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Abstract—This paper studies fast downlink beamforming algorithms using deep learning in multiuser multiple-input-single-output systems where each transmit antenna at the base station has its own power constraint. We focus on the signal-to-interference-plus-noise ratio (SINR) balancing problem which is quasi-convex but there is no efficient solution available. We first design a fast subgradient algorithm that can achieve near-optimal solution with reduced complexity. We then propose a deep neural network structure to learn the optimal beamforming based on convolutional networks and exploitation of the duality of the original problem. Two strategies of learning various dual variables are investigated with different accuracies, and the corresponding recovery of the original solution is facilitated by the subgradient algorithm. We also develop a generalization method of the proposed algorithms so that they can adapt to the varying number of users and antennas without re-training. We carry out intensive numerical simulations and testbed experiments to evaluate the performance of the proposed algorithms. Results show that the proposed algorithms achieve close to optimal solution in simulations with perfect channel information and outperform the alleged theoretically optimal solution in experiments, illustrating a better performance-complexity tradeoff than existing schemes.

Index Terms—Deep learning, beamforming, multiple-input-single-output (MISO), signal-to-interference-plus-noise ratio (SINR) balancing, per-antenna power constraints.

MULTIUSER multi-antenna techniques (or multiple-input multiple-output, MIMO) techniques can significantly improve the spectral and energy efficiency of wireless communications by exploiting the degree of freedom in the spatial domain. They have been widely adopted in modern wireless communications systems such as the fourth and the fifth-generation (4G and 5G) of cellular networks [1], [2], the high efficiency wireless local area (WiFi) networks standard 802.11ax [3], and the latest satellite digital video broadcasting standard DVB-S2X [4]. Among the multiuser MIMO techniques, beamforming is one of the most promising and practical schemes to mitigate multiuser interference and exploit the gain of MIMO antennas.

In the last two decades, the optimal beamforming strategies have been intensively studied for the multiple-input single-output (MISO) downlink where a base station with multiple antennas serves multiple single-antenna users. For instance, the problem of signal-to-interference-plus-noise ratio (SINR) balancing or maximization of the minimum SINR of all users, under a total power constraint was studied in [5], [6], the total BS transmit power minimization problem under quality of service (QoS) constraints was investigated in [7]–[10], and the sum rate maximization problem under the total power constraint was tackled in [6], [11]–[13]. The existing approaches mainly make use of the advances of convex optimization techniques such as second-order cone programming (SOCP) [8], [9] and semidefinite programming (SDP) [14], and the uplink-downlink duality which indicates that under the sum power constraint, the achievable SINR region and the normalized beamforming in the downlink are the same as those in the dual uplink channel.

Early works mostly focus on the optimal beamforming design under the sum power constraint across all antennas of a transmitter. This constraint does not take into account the fact that each transmit antenna has its own power amplifier, and therefore its power is individually limited. The per-antenna power constraints were first systematically studied in [15] where a dual framework was proposed to minimize the...
maximum transmit power of each antenna under users’ SINR constraints. This work has sparked much research interest in optimizing beamforming under per-antenna power constraints. The work in [16] studied the optimization of the nonlinear zero forcing (ZF) dirty paper coding based beamforming under per-antenna power constraints. Generic optimization of beamforming for multibeam satellite systems was studied in [17] under general linear and nonlinear power constraints. The per-antenna constant envelope precoding for large multiuser MIMO systems was investigated in [18]. The transceiver designs for multi-antenna multi-hop cooperative communications under per-antenna power constraints were proposed in [19] and both linear and nonlinear transceivers were investigated. The signal-to-leakage-plus-noise ratio (SLNR) maximized precoding for the downlink under per-antenna power constraints was considered in [20] where a semi-closed form optimal solution was proposed. A general framework for covariance matrix optimization of MIMO systems under different types of power constraints was proposed in [21]. More recently, the optimal MIMO precoding under the constraints of both the total consumed power constraint and the individual radiated power constraints was studied in [22] and numerical algorithms were developed to maximize the mutual information.

The problem of interest in this paper is to efficiently maximize the minimum received SINR or to balance SINR, in the multiuser MISO downlink under per-antenna power constraints at the BS. This problem, although being quasiconvex, is more challenging than the counterpart with the total power constraint and the problem of minimizing the per-antenna power in [15], and until now there does not exist efficient algorithms. Consequently, existing beamforming techniques are unable to support real-time applications because the small-scale fading channel varies considerably fast. For instance, in a WLAN 802.11n system operating at 2.4 GHz with a pedestrian speed of 1.4 m/s, the coherence time is 89 ms; and in a Long-Term Evolution (LTE) downlink operating at 2.6 GHz with a residential area vehicle velocity of 10 m/s, the coherence time is only 11.5 ms. Traditional time-consuming optimization routines will produce obsolete beamforming solution that is not timely for the current channel state and lead to significant performance degradation which will be demonstrated in our experiment. In [23], the dual problem was derived and the optimal solution at much reduced computational cost was developed. However, it was found out that the best solution is obtained by a commercial nonlinear solver [24], which does not explore the structure of the problem and is still not efficient. Although there are simple heuristic beamforming solutions which have closed-form solutions such as the ZF beamforming and the regularized ZF (RZF) beamforming, the reduced complexity often leads to performance loss. Even worse, the work in [25] showed that the conventional ZF beamforming under per-antenna power constraints no longer admits a simple pseudo-inverse form as the case under the total power constraint, and instead the optimal ZF beamforming requires solving an SOCP problem which has much higher complexity.

In this paper, we take a different approach and develop deep learning (DL) enabled beamforming solutions to dramatically improve the computational efficiency. Recently DL has been recognized as a promising solution for addressing various problems in several areas of wireless networks. This is because deep neural networks have the ability to model highly nonlinear functions at considerably low complexity. One of the areas of interest is to deal with scenarios in which the channel model does not exist, e.g., in underwater and molecular communications [26] or is difficult to characterize analytically due to imperfections and nonlinearities [27]. In these situations, DL based detection has been proposed to tackle the underlying unknown nonlinearities [28]. Another area of interest is to optimize the end-to-end system performance which is the focus of our paper. In this aspect, DL has found many exciting applications in wireless communications such as channel decoding [31], [32], MIMO detection [33], [34], channel estimation [35], [36]. The current work belongs to the framework of learning to optimize in wireless resource allocation. The rationale is that the DL technique bypasses the complex optimization procedures, and learns the optimal mapping from the channel state to produce the beamforming solution directly by training a neural network. The result is that the trained neural network can be used as a function mapping to obtain the real-time beamforming solution with channel state as input. As a result, the computational complexity is transferred to offline training phase, and hence the complexity during the online transmission phase is greatly reduced. The mostly successful applications of DL in this framework far is power allocation [37]–[41], in which the power vector is treated as the training output, while the channel gains are taken into the input of the DL network. In this case, the power variables only take positive values and the number of power variables is normally the number of users and therefore relatively small and easy to handle. However, there are few works that focus on the learning approach to optimize the beamforming design in multi-antenna communications, with the exception of [42]–[47]. The difficulty is partly due to the large number of complex variables contained in the beamforming matrix that need to be optimized. An outage-based approach to transmit beamforming was studied in [42] to deal with the channel uncertainty at the BS, however, only a single user was considered. The work

1To the best of our knowledge, the computational complexity of the training phase is not well understood, due to the complex implementation of the backpropagation process and that it depends very much on the specific application regarding the required number of training examples for satisfactory generalization. That said, this is usually not a concern in most applications because training takes place offline given sufficient computational capability and retraining is only performed infrequently when the specific applications depart considerably from those training examples.
in [43] designed a decentralized robust precoding scheme based on a deep neural network (DNN). The projection over a finite dimensional subspace in [43] reduced the difficulty, but also limited the performance. A DL model was used in [44] to predict the beamforming matrix directly from the signals received at the distributed BSs in millimeter wave systems. However, both [43] and [44] predicted the beamforming matrix in the finite solution space at the cost of performance loss. The works in [42], [45] directly estimated the beamforming matrix without exploiting the problem structure in which the number of variables to predict increases significantly as the numbers of transmit antennas and users increase. This will lead to high training complexity and low learning accuracy of the neural networks when the numbers of transmit antennas and users are large. In our previous works [46], [47], we proposed a beamforming neural network to optimize the beamforming vectors, but it is restricted to the total power constraint. We notice that none of existing works addressed the SINR balancing problem under the practical per-antenna power constraints, for which DL solution becomes even more attractive.

In this paper, we propose a DL enabled beamforming optimization approach for SINR balancing to provide an improved performance-complexity tradeoff under per-antenna power constraints. Inspired by the model driven learning philosophy [48], we propose to first learn the dual variables with reduced dimension rather than the original large beamforming matrix and then recover the beamforming solution from the learned dual solution, by exploiting the structure or model of the beamforming optimization problem. Our main contributions are summarized as follows:

- A subgradient algorithm is first proposed which not only demonstrates faster convergence than the best known algorithm in [23], but also facilitates the development of the DL solutions.
- A general DL structure to learn the dual variables is proposed, and two learning strategies are proposed to achieve the performance-complexity tradeoff. A heuristic method is developed to facilitate the generalization of the proposed DL algorithms by augmenting the training set so that they can adapt to the varying number of active users and antennas without re-training.
- Both software simulations and testbed experiments using software defined radio (SDR) are carried out to validate the performance of the proposed algorithms. To the best of our knowledge, this is the first testbed demonstration of deep learning enabled multiuser beamforming.

The remainder of this paper is organized as follows. Section II introduces the system model and formulates the SINR balancing problem and its dual formulation. Section III proposes the subgradient algorithm. Section IV provides the general structure framework for the beamforming optimization based on learning the dual variables and the recovery algorithms. Numerical and experimental results are presented in Section V. Finally, conclusion is drawn in Section VI.

Notations: The notations are given as follows. Matrices and vectors are denoted by bold capital and lowercase symbols, respectively. \((\cdot)^T\), \((\cdot)^*\), \((\cdot)^{\dagger}\) and \((\cdot)^{-1}\) stand for transpose, conjugate transpose and inverse/pseudo inverse (when applicable) operations of a matrix, respectively. \(\mathbf{A} \succ 0\) indicates that the matrix \(\mathbf{A}\) is positive definite. The operator \(\text{diag}(\mathbf{a})\) denotes the operation to diagonalize the vector \(\mathbf{a}\) into a matrix whose main diagonal elements are from \(\mathbf{a}\). Finally, \(\mathbf{a} \sim \mathcal{CN}(0, \mathbf{Σ})\) represents a complex Gaussian vector with zero-mean and covariance matrix \(\mathbf{Σ}\). \(\mathbf{Z}\) denotes the non-negative field.

II. SYSTEM MODEL AND PROBLEM FORMULATION

Consider an MISO downlink channel where an \(N_t\)-antenna BS transmits signals to \(K\) single-antenna users. For the user \(k\), its channel vector, beamforming vector, and data symbol are denoted as \(\mathbf{h}_k^T, \mathbf{w}_k, s_k\), respectively, where \(E(|s_k|^2) = 1\). The additive white Gaussian noise (AWGN) at the received is denoted as \(n_k \sim \mathcal{CN}(0, N_0)\). All wireless links exhibit independent frequency non-selective Rayleigh block fading. The received signal at user \(k\) is

\[
y_k = \mathbf{h}_k^T \sum_{i=1}^{K} \mathbf{w}_i s_i + n_k.
\]

The SINR at the receiver of user \(k\) is given by

\[
\gamma_k = \frac{\left| \mathbf{h}_k^T \mathbf{w}_k \right|^2}{\sum_{i=1, i \neq k}^{K} \left| \mathbf{h}_k^T \mathbf{w}_i \right|^2 + N_0}.
\]

The beamforming matrix is collected in \(\mathbf{W} = [\mathbf{w}_1, \mathbf{w}_2, \ldots, \mathbf{w}_K] \in \mathbb{C}^{N_t \times K}\). Then the per-antenna power at antenna \(n\) can be expressed as

\[
p_n = \|\mathbf{W}(n,:)\|^2 = \|e_n^T \mathbf{W}\|^2,
\]

where \(e_n\) is a zero vector except its \(n\)-th element being 1.

The problem of interest is to maximize the minimum user SINR, i.e., SINR balancing, under per-antenna power constraints \(\{P_n\}\). Mathematically, it can be formulated as follows:

\[
P_1: \max_{\mathbf{W}, \Gamma} \Gamma
\]

s.t. \(\gamma_k = \frac{\left| \mathbf{h}_k^T \mathbf{w}_k \right|^2}{\sum_{i=1, i \neq k}^{K} \left| \mathbf{h}_k^T \mathbf{w}_i \right|^2 + N_0} \geq \Gamma, \ \forall k\),

\[
p_n = \|e_n^T \mathbf{W}\|^2 \leq P_n, \ \forall n.
\]

The SINR balancing problem is in general quasi-convex, so it can be solved via methods such as bisection search and generalized eigenvalue programming [8], [23]. However, these methods suffer from high complexity and computational delay, and are not practical for real-time data transmissions.
In [23], a useful dual formulation of P1 is derived as

\[
P2: \max_{\beta, \lambda, \mu} \beta
\]

\[
s.t. \beta \lambda_k h_k^T G(\lambda, \mu)^{-1} h_k^* \leq 1, \quad \forall k,
\]

\[
\sum_{i=1}^K \lambda_i = 1,
\]

\[
\sum_{n=1}^{N_t} \mu_n P_n = 1,
\]

\[
\lambda, \mu, \beta \geq 0.
\]

where \( G(\lambda, \mu) \triangleq \sum_{i=1}^K \lambda_i h_i^* h_i^T + \text{Diag}(\mu), \lambda \in \mathbb{Z}^K, \mu \in \mathbb{Z}^{N_t} \) are dual variables associated with the SINR constraint (4) and the per-antenna power constraint (5) in P1, and \( \beta \) is related to the minimum SINR \( \Gamma \) in P1 by the relation \( \beta = 1 + \frac{1}{\Gamma}. \) For the solution of P2, it is assumed that \( G(\lambda, \mu) > 0. \)

Although the problem P2 is still a quasi-convex problem, compared to the original problem P1, it can be more efficiently solved because it only involves \( K + N_t + 1 \) non-negative variables while P1 needs to optimize \( 2KN_t \) real variables. The problem P2 can be solved using standard nonlinear solvers such as Matlab’s built-in function ‘fmincon’. Currently the fastest optimal solution is known to be achieved by Ziena’s nonlinear solver Knitro [24], which is compared and shown in [23]. However, the general solvers do not exploit the special analytical properties of the problem P2, so they are not efficient. In addition, it is not known how to recover the optimal solution to the beamforming matrix \( W \) once P2 is solved. These issues will be studied in the next section.

III. A SUBGRADIENT ALGORITHM TO SOLVE P2

In this section, we derive a fast subgradient algorithm to solve P2, based on the downlink-uplink duality results derived in [15]. According to [15, Theorem 1], the problem P2 can be equivalently written as the following max-max problem:

\[
P3: \max_{\mu, \Gamma, \lambda} \Gamma
\]

\[
s.t. \max_{w_k} \lambda_k |w_k^* h_k^T|^2 \geq \Gamma, \quad \forall k,
\]

\[
\sum_{k=1}^K \lambda_k = 1,
\]

\[
\sum_{n=1}^{N_t} \mu_n P_n = 1,
\]

\[
\lambda, \mu, \Gamma \geq 0.
\]

P3 can be interpreted as the maximization of the minimum user SINR in the virtual uplink in which \( K \) single-antenna users transmit signals to the BS with the total power constraint \( \frac{1}{N_0}. \) The uncertain covariance matrix of the received noise vector is characterized by \( \text{Diag}(\mu). \) The normalized receive beamforming at the BS for user \( k \) is denoted by \( \bar{w}_k = \frac{w_k}{\|w_k\|} \) which has the same direction as the downlink transmit beamforming, while \( \lambda_k \) denotes the uplink transmit power of user \( k. \) Because the covariance matrix of the received noise vector \( \text{Diag}(\mu) \) is also a variable, P3 is still difficult to solve. To tackle this problem, we first keep the variable \( \mu \) fixed, and then reach the sub-problem below:

\[
P4: f(\mu) = \max_{\Gamma, \lambda} \Gamma
\]

\[
s.t. \max_{w_k} \lambda_k |w_k^* h_k^T|^2 \geq \Gamma, \quad \forall k,
\]

\[
\sum_{k=1}^K \lambda_k = 1,
\]

\[
\lambda, \Gamma \geq 0.
\]

P4 can be interpreted as the nonlinear SINR balancing problem with a total power constraint and colored noise with covariance matrix \( \text{Diag}(\mu). \) In the following, we propose an efficient fixed-point iteration in Algorithm 1 below to solve P4.

Algorithm 1 to Solve P4:

1) Initialize \( \lambda \) that satisfies \( \sum_{k=1}^K \lambda_k N_0 = 1. \) Suppose \( j \) is the iteration index, and the achievable SINR in the uplink is \( \gamma^{(j)}. \) Repeat the following steps 2)-5) until convergence.

2) For each \( k, \) define \( G_k(\lambda, \mu) \triangleq \sum_{i=1, i \neq k}^K \lambda_i h_i^* h_i^T + \text{Diag}(\mu). \)

3) Solve an auxiliary variable \( \bar{\lambda}_k \) as

\[
\bar{\lambda}_k = \mathcal{I}_k(\gamma^{(j-1)} \triangleq \gamma^{(j)} \frac{1}{h_k^T G_k(\lambda, \mu)^{-1} h_k^*}, \quad \forall k.
\]

4) Normalize \( \{\lambda_k\} \) to obtain \( \{\bar{\lambda}_k\} \) as:

\[
\lambda_k = \bar{\lambda}_k \eta, \quad \text{where} \quad \eta = \frac{1}{\sum_{k=1}^K \lambda_k N_0}.
\]

5) Calculate \( \beta_k = \frac{\lambda_k h_k^* G_k(\lambda, \mu)^{-1} h_k^*}{\|G_k(\lambda, \mu)^{-1} h_k^*\|^2}. \) Then update the achievable SINR in the uplink as

\[
\gamma^{(j)} = \min_k \frac{1}{\beta_k}.
\]

It can be proved that Algorithm 1 converges to the optimal solution of P4. The proof is similar to [53, Theorem 11.1] and a refined version is provided in Appendix A for completeness.

The optimal uplink beamforming for a given \( \mu \) can be derived according to the minimum mean square error (MMSE) criterion:

\[
\bar{w}_k = \frac{G_k(\lambda, \mu)^{-1} h_k^*}{\|G_k(\lambda, \mu)^{-1} h_k^*\|}, \quad \forall k.
\]

With the inner maximization problem P4 solved for given \( \mu, \) we can obtain the objective function value \( f(\mu). \) Next we solve the outer maximization of \( \mu \) using a subgradient projection algorithm, where the subgradient can be found using the downlink beamforming obtained from the normalized uplink beamforming.

As proved in Appendix B, \( f(\mu) \) is a concave function in \( \mu. \) A subgradient of \( \mu_k \) can be expressed as \( \|e_k^T W \|^2 \) because \( \mu_k \) is the dual variable associated with the \( n \)-th antenna power constraint. The proof is omitted. Based on this result, we propose the following subgradient based algorithm [15, 49] to solve P3.
Algorithm 2 to Solve P3:

1) Initialize $\mu$. Suppose $j$ is the iteration index. Repeat the following steps 2)-7) until convergence.
2) Given $\mu$, call Algorithm 1 to find the optimal $\tilde{\lambda}$.
3) Calculate $\beta_k = \frac{\lambda_k h_k^T G_k(\tilde{\lambda}, \mu)^{-1} h_k^*}{|G_k(\tilde{\lambda}, \mu)^{-1} h_k^*|}$. Then update the achievable SINR in the uplink as
   \[ \gamma^{(j)} = \min_k \frac{1}{\beta_k - 1}. \] (16)
4) Find the optimal normalized uplink beamforming
   \[ \bar{w}_k = \frac{G_k(\tilde{\lambda}, \mu)^{-1} h_k^*}{||G_k(\tilde{\lambda}, \mu)^{-1} h_k^*||}, \forall k. \] (17)
5) Find the downlink power $\{p_k\}$ to achieve the SINR $\gamma^{(j)}$, i.e., to solve the following linear equation set:
   \[ \sum_{k=1, k \neq j}^K p_k |h_k^T \bar{w}_k|^2 = \gamma^{(j)}, \quad k = 1, \ldots, K. \] (18)
6) Update the downlink beamforming vector as $w_k = \sqrt{p_k} \bar{w}_k$ and $W = [w_1, \ldots, w_K]$.
7) Update $\mu$ using the subgradient Euclidean projection method with step size $\alpha_j$:
   \[ \mu^{(j+1)} = P_S(\mu_j^{(j)} + \alpha_j \text{Diag}\{||e_n^T W||^2\}), \] (19)
   where $S = \{\mu| \sum_{n=1}^{N_t} \mu_n P_n = 1\}$.
   This projection $P_S$ can be solved efficiently using the bisection search. The detailed projection algorithm is provided in Algorithm 3 of Appendix C.
8) Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:
   \[ W(n,:) = W(n,:) \sqrt{\min_n \frac{P_n}{||e_n^T W||^2}}, \forall n. \] (20)

Remarks: The subgradient algorithm exploits the structure of the original problem $P1$, so it is more efficient than a general nonlinear solver. However, the step size $\alpha_j$ is a critical parameter. We find that $\alpha_j = 0.01 \times 2^{-j}$ gives a satisfactory performance. We observe that in general Algorithm 2 can solve $P3$ faster than the available numerical solver such as Knitro and achieve close to optimal performance. It may only converge to the neighbourhood of the optimal solution, and its convergence may be slow, as seen in Fig. 1(a). In addition, the subgradient algorithm may not guarantee that the per-antenna power constraints will be satisfied, and that is why Step 8 of Algorithm 2 is necessary to regulate the per-antenna power.

Another important implication of the development of Algorithm 2 is that it provides an efficient way to recover the primal variable, i.e., the downlink beamforming vectors, given various dual variables either $\mu$ and $\tilde{\lambda}$, or only $\mu$. The details will be given in the next section.

IV. THE PROPOSED DEEP LEARNING STRUCTURE AND STRATEGIES

In this section, we develop DL based solutions of $P1$ that can achieve better performance-efficiency tradeoff than the currently available solutions. Instead of learning to optimize the original beamforming matrix $W$ directly, we will learn the optimization of the dual variables in $P2$. This will dramatically reduce the number of variables that need to be learned. In the sequel, we will first introduce a general DL structure that takes the channel $h = [h_1^T, \ldots, h_K^T]^T$ as the input, and the output is the dual variable(s) in $P2$. We will also devise a generalized learning solution such that the proposed DL structure can deal with varying number of users and antennas and transmit power without re-training. We will then propose two learning strategies, i.e., one is to learn the dual variables $\mu$ and $\tilde{\lambda}$ with fast recovery of the original beamforming solution, and the other is to learn only the dual variable $\mu$ with improved learning accuracy, to achieve various tradeoffs.

A. A General DL Structure

We first show the existence of a neural network that can approximate the solution of the optimization problem $P2$. To this end, we define $\mu^{opt}$ and $\tilde{\lambda}^{opt}$ as two tensors with the...
optimized dual variables $\mu$ and $\lambda$, respectively. The neural network aims to learn the continuous mapping

$$F(h, \mu^0) = \{\mu^{opt}, \lambda^{opt}\},$$

(21)

where $\mu^0$ is the initialization set of dual variables and $F(\cdot, \cdot)$ denotes the continuous mapping process in Algorithm 2 to achieve the stationary point from the input set of channel coefficients together with the initialization set of dual variables. The following theorem will prove the existence of a feedforward network which imitates the continuous mapping in (21).

**Theorem 1:** For any given accuracy $\varepsilon > 0$, there exists a positive constant $L$ large enough such that a feedforward neural network with $L$ layers can produce similar performance to the mapping process in (21), i.e.,

$$\sup_{h, \psi} \|\text{NET}_L(h, \psi) - F(h, \mu^0)\|_F \leq \varepsilon,$$

(22)

where $\psi$ is the set of the neural network parameters including weights and biases.

**Proof:** The result in Theorem 1 can be obtained directly by applying the universal approximation theorem in [50] to the continuous mapping in Algorithm 2. \qed

Based on results in Theorem 1, next we find solutions through designing the neural networks with the DL technique. Similar to our previous work [47], we introduce a general DL structure to approximate the mapping function from the channel coefficients to the beamforming solutions, as shown in Fig. 2. In addition to the conventional neural network module, the adopted DL structure also introduces a signal processing module based on expert knowledge for beamforming recovery from the key features, such as the dual variables $\lambda$ and $\mu$ in problem P2. Predicting the beamforming matrix directly may lead to high complexity since the number of the variables in the beamforming matrix depends on both the number of users $K$ and the number of BS antennas $N_t$. Thus instead of predicting the beamforming matrix directly, we predict some key features (i.e., the dual variables $\mu$ and $\lambda$) whose variables are much less than those in the beamforming matrix. Then these key features are used to recover the beamforming matrix in the signal processing module.

The adopted DL structure takes the convolutional neural network (CNN) architecture as the backbone because the parameter sharing adopted in the CNN can reduce the number of the learned parameters when compared to a fully-connected DNN. Moreover, CNN is well known to be effective for extracting features, which will benefit the generation of the beamforming solution using the channel features. The adopted DL structure includes two main modules: the neural network module and the signal processing module [51]. Here we give a short description about the two modules, and for more details readers are referred to [47].

1) **Neural Network Module:** The neural network module is a data-driven approach to approximate the mapping function from the complex channels to the key features. In addition to the input and output layers, the neural network module also includes convolutional (CL) layers, batch normalization (BN) layers, activation (AC) layers, a flatten layer, and a fully-connected (FC) layer. The input of the neural network module is the complex channel coefficients, which are not supported by the current neural network software. To address this issue, we separate the complex channel vector $h = [h_1^T, \cdots, h_K^T]^T \in C^{NK \times 1}$ into two components $R(h)$ and $I(h)$ and form the new input $[R(h), I(h)]^T \in \mathbb{R}^{2 \times NK}$, where $R(h)$ and $I(h)$ contain the real and imaginary parts of each element in $h$, respectively. Each CL layer consists of many filters which apply convolution operations to the layer input, capture special patterns and pass the result to the next layer. The parameters of the filters are shared among different channel coefficients. The main function of the BN layers is to normalize the output of the CL layers by two trainable parameters, i.e., a “mean” parameter and a “standard deviation” parameter. Besides, the BN layers can reduce the probability of over-fitting and enable a higher learning rate.

AC layers help neural networks extract the useful information and suppress the insignificant points of the input data. The rectified linear unit (ReLU) and sigmoid functions are suitable choices for the last AC layer, since the predicted variables are continuous and positive numbers. The function of the flatten layer is to change the shape of its input into a vector for the FC layer to interpret. In addition to these functional layers, the loss function, marked ‘MSE/MAE’ on the output layer in Fig. 2, is also very important in the introduced DL structure. The mean absolute error (MAE) or the mean square error (MSE) is used in the loss function to update parameters. The loss function together with the learning rate determines how to update the parameters of the neural network module.

2) **Signal Processing Module:** The neural network module offers universality in learning the key features from data, while the signal processing module aims to recover the beamforming matrix from the predicted key features at the output layer. Different from the neural network module whose model is unknown, the signal processing module utilizes the (partially) known models of the data to recover the beamforming matrix. The learned key features and the functionalities in the signal processing module are designated according to the expert knowledge. Note that the expert knowledge is problem-dependent and has no unified form, but what is in common is that the expert knowledge can significantly reduce the number of variables to be predicted compared to the beamforming matrix [47]. For example, the dual forms of the original problems are the typical expert knowledge for beamforming...
optimal. The details of the signal processing module used to recover the beamforming matrix is provided in the next two subsections.

B. To Learn \( \lambda \) and \( \mu \) and the Recovery Algorithm

With the above proposed general DL structure, we need to decide which features of dual optimization variables in \( P2 \) will be learned, and what signal processing function is needed to recover the beamforming matrix. The first option is to learn both \( \lambda \) and \( \mu \), so the output has \( K + N_t \) variables. Once they are learned, the following algorithm with steps taken from Algorithm 2 can be used to find a feasible beamforming solution that satisfies the per-antenna power constraints.

Algorithm 4: To recover \( W \) from \( \lambda \) and \( \mu \)
1) Given the learned solution of \( \lambda \) and \( \mu \), Calculate \( \beta_k = \frac{1}{\lambda_k h_k^T G_k(\lambda, \mu)^{-1}h_k} \). Then update the achievable SINR in the uplink as
\[
\gamma = \min_k \frac{1}{\beta_k - 1}. \tag{23}
\]
2) Find the optimal normalized uplink beamforming as
\[
\bar{w}_k = \frac{G_k(\lambda, \mu)^{-1}h_k}{\|G_k(\lambda, \mu)^{-1}h_k\|}, \forall k. \tag{24}
\]
3) Find the downlink power \( \{p_k\} \) to achieve the SINR \( \gamma \), i.e., to solve the following linear equation set:
\[
\sum_{i=1, i\neq k}^K p_i |h_i^T \bar{w}_k|^2 + N_0 = \gamma^*, \quad k = 1, \ldots, K. \tag{25}
\]
4) Update the downlink beamforming vector as \( w_k = \sqrt{p_k} \bar{w}_k \) and \( W = [w_1, \cdots, w_K] \).
5) Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:
\[
W(n,:) = W(n,:) \sqrt{\min_n \frac{P_n}{\|e_n^T W\|^2}}, \quad \forall n. \tag{26}
\]

C. To Learn \( \mu \) Only and the Recovery Algorithm

The above learning strategy is straightforward and fast if the learning result is satisfactory, however, the learning accuracy can be much improved if the number of variables is reduced. This motivates us to use the proposed DL structure to learn only the dual variable \( \mu \) with output size of \( N_t \), which contains \( K \) less variables than the above approach that learns both \( \lambda \) and \( \mu \). The idea of this approach is that given \( \mu \), the optimal \( \lambda \) can be efficiently optimized using Algorithm 2, which is more accurate than the learning approach above. An additional advantage is that the output size does not depend on the number of users, so it can more easily adapt to the varying number of users. Once \( \mu \) is learned, the following algorithm with steps taken from Algorithm 2 can be used to derive a feasible beamforming solution to the original problem \( P1 \).

Algorithm 5: To recover \( W \) from \( \mu \)
1) Given the learned solution \( \mu \), call Algorithm 1 to find the optimal \( \lambda \).
2) Calculate \( \beta_k = \frac{1}{\lambda_k h_k^T G_k(\lambda, \mu)^{-1}h_k} \). Then update the achievable SINR in the uplink as
\[
\gamma = \min_k \frac{1}{\beta_k - 1}. \tag{27}
\]
3) Find the optimal normalized uplink beamforming as
\[
\bar{w}_k = \frac{G_k(\lambda, \mu)^{-1}h_k}{\|G_k(\lambda, \mu)^{-1}h_k\|}, \forall k. \tag{28}
\]
4) Find the downlink power \( \{p_k\} \) to achieve the SINR \( \gamma \), i.e., to solve the following linear equation set:
\[
\sum_{i=1, i\neq k}^K p_i |h_i^T \bar{w}_k|^2 + N_0 = \gamma^*, \quad k = 1, \ldots, K. \tag{29}
\]
5) Update the downlink beamforming vector as \( w_k = \sqrt{p_k} \bar{w}_k \) and \( W = [w_1, \cdots, w_K] \).
6) Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:
\[
W(n,:) = W(n,:) \sqrt{\min_n \frac{P_n}{\|e_n^T W\|^2}}, \quad \forall n. \tag{30}
\]

D. Generalization of the Proposed DL Structure

In this section, we will generalize the proposed universal DL so that it can adapt to the change of the number of users and antennas. Although the above DL approaches can achieve satisfactory performance for beamforming design, applying the DL approaches to practical applications faces the difficulties caused by the dynamic wireless networks. In other words, when the number of transmit antennas \( N_t \) or the number of users \( K \) changes, a new model should be trained for prediction. This fact suggests that the applicability of the DL approaches is limited. Transfer learning and training set augmentation are effective ways to improve the generalization. The former transfers an existing model to a new scenario with some additional training and labelling effort [52], whereas the latter aims to train a large-scale model which adapts to different \( N_t \) and \( K \) by adding more samples into the training set, so that the training set can cover more possible scenarios. In this work, we adopt the latter method for simplicity. Without losing generality, we take the DL approach to learning \( \mu \) only as an example and give more details about the training set augmentation method.

In the training set augmentation method, we aim to train a large-scale model with \( 2N_t K' \)-input and \( N_t' \)-output. In order to make the large-scale model adaptable to different \( N_t \) and \( K \) values, we generate an augmented training set. Different from the training set whose samples have the same \( N_t \) and \( K \) values, the samples in the augmented training set are diverse, i.e., the numbers of the transmit antennas and the numbers of users in different samples could vary. However, the size of each sample is fixed as \( 2N_t K' \)-input and \( N_t' \)-output. For the
cases where \(N_t < N'_t\) (or \(K < K'\)), the redundant \(N'_t - N_t\) rows (or \(K - K_0\) columns) of the channel matrix are filled with 0’s. Similarly, the redundant \(N'_t - N_t\) elements of output are set as 0 when \(N_t < N'_t\). In each sample, we assume each \(K \in \{1, 2, \ldots, K'\}\) is generated with the equal probability of \(\frac{K}{K'}\) and each \(N_t \in \{1, 2, \ldots, N'_t\}\) is generated with the equal probability of \(\frac{N_t}{N'_t}\). Therefore, the occurrence probabilities of different \(K\) values are statistically equal among all samples and so are different \(N_t\) values. It is suggested that the number of the samples in the augmented training set for the large-scale model should be 5-10 times as many as that in the training set with fixed \(N_t\) and \(K\) values. However, this approach works only if the number of users or antennas does not exceed the maximum values used in the training set, otherwise re-training will be needed.

V. PERFORMANCE EVALUATION

Both simulations and experiments are carried out to evaluate the performance of the proposed DL enabled beamforming optimization. We assume that all channel entries undergo independent and identically distributed Rayleigh flat-fading with zero mean and unit variance unless otherwise specified, and perfect CSI is available at the BS. All transmit power is normalized by the noise power.

The training samples (dual variables) are generated by solving the problem \(P2\) using Knitro for its stability and efficiency, but can also be generated by solving the problem \(P1\) using the bisection search method at the cost of more computational time during the offline training. In our simulation, we use 20000 training samples and 5000 testing samples, respectively. All of proposed DL networks have one input layer, two CL layers, two BN layers, three AC layers, one flatten layer, one FC layer, and one output layer. Besides, each CL layer has 8 kernels of size 3 x 3 and the first two AC layers adopt the ReLU function. Each CL applies stride 1 and zero padding 1 such that the output width and height of all CLs remain the same as those of the input [56]. To be specific, the input size of the first CL is \(2 \times N_t K \times 1\) and the output size is \(2 \times N_t K \times 8\). Both the input size and output size of the second CL are \(2 \times N_t K \times 8\). When parameter sharing is considered, the numbers of parameters in the first and second CL are \(3 \times 3 \times 1 \times 8 = 72\) (weights) + 8 (bias) = 80, and \(3 \times 3 \times 8 \times 8 \times 8 = 584\), respectively, with a total of 664. When no parameter sharing is considered, the numbers of parameters in the two CLs are \((2 \times N_t K \times 8) \times (3 \times 3 \times 1 + 1) = 160N_t K\) and \((2 \times N_t K \times 8) \times (3 \times 3 \times 8 + 1) = 1168N_t K\), respectively, with a total of 1328N_t K. Adam optimizer [57] is used with the mean squared error based loss function. We adopt the sigmoid function in the last AC layer.

We will compare the performance and running time of the following schemes when possible:

1) The optimal solution to solve \(P2\) using Knitro.
2) The proposed subgradient algorithm (Algorithm 2) in Section III.
3) The proposed solution based on learned \(\lambda\) and \(\mu\).
4) The proposed solution based on learned \(\mu\) only.
5) ZF Solution [25].

a) When \(N_t = K\), pseudo inverse of the channel is the optimal beamforming direction, i.e.,
\[
\mathbf{W} = \mathbf{H}^\dagger (\mathbf{H}^T \mathbf{H})^{-1},
\]
and the achievable SINR is \(\Gamma_{ZF} = \min_k \frac{P_k}{\|\mathbf{e}_k^W\|^2}\). The overall optimal beamforming matrix is given by \(\mathbf{W} = \sqrt{\Gamma_{ZF}} \mathbf{W}\).

b) However, when \(N_t > K\), the optimal solution relies on solving the following SOCP problem \(P7\), so the associated complexity is high:
\[
P7: \max_{\mathbf{W}, \Gamma} \Gamma\\
\text{s.t.} |h_k^T \mathbf{w}_j|^2 \geq \Gamma, \forall k,\]
\[
\mathbf{h}_k^T \mathbf{w}_j = 0, \forall k \neq j,\]
\[
p_n = \|e_n^W\|^2 \leq P_n, \forall n. \quad (32)
\]

c) When \(N_t < K\), there is no feasible ZF solution. 6) RZF Solution [58]. This is a low-complexity heuristic solution that improves the performance of ZF especially at the low SNR region. The beamforming direction is given by:
\[
\mathbf{W} = \mathbf{H}^\dagger (\mathbf{H}^T \mathbf{H}^\dagger + \alpha \mathbf{I}_{K \times K})^{-1},
\]
where \(\alpha = \frac{K N_t}{\sum_{n=1}^{N_t} P_n}\) and the overall beamforming matrix is given by \(\mathbf{W} = \sqrt{\min_n \frac{P_n}{\|e_n^W\|^2}} \mathbf{W}\).
For fair comparison, the convergence of all iterative algorithms is achieved when the relative change of the objective function values is below \(10^{-8}\). All algorithms are implemented on an Intel i7-7700U CPU with 32 GB RAM using Matlab R2017b. One NVIDIA Titan Xp GPU is used to train the neural network.

A. Simulation Results

We first compare the SINR and running time results for a system with \(N_t = K = 4\) in Fig. 3. In Fig. 3 (a), we can see that both the proposed subgradient solution and the solution based on learned \(\mu\) can achieve close to optimal solution and outperform the RZF solution and the ZF solution especially at the low signal to noise (SNR) regime. As the SNR increases, all solutions converge to the optimal solution. Fig. 3 (b) shows that both of the proposed learning based solutions can achieve more than an order of magnitude gain in terms of computational time when compared to the optimal algorithm. The proposed subgradient algorithm is more efficient than the optimal solution using Knitro. ZF and RZF solutions have the lowest possible complexity because there is no optimization involved. In addition, we compare the robustness of various schemes against channel errors in Fig. 4. The channel vectors are modelled as \(\mathbf{h}_k = \mathbf{h}_k + \sigma \mathbf{e}_k\), \(\forall k\), where \(\mathbf{h}_k\) is the imperfect channel estimate, \(\mathbf{e}_k \sim \mathcal{C}\mathcal{N}(0, I_K)\) is the channel error vector and \(\sigma^2\) is the variance of channel estimation error. As expected, we can see that the channel estimation error causes degradation of the SINR performance for all the solutions. However, the results show that the proposed learning based solutions and the optimal solution are very robust, but the performance loss of the ZF and RZF beamforming is severe.
Next we demonstrate the scalability of the algorithms when $N_t = K$ and the number of users varies from 2 to 10 when $P_n = 10$ dB in Fig. 5. As can be seen from Fig. 5 (a), as both the numbers of users and antennas increase, the achievable SINR first decreases and then increases. The performance of the ZF and RZF solutions drops quickly. As the number of users increases, both learning based solutions significantly outperform the ZF solution and the performance gap is enlarged while their gap to the optimal solution remains constant. Fig. 5 (b) shows the complexity performance. The proposed algorithm that learns both $\lambda$ and $\mu$ has a lower complexity. As the number of users increases, e.g., when $K = 10$, it can achieve nearly 50-fold gain in terms of computational time when compared to the optimal algorithm. The proposed algorithm that learns only $\mu$ achieves 0.5 dB higher SINR than that learns both $\lambda$ and $\mu$ at the cost of slightly increased time complexity. Next we examine the SINR performance of the system using a more realistic 3GPP Spatial Channel Model (3GPP TR 25.996) [59] as shown in Fig. 6. We consider a scenario of urban micro cells and assume the distances between the BS and the users are between 50 m and 300 m and distributed uniformly. The total system bandwidth is 20 MHz. Similar trends of the algorithms are observed in Fig. 6 as those in Fig. 5 (a), and both learning based solutions still significantly outperform the RZF and the ZF solutions.

We then consider the performance of a system with $N_t = 10$ transmit antennas at the BS, and vary the number of users $K$ when $P_n = 10$ dB in Fig. 7 (a). It is noticed that there is about 1 to 2 dB gap between the learned solutions and the optimal solution, while the ZF solution is almost optimal when $N_t > K$. However, from Fig. 7 (b), we can see that the ZF solution has the highest complexity in this case because its solution needs to be optimized via solving the SOCP problem P7. The proposed algorithm that learns both $\lambda$ and $\mu$ achieves more than two orders of magnitude gain in terms of computational complexity when compared to the ZF solution.

Next we demonstrate the generalization property of our proposed algorithm that learns only $\mu$. We train a model with $N_t = K = 10$ only once, and then use it when $N_t \leq 10$. 
Fig. 6. The SINR performance an $N_t = K$ system averaged over 5000 samples using the 3GPP Spatial Channel Model in an urban micro cell environment when $P_n = 30$ dBm.

Fig. 7. The performance and complexity of a system with $N_t = 10$ and varying $K$ when $P_n = 10$ dB, over 5000 samples: (a) minimum SINR and (b) time consumption per channel realization.

Fig. 8. The SINR performance with varying $K$ and $N_t$ using the same trained system under $N_t = K = 10$.

Fig. 9. The implemented multiuser beamforming testbed system, where two USRPs are combined to make a four-antenna transmitter and two USRPs are used to emulate four single-antenna users.

and $K \leq 10$ vary. As shown in Fig. 8, it is observed the SINR performances of the optimal solution and the proposed generalization algorithm using the same model not only has the same trend with respect to the number of users, but also are close to each other. More specifically, the achieved SINRs of the two schemes decrease with the increase of the user number when the number of BS antennas is fixed. Such observation validates the feasibility of the training set augment method and motivates further research on improving the generalization of the proposed DL-based algorithms. Besides, we find that adding more antennas can improve the SINR performance because of the spatial gain.

B. Testbed Results

To evaluate the proposed learning-based algorithm in a real-world scenario, we have implemented a multi-user beamforming testbed system based on SDR in our lab environment.

1) Testbed Setup: The multi-user beamforming testbed system is based on the SDR structure, which consists of one PC hosting Matlab, a Gigabit Ethernet switch, four NI’s USRP devices as transmitters or receivers and a CDA-2990 Clock Distribution Device. The USRP devices and the Clock Distribution Device for synchronization are illustrated in Fig. 9.

We adopt the SDR system since it provides a flexible development environment as well as a practical prototype. The USRP devices are exploited as the radio fronts in the SDR system, which can support different interfacing methods including PCIe and Gigabit Ethernet connections. Besides, the USRP devices can support a wide range of baseband signal processing platforms, including Matlab, Labview and GNU Radio. The transmitters and receivers are implemented using USRP-2950 devices, which support the Radio Frequency (RF) range from 50MHz to 2.2GHz [60]. For the evaluation purpose, the 900 MHz Industrial, Scientific and Medical (ISM) frequency band is used. The key parameters of the multi-user beamforming system are listed in Table I.

In the experiment, we consider the scenario consisting of one BS with four transmit antennas and four single-antenna
users, i.e., $N_t = K = 4$. We combine two USRP-2950 devices as a cooperative four-antenna transmitter and employ two USRP-2950 devices as four individual single-antenna users. All channels on the USRP devices are synchronized using the CDA-2990 Clock Distribution Device. The omnidirectional tri-band SMA-703 antennas are used for both the transmitters and the receivers, while the receiver antennas are extended using RF cables. Specifically, both static and dynamic channel conditions are examined to evaluate the proposed learning-based beamforming algorithms. For the static channel scenario, the transmitter antennas are placed next to each other with a space of 0.1 m, while the receiver antennas are placed 1.5 m away from the transmitter antennas as well as from each other. For the dynamic scenario, a low-mobility scenario is simulated, where one of the receiving antennas is moving at the speed of 0.6 m/s. Besides, the experiment also exploits different transmitter powers to evaluate the algorithms’ performance in different SNR configurations, where 0 dB of transmit power gain corresponds to a transmit power of $-70$ dBm.

Since the multi-user beamforming system coordinates several USRP devices as transmitters and receivers at the same time, a Gigabit Ethernet switch is used to enable multiple USRP interfacing.

The baseband signal processing modules and the proposed learning-based beamforming algorithms are implemented as Matlab function scripts on a PC with 1 Intel i7-4790 CPU Core, and RAM of 32GB. In the experiment, all users are sharing the same channel and they all use the Quadrature Phase Shift Keying (QPSK) modulation. The payloads are prefixed with different Gold sequences for each user, which are exploited for both synchronization and channel estimation. Besides, all baseband signals are shaped using a Raised Cosine Filter. During the experiment, each user decodes its own payload and provides channel estimation as feedback to the transmitter. The transmitters and receivers are controlled using different Matlab sessions, while the channel estimation information is exchanged locally on the PC’s cache storage. The beamforming algorithms optimize the beam weight vectors using the aggregated channel estimation information. The transmitter applies the optimized beam weight vectors to generate the signals for each antenna before transmission.

2) Experiment Results and Discussions: To demonstrate the performance of the proposed learning algorithm (based on learned $\lambda$ and $\mu$), three benchmark algorithms are implemented on the multi-user beamforming system, which are the theoretically optimal solution, the ZF solution and the RZF solution. Each algorithm is evaluated under both static and dynamic conditions, and we choose bit error rate (BER) as the performance metric. In order to generate the BER performance of each solution, a real-time experiment is conducted using the testbed illustrated in Fig. 9 with different transmitter power. For each transmit power, the BS sends $10^4$ packets each containing 256 QPSK symbols and the BER is calculated based on the averaged bit error of all packets.

Fig. 10 depicts the BER results in the static and dynamic channel conditions as the transmit power gain varies. Under the static condition as shown in Fig. 10 (a), the proposed learning-based algorithm outperforms the ZF solution and RZF solution across the considered transmit power range. Specifically, the BER performance gain of the learning based algorithm is approximately 4 dB over the ZF solution and 3 dB over the RZF solution in the relatively low transmit SNR regime, and this performance gain reduces as the transmit SNR grows. Compared to the theoretically optimal solution, the learning-based algorithm has a close performance in the low transmit SNR regime but becomes inferior for high transmit SNR conditions. This is expected because under

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Descriptions</th>
</tr>
</thead>
<tbody>
<tr>
<td>Clock and PPS source</td>
<td>CDA-2990 10MHz and 1PPS</td>
</tr>
<tr>
<td>Radio Front</td>
<td>USRP 2950, 50MHz-2.3GHz</td>
</tr>
<tr>
<td>Antennas</td>
<td>Tri-band SMA-703</td>
</tr>
<tr>
<td>Modulation</td>
<td>QPSK</td>
</tr>
<tr>
<td>Prefix</td>
<td>Gold code of length 127</td>
</tr>
<tr>
<td>Baseband Sample Rate</td>
<td>40 kilo samples/second (kps)</td>
</tr>
<tr>
<td>Pulse Shaping</td>
<td>Raised Cosine Filter of squared root shape with rolloff factor 0.8 and decimation factor 8</td>
</tr>
<tr>
<td>Channel Estimation</td>
<td>MMSE Estimator</td>
</tr>
</tbody>
</table>

Table 1: Testbed System Configuration
VI. CONCLUSION AND FUTURE DIRECTIONS

In this paper, we have developed deep learning enabled solutions for fast optimization of downlink beamforming under the per-antenna power constraints. Our solutions are both model driven and data driven, and are achieved by exploiting the structure of the beamforming problem, learning the dual variables from labelled data and then recovering the original beamforming solutions. Our solutions can naturally adapt to the varying number of active users in dynamic environments without re-training thus making it more general. The simulation results have shown the superior performance-complexity tradeoff achieved by the proposed solutions, and the results have been further verified by the tested experiments using software defined radio.

We would like to point out a few promising future directions. This paper assumes that perfect CSI is available; however in practice, CSI estimation is never perfect. One future direction would be to investigate a more advanced robust learning framework to mitigate channel estimation errors or other types of impairments. As a step further, another promising future direction will be to study how to use deep learning to map directly from the pilot signals to the beamformed signals, bypassing the explicit channel estimation step.

In order to reduce computational complexity of the training process when the channel conditions change, one possible method is to use a wide range of channel realizations during the off-line training phase, in order that the neural network can learn to generalize from a wider range of channel variations. Another approach is to employ transfer learning [52]. The main idea is that knowledge learned from one training task for a given channel condition may be transferred to a similar task with a different channel condition, and this will greatly improve the off-line training phase, in order that the neural network can learn to generalize from a wider range of channel variations. Another approach is to employ transfer learning [52]. The main idea is that knowledge learned from one training task for a given channel condition may be transferred to a similar task with a different channel condition, and this will greatly improve the off-line training phase, in order that the neural network can learn to generalize from a wider range of channel variations.

### APPENDIX

#### A. Proof of the Convergence and Optimality of Algorithm 1 to Solve P4

The proof has two parts. The first part is devoted to the proof of convergence and the second part addresses the uniqueness and optimality of the fixed point after convergence.

Let us start with $\gamma^{(j)} (\ For j \geq 1$) which is achievable for the power vector $\lambda^{(j)}$. It is easily seen that given $\gamma^{(j)}$, $\gamma(\lambda^{(j)})$ (user index $k$ is omitted for convenience) is a standard interference function, which satisfies the following properties [54], [55]:

(P1) $\lambda^{(j)}$ is component-wise monotonically decreasing;

(P2) If $\lambda \geq \lambda^{(j)}$, then $\gamma(\lambda) \geq \gamma(\lambda^{(j)})$;

(P3) $\lambda^{(j)}$, for all $j$, are all feasible solutions given the SNR constraint $\gamma^{(j)}$.

Assume that at the $j$-th iteration, the dual variable is $\lambda^{(j)}$ and the achievable SNR is $\gamma^{(j)}$. Then at the $(j + 1)$-th iteration, according to (P1), $\lambda^{(j+1)} \leq \lambda^{(j)} \forall k$, and as such $\eta \geq 1$ and $\lambda^{(j+1)} \leq \lambda^{(j)} \forall k$ (in Step 4). According to P2, in Step 5 we have the SNR result $\gamma_k(\lambda^{(j+1)}) \geq \gamma_k(\lambda^{(j+1)})$. Then, according to (P3), $\gamma_k(\lambda^{(j+1)}) \geq \gamma_k(\lambda^{(j)}) \forall k$, and therefore $\gamma^{(j+1)} = \min_k \gamma_k(\lambda^{(j+1)}) \geq \min_k \gamma_k(\lambda^{(j+1)}) \geq \gamma^{(j)}$, i.e., the balanced SNR $\gamma^{(j)}$ is increasing as the iteration goes. Since $\gamma^{(j)}$ is upper bounded, the algorithm converges to a...
fixed point \( \lambda^{(\infty)} \). Next, we prove that the fixed point is also optimal.  

We see that \( \lambda_k^{(\infty)} \) satisfies the following fixed-point equation:

\[
\lambda_k^{(\infty)} = \gamma^{(\infty)} \tilde{f}_k(\lambda_k^{(\infty)}) \quad \forall k, \tag{34}
\]

and it satisfies the total virtual uplink power is \( \sum_k \lambda_k^{(\infty)} = \frac{1}{N_0} \). Clearly, the total uplink transmit power is a monotonic non-decreasing function of the SINR constraint. This implies that there is no solution \( \lambda^{*} \) which provides a strictly higher SINR \( \gamma^{*} > \gamma^{(\infty)} \) but still maintains the power constraint \( \sum_k \lambda_k^{(\infty)} = \frac{1}{N_0} \).

**B. Proof that \( f(\mu) \) of \( P4 \) is a Concave Function in \( \mu \)**

**Proof:** First note that Algorithm 1 to solve \( P4 \) belongs to a fixed-point iteration, which means a solution \( \{ \Gamma, \lambda \} \) that satisfies the first two constraints (9) and (10) with equality ensuring an optimal solution. This indicates there is no local optimum, and the gap between \( P4 \) and its dual problem is zero. Then it suffices to prove that the objective function of the dual problem of \( P4 \) is concave in \( \mu \).

By using (11) of [23], we can rewrite \( P4 \) as

\[
P4^*: \quad f(\mu) = \max_{\Gamma, \lambda} \Gamma \\
\text{s.t.} \quad \sum_{i=1}^{K} \lambda_i h_i^T h_i + \text{Diag}(\mu) - \left( 1 + \frac{1}{\Gamma} \right) \lambda_k h_k^T h_k \geq 0, \quad \forall k, \\
\sum_{k=1}^{K} \lambda_k N_0 = 1, \\
\lambda \geq 0. \tag{35}
\]

Its Lagrangian is

\[
L_\mu(\Gamma, \lambda, a, b, \{ C_k \}) = \Gamma + a \left( \sum_{k=1}^{K} \lambda_k N_0 - 1 \right) + b^T \lambda \\
- \sum_{k=1}^{K} \text{trace} \left( \sum_{i=1}^{K} \lambda_i h_i^T h_i + \text{Diag}(\mu) \\
- \left( 1 + \frac{1}{\Gamma} \right) \lambda_k h_k^T h_k \right) C_k, \tag{36}
\]

where \( a, b, \{ C_k \} \) are dual variables. Note that it is derived based on the maximization rather than the commonly used minimization of an objective function.

The dual objective function is expressed as

\[
G_\mu(a, b, \{ C_k \}) = \min_{\Gamma, \lambda} L_\mu(\Gamma, \lambda, a, b, \{ C_k \})
\]

which is to be minimized over \( (a, b, \{ C_k \}) \) and only contains a linear term of \( \sum_{k=1}^{K} \text{trace} \left( \sum_{i=1}^{K} \lambda_i h_i^T h_i + \text{Diag}(\mu) \right) C_k \) about \( \mu \), and the constraints of the dual problem (although not derived here) do not involve \( \mu \). Therefore the dual objective function \( \min G_\mu(a, b, \{ C_k \}) \) is a point-wise minimum of a family of affine functions about \( \mu \) and as a result concave [61, Sec.3.2.2], so is \( f(\mu) \). This completes the proof. \( \square \)

**C. To Find the Subgradient Euclidean Projection in Algorithm 2**

The Euclidean projection is needed when the update of \( \mu \) based on the subgradient in Algorithm 2 does not fall into the feasible set \( S \). It needs to solve the following optimization problem:

\[
P5: \quad \min_{\nu} \| \nu - \mu \|^2 \quad \text{s.t.} \sum_{n=1}^{N_1} \nu_n P_n = 1, \quad \nu \geq 0, \tag{37}
\]

where \( \mu = \mu_k^{(i)} + \alpha_k \text{Diag}([|e_n|^T W|^2]) \). Although \( P5 \) is a convex problem and can be solved by a standard numerical algorithm, below we derive its analytical property and propose a more efficient bisection algorithm to solve it.

Its Lagrangian can be expressed as

\[
L = \| \nu - \mu \|^2 + x(\sum_{n=1}^{N_1} \nu_n P_n - 1) - \sum_n y_n \nu_n, \tag{38}
\]

where \( x \) and \( y_n \geq 0 \) are dual variables. Setting its first-order derivative to be zero leads to

\[
\nu_n = \frac{2 \mu_n + y_n - x P_n}{2}, \tag{39}
\]

Substitute it to \( \sum_{n=1}^{N_1} \sum_{n=1}^{N_1} \nu_n P_n = 1 \) and we get

\[
\sum_{n=1}^{N_1} \max \left( \frac{2 \mu_n - x P_n}{2}, 0 \right) P_n = 1. \tag{40}
\]

Therefore the remaining task is to find \( x \) that satisfies (40). Obviously the left hand side of (40) is monotonic in \( x \), so we propose the following bisection method to find the optimal \( x \).

**Algorithm 3 to Solve P5:**

1. Set the upper and lower bounds of \( x \) as \( x^U \) and \( x^L \).
2. Repeat the following steps until convergence.
3. If \( \sum_{n=1}^{N_1} \max \left( \frac{2 \mu_n - x^2 P_n}{2}, 0 \right) P_n > 1, \quad x^L = x^U; \) otherwise \( x^U = x^L \).

**REFERENCES**


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Abstract—This paper studies fast downlink beamforming algorithms using deep learning in multiuser multiple-input-single-output systems where each transmit antenna at the base station has its own power constraint. We focus on the signal-to-interference-plus-noise ratio (SINR) balancing problem which is quasi-convex but there is no efficient solution available. We first design a fast subgradient algorithm that can achieve near-optimal solution with reduced complexity. We then propose a deep neural network structure to learn the optimal beamforming based on convolutional networks and exploitation of the duality of the original problem. Two strategies of learning various dual variables are investigated with different accuracies, and the corresponding recovery of the original solution is facilitated by the subgradient algorithm. We also develop a generalization method of the proposed algorithms so that they can adapt to the varying number of users and antennas without re-training. We carry out intensive numerical simulations and testbed experiments to evaluate the performance of the proposed algorithms. Results show that the proposed algorithms achieve close to optimal solution in simulations with perfect channel information and outperform the alleged theoretically optimal solution in experiments, illustrating a better performance-complexity tradeoff than existing schemes.

Index Terms—Deep learning, beamforming, multiple-input-single-output (MISO), signal-to-interference-plus-noise ratio (SINR) balancing, per-antenna power constraints.

I. INTRODUCTION

MULTIUSER multi-antenna techniques (or multiple-input multiple-output, MIMO) techniques can significantly improve the spectral and energy efficiency of wireless communications by exploiting the degree of freedom in the spatial domain. They have been widely adopted in modern wireless communications systems such as the fourth and the fifth-generation (4G and 5G) of cellular networks [1], [2], the high efficiency wireless local area (WiFi) networks standard 802.11ax [3], and the latest satellite digital video broadcasting standard DVB-S2X [4]. Among the multiuser MIMO techniques, beamforming is one of the most promising and practical schemes to mitigate multiuser interference and exploit the gain of MIMO antennas.

In the last two decades, the optimal beamforming strategies have been intensively studied for the multiple-input single-output (MISO) downlink where a base station with multiple antennas serves multiple single-antenna users. For instance, the problem of signal-to-interference-plus-noise ratio (SINR) balancing or maximization of the minimum SINR of all users, under a total power constraint was studied in [5], [6], the total BS transmit power minimization problem under quality of service (QoS) constraints was investigated in [7]–[10], and the sum rate maximization problem under the total power constraint was tackled in [6], [11]–[13]. The existing approaches mainly make use of the advances of convex optimization techniques such as second-order cone programming (SOCP) [8], [9] and semidefinite programming (SDP) [14], and the uplink-downlink duality which indicates that under the sum power constraint, the achievable SINR region and the normalized beamforming in the downlink are the same as those in the dual uplink channel.

Early works mostly focus on the optimal beamforming design under the sum power constraint across all antennas of a transmitter. This constraint does not take into account the fact that each transmit antenna has its own power amplifier, and therefore its power is individually limited. The per-antenna power constraints were first systematically studied in [15] where a dual framework was proposed to minimize the

maximum transmit power of each antenna under users’ SINR constraints. This work has sparked much research interest in optimizing beamforming under per-antenna power constraints. The work in [16] studied the optimization of the nonlinear zero forcing (ZF) dirty paper coding based beamforming under per-antenna power constraints. Generic optimization of beamforming for multibeam satellite systems was studied in [17] under general linear and nonlinear power constraints. The per-antenna constant envelope precoding for large multiser MIMO systems was investigated in [18]. The transceiver designs for multi-antenna multi-hop cooperative communications under per-antenna power constraints were proposed in [19] and both linear and nonlinear transceivers were investigated. The signal-to-leakage-plus-noise ratio (SLNR) maximized precoding for the downlink under per-antenna power constraints was considered in [20] where a semi-closed form optimal solution was proposed. A general framework for covariance matrix optimization of MIMO systems under different types of power constraints was proposed in [21]. More recently, the optimal MIMO precoding under the constraints of both the total consumed power constraint and the individual radiated power constraints was studied in [22] and numerical algorithms were developed to maximize the mutual information.

The problem of interest in this paper is to efficiently maximize the minimum received SINR or to balance SINR, in the multiuser MISO downlink under per-antenna power constraints at the BS. This problem, although being quasiconvex, is more challenging than the counterpart with the total power constraint and the problem of minimizing the per-antenna power in [15], and until now there does not exist efficient algorithms. Consequently, existing beamforming techniques are unable to support real-time applications because the small-scale fading channel varies considerably fast. For instance, in a WLAN 802.11n system operating at 2.4 GHz with a pedestrian speed of 1.4 m/s, the coherence time is 89 ms; and in a Long-Term Evolution (LTE) downlink operating at 2.6 GHz with a residential area vehicle velocity of 10 m/s, the coherence time is only 11.5 ms. Traditional time-consuming optimization routines will produce obsolete beamforming solution that is not timely for the current channel state and lead to significant performance degradation which will be demonstrated in our experiment. In [23], the dual problem was derived and the optimal solution at much reduced computational cost was developed. However, it was found out that the best solution is obtained by a commercial nonlinear solver [24], which does not explore the structure of the problem and is still not efficient. Although there are simple heuristic beamforming solutions which have closed-form solutions such as the ZF beamforming and the regularized ZF (RZF) beamforming, the reduced complexity often leads to performance loss. Even worse, the work in [25] showed that the conventional ZF beamforming under per-antenna power constraints no longer admits a simple pseudo-inverse form as the case under the total power constraint, and instead the optimal ZF beamforming requires solving an SOCP problem which has much higher complexity.

In this paper, we take a different approach and develop deep learning (DL) enabled beamforming solutions to dramatically improve the computational efficiency. Recently DL has been recognized as a promising solution for addressing various problems in several areas of wireless networks. This is because deep neural networks have the ability to model highly nonlinear functions at considerably low complexity. One of the areas of interest is to deal with scenarios in which the channel model does not exist, e.g., in underwater and molecular communications [26] or is difficult to characterize analytically due to imperfections and nonlinearities [27]. In these situations, DL based detection has been proposed to tackle the underlying unknown nonlinearities [28]. Another area of interest is to optimize the end-to-end system performance [29], [30]. Conventional communication systems are based on the modular design and each block (e.g., coding, modulation) is optimized independently, which can not guarantee the optimal overall performance. However, DL holds great promises for further improvement by considering end-to-end performance optimization. The third area of interest is to overcome the complexity of wireless networks [27] which is the focus of our paper. In this aspect, DL has found many exciting applications in wireless communications such as channel decoding [31], [32], MIMO detection [33], [34], channel estimation [35], [36]. The current work belongs to the framework of learning to optimize in wireless resource allocation. The rationale is that the DL technique bypasses the complex optimization procedures, and learns the optimal mapping from the channel state to produce the beamforming solution directly by training a neural network. The result is that the trained neural network can be used as a function mapping to obtain the real-time beamforming solution with channel state as input. As a result, the computational complexity is transferred to offline training phase, and hence the complexity during the online transmission phase is greatly reduced. The mostly successful applications of DL in this framework by far is power allocation [37]–[41], in which the power vector is treated as the training output, while the channel gains are taken into the input of the DL network. In this case, the power variables only take positive values and the number of power variables is normally the number of users and therefore relatively small and easy to handle.

However, there are few works that focus on the learning approach to optimize the beamforming design in multi-antenna communications, with the exception of [42]–[47]. The difficulty is partly due to the large number of complex variables contained in the beamforming matrix that need to be optimized. An outage-based approach to transmit beamforming was studied in [42] to deal with the channel uncertainty at the BS, however, only a single user was considered. The work

1To the best of our knowledge, the computational complexity of the training phase is not well understood, due to the complex implementation of the backpropagation process and that it depends very much on the specific application regarding the required number of training examples for satisfactory generalization. That said, this is usually not a concern in most applications because training takes place offline given sufficient computational capability and retraining is only performed infrequently when the specific applications depart considerably from those training examples.
in [43] designed a decentralized robust precoding scheme based on a deep neural network (DNN). The projection over a finite dimensional subspace in [43] reduced the difficulty, but also limited the performance. A DL model was used in [44] to predict the beamforming matrix directly from the signals received at the distributed BSs in millimeter wave systems. However, both [43] and [44] predicted the beamforming matrix in the finite solution space at the cost of performance loss. The works in [42], [45] directly estimated the beamforming matrix without exploiting the problem structure in which the number of variables to predict increases significantly as the numbers of transmit antennas and users increase. This will lead to high training complexity and low learning accuracy of the neural networks when the numbers of transmit antennas and users are large. In our previous works [46], [47], we proposed a beamforming neural network to optimize the beamforming vectors, but it is restricted to the total power constraint. We notice that none of existing works addressed the SINR balancing problem under the practical per-antenna power constraints, for which DL solution becomes even more attractive.

In this paper, we propose a DL enabled beamforming optimization approach for SINR balancing to provide an improved performance-complexity tradeoff under per-antenna power constraints. Inspired by the model driven learning philosophy [48], we propose to first learn the dual variables with reduced dimension rather than the original large beamforming matrix and then recover the beamforming solution from the learned dual solution, by exploiting the structure or model of the beamforming optimization problem. Our main contributions are summarized as follows:

- A subgradient algorithm is first proposed which not only demonstrates faster convergence than the best known algorithm in [23], but also facilitates the development of the DL solutions.
- A general DL structure to learn the dual variables is proposed, and two learning strategies are proposed to achieve the performance-complexity tradeoff. A heuristic method is developed to facilitate the generalization of the proposed DL algorithms by augmenting the training set so that they can adapt to the varying number of active users and antennas without re-training.
- Both software simulations and testbed experiments using software defined radio (SDR) are carried out to validate the performance of the proposed algorithms. To the best of our knowledge, this is the first testbed demonstration of deep learning enabled multiuser beamforming.

The remainder of this paper is organized as follows. Section II introduces the system model and formulates the SINR balancing problem and its dual formulation. Section III proposes the subgradient algorithm. Section IV provides the general structure framework for the beamforming optimization based on learning the dual variables and the recovery algorithms. Numerical and experimental results are presented in Section V. Finally, conclusion is drawn in Section VI.

Notations: The notations are given as follows. Matrices and vectors are denoted by bold capital and lowercase symbols, respectively. $(\cdot)^T$, $(\cdot)^*$, $(\cdot)^\dagger$ and $(\cdot)^{-1}$ stand for transpose, conjugate transpose and inverse/pseudo inverse (when applicable) operations of a matrix, respectively. $A > 0$ indicates that the matrix $A$ is positive definite. The operator ${\rm diag}(\mathbf{a})$ denotes the operation to diagonalize the vector $\mathbf{a}$ into a matrix whose main diagonal elements are from $\mathbf{a}$. Finally, $\mathbf{a} \sim \mathcal{CN}(0, \Sigma)$ represents a complex Gaussian vector with zero-mean and covariance matrix $\Sigma$. $Z$ denotes the non-negative field.

II. SYSTEM MODEL AND PROBLEM FORMULATION

Consider an MISO downlink channel where an $N_t$-antenna BS transmits signals to $K$ single-antenna users. For the user $k$, its channel vector, beamforming vector, and data symbol are denoted as $\mathbf{h}_k^T, \mathbf{w}_k$, $s_k$, respectively, where $E(|s_k|^2) = 1$. The additive white Gaussian noise (AWGN) at the receiver is denoted as $n_k \sim \mathcal{CN}(0, N_0)$. All wireless links exhibit independent frequency non-selective Rayleigh block fading. The received signal at user $k$ is

$$y_k = \mathbf{h}_k^T \sum_{i=1}^K \mathbf{w}_i s_i + n_k.$$  \hfill (1)

The SINR at the receiver of user $k$ is given by

$$\gamma_k = \frac{\left| \mathbf{h}_k^T \mathbf{w}_k \right|^2}{\sum_{i=1, i \neq k}^K |\mathbf{h}_k^T \mathbf{w}_i|^2 + N_0}. \hfill (2)$$

The beamforming matrix is collected in $\mathbf{W} = [\mathbf{w}_1, \mathbf{w}_2, \ldots, \mathbf{w}_K] \in \mathbb{C}^{N_t \times K}$. Then the per-antenna power at antenna $n$ can be expressed as

$$p_n = \| \mathbf{W}(n,:) \|^2 = \| e_n^T \mathbf{W} \|^2, \hfill (3)$$

where $e_n$ is a zero vector except its $n$-th element being 1.

The problem of interest is to maximize the minimum user SINR, i.e., SINR balancing, under per-antenna power constraints $\{P_n\}$. Mathematically, it can be formulated as follows:

**P1:** $\max_{\mathbf{W}^T} \quad \Gamma$

$$\text{s.t. } \gamma_k = \frac{\left| \mathbf{h}_k^T \mathbf{w}_k \right|^2}{\sum_{i=1, i \neq k}^K |\mathbf{h}_k^T \mathbf{w}_i|^2 + N_0} \geq \Gamma, \quad \forall k; \hfill (4)$$

$$p_n = \| e_n^T \mathbf{W} \|^2 \leq P_n, \quad \forall n. \hfill (5)$$

The SINR balancing problem is in general quasi-convex, so it can be solved via methods such as bisection search and generalized eigenvalue programming [8], [23]. However, these methods suffer from high complexity and computational delay, and are not practical for real-time data transmissions.
In [23], a useful dual formulation of P1 is derived as

\[ \begin{align*}
\text{P2:} & \quad \max_{\beta, \lambda, \mu} \beta \\
\text{s.t.} & \quad \lambda_k h_k^T G(\lambda, \mu)^{-1} h_k^* \leq 1, \quad \forall k, \\
& \quad \sum_{k=1}^{N_t} \lambda_k N_0 = 1, \\
& \quad \sum_{n=1}^{N_r} \mu_n P_n = 1, \\
& \quad \lambda, \mu, \beta \geq 0.
\end{align*} \]  

(6)

where \( G(\lambda, \mu) \triangleq \sum_{k=1}^{K} \lambda_k h_k^T h_k^T + \text{Diag}(\mu), \) \( \lambda \in \mathbb{Z}^K, \mu \in \mathbb{Z}^{N_r} \) are dual variables associated with the SINR constraint (4) and the per-antenna power constraint (5) in P1, and \( \beta \) is related to the minimum SINR \( \Gamma \) in P1 by the relation \( \beta = 1 + \frac{1}{P_1}. \) For the solution of P2, it is assumed that \( G(\lambda, \mu) \succ 0. \)

Although the problem P2 is still a quasi-convex problem, compared to the original problem P1, it can be more efficiently solved because it only involves \( K + N_t + 1 \) non-negative variables while P1 needs to optimize \( 2KN_r \) real variables. The problem P2 can be solved using standard nonlinear solvers such as Matlab’s built-in function ‘fmincon’. Currently the fastest optimal solution is known to be achieved by Ziena’s nonlinear solver Knitro [24], which is compared and shown in [23]. However, the general solvers do not exploit the special analytical properties of the problem P2, so they are not efficient. In addition, it is not known how to recover the optimal solution to the beamforming matrix \( \mathbf{W}^* \) once P2 is solved. These issues will be studied in the next section.

III. A SUBGRADIENT ALGORITHM TO SOLVE P2

In this section, we derive a fast subgradient algorithm to solve P2, based on the downlink-uplink duality results derived in [15]. According to [15, Theorem 1], the problem P2 can be equivalently written as the following max-max problem:

\[ \begin{align*}
\text{P3:} & \quad \max_{\mu} \max_{\Gamma, \lambda} \Gamma \\
\text{s.t.} & \quad \lambda_k = \frac{1}{N_0}, \\
& \quad \sum_{k=1}^{K} \lambda_k = 1, \\
& \quad \sum_{n=1}^{N_r} \mu_n P_n = 1, \\
& \quad \lambda, \mu, \Gamma \geq 0.
\end{align*} \]  

(8)
Algorithm 2 to Solve P3:

1) Initialize $\mu$. Suppose $j$ is the iteration index. Repeat the following steps 2)-7) until convergence.
2) Given $\mu$, call Algorithm 1 to find the optimal $\bar{\lambda}$. 
3) Calculate $\beta_k = \frac{\lambda_k h_k^T G_k(\lambda, \mu)^{-1} h_k^*}{\|G_k(\lambda, \mu)^{-1} h_k^*\|^2}$. Then update the achievable SINR in the uplink as
   \[
   \gamma^{(j)} = \min_k \frac{1}{\beta_k - 1}. \tag{16}
   \]
4) Find the optimal normalized uplink beamforming
   \[
   w_k = \frac{G_k(\bar{\lambda}, \mu)^{-1} h_k^*}{\|G_k(\bar{\lambda}, \mu)^{-1} h_k^*\|}, \quad \forall k. \tag{17}
   \]
5) Find the downlink power $\{p_k\}$ to achieve the SINR $\gamma^{(j)}$, i.e., to solve the following linear equation set:
   \[
   \sum_{i=1, i \neq k}^K p_i h_i^T w_k |^2 + N_0 = \gamma^{(j)}, \quad k = 1, \ldots, K. \tag{18}
   \]
6) Update the downlink beamforming vector as $w_k = \sqrt{p_k}w_k$, and $W = [w_1, \ldots, w_K]$.
7) Update $\mu$ using the subgradient Euclidean projection method with step size $\alpha_j$:
   \[
   \mu^{(j+1)} = \mathcal{P}_S(\mu^{(j)} + \alpha_j \text{Diag}(\|e^T W\|^2)), \tag{19}
   \]
   where $S = \{\mu \mid \sum_{n=1}^{N_1} \mu_n P_n = 1\}$.
   This projection $\mathcal{P}_S$ can be solved efficiently using the bisection search. The detailed projection algorithm is provided in Algorithm 3 of Appendix C.
8) Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:
   \[
   W(n, :) = W(n, :). \min_n \frac{P_n}{\|\bar{h}_n^T W\|^2}, \quad \forall n. \tag{20}
   
Remarks: The subgradient algorithm exploits the structure of the original problem $\text{P}_1$, so it is more efficient than a general nonlinear solver. However, the step size $\alpha_j$ is a critical parameter. We find that $\alpha_j = 0.01 \times 2^{-j}$ gives a satisfactory performance. We observe that in general Algorithm 2 can solve $\text{P}_3$ faster than the available numerical solver such as Knitro and achieve close to optimal performance. It may only converge to the neighbourhood of the optimal solution, and its convergence may be slow, as seen in Fig. 1(b). In addition, the subgradient algorithm may not guarantee that the per-antenna power constraints will be satisfied, and that is why Step 8) of Algorithm 2 is necessary to regulate the per-antenna power.

Another important implication of the development of Algorithm 2 is that it provides an efficient way to recover the primal variable, i.e., the downlink beamforming vectors, given various dual variables either $\mu$ and $\lambda$, or only $\mu$. The details will be given in the next section.

IV. THE PROPOSED DEEP LEARNING STRUCTURE AND STRATEGIES

In this section, we develop DL based solutions of $\text{P}_1$ that can achieve better performance-efficiency tradeoff than the currently available solutions. Instead of learning to optimize the original beamforming matrix $W$ directly, we will learn the optimization of the dual variables in $\text{P}_2$. This will dramatically reduce the number of variables that need to be learned. In the sequel, we will first introduce a general DL structure that takes the channel $h = [h_1^T, \ldots, h_K^T]^T$ as the input, and the output is the dual variable(s) in $\text{P}_2$. We will also devise a generalized learning solution such that the proposed DL structure can deal with varying number of users and antennas and transmit power without re-training. We will then propose two learning strategies, i.e., one is to learn the dual variables $\mu$ and $\lambda$ with fast recovery of the original beamforming solution, and the other is to learn only the dual variable $\mu$ with improved learning accuracy, to achieve various tradeoffs.

A. A General DL Structure

We first show the existence of a neural network that can approximate the solution of the optimization problem $\text{P}_2$. To this end, we define $\mu^{\text{opt}}$ and $\lambda^{\text{opt}}$ as two tensors with the
The neural network aims to learn the continuous mapping
\[ \mathcal{F}(h, \mu^0) = \{\mu^{opt}, \lambda^{opt}\}, \] (21)
where \(\mu^0\) is the initialization set of dual variables and \(\mathcal{F}(\cdot, \cdot)\) denotes the continuous mapping process in Algorithm 2 to achieve the stationary point from the input set of channel coefficients together with the initialization set of dual variables. The following theorem will prove the existence of a feedforward network which imitates the continuous mapping in (21).

**Theorem 1:** For any given accuracy \(\varepsilon > 0\), there exists a positive constant \(L\) large enough such that a feedforward neural network with \(L\) layers can produce similar performance to the mapping process in (21), i.e.,
\[ \sup_{\mu, \psi}||\operatorname{NET}_\mathcal{L}(h, \psi) - \mathcal{F}(h, \mu^0)||_F \leq \varepsilon, \] (22)
where \(\psi\) is the set of the neural network parameters including weights and biases.

**Proof:** The result in Theorem 1 can be obtained directly by applying the universal approximation theorem in [50] to the continuous mapping process in Algorithm 2. □

Based on results in Theorem 1, next we find solutions through designing the neural networks with the DL technique. Similar to our previous work [47], we introduce a general DL structure to approximate the mapping function from the channel coefficients to the beamforming solutions, as shown in Fig. 2. In addition to the conventional neural network module, the adopted DL structure also introduces a signal processing module based on expert knowledge for beamforming recovery from the key features, such as the dual variables \(\lambda\) and \(\mu\) in problem P2. Predicting the beamforming matrix directly may lead to high complexity since the number of the variables in the beamforming matrix depends on both the number of users \(K\) and the number of BS antennas \(N_t\).

Thus instead of predicting the beamforming matrix directly, we predict some key features (i.e., the dual variables \(\mu\) and \(\lambda\)) whose variables are much less than those in the beamforming matrix. Then these key features are used to recover the beamforming matrix in the signal processing module.

The adopted DL structure takes the convolutional neural network (CNN) architecture as the backbone because the parameter sharing adopted in the CNN can reduce the number of the learned parameters when compared to a fully-connected DNN. Moreover, CNN is well known to be effective for extracting features, which will benefit the generation of the beamforming solution using the channel features. The adopted DL structure includes two main modules: the neural network module and the signal processing module [51]. Here we give a short description about the two modules, and for more details readers are referred to [47].

1) **Neural Network Module:** The neural network module is a data-driven approach to approximate the mapping function from the complex channels to the key features. In addition to the input and output layers, the neural network module also includes convolutional (CL) layers, batch normalization (BN) layers, activation (AC) layers, a flatten layer, and a fully-connected (FC) layer. The input of the neural network module is the complex channel coefficients, which are not supported by the current neural network software. To address this issue, we separate the complex channel vector \(h = [h_1, \cdots, h_K]^T \in \mathbb{C}^{N_k \times 1}\) into two components \(\Re(h)\) and \(\Im(h)\) and form the new input \(\{\Re(h), \Im(h)\}^T \in \mathbb{R}^{2 \times N_k K}\), where \(\Re(h)\) and \(\Im(h)\) contain the real and imaginary parts of each element in \(h\), respectively. Each CL layer consists of many filters which apply convolution operations to the layer input, capture special patterns and pass the result to the next layer. The parameters of the filters are shared among different channel coefficients. The main function of the BN layer is to normalize the output of the CL layers by two trainable parameters, i.e., a “mean” parameter and a “standard deviation” parameter. Besides, the BN layers can reduce the probability of over-fitting and enable a higher learning rate.

AC layers help neural networks extract the useful information and suppress the insignificant points of the input data. The rectified linear unit (ReLU) and sigmoid functions are suitable choices for the last AC layer, since the predicted variables are continuous and positive numbers. The function of the flatten layer is to change the shape of its input into a vector for the FC layer to interpret. In addition to these functional layers, the loss function, marked ‘MSE/MAE’ on the output layer in Fig. 2, is also very important in the introduced DL structure. The mean absolute error (MAE) or the mean square error (MSE) is used in the loss function to update parameters. The loss function together with the learning rate determines how to update the parameters of the neural network module.

2) **Signal Processing Module:** The neural network module offers universality in learning the key features from data, while the signal processing module aims to recover the beamforming matrix from the predicted key features at the output layer. Different from the neural network module whose model is unknown, the signal processing module utilizes the (partially) known models of the data to recover the beamforming matrix. The learned key features and the functionalities in the signal processing module are designated according to the expert knowledge. Note that the expert knowledge is problem-dependent and has no unified form, but what is in common is that the expert knowledge can significantly reduce the number of variables to be predicted compared to the beamforming matrix [47]. For example, the dual forms of the original problems are the typical expert knowledge for beamforming.
optimization. The details of the signal processing module used to recover the beamforming matrix is provided in the next two subsections.

B. To Learn λ and μ and the Recovery Algorithm

With the above proposed general DL structure, we need to decide which features of dual optimization variables in \( \mathbf{P2} \) will be learned, and what signal processing function is needed to recover the beamforming matrix. The first option is to learn both \( \lambda \) and \( \mu \), so the output has \( K + N_t \) variables. Once they are learned, the following algorithm with steps taken from Algorithm 2 can be used to find a feasible beamforming solution that satisfies the per-antenna power constraints.

**Algorithm 4: To recover \( W \) from \( \lambda \) and \( \mu \)**

1. Given the learned solution \( \lambda \) and \(\mu\), Calculate \( \beta_k = \frac{1}{\lambda_k h_k^H G_k(\lambda, \mu)^{-1} h_k} \). Then update the achievable SINR in the uplink as

\[
\gamma = \min_k \frac{1}{\beta_k - 1}. \tag{23}
\]

2. Find the optimal normalized uplink beamforming as

\[
\bar{w}_k = \frac{G_k(\lambda, \mu)^{-1} h_k^*}{\|G_k(\lambda, \mu)^{-1} h_k^*\|}, \forall k. \tag{24}
\]

3. Find the downlink power \( \{p_k\} \) to achieve the SINR \( \gamma \), i.e., to solve the following linear equation set:

\[
\frac{\sum_{i=1,i\neq k}^K p_i |h_i^T \bar{w}_k|^2}{\sum_{i=1,i\neq k}^K p_i |h_i^T \bar{w}_k|^2 + N_0} = \gamma^*, \quad k = 1, \ldots, K. \tag{25}
\]

4. Update the downlink beamforming vector as \( \bar{w}_k = \sqrt{p_k} \bar{w}_k \) and \( W = [\bar{w}_1, \cdots, \bar{w}_K] \).

5. Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:

\[
W(n,:) = W(n,:) \sqrt{\min_n \frac{P_n}{\| e_n^T \bar{W}\|^2}}, \forall n. \tag{26}
\]

C. To Learn μ Only and the Recovery Algorithm

The above learning strategy is straightforward and fast if the learning result is satisfactory, however, the learning accuracy can be much improved if the number of variables is reduced. This motivates us to use the proposed DL structure to learn only the dual variable \( \mu \) with output size of \( N_t \), which contains \( K \) less variables than the above approach that learns both \( \lambda \) and \( \mu \). The idea of this approach is that given \( \mu \), the optimal \( \lambda \) can be efficiently optimized using Algorithm 2, which is more accurate than the learning approach above. An additional advantage is that the output size does not depend on the number of users, so it can more easily adapt to the varying number of users. Once \( \mu \) is learned, the following algorithm with steps taken from Algorithm 2 can be used to derive a feasible beamforming solution to the original problem \( \mathbf{P1} \).

**Algorithm 5: To recover \( W \) from \( \mu \)**

1. Given the learned solution \( \mu \), call Algorithm 1 to find the optimal \( \lambda \).

2. Calculate \( \beta_k = \frac{1}{\lambda_k h_k^H G_k(\lambda, \mu)^{-1} h_k} \). Then update the achievable SINR in the uplink as

\[
\gamma = \min_k \frac{1}{\beta_k - 1}. \tag{27}
\]

3. Find the optimal normalized uplink beamforming as

\[
\bar{w}_k = \frac{G_k(\lambda, \mu)^{-1} h_k^*}{\|G_k(\lambda, \mu)^{-1} h_k^*\|}, \forall k. \tag{28}
\]

4. Find the downlink power \( \{p_k\} \) to achieve the SINR \( \gamma \), i.e., to solve the following linear equation set:

\[
\frac{\sum_{i=1,i\neq k}^K p_i |h_i^T \bar{w}_k|^2}{\sum_{i=1,i\neq k}^K p_i |h_i^T \bar{w}_k|^2 + N_0} = \gamma^*, \quad k = 1, \ldots, K. \tag{29}
\]

5. Update the downlink beamforming vector as \( \bar{w}_k = \sqrt{p_k} \bar{w}_k \) and \( W = [\bar{w}_1, \cdots, \bar{w}_K] \).

6. Regulate the downlink beamforming. Update the beamforming vector as follows to satisfy all per-antenna power constraints:

\[
W(n,:) = W(n,:) \sqrt{\min_n \frac{P_n}{\| e_n^T \bar{W}\|^2}}, \forall n. \tag{30}
\]

D. Generalization of the Proposed DL Structure

In this section, we will generalize the proposed universal DL so that it can adapt to the change of the number of users and antennas. Although the above DL approaches can achieve satisfactory performance for beamforming design, applying the DL approaches to practical applications faces the difficulties caused by the dynamic wireless networks. In other words, when the number of transmit antennas \( N_t \) or the number of users \( K \) changes, a new model should be trained for prediction. This fact suggests that the applicability of the DL approaches is limited. Transfer learning and training set augmentation are effective ways to improve the generalization. The former transfers an existing model to a new scenario with some additional training and labelling effort [52], whereas the latter aims to train a large-scale model which adapts to different \( N_t \) and \( K \) by adding more samples into the training set, so that the training set can cover more possible scenarios. In this work, we adopt the latter method for simplicity. Without losing generality, we take the DL approach to learning \( \mu \) only as an example and give more details about the training set augmentation method.

In the training set augmentation method, we aim to train a large-scale model with \( 2N_t'K' \)-input and \( N_t' \)-output. In order to make the large-scale model adaptable to different \( N_t \) and \( K \) values, we generate an augmented training set. Different from the training set whose samples have the same \( N_t \) and \( K \) values, the samples in the augmented training set are diverse, i.e., the numbers of the transmit antennas and the numbers of users in different samples could vary. However, the size of each sample is fixed as \( 2N_t'K' \)-input and \( N_t' \)-output. For the
cases where \( N_t < N'_t \) (or \( K < K' \)), the redundant \( N'_t - N_t \) rows (or \( K - K_0 \) columns) of the channel matrix are filled with 0's. Similarly, the redundant \( N'_t - N_t \) elements of output are set as 0 when \( N_t < N'_t \). In each sample, we assume each \( K \in \{1, 2, \ldots, K'\} \) is generated with the equal probability of \( \frac{1}{K'} \) and each \( N_t \in \{1, 2, \ldots, N'_t\} \) is generated with the equal probability of \( \frac{1}{N'_t} \). Therefore, the occurrence probabilities of different \( K \) values are statistically equal among all samples and so are different \( N_t \) values. It is suggested that the number of samples in the augmented training set for the large-scale model should be 5-10 times as many as that in the training set with fixed \( N_t \) and \( K \) values. However, this approach works only if the number of users or antennas does not exceed the maximum values used in the training set, otherwise re-training will be needed.

V. PERFORMANCE EVALUATION

Both simulations and experiments are carried out to evaluate the performance of the proposed DL enabled beamforming optimization. We assume that all channel entries undergo independent and identically distributed Rayleigh flat-fading with zero mean and unit variance unless otherwise specified, and perfect CSI is available at the BS. All transmit power is normalized by the noise power.

The training samples (dual variables) are generated by solving the problem \( P2 \) using Knitro for its stability and efficiency, but can also be generated by solving the problem \( P1 \) using the bisection search method at the cost of more computational time during the offline training. In our simulation, we use 20000 training samples and 5000 testing samples, respectively. All of proposed DL networks have one input layer, two CL layers, two BN layers, three AC layers, one flatten layer, one FC layer, and one output layer. Besides, each CL layer has 8 kernels of size 3 × 3 and the first two AC layers adopt the ReLU function. Each CL applies stride 1 and zero padding 1 such that the output width and height of all CLs remain the same as those of the input [56]. To be specific, the input size of the first CL is \( 2 \times N_t \times K \times 1 \) and the output size is \( 2 \times N'_t \times K \times 8 \). Both the input size and output size of the second CL are \( 2 \times N'_t \times K \times 8 \). When parameter sharing is considered, the numbers of parameters in the first and second CL are \( 3 \times 3 \times 1 \times 8 = 72 \) (weights) + 8 (bias) = 80, and \( 3 \times 3 \times 8 \times 8 = 584 \), respectively, with a total of 664. When no parameter sharing is considered, the numbers of parameters in the two CLs are \( (2 \times N_t \times K \times 8) \times (3 \times 3 \times 1 + 1) = 160 N_t K \) and \( 2 \times N'_t \times K \times (3 \times 3 \times 8 + 1) = 1168 N'_t K \), respectively, with a total of 1328 \( N_t K \). Adam optimizer [57] is used with the mean squared error based loss function. We adopt the sigmoid function in the last AC layer.

We will compare the performance and running time of the following schemes when possible:

1) The optimal solution to solve \( P2 \) using Knitro.
2) The proposed subgradient algorithm (Algorithm 2) in Section III.
3) The proposed solution based on learned \( \lambda \) and \( \mu \).
4) The proposed solution based on learned \( \mu \) only.
5) ZF Solution [25].

a) When \( N_t = K \), pseudo inverse of the channel is the optimal beamforming direction, i.e.,

\[
\hat{W} = H^\top (H^\top H)^{-1}, \tag{31}
\]

and the achievable SINR is \( \Gamma_{ZF} = \min_n \frac{P}{\|\hat{w}_n\|^2} \).

The overall optimal beamforming matrix is given by \( W = \sqrt{\Gamma_{ZF}} \hat{W} \).

b) However, when \( N_t > K \), the optimal solution relies on solving the following SOCP problem \( P7 \), so the associated complexity is high:

\[
P7: \max_{W, \Gamma} \Gamma \quad \text{s.t.} |h^T_k w_k|^2 \geq \Gamma, \forall k, \h_k^T w_j = 0, \forall k \neq j, \quad p_n = \|\hat{e}_n^T W\|^2 \leq P_n, \forall n. \tag{32}
\]

c) When \( N_t < K \), there is no feasible ZF solution.

6) RZF Solution [58]. This is a low-complexity heuristic solution that improves the performance of ZF especially at the low SNR region. The beamforming direction is given by:

\[
\hat{W} = H^\top (H^\top H + \alpha I_{K \times K})^{-1}, \tag{33}
\]

where \( \alpha = \frac{K N_t}{\sum_{n=1}^{N} P_n} \) and the overall beamforming matrix is given by \( W = \sqrt{\min_n \frac{P}{\|\hat{w}_n\|^2}} \).

For fair comparison, the convergence of all iterative algorithms is achieved when the relative change of the objective function values is below \( 10^{-8} \). All algorithms are implemented on an Intel i7-7700U CPU with 32 GB RAM using Matlab R2017b. One NVIDIA Titan Xp GPU is used to train the neural network.

A. Simulation Results

We first compare the SINR and running time results for a system with \( N_t = K = 4 \) in Fig. 3. In Fig. 3 (a), we can see that both the proposed subgradient solution and the solution based on learned \( \mu \) can achieve close to optimal solution and outperform the RZF solution and the ZF solution especially at the low signal to noise (SNR) regime. As the SNR increases, all solutions converge to the optimal solution. Fig. 3 (b) shows that both of the proposed learning based solutions can achieve more than an order of magnitude gain in terms of computational time when compared to the optimal algorithm. The proposed subgradient algorithm is more efficient than the optimal solution using Knitro. ZF and RZF solutions have the lowest possible complexity because there is no optimization involved. In addition, we compare the robustness of various schemes against channel errors in Fig. 4. The channel vectors are modelled as \( \hat{h}_k = \hat{h}_k + \sigma_e \hat{h}_k, \forall k \), where \( \hat{h}_k \) is the imperfect channel estimate, \( e_k \sim \mathcal{CN}(0, I_N) \) is the channel error vector and \( \sigma^2 \) is the variance of channel estimation error. As expected, we can see that the channel estimation error causes degradation of the SINR performance for all the solutions. However, the results show that the proposed learning based solutions and the optimal solution are very robust, but the performance loss of the ZF and RZF beamforming is severe.
Next we demonstrate the scalability of the algorithms when $N_t = K$ and the number of users varies from 2 to 10 when $P_n = 10$ dB in Fig. 5. As can be seen from Fig. 5 (a), as both the numbers of users and antennas increase, the achievable SINR first decreases and then increases. The performance of the ZF and RZF solutions drops quickly. As the number of users increases, both learning based solutions significantly outperform the ZF solution and the performance gap is enlarged while their gap to the optimal solution remains constant. Fig. 5 (b) shows the complexity performance. The proposed algorithm that learns both $\lambda$ and $\mu$ has a lower complexity. As the number of users increases, e.g., when $K = 10$, it can achieve nearly 50-fold gain in terms of computational time when compared to the optimal algorithm. The proposed algorithm that learns only $\mu$ achieves 0.5 dB higher SINR than that learns both $\lambda$ and $\mu$ at the cost of slightly increased time complexity. Next we examine the SINR performance of the system using a more realistic 3GPP Spatial Channel Model (3GPP TR 25.996) [59] as shown in Fig. 6. We consider a scenario of urban micro cells and assume the distances between the BS and the users are between 50 m and 300 m and distributed uniformly. The total system bandwidth is 20 MHz. Similar trends of the algorithms are observed in Fig. 6 as those in Fig. 5 (a), and both learning based solutions still significantly outperform the RZF and the ZF solutions.

We then consider the performance of a system with $N_t = 10$ transmit antennas at the BS, and vary the number of users $K$ when $P_n = 10$ dB in Fig. 7 (a). It is noticed that there is about 1 to 2 dB gap between the learned solutions and the optimal solution, while the ZF solution is almost optimal when $N_t > K$. However, from Fig. 7 (b), we can see that the ZF solution has the highest complexity in this case because its solution needs to be optimized via solving the SOCP problem $P7$. The proposed algorithm that learns both $\lambda$ and $\mu$ achieves more than two orders of magnitude gain in terms of computational complexity when compared to the ZF solution.

Next we demonstrate the generalization property of our proposed algorithm that learns only $\mu$. We train a model with $N_t = K = 10$ only once, and then use it when $N_t \leq 10$
adding more antennas can improve the SINR performance because of the spatial gain.

### B. Testbed Results

To evaluate the proposed learning-based algorithm in a real-world scenario, we have implemented a multi-user beamforming testbed system based on SDR in our lab environment.

1) **Testbed Setup:** The multi-user beamforming testbed system is based on the SDR structure, which consists of one PC hosting Matlab, a Gigabit Ethernet switch, four NI’s USRP devices as transmitters or receivers and a CDA-2990 Clock Distribution Device. The USRP devices and the Clock Distribution Device for synchronization are illustrated in Fig. 9. We adopt the SDR system since it provides a flexible development environment as well as a practical prototype. The USRP devices are exploited as the radio fronts in the SDR system, which can support different interfacing methods including PCIe and Gigabit Ethernet connections. Besides, the USRP devices can support a wide range of baseband signal processing platforms, including Matlab, Labview and GNU Radio. The transmitters and receivers are implemented using USRP-2950 devices, which support the Radio Frequency (RF) range from 50MHz to 2.2GHz [60]. For the evaluation purpose, the 900 MHz Industrial, Scientific and Medical (ISM) frequency band is used. The key parameters of the multi-user beamforming system are listed in Table I.

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In the experiment, we consider the scenario consisting of one BS with four transmit antennas and four single-antenna users.
TABLE I
TESTBED SYSTEM CONFIGURATION

<table>
<thead>
<tr>
<th>Parameters</th>
<th>Descriptions</th>
</tr>
</thead>
<tbody>
<tr>
<td>Clock and PPS source</td>
<td>CDA-2990 10MHz and 1PPS</td>
</tr>
<tr>
<td>Radio Front</td>
<td>USRP 2950, 50MHz-2.2GHz</td>
</tr>
<tr>
<td>Antennas</td>
<td>Tri-band SMA-703</td>
</tr>
<tr>
<td>Modulation</td>
<td>QPSK</td>
</tr>
<tr>
<td>Prefix</td>
<td>Gold code of length 127</td>
</tr>
<tr>
<td>Baseband Sample Rate</td>
<td>40 kilosamples/second (kops)</td>
</tr>
<tr>
<td>Pulse Shaping</td>
<td>Raised Cosine Filter of squared root shape with rolloff factor 0.8 and decimation factor 8</td>
</tr>
<tr>
<td>Channel Estimation</td>
<td>MMSE Estimator</td>
</tr>
</tbody>
</table>

users, i.e., $N_t = K = 4$. We combine two USRP-2950 devices as a cooperative four-antenna transmitter and employ two USRP-2950 devices as four individual single-antenna users. All channels on the USRP devices are synchronized using the CDA-2990 Clock Distribution Device. The omnidirectional tri-band SMA-703 antennas are used for both the transmitters and the receivers, while the receiver antennas are extended using RF cables. Specifically, both static and dynamic channel conditions are examined to evaluate the proposed learning-based beamforming algorithms. For the static channel scenario, the transmitter antennas are placed next to each other with a space of 0.1 m, while the receiver antennas are placed 1.5 m away from the transmitter antennas as well as from each other. For the dynamic scenario, a low-mobility scenario is simulated, where one of the receiving antennas is moving at the speed of 0.6 m/s. Besides, the experiment also exploits different transmitter powers to evaluate the algorithms’ performance in different SNR configurations, where 0 dB of transmit power gain corresponds to a transmit power of $-70$ dBm. Since the multi-user beamforming system coordinates several USRP devices as transmitters and receivers at the same time, a Gigabit Ethernet switch is used to enable multiple USRP interfacing.

The baseband signal processing modules and the proposed learning-based beamforming algorithms are implemented as Matlab function scripts on a PC with 1 Intel i7-4790 CPU Core, and RAM of 32GB. In the experiment, all users are sharing the same channel and they all use the Quadrature Phase Shift Keying (QPSK) modulation. The payloads are prefixed with different Gold sequences for each user, which are exploited for both synchronization and channel estimation. Besides, all baseband signals are shaped using a Raised Cosine Filter. During the experiment, each user decodes its own payload and provides channel estimation as feedback to the transmitter. The transmitters and receivers are controlled using different Matlab sessions, while the channel estimation information is exchanged locally on the PC’s cache storage. The beamforming algorithms optimize the beam weight vectors using the aggregated channel estimation information. The transmitter applies the optimized beam weight vectors to generate the signals for each antenna before transmission.

2) Experiment Results and Discussions: To demonstrate the performance of the proposed learning algorithm (based on learned $\lambda$ and $\mu$), three benchmark algorithms are implemented on the multi-user beamforming system, which are the theoretically optimal solution, the ZF solution and the RZF solution. Each algorithm is evaluated under both static and dynamic conditions, and we choose bit error rate (BER) as the performance metric. In order to generate the BER performance of each solution, a real-time experiment is conducted using the testbed illustrated in Fig. 9 with different transmitter power. For each transmit power, the BS sends $10^4$ packets each containing 256 QPSK symbols and the BER is calculated based on the averaged bit error of all packets.

Fig. 10 depicts the BER results in the static and dynamic channel conditions as the transmit power gain varies. Under the static condition as shown in Fig. 10 (a), the proposed learning-based algorithm outperforms the ZF solution and RZF solution across the considered transmit power range. Specifically, the BER performance gain of the learning based algorithm is approximately 4 dB over the ZF solution and 3 dB over the RZF solution in the relatively low transmit SNR regime, and this performance gain reduces as the transmit SNR grows. Compared to the theoretically optimal solution, the learning-based algorithm has a close performance in the low transmit SNR regime but becomes inferior for high transmit SNR conditions. This is expected because under
static channel conditions, there is sufficient time to implement
the theoretically optimal algorithm, therefore it achieves the
best performance. However, the algorithms show difference
BER performance under the dynamic channel conditions,
as depicted in Fig. 10 (b). The learning-based algorithm out-
performs all benchmark algorithms in the relatively medium to
high SNR ranges, which corresponds to 0 to 12 dB in Fig. 10
(b). It is worth noting that the learning-based algorithm is
superior to the alleged theoretically optimal solution under
dynamic channel conditions and in particular, the maximum
achieved BER performance gain is approximately 1 dB over
the theoretically optimal solution. This result is expected, and
can be explained as follows. The beamforming algorithms
require up to date CSI for optimization, but the computational
delay of the theoretically optimal solution is considerably
long, and by the time the solution is found, the channel
would have changed. In other words, the theoretically optimal
beamforming solution is optimized only based on the outdated
CSI, and therefore the mismatch leads to performance
degradation, and the theoretically optimal performance can no
longer be guaranteed. This can be verified by the typical time-
consumption performance for the considered algorithms as
illustrated in Table II. This performance degradation becomes
worse when the channel conditions are dynamic than that
in the static channel conditions as shown by Fig. 10 (a) and
Fig. 10(b). It is seen from Table II that the ZF and RZF
solutions require much less computational time when optimiz-
ing the beamforming weights, so the performance of the ZF
solution is close to that of the theoretically optimal solution
(degraded by operating on outdated CSI) in the experiment,
and the RZF solution even outperforms the optimal solution.
However, the BER performance of the ZF and RZF solutions is
still inferior to that of the proposed learning-based algorithm.
It is worth noticing that under both the static and dynamic
channel conditions, the precise channel models are not known,
so in the experiment, we resort to the trained neural network
based on the small-scale fading for online learning of the
beamforming solution. The results in Fig. 10 show that the
trained network for one channel model generalizes well to
cope with different channel conditions and this will greatly
reduce the need to re-train the neural network.

VI. CONCLUSION AND FUTURE DIRECTIONS

In this paper, we have developed deep learning enabled
solutions for fast optimization of downlink beamforming under
the per-antenna power constraints. Our solutions are both
model driven and data driven, and are achieved by exploiting
the structure of the beamforming problem, learning the dual
variables from labelled data and then recovering the original
beamforming solutions. Our solutions can naturally adapt to
the varying number of active users in dynamic environments
without re-training thus making it more general. The simulation
results have shown the superior performance-complexity
tradeoff achieved by the proposed solutions, and the results
have been further verified by the testbed experiments using
software defined radio.

We would like to point out a few promising future direc-
tions. This paper assumes that perfect CSI is available; how-
ever in practice, CSI estimation is never perfect. One future
direction would be to investigate a more advanced robust learn-
ing framework to mitigate channel estimation errors or other
types of impairments. As a step further, another promising
future direction will be to study how to use deep learning to
map directly from the pilot signals to the beamformed signals,
bypassing the explicit channel estimation step.

In order to reduce computational complexity of the training
process when the channel conditions change, one possible
method is to use a wide range of channel realizations during
the off-line training phase, in order that the neural network can
learn to generalize from a wider range of channel variations.
Another approach is to employ transfer learning [52]. The
main idea is that knowledge learned from one training task
for a given channel condition may be transferred to a similar
training task for a different channel condition, and can help
train a new model with additional examples, which is worthy
of further study.

APPENDIX

A. Proof of the Convergence and Optimality of Algorithm 1
to Solve P4

The proof has two parts. The first part is devoted to the proof
of convergence and the second part addresses the uniqueness
and optimality of the fixed point after convergence.

Let us start with \( \gamma^{(j)} (j \geq 1) \) which is achievable for
the power vector \( \lambda^{(j)} \). It is easily seen that given \( \gamma^{(j)} \),
\( \mathcal{I}(\lambda^{(j)}) \) (user index \( k \) is omitted for convenience) is a standard
interference function, which satisfies the following properties
\[54], [55]:
\( \lambda^{(j)} \) is component-wise monotonically decreasing;
\( \mathcal{I}(\lambda) \geq \mathcal{I}(\lambda^{(j)}) \) for all \( j \);
\( \lambda^{(j)} \) for all \( j \), are all feasible solutions given the SINR
constraint \( \gamma^{(j)} \).

Assume that at the \( j \)-th iteration, the dual variable is \( \lambda^{(j)} \)
and the achievable SINR is \( \gamma^{(j)} \). Then at the \( (j+1) \)-th
iteration, according to (P1), \( \lambda^{(j+1)} \leq \lambda^{(j)} \) \( \forall k \), and as such
\( \gamma^{(j)} \geq 1 \) and \( \lambda^{(j+1)} \leq \lambda^{(j)} \) \( \forall k \) in Step 4. According to P2,
in Step 5 we have the SINR result \( \gamma^{(j)}(\lambda^{(j+1)}) < \gamma^{(j)}(\lambda^{(j)}). \)
Then, according to (P3), \( \gamma^{(j)}(\lambda^{(j+1)}) \geq \gamma^{(j)} \) \( \forall k \), and therefore
\( \gamma^{(j+1)} = \min_{\lambda^{(j+1)}} \gamma^{(j)}(\lambda^{(j+1)}) \geq \min_{k} \gamma^{(j)}(\lambda^{(j+1)}) \geq \gamma^{(j)} \),
i.e., the balanced SINR \( \gamma^{(j)} \) is increasing as the iteration goes.
Since \( \gamma^{(j)} \) is upper bounded, the algorithm converges to a

<table>
<thead>
<tr>
<th>Processing/Solution</th>
<th>CSI Feedback Period</th>
<th>Learned Solution</th>
<th>Theoretically Optimal Solution</th>
<th>ZF Solution</th>
<th>RZF Solution</th>
</tr>
</thead>
<tbody>
<tr>
<td>Typical Time (second)</td>
<td>2x10^{-2}</td>
<td>5x10^{-3}</td>
<td>8x10^{-2}</td>
<td>2x10^{-4}</td>
<td>2x10^{-4}</td>
</tr>
</tbody>
</table>

TABLE II
TRIAL TIME PERFORMANCE IN THE EXPERIMENT SCENARIO
fixed point $\lambda^{(\infty)}$. Next, we prove that the fixed point is also optimal.

We see that $\lambda_k^{(\infty)}$ satisfies the following fixed-point equation:

$$\lambda_k^{(\infty)} = \gamma^{(\infty)} \tilde{f}_k(\lambda_k^{(\infty)}) \forall k.$$  \hfill (34)

and it satisfies the total virtual uplink power is $\sum_k \lambda_k^{(\infty)} = \frac{1}{N_0}$. Clearly, the total uplink transmit power is a monotonic non-decreasing function of the SINR constraint. This implies that there is no solution $\lambda^*$ which provides a strictly higher SINR $\gamma^* > \gamma^{(\infty)}$ but still maintains the power constraint $\sum_k \lambda_k^{(\infty)} = \frac{1}{N_0}$. \hfill $\square$

B. Proof that $f(\mu)$ of P4 is a Concave Function in $\mu$

Proof: First note that Algorithm 1 to solve P4 belongs to a fixed-point iteration, which means a solution $\{\Gamma, \lambda\}$ that satisfies the first two constraints (9) and (10) with equality ensuring an optimal solution. This indicates there is no local optimum, and the gap between P4 and its dual problem is zero. Then it suffices to prove that the objective function of the dual problem of P4 is concave in $\mu$.

By using (11) of [23], we can rewrite P4 as

P4': $f(\mu) = \max_{\Gamma, \lambda} \Gamma$

s.t. $\sum_{i=1}^K \lambda_i h_i^T h_i + \text{Diag}(\mu) - \left(1 + \frac{1}{\Gamma}\right) \lambda_k h_k^T h_k > 0, \forall k$,$ $K

$\sum_{k=1}^K \lambda_k N_0 = 1$,$ \lambda \geq 0$.

(35)

Its Lagrangian is

$L(\mu, \lambda, a, b, \{C_k\}) = \Gamma + a \left(\sum_{k=1}^K \lambda_k N_0 - 1\right) + b^T \lambda$

$- \sum_{k=1}^K \text{trace} \left(\sum_{i=1}^K \lambda_i h_i^T h_i^T + \text{Diag}(\mu) - \left(1 + \frac{1}{\Gamma}\right) \lambda_k h_k^T h_k^T \right) C_k$, \hfill (36)

where $a, b, \{C_k\}$ are dual variables. Note that it is derived based on the maximization rather than the commonly used minimization of an objective function.

The dual objective function is expressed as

$G(\mu, a, \{C_k\}) = \min_{\Gamma, \lambda} L(\mu, \lambda, a, b, \{C_k\})$ which is to be minimized over $(a, b, \{C_k\})$ and only contains a linear term of $\sum_{k=1}^K \text{trace} \left(\sum_{i=1}^K \lambda_i h_i^T h_i^T + \text{Diag}(\mu) \right)$ about $\mu$, and the constraints of the dual problem (although not derived here) do not involve $\mu$. Therefore the dual objective function $\min G(\mu, a, \{C_k\})$ is a point-wise minimum of a family of affine functions about $\mu$ and as a result concave [61, Sec.3.2.2], so is $f(\mu)$. This completes the proof. \hfill $\square$

C. To Find the Subgradient Euclidean Projection in Algorithm 2

The Euclidean projection is needed when the update of $\mu$ based on the subgradient in Algorithm 2 does not fall into the feasible set $S$. It needs to solve the following optimization problem:

\[ \text{P5:} \min_{\nu} \| \nu - \mu \|^2 \quad \text{s.t.} \quad \sum_{n=1}^{N_t} \nu_n P_n = 1, \quad \nu \geq 0, \]  \hfill (37)

where $\mu = \mu_k^{(i)} + \alpha_i \text{Diag}([e_n^T W]^2)$. Although P5 is a convex problem and can be solved by a standard numerical algorithm, below we derive its analytical property and propose a more efficient bisection algorithm to solve it.

Its Lagrangian can be expressed as

$$L = \| \nu - \mu \|^2 + x(\sum_{n=1}^{N_t} \nu_n P_n - 1) - \sum_{n} \nu_n x_n,$$  \hfill (38)

where $x$ and $y_n \geq 0$ are dual variables.

Setting its first-order derivative to be zero leads to

$$\nu_n = \frac{2\mu_n + y_n - x_n P_n}{2} = \max \left(\frac{2\mu_n - x_n P_n}{2}, 0\right).$$  \hfill (39)

Substitute it to $\sum_{n=1}^{N_t} \nu_n P_n = 1$ and we get

$$\sum_{n=1}^{N_t} \max\left(\frac{2\mu_n - x_n P_n}{2}, 0\right) P_n = 1.$$  \hfill (40)

Therefore the remaining task is to find $x$ that satisfies (40). Obviously the left hand side of (40) is monotonic in $x$, so we propose the following bisection method to find the optimal $x$.

Algorithm 3 to Solve P5:

1) Set the upper and lower bounds of $x$ as $x^U$ and $x^L$.

Repeat the following steps until convergence.

2) Calculate $x^* = \frac{x^U + x^L}{2}$.

3) If $\sum_{n=1}^{N_t} \max\left(\frac{2\mu_n - x_n P_n}{2}, 0\right) P_n > 1$, $x^L = x^*$; otherwise $x^U = x^*$.

REFERENCES


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