))$, which 204 aligns data samples from source and target domains as well as transfers the discriminative information 205 to the classifier f in the target domain. 206

5.3 Related Work

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Learning generative models for domain transfer is an open problem. There are several topics closely related to our work, but with different definitions.

Deep Generative Modeling The original objective of this topic is to generate new samples from scratch by learning complicated data distributions in an unsupervised way. At test time, it takes random noise as input and outputs realistic samples. In some cases, it can also take in conditional information to produce user-specified output. There are three main frameworks of deep generative modeling: generative adversarial networks (GANs) [15], variational auto-encoders (VAEs) [16], and auto-regressive models [17].

GANs build the generative model on the top of a discriminative network to force the output to be indistinguishable from the real samples. This model works pretty well for generating images with impressive visual quality [18] and high resolution [19]. Variations under this framework include conditional GAN [20] that generate samples conditioned on class labels, LAPGAN [21] that generates images in a coarse-to-fine fashion, WGAN-GP [22] that enables stable training of GANs without hyperparameter tuning.

VAEs use an encoder-decoder framework to model data in a latent space and optimize the reconstruction loss plus a regularizer. The generative process has two steps of sampling: first draw latent variables from p(z) and then draw datapoints from the conditional distribution p(x|z). At test time, the encoder part is discarded and the decoder takes random noise as input to generate new samples. However, the reconstructed samples are blurry. This is because the VAE decoder assumes p(x|z) to be an isotropic Gaussian, which leads to the use of L2 loss. To remedy this, VAE-GAN [23] suggests learning the loss through a GAN discriminator.

Auto-regressive model is quite different from the above two. It aims at modeling time-varying processes by assuming that the value of a time series depends on its previous values and a stochastic term. For a sequential data sample $x=(x_1,x_2,\ldots,x_T)$, the joint distribution p(x) is factorised as a product of conditional distributions

$$p(x) = \prod_{t=1}^{T} p(x_t | x_1, \dots, x_{t-1})$$
(1)

This idea is quite straightforward for modeling audio sequence [24], but it also works for images. In PixelRNN [25], each image is written as a sequence, in which pixels are taken row by row from the image. The two-dimensional spatial autocorrelation of pixels is modeled by one-dimensional temporal correlations. Since the generation process is sequential, it requires a lot of GPU memory and computation time (200K updates over 32 GPUs) even after some modifications [26].

Image Style Transfer There are two types of style transfer problems: example-based style transfer where the style comes from one image, and domain-based style transfer where the style is learnt from a collection of images in a specific domain. The former problem originates from nonphoto-realistic rendering (NPR) [27] in computer graphic, and has the similar meaning of realistic image manipulation. The goal is to edit image in a user-specified way and keep it as realistic as possible.

Practical issues include texture synthesis and transfer [28], photo manipulation of shape and color [29], photorealistic image stylization [30], etc. In general, the output should be similar to the input in high-level structures and varies in low-level details such as color and texture.

Recently, Gatys et al. [31] claimed the image content and style information are separable in Convolutional Neural Network representations. They introduced a method [32] to separate and recombine content and style of natural images by matching feature correlations (Gram matrix) in different convolutional layers. However, their synthesis process is slow (an hour for a 512*512 image). Moreover, the style from a single image is ambiguous and may not capture the general theme of an entire domain of images.

The second problem, also named as image-to-image translation, learns a mapping to transfer images from one domain to another. For example, super-resolution [39] maps low-dimentional images to high-dimention, colorization [40] maps gray images to color; other cases include day to night, dog to cat, yong to old, summer to winter, photographs to paintings, aerial photos to maps [30,34,35,36,37,38,41]. The mapping can be learnt in a supervised or unsupervised manner. In supervised settings [33,42,43], corresponding image pairs across domains are available for training. In unsupervised settings [2,4,8,9,10], there's no paired data and the training set only contains independent set of images for each domain. Our work is under the unsupervised settings because it is more applicable, and the training data is almost free and unlimited.

Domain Adaptation Most recognition algorithms are developed and evaluated on the same data distributions, e.g, the public datasets ImageNet, MS-COCO, CIFAR-10, MNIST. In real applications, people often confront performance degradation when apply a classifier trained on a source domain to a target domain.

In unsupervised domain adaptation, source domain has labeled data $\mathcal{D}_s = \{x_i^s, y_i^s\}_{i=1}^{n_s}$ while target domain contains data without labels $\mathcal{D}_t = \{x_i^t\}_{i=1}^{n_t}$. The goal is to learn a classifier $f: x_i^t \mapsto y_i^t$ for the unseen target samples by exploring the knowledge learnt from the source domain. Domain adaptation algorithms attempts to transfer knowledge across domains by solving the domain shift problem, i.e., the data-label distributions $p(x^s, y^s)$ and $p(x^t, y^t)$ are different.

There are many approaches to address this issue. One is to extract transferable features that are 270 invariant across domains [45,46], or learn representative hash codes [47] to find a common latent 271 space where the classifier can be used without considering the data's origin. Another trend is to learn the transformation between domains [48] to align the source and target datapoints through barycentric mapping, and train a classifier on the transferred source data. Courty [49] and Damodaran[14] proposed to look for a transformation that matches the data-label joint distributions $p(x^s, y^s)$ in 275 source domain to its equivalent version $p(x^t, y^t)$ in target domain. The predictive function f is learnt 276 by minimizing the optimal transport loss between the distributions $p(x^s, y^s)$ and $p(x^t, f(x^t))$. As a 277 by-product, minimizing the optimal transport cost is equivalent to mapping a source domain sample 278 to a target domain sample with similar semantic content, and this is the domain transfer problem. 279

Voice Conversion Voice conversion (VC) aims to change a speaker's voice to make it sounds like spoken by another person. It is a special case of voice transformation (VT), whose goal is to modify human speech without changing its content. VC transforms speacker identity by replacing speaker-dependent components of the signal while maintaining the linguistic information. Speech quality and speaker similarity are two important factors to evaluate a VC system. There are a bunch of VC applications, such as movie dubbing, personalized TTS (Text To Speach) systems, speaker accent or emotion transformation, speaking-aid devices, call quality enhancement, etc.

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Most VC frameworks involve three steps: feature extraction, feature conversion, waveform generation. In speech analysis, waveform signals are encoded into feature representations that are easy to control and modify. Spectral envelope, mel-cepstrum, fundamental frequency (f0), formant frequencies and bandwidths are the most widely used features to represent speech in short-time segments. To capture contextual information across frames, implicit methods such as hidden Markov models (HMMs), Long Short-Term Memory (LSTM) and recurrent neural networks (RNNs) [63] were developed.

The main work in VC is to transform the source feature sequences to target feature sequences that capture the speaker identity. Most traditional VC systems perform frame-by-frame mapping under the assumption that speech segments are independent from each other. Some recent models such as HMM and RNN incorporate speech dynamics implicitly. There are four typical approaches to learn the

mapping function: codebook mapping (e.g., Vector quantization (VQ) [56]), mixed linear mappings (e.g., Gaussian mixture model (GMM) [57]), neural network mapping (e.g., RBM, DNN, RNN [55]), and exemplar-based mapping (e.g., non-negative matrix factorization (NMF) [58]). Beyond these, an autoregressive neural network model called WaveNet [24] was proposed. It can directly learn the mapping based on raw audio and generate speech waveforms conditioning on the speaker identity.

There are various assumptions in speech analysis and waveform generation. Source-filter models assume speech to be generated by excitation signals passing through a vocal tract, and encode speech waveforms as acoustic features that represent sound source and vocal tract independently. However, the original phase information will lose under this assumption. At conversion time, the converted target features are passed through a vocoder based on the source filter model to reconstruct the waveform. Quality degradation may happen due to the inaccurate assumption. Iterative phase reconstruction algorithm Griffin-Lim [59] was adopted to aleviate this issue. Harmonic plus noise models (HNM) [54] assume speech to be a combination of a noise component and a harmonic component, i.e., sinusoidal waves with frequencies relevant to pitch. Speech is parameterized by the fundamental frequency f_0 and a spectrum which consists of a lower band of harmonic and a higher band of noise. Other assumptions include stationary speech signal, frame-by-frame mapping, time-invariant linear filter, etc. Recently, Tamamori et. al [53] proposed a speaker-dependent WaveNet vocoder that does not require explicit modeling of excitation signals and those assumptions.

Software: Merlin baseline system, sprocket baseline system, STRAIGHT, WORLD, Griffin-Lim, Speech Signal Processing Toolkit (SPTK)

In terms of conversion conditions, VC can be categorized into parallel and non-parallel, text-dependent and text-independent systems [50]. In parallel systems, the training corpus consists of paired recodings from the source and target spearkers with same liguistic contents. The shared acoustic features can be used to train the mapping model. To get parallel feature sequences of equal length, a time-alignment step must be included to remove the temporal differences in the recordings, for example, the dynamic time warping (DTW) [56] algorithm. Phoneme transcriptions are also useful for time alignment. Non-parallel system does not require sentences with the same linguistic contents. It is much more useful and practical because non-parallel speech data is easier to collect and therefore can get larger training sets. There are several ways to learn the mapping without paired data: (1) use unit selection method [60] to choose matched linguistic feature pairs; (2) build pseudo parallel sentences using an extra automatic speech recognition (ASR) module [61]; (3) extract speaker-independent features in a shared latent space [62]; (4) use unpaired image-to-image translation approaches such as CycleGAN [10].

Parallel, text-dependent systems are supposed to have better performancee. However, parallel utterance pairs are difficult to get. Most parallel VC systems require time alignment to extract parallel source-target features. The misalignment in automatic time alignment algorithms often leads to degradation in speech quality, while manual correction is arduous. Recently, the winner of VC Challenge 2018 [51] showed their algorithm can achieve similar results in both parallel and non-parallel settings. It first uses a lot of external speech data with phonetic transcriptions to train a speaker-independent content-posterior-feature extractor, followed by a speaker-dependent LSTM-RNN to predict fundamental frequency f_0 and STRAIGHT spectral features [52], and then reconstruct the waveforms with a speaker-dependent WaveNet vocoder [53]. Moreover, Kaneko et.al [64] and Fang et. al [54] claimed their nonparallel, text-independent VC algorithms based on CycleGAN [10] perform comparable to or better than the state-of-the-art parallel approaches.

341 Music Style Transfer

5.4 Method

- 343 5.5 Case Study: Image Style Transfer
- 5.6 Case Study: Emotional Voice Conversion

345 6 Dissertation Outline

The research will be split into the following four stages:

347 6.1 Introduction

348 6.2 Related Work

349 6.3 Distributionally Robust Games

In this part we introduce distributionally robust games and develop new filtering and learning 350 architectures under this framework. The system may contain several competing neural networks: the 351 attackers learn to generate synthetic samples that are supposed to have the same distribution as the 352 original ones, while the defenders try to find counter-examples and create difficulties for the other 353 side. Each player tries to perform better and beat the others, which forms a multi-agent zero-sum 354 game with uncertain payoffs. The players use a robust optimization approach to contend with the 355 worst-case scenario payoff. The attacker network is constructed based on the outcome of defender 356 357 networks, and vice versa. The competing networks are trained together iteratively until achieving the distributional robust Nash equilibrium. 358

359 6.4 Wasserstein Metric

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The loss function is designed to measure the similarity of two probability distributions. Unsupervised learning is conducted by minimizing the loss. We plan to study the properties of several widely used loss metrics:

- Compare L1, L2-loss, KL-divergence, f-divergence, and Wasserstein distance
- Study the time-dependent formulation of the optimal transportation cost
- Test the effect of translation and perturbation for a certain loss metric

366 6.5 Learning Algorithms for Robust Optimization

- Develop a specific learning algorithm to find robust Nash equilibria, which should be stable and efficient
 - Compare with existing numerical optimization approaches in large-scale machine learning: SGD, Adam, Momentum, Ishikawa-Nesterov, Newton's method, conjugate gradient, natural gradient, etc.
- Compare with existing deep generative models: RBM, VAE, GAN, WGAN, etc.

373 6.6 Generative Modeling for Vehicle Tracking

- 374 6.7 Generative Modeling for Image Synthesis
- 375 6.8 Generative Modeling for Voice Conversion

376 6.9 Experiments on Large-scale Machine Learning datasets

- Maryland Traffic Surveillance Dataset (Vehicle tracking)
 - Large-scale CelebFaces Attributes Dataset (CelebA, image synthesis)
- Large-scale Scene Understanding Challenge (LSUN, image synthesis)
- Interactive Emotional Dyadic Motion Capture (IEMOCAP, emotional voice conversion)

381 6.10 Discussion

382 6.11 Conclusion and Future Work

7 Research Plan

384 7.1 Research Progress

- Literature review, planning
- Theory part on distributional robust games, Bregman learning and convex optimization

- Theoretical analysis and comparison for L2 distance, f-divergence and Wasserstein metric
- Algorithm design, overall integration, simulations, specific implementations on real problems
- Application part on large-scale machine learning: experiments, evaluation and revision
- Documentation and Defence

392 7.2 Application on Image and Audio Synthesis

- Test on large-scale image dataset MNIST, CelebA and LSUN
- Literature review on emotional speech classification and audio synthesis
- Compare two sound representations in generative learning: waveform and spectrogram
- Design deep generative models for emotional speech generation
 - Test on voice conversion or music style transfer if possible

398 **7.3 Timeline**

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Fall 2018	Study on Generative Models for Voice Conversion
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Spring 2019 Write Thesis
August 2019 Thesis Defense

399 8 Conclusion

- Tackling the aforementioned problems would take us much closer to real intelligent systems, and
- defines three core pillars of Artificial Intelligence. However, there are many other problems which
- 402 need to be solved and integrated to achieve a fully intelligent system, e.g. navigation, learning by
- imitation, cooperation, and many others.

404 9 List of Publications

9.1 Thesis Related Publications

- Jian Gao and Hamidou Tembine, Distributionally Robust Games for Deep Generative Learning, July 2018. DOI: 10.13140/RG.2.2.15305.44644
- Jian Gao, Yida Xu, Julian Barreiro-Gomez, Massa Ndong, Michalis Smyrnakis and Hamidou
- Tembine (September 5th 2018) Distributionally Robust Optimization. In Jan Valdman,
- 410 Optimization Algorithms, IntechOpen. DOI: 10.5772/intechopen.76686. ISBN: 978-1-
- 411 78923-677-4
- Jian Gao and Hamidou Tembine, Distributionally Robust Games: Wasserstein Metric,
- International Joint Conference on Neural Networks (IJCNN), Rio de Janeiro, Brazil, July
- 414 2018
- Jian Gao and Hamidou Tembine, Bregman Learning for Generative Adversarial Networks,
- Chinese Control and Decision Conference (CCDC), Shenyang, China, June 2018 (Best
- 417 Paper Finalist Award)
- Jian Gao and Hamidou Tembine, Distributed Mean-Field-Type Filter for Vehicle Tracking,
- in American Control Conference (ACC), Seattle, USA, May 2017 (Student Travel Award)
- Dario Bauso, Jian Gao and Hamidou Tembine, Distributionally Robust Games: f-Divergence
- and Learning, 11th EAI International Conference on Performance Evaluation Methodologies
- and Tools (VALUETOOLS), Venice, Italy, Dec 2017

9.2 Other Publications

- J. Gao and H. Tembine, "Distributed Mean-Field-Type Filters for Traffic Networks," in IEEE Transactions on Intelligent Transportation Systems. doi: 10.1109/TITS.2018.2816811
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