Machine Learning for Wireless Networks with Artificial Intelligence: A Tutorial on Neural Networks

Mingzhe Chen*, Ursula Challita†, Walid Saad‡, Changchuan Yin*, and Mérouane Debbah §

*Beijing Laboratory of Advanced Information Network, Beijing University of Posts and Telecommunications, Beijing, China 100876, Email: chenmingzhe@bupt.edu.cn and ccyin@bupt.edu.cn.
† School of Informatics, The University of Edinburgh, Edinburgh, UK. Email: ursula.challita@ed.ac.uk.
‡Wireless@VT, Bradley Department of Electrical and Computer Engineering, Virginia Tech, Blacksburg, VA, USA, Email: walids@vt.edu.
§Mathematical and Algorithmic Sciences Lab, Huawei France R & D, Paris, France, Email: merouane.debbah@huawei.com.

Abstract

Next-generation wireless networks must be able to support ultra-reliable, low-latency communication and intelligently manage the internet of things (IoT) devices in real-time dynamic environment. Such communication requirements and mobile edge and core intelligence can only be realized by integrating fundamental notions of artificial intelligence (AI) and machine learning across the wireless infrastructure and end-user devices. In this context, this tutorial introduces the use of comprehensive concepts of machine learning, in general, and artificial neural networks (ANNs), in particular, and their potential applications in wireless communications. For this purpose, we present a comprehensive overview on a number of key types of neural networks that include feed-forward, recurrent, spiking, and deep neural networks. For each type of neural network, we present the basic architecture and training procedure, as well as the associated challenges and opportunities. Then, we provide a panoramic overview on the variety of wireless communication problems that can be addressed using ANNs. For each application, we present the main motivation for using ANNs along with their associated challenges while also providing a detailed example for a use case scenario. Meanwhile, for each individual application, we present a broad overview on future works that can be addressed using ANNs. In a nutshell, this article constitutes a comprehensive overview of machine learning tailored to the demands of communications and network engineers.
I. INTRODUCTION

The wireless networking landscape is undergoing a major revolution. The smartphone-centric networks of yesteryears are gradually morphing into a massive Internet of Things (IoT) ecosystem [1]–[4] that integrates a heterogeneous mix of wireless-enabled devices ranging from smartphones, to drones, connected vehicles, wearables, sensors, and virtual reality apparatus. This unprecedented transformation will not only drive an exponential growth in wireless traffic in the foreseeable future, but it will also lead to the emergence of new and untested wireless service use cases, that substantially differ from conventional multimedia or voice-based services. For instance, beyond the need for high data rates – which has been the main driver of the wireless network evolution in the past decade – next-generation wireless networks must be able to deliver ultra-reliable, low-latency communication [5]–[8], that is adaptive, in real-time to the rich and dynamic IoT environment. For example, drones and connected vehicles [9]–[12] will place autonomy at the heart of the IoT. This, in turn, will necessitate the deployment of ultra-reliable wireless links that can provide real-time, low-latency control for such autonomous systems. Meanwhile, in tomorrow’s wireless networks, large volumes of data must be collected, periodically and in real-time, across a massive number of sensing and wearable devices. Such massive short-packet transmissions will lead to a substantial traffic over the wireless uplink, which has traditionally been much less congested than the downlink. This same wireless network must also support cloud-based gaming [13], immersive virtual reality services, real-time HD streaming, and conventional multimedia services. This will ultimately create a radically different networking environment whose novel applications and their diverse quality-of-service (QoS) and reliability requirements mandate a fundamental change in the way in which wireless networks are modeled, analyzed, designed, and optimized.

The need to cope with this ongoing and rapid evolution of wireless services has led to much research that investigates what the optimal cellular network architecture will be within the context of the emerging fifth generation (5G) wireless networks (e.g., see [14] and references therein). While the main ingredients for 5G – such as dense small cell deployments, millimeter wave (mmWave) communications, and device-to-device (D2D) communications – have been identified, integrating them into a truly harmonious wireless system that can meet the IoT challenges requires instilling intelligent functions across both the edge and core of the network.
These intelligent functions must be able to adaptively exploit the wireless system resources and generated data, in order to optimize network operation and guarantee, in real-time, the QoS needs of emerging wireless and IoT services. Such mobile edge and core intelligence can only be realized by integrating fundamental notions of artificial intelligence (AI) [15] across the wireless infrastructure and end-user devices.

While the notion of artificial intelligence can be traced back to the mythological Greek bronze man Talos [16] – an artificially intelligent man-machine created to protect the island of Crete from invaders, its true potential has only been recently realized owing to the substantial developments in machine learning [17], in general, and neural networks [18], in particular. Indeed, machine learning tools are undoubtedly the primary vessel that is carrying artificial intelligence to fruition across a myriad of applications [19]–[26] that range from computer vision to natural language processing, robotics, and autonomous systems. As such, creating AI-enabled wireless networks is contingent upon developing the right set of machine learning and neural network tools that can provide the plethora of AI function needed in future wireless networks and unleash the true potential of the IoT. Such tools must naturally be tailored to the unique features of the wireless environment, which is evidently quite different from the traditional applications of AI, such as in robotics and computer vision.

For instance, artificial intelligence is expected to play several roles in the next-generation of wireless networks [27]. First, the most natural application of AI and machine learning is to exploit big data analytics to enhance situational awareness and overall network operation. In this context, AI will provide the wireless network with the ability to parse through a massive amount of data, generated from multiple sources that range from wireless channel measurements and sensor readings to drones and surveillance images, in order to create a comprehensive operational map of the massive number of devices within the network. This map can, in turn, be exploited to optimize various functions, such as fault monitoring and user tracking, across the wireless network.

Second, beyond its powerful prediction and data analytics functions, AI will be a major driver of intelligent and data-driven wireless network optimization. For instance, machine learning tools will enable the introduction of intelligent resource management tools, that can be used to address a variety of problems ranging from cell association and radio access technology selection to frequency allocation, spectrum management, power control, and intelligent beamforming. In
contrast to conventional distributed optimization technique, that are often done iteratively in an offline or semi-offline manner, AI-guided resource management mechanisms will be able to operate in a fully online manner by learning, in real time, the states of the wireless environment and the network’s users. Such mechanisms will therefore be able to continuously improve their own performance over time which, in turn, will enable more intelligent and dynamic network decision making. Such intelligent decision making is essential for much of the envisioned IoT and 5G services, particularly those that require real-time, low latency operation, such as autonomous driving, drone guidance, and industrial control. In fact, if properly designed, machine learning-based AI optimization algorithms will provide inherently self-organizing, self-healing, and self-optimizing solutions for a broad range of problems within the context of network optimization and resource management. Such AI-driven self-organizing solutions are particularly apropos for ultra dense wireless networks in which classical centralized and distributed optimization approaches can no longer cope with the scale and heterogeneity of the network.

Third, beyond its system-level functions, AI can play a key role at the physical layer of a wireless network. As seen in recent works [28]–[33], neural networks and machine learning tools can be used to redefine the way in which physical layer functions, such as coding and modulation, are designed and operated, at both transmitter and receiver levels, within a generic communication system. Such an AI-driven approach has been shown [28]–[33] a lot of promise in delivering lower bit error rates and better robustness to the wireless channel impediments.

Last, but not least, the rapid deployment of highly user-centric wireless services, such as virtual reality [34]–[36], in which the gap between the end-user and the network functions is almost minimal, strongly motivate the need for wireless networks that can track and adapt to their human user behavior. In this regard, machine learning is perhaps the only tool that is capable to learn and mimic human behavior, which will help in creating the wireless network to adapt its functions to its human users, thus creating a truly immersive environment and maximizing the overall quality-of-experience (QoE) of the users.

Clearly, AI-based system operation is now longer a privilege, but rather a necessity for future wireless networks. AI-driven wireless network designs will pave the way towards an unimaginably rich set of new network functions and wireless services. For instance, even though 5G networks may not be fully AI capable, we envision that the subsequent, sixth generation (6G) of wireless cellular networks will be almost completely reliant on AI and machine learning, as
evidenced by recent developments of AI enabled mobile networks proposed by Huawei [37] and “big innovation house” proposed by Qualcomm [38]. As such, the question is no longer if machine learning tools are going to be integrated into wireless networks but rather when such an integration will happen. In fact, the importance of an AI-enabled wireless network has already been motivated by a number of recent wireless networking paradigms such as mobile edge caching, context-aware networking, big data analytics, location-based services, and mobile edge computing [39]–[45], the majority of which rely on machine learning or AI-inspired tools. However, despite their importance, these works remain rather focused on the wireless challenges, rather than on the synergistic integration of wireless and AI. In consequence, these works are of narrow focus and do not provide any broad, tutorial-like material that can shed light on the challenges and opportunities associated with the use of machine learning for designing wireless networks with AI.

Meanwhile, a number of surveys on machine learning and neural network applications in wireless networking have emerged, such as in [2], [4], [28], and [46]–[59]. Despite being interesting, these surveys are limited in a number of ways. First, they are often focused on one type of machine learning techniques (often deep learning such as in [28], [46], and [49]) and, as such, they do not capture the rich spectrum of available AI and machine learning frameworks. Second, they mostly restrict their scope to a single wireless application such as sensor networks [50]–[52] or cognitive radio networks [48], machine-to-machine communication [4], or physical layer design [28] and, hence, they do not cover the broad range of applications that can adopt machine learning in future networks. Third, a large number of the existing surveys, such as [2], [4], and [47]–[52], are highly qualitative and do not provide an in-depth technical and quantitative description on the variety of existing machine learning tools that are suitable for wireless communications. Last, but not least, a number of surveys [53]–[57] discuss the basics of neural networks with applications outside of wireless communications. However, these surveys are largely inaccessible for the wireless community, due to their reliance on examples from rather orthogonal disciplines such as computer vision. Moreover, most of the existing tutorials or surveys [57]–[59] do not provide concrete guidelines on how, when, and where to use different machine learning tools in the context of wireless networks. Finally, this introductory literature on machine learning for wireless networks such as in [2], [4], [28], and [46]–[59], is largely sparse and fragmented, hence, making it difficult to understand the intrinsic details of this broad
and far reaching area. In fact, there is a clear lack of a comprehensive and holistic tutorial that can explain, pedagogically and, in detail, how to develop machine learning solutions to endow wireless networks with artificial intelligence and realize the full potential of IoT systems, and beyond.

The main contribution of this paper is, thus, to provide one of the first holistic tutorials on the topic of machine learning for wireless network design. The overarching goal is to gather the state-of-the-art and emerging research contributions, from machine learning and wireless communications, that address the major opportunities and challenges in developing machine learning frameworks for understanding and designing wireless systems with artificial intelligence. In particular, we will provide a comprehensive treatment of artificial neural networks, which are one of the most important pillars of machine learning, with an emphasis on both new analytical techniques from neural networks and novel wireless application scenarios. After providing a substantial introduction to the basics of machine learning using neural networks, we introduce a classification of the various types of neural networks. For each type, we provide an introduction on their basic components, their training processes, and their use cases with specific example neural networks. Then, we overview a broad range of wireless applications that can make use of neural network designs. These applications include drone-based communications, spectrum management, wireless virtual reality, mobile edge caching and computing, and the IoT, among others. For each application, we first outline the main rationale for applying machine learning while pinpointing illustrative scenarios. Then, we overview the challenges and opportunities brought forward by the use of neural networks in the specific wireless application. We complement this overview with a detailed example drawn from the state-of-the-art and, then, we conclude by shedding light on the potential future works within each specific area. Overall, we anticipate that this tutorial will provide an invaluable reference on the use of neural networks in wireless communications and, as a result, will help in driving further research in this area.

The rest of this paper is organized as follows. In Section II, we introduce the basic of machine learning and artificial neural networks. Section III presents several key types of artificial neural networks such as feed-forward neural networks, recurrent neural networks, spiking neural networks, and deep neural networks. In Section IV, we introduce the use of artificial neural networks for wireless communication and corresponding challenges and opportunities. Finally, conclusions are draw in Section V.
II. ARTIFICIAL NEURAL NETWORKS: PRELIMINARIES

In this section, we first provide a brief overview on the basics of machine learning, while motivating the importance of neural networks. Then, we expose the fundamentals of a suite of neural network algorithms and techniques.

A. Machine Learning: A Brief Overview

Machine learning was born from pattern recognition and it is essentially based on the premise that machines should be endowed with artificial intelligence that enables them to learn from previous computations and adapt to their environment through experience [19]–[26]. Due to growing volumes of generated data – across critical infrastructure, communication networks, and smart cities – and the need for intelligent data analytics, the use of machine learning algorithms has become ubiquitous [60] across many sectors such as financial services, government, health care, technology, marketing and sales. Using machine learning algorithms to build models that uncover connections and predict dynamic system or human behavior, system operators can make intelligence decisions without human intervention. For example, using machine learning enables a system can grasp the entire knowledge of social relationships between individuals and can recognize individuals’ speech, face, and writing. Machine learning tasks often depend on the nature of their training data. In machine learning, training means the process that teaches the machine learning framework to achieve a goal such speech recognition. In other words, training enables the machine learning framework to discover potentially relationships between the input data and output data of this machine learning framework. Hence, they can be classified as follows [61]: a) supervised learning, b) unsupervised learning, c) semisupervised learning, and d) reinforcement learning.

Supervised learning algorithms are trained using labeled data. The labeled data represents that both the input data and its desired output data is known to the system. Supervised learning is commonly used in applications that have enough historical data. In contrast, training of unsupervised learning tasks is done without labeled data. The goal of unsupervised learning is to explore the data and infer some structure directly from the unlabeled data. Semisupervised learning is used for the same applications as supervised learning but it uses both labeled and unlabeled data for training. This type of learning can be used with methods such as classification, regression and prediction. Semisupervised learning is useful when the cost of a fully labeled
training process is too high. In contrast to the previously discussed learning that need to train with historical data, reinforcement learning (RL) is trained by the data from implementation. The goal of RL is to learn the environment and find the best strategy for different environments. RL algorithms are used for robotics, gaming, and navigation [62]. To perform these learning tasks, numerous types of machine learning frameworks have been developed. Among those frameworks, artificial neural networks (ANNs) [55] constitute one of the most important pillars of machine learning, as they are able to mimic human intelligence, to model complex relationships between inputs and outputs, to find patterns in data, or to extract the statistical structure in an unknown joint probability distribution from the observed data.

ANNs are inspired by the structure and functional aspects of biological neural networks, which can learn from observational complicated or imprecise data. ANNs process information in a manner that is analogous to the human brain. A given ANN is composed of a large number of highly interconnected processing elements working in parallel to solve a specific problem. ANNs can be used to extract patterns and detect trends that are too complex to be noticed...
by either humans or other computer techniques. An ANN can create its own organisation or representation of the information it receives during learning time. Moreover, an ANN can be used in a self-organized manner to learn how to do tasks based on the data given for training or initial experience. Within the context of wireless communications, as will be clearer from the latter sections, ANNs can be used to investigate and predict network and user behavior so as to provide users’ information for solving wireless network problems such as users’ associations, spectrum management, and power allocation. Moreover, recent developments of smart devices and mobile applications have significantly increased the level at which human users interact with mobile systems. Therefore, using ANNs to exact information from the users behaviors enables the wireless network can know the users’ future behaviors and, hence, design an optimal strategy so as to improve the QoS and reliability. A trained ANN can be thought of as an “expert” in dealing with human related data, which are a promising approach for solving the wireless communications problems such as resource allocation and power control. Next, we will introduce the concept and architecture of ANNs.

B. General Architecture of Artificial Neural Networks

The architecture of ANNs consists of a number of simple, highly interconnected processing elements known as neurons, which are used to mimic how the human brain learns. ANNs are essentially an artificial model of a human nervous system whose base elements are also neurons used to process information in the sense of cognition and transmit this information signal in the nervous system [64]. A neuron consists of nucleus, dendrites and axons [64]. Neurons are connected to each other by dendrites and axons as shown in Fig. 2. The connection point between two neurons is known as a synapse. The information signal transmitting to a neuron will change
its membrane potential. If this change makes the neuron’s membrane potential exceed a certain value, the neuron will send a signal to all of its connected neurons. This is how signals propagate through the human nervous system. ANNs use artificial neurons to mimic this operation of the human nervous system, thus enabling artificial intelligence. Mathematically, an artificial neuron consists of the following components: (a) A number of incoming connections, analogous to synapses on the dendrites; (b) A number of outcoming connections, analogous to synapses on the axon; and (c) An activation value assigned to each neuron, analogous to a biological neuron’s membrane potential. To capture the synapses’ strength for biological neurons, each connection will have a connection strength. The connection strength between two neurons is defined as weight value. The basic model for a neuron $j$ is shown in Fig. 3 and mathematically given by:

$$o_j = f \left( b_j + \sum_{i=k}^{N} n_{jk} \cdot w_{jk} \right),$$

where $n_{jk}$ is the input signal from neuron $j$ to neuron $i$ and $w_{jk}$ is the corresponding input weight value, $o_j$ is the output signal of neuron $j$, $b_j$ is the bias of neuron $j$, and $f(\cdot)$ is a nonlinear activation function. A bias value can shift the activation function, which is critical for successful learning. The activation function in a neural network will represent the rate of action potential firing in the cell of a neuron. The simplest form of an activation function is binary such as a Heaviside step function, which indicates whether a neuron is firing or not. However, using linear activation functions, many neurons must be used in computation beyond linear separation. Meanwhile, an ANN constructed using the activation function in (1) cannot reach a stable state after training since the value of activation function will increase without bound. Therefore, a normalizable activation function such as a sigmoid activation function can be used for each neuron instead of the activation function in (1). The selection of a type of activation functions in ANNs depends on the sought objectives such as analytic tractability, computational power, and type of desired output signal (logistic or continuous). For example, a given activation function in
is used due to its analytic tractability. Indeed, an ANN is a composition of multiple neurons connected in different ways and operating on different activation functions.

In general, the main components of an ANN that consists of multiple neurons will include the following:

- **Input layer** that consists of a number of neurons used to represent the input signal which will be transmitted in the neurons.
- **Output layer** that consists of a number of neurons used to represent the output signal.
- **Hidden layer** that consists of a number of neurons used to mimic human brain.
- **Input weight matrix** that represents the strength of the connections between the neurons in the input layer and the neurons in the hidden layer.
- **Neuron weight matrix** that represents the strength of the connections between the neurons in the hidden layer.
- **Output weight matrix** that represents the strength of the connections between the neurons in the output layer and the neurons in the hidden layer.

The connection strength in all weight matrices can be used to calculate the value of the activation function as shown in (1). For example, if $n_j$ in (1) represents a vector of input signal, then $w_j = [w_{j1}, w_{j2}, \ldots, w_{jN}]$ represents the value of the input weight, and, thus, the value of the activation function can be calculated by (1). The hidden layer is used to analyze the relationship between the input signal in the input layer and the output signal in the output layer. We can consider the neurons in the hidden layer as a black box that can be used to find the relationship between the input signal and the output signal without having any specifics on the activations of neurons in hidden layer. This is why this layer is known as a “hidden” layer. Given the basic components of an ANN, next, we will use an ANN example to explain specifically how these components consist of an ANN.

One of the simplest forms of an artificial neural network is the *feed-forward neural network (FNN)* [65], as shown in Fig. 4. An FNN consists of the following components: (a) Input layer, (b) Hidden layer(s), and (c) Output layer. In this FNN architecture, the connection strength between the neurons in the input layer and the hidden layer is captured by the input weight matrix. The connection strength between the neurons in the hidden layer is captured by the neuron by neuron weight matrix and the connection strength between the neurons in the output layer and hidden layer is output weight matrix. In an FNN, the connection between the neurons is unidirectional
and there is no connection between the neurons in a layer. Each neuron in the hidden layer calculates its output using an activation function such as the function in (1). Moreover, each neuron in the hidden layer has incoming connections only from the previous layer and outgoing connections only to the next layer, and, hence, this architecture is named feed-forward neural network. Given this connection method, the output of each neuron propagates from the input layer through the hidden layers to the output layer.

In fact, an ANN model is typically determined by the connection method, activation function, and layers. For instance, if each neuron in the hidden layer has incoming connections only from the previous layer and outgoing connections only to the next layer, the network is an FFN. If the connections between neurons form a loop, the network is called recurrent neural networks (RNNs) [66]. The network is called deep neural network [67], if there is more than one hidden layer in any given ANN model such as FFNs or RNNs. Having introduced the general architecture of an ANN, next, we will introduce the training methods for ANNs so as to perform learning tasks.

C. Training in Neural Networks

To learn information from their input data, ANNs must adjust the weights of the connections between the neurons in the system. The process of adjusting and updating the weights is known as the training process. Different learning tasks require different training algorithms. For example, to perform supervised learning tasks such as mobility prediction, ANNs are trained using labeled data. For unsupervised learning tasks such as clustering data into different groups, ANNs can be trained without the use of labeled data. In other words, for the unsupervised case, the desired
output for each input is not known. Next, we mainly introduce the general training algorithms for supervised learning tasks, as they constitute the foundation for other types.

For supervised learning tasks, the objective of training ANNs is to minimize the errors between the desired output signal and actual output signal. This error can be typically defined as:

\[ E(W, b) = 0.5 \cdot \sum (\|y(W, b, x) - y_D\|), \] (2)

where \( x \) is a vector of input signals, \( W \) is the weight matrix that is a combination of input weight matrix, hidden weight matrix, and output weight matrix, \( b \) is a vector of bias factor, and \( y_D \) is the desired output. \( y(W, b, x) \) is the actual output signal for each neuron, which can be calculated based on (1). In (1), the error is scaled by \( \frac{1}{2} \) to facilitate differentiation. In general, the most commonly used supervised learning algorithms for ANNs include gradient descent and backpropagation [68], which is a special case of gradient descent. Next, we will introduce these learning algorithms as they constitute a building block for any other learning algorithm.

In order to minimize \( E(W, b) \), we need to update the weights related with each neuron. One common approach to do this weight update is via the use of a gradient descent algorithm, as introduced next. For a given neuron \( j \), the error between the desired output signal \( o_{D,j} \) and actual output signal \( o_j \) is given by \( E_j(w_j, b_j) = 0.5 \cdot \sum (\|o_j(w_j, b_j, n_j) - o_{D,j}\|) \). The gradient descent algorithm is used to minimize \( E_j(w_j, b_j) \), which relies on the following equations:

\[ w_{jk,new} = w_{jk,old} - \gamma \frac{\partial E_j(w_j, b_j)}{\partial w_{jk}}, \] (3)

\[ b_{j,new} = b_{j,old} - \gamma \frac{\partial E_j(w_j, b_j)}{\partial b_j}, \] (4)

where \( \gamma \) is the learning rate and \( w_{jk} \) represents an element in \( w_j \). The first order derivative \( \frac{\partial E_j(w_j, b_j)}{\partial w_{jk}} \), will enable us to determine whether the error \( E_j(w_j, b_j) \) is decreasing or increasing when the weight value is \( w_{jk} \). Based on (3) and (4), ANNs can update the weight matrix and bias to find the optimal \( w_j \) and \( b_j \) that will minimize \( E_j(w_j, b_j) \). From (3) and (4), we can see that, the update of \( w_{jk} \) and \( b_j \) is easy to compute and, hence, the gradient descent algorithm is considered to be computationally fast, even on large datasets [69]. The gradient descent algorithm mentioned above only focuses on the update of a neuron. However, in an ANN, the signal is transmitted from one neuron to another neuron and, hence, we must design a rule to train these neurons. Backpropagation is the most widely used algorithm to calculate the gradient of the error such as \( \frac{\partial E_j(w_j, b_j)}{\partial w_{jk}} \) in (3) and \( \frac{\partial E_j(w_j, b_j)}{\partial b_j} \) in (4), so as to effectively minimize \( E(W, b) \) for
an ANN. In fact, backpropagation is a method to compute the gradient of each neuron for an ANN, which is just a chain rule, which is just a chain rule.

Next, we will introduce the backpropagation algorithm step by step. We first assume that neuron $j$ is at layer $L$ and neuron $i$ is at layer $L + 1$, which is closer to the output layer than layer $L$. The backpropagation procedure that can be used to update the weight value of $w_{ij}$ will proceed as follows:

- An input signal is transmitted from the input layer to the hidden layer of the ANN, until it reaches the output layer. If an ANN does not have hidden layers, (i.e., perceptron), the input signal will be directly transmitted to the output layer. Then, the activations of the neurons in all of the layers will be computed based on (1).

- The ANN will next compute the error between the targeted output and actual output based on (2) and will derive an error propagation value $\delta_i$ for each neuron. $\delta_i$ of neuron $i$ can be given by:

$$\delta_i = \frac{\partial E(W, b)}{\partial o_i} \cdot \frac{\partial o_i}{\partial n_{\text{sum},i}},$$

where $n_{\text{sum},i} = b_i + \sum_{k=1}^{N} n_{ik} \cdot w_{ik}$ is the summation of the input signal of neuron $i$ and its bias. In particular, if the activation function is a logistic function, $f(x) = \frac{1}{1+e^{-x}}$, then the error propagation of neuron $j$ can be given by [68]:

$$\delta_i = \begin{cases} (o_i - o_{D,i}) o_i (1 - o_i), & \text{neuron } i \text{ in the output layer,} \\ \left( \sum_{l \in \mathcal{L}_{L+1}} \delta_l w_{li} \right) o_i (1 - o_i), & \text{neuron } i \text{ in the hidden layer,} \end{cases}$$

where $\mathcal{L}_{L+1}$ represents the set of neurons at layer $L + 1$ (layer $L + 1$ is closer to the output layer than layer $L$). From (6), we can see that error propagation $\delta_i$ of a neuron in layer $L$ depends on the error propagation $\delta_l, l \in \mathcal{L}_{L+1}$, of a neuron at layer $L + 1$. Therefore, each neuron must transmit its error propagation parameter to the neurons at the formal layer. This is the central definition of backpropagation.

- Next, the ANN updates the weight value of $w_{ij}$, which can be given by $w_{ij,\text{new}} = w_{ij,\text{old}} - \gamma \delta_i o_j$.

- Repeat the above process until all weight values reach the minimum of $E(W, b)$. Note that backpropagation is not guaranteed to find a global minimum as it typically converges to
a local optimum since the training data set is finite and, hence, it must have some blindness in exploration.

In backpropagation, the gradient is computed based on the complete labeled data. However, if the size of the labeled data is very large, then using backpropagation may be time consuming. To reduce the time used for training when the size of the labeled data is very large, a stochastic gradient descent (SGD) algorithm [70] can be employed to update the weight values and bias. The stochastic gradient descent performs a weight value update for each training example. However, the SGD algorithm will often update frequently, which leads to the SGD algorithm overshooting – the weight values are larger or smaller than the optimum. To overcome these drawbacks of SGD, the mini batch gradient descent [71] can be used. The mini batch gradient descent is an algorithm that strikes a balance between stochastic gradient descent and batch gradient descent [71]. In mini-batch gradient descent, the gradient is computed based on a small number of samples, e.g., of around 10-500. One benefit of mini-batch gradient descent is that it can be performed in a distributed manner and, hence, it is time efficient.

In summary, gradient descent algorithms enable an ANN to be trained in a computational simple manner, and hence, they can quickly converge to a local minimum value, even on a large dataset. However, choosing a proper learning rate for the update of the weights and bias can be difficult. In fact, the learning rate determines the step size the algorithm uses to reach the minimizer and, thus, it has an impact on the convergence rate. In particular, a learning rate that is too large can cause the algorithm to diverge from the optimal solution. This is due to the fact that choosing very large initial learning rates will decay the loss function faster thus not allowing the model to explore better the optimization space. On the other hand, a learning rate that is too small will result in a low speed of convergence. In particular, the optimal value of the initial learning rate is dependent on the dataset under study, where for each dataset, there exists an interval of good learning rates at which the performance does not vary much [72]. Moreover, gradient descent algorithms often converge to a sub-optimal local minimum rather than the global minimum. To solve these challenges, several algorithms have been proposed, such as momentum SGD [71], nesterov accelerated gradient [73], Adagrad [74], and AdaDelta [75]. For instance, Adagrad and Adam are independent of the initial value of the learning rate while RMSProp relies heavily on a good choice of an initial learning rate.

It is worth noting that two central problems in training ANNs are overfitting and underfitting.
In particular, overfitting corresponds to the case in which the model learns the random fluctuations and noise in the training dataset to the extent that it negatively impacts the model’s ability to generalize when fed with new data. This occurs mainly when the dataset is too small compared to the number of model parameters that must be learned. On the other hand, underfitting occurs when a learning algorithm cannot capture the underlying trend of the data. Intuitively, underfitting occurs when the learning algorithm does not fit the data well enough. Therefore, one must carefully choose the architecture of an ANN along with the proper training methods to avoid overfitting and underfitting.

Using the aforementioned training algorithms, the values of the weight matrix and bias can be updated to their optimal values, and, hence, a trained ANN can output the desired output signal. However, each type of ANNs is suitable for a particular type of data. For instance, RNNs are more convenient for time series data while spiking neural networks are good at modeling continuous data. Therefore, in Section III, we will introduce specific types of ANNs and we discuss their properties and use cases.

III. TYPES OF ARTIFICIAL NEURAL NETWORKS

Beyond the simple FNN architecture introduced in this section, we specifically introduce three more advanced types of ANNs: recurrent neural networks, spiking neural networks, and deep neural networks. For each kind of neural network, we will introduce its architecture, advantages and properties, training and learning method, and specific examples of different types of ANNs. We conclude this section with an overview on other types of ANNs that can be also of potential interest for diverse wireless networking applications.

A. Recurrent Neural Networks

1) Architecture of Recurrent Neural Networks: In a traditional ANN, it is assumed that all inputs are independent from each other or all outputs are independent from each other. However, for many tasks, the inputs (outputs) are related. For example, for mobility pattern prediction, the input data that is the users’ locations are actually related. To this end, recurrent neural networks (RNNs) [76], which are ANN architectures that allow neuron connections from a neuron in one layer to neurons in previous layers, have been introduced, as shown in Fig. 5. This seemingly simple change enables the output of a neural network to not only depend on the current input
but also on the historical input, as shown in Fig. 6. This allows RNNs to make use of sequential information and exploit dynamic temporal behaviors such as those faced in mobility prediction, handwriting recognition, or speech recognition. For example, an RNN can be used to recognize an individual person’s speech when they pronounce only one word at each step. Clearly, this task cannot be done in one step without combining different words from different steps. Another application example for RNN is mobility prediction. An RNN can be used to predict the mobility patterns of certain vehicles. These patterns are related to the historical locations that the vehicles have been visited. This task also cannot be done in one step without combing historical locations from previous steps. Therefore, the ANNs whose output depends only on the current input, such as FNNs, cannot perform tasks such as speech recognition. RNNs can also be taken as an ANN that have a “memory”, which allows RNNs to store historical information. Note that, in theory, RNNs can make use of historical information in arbitrarily long sequences, but in practice they are limited to only a subset of historical information [66]. Given their ability to store historical
information, RNNs can more easily perform time-related tasks such as users’ mobility pattern predictions compared to traditional ANNs (e.g., FFNs). However, RNNs can require more time to train since each value of the activation function depends on the series data recorded in RNNs.

In terms of architecture, the key components of a given RNN can be specified as follows:

- **Input signal** $x_t$: this signal represents the input data to a given RNN at time $t$.
- **Input weight matrix** $W_{in}$: this matrix represents the strength of the connections between the neurons in the input layer and the neurons in the hidden layers.
- **Output weight matrix** $W_{out}$: this matrix is used to represent the strength of the connections between the neurons in the output layer and the neurons in the hidden layers.
- **Recurrent weight matrix** $W$: The hidden weight matrix is defined as the recurrent weight matrix, which represents the strength of the connections between the neurons in the hidden layers.
- **Hidden state** $s_t$: this is effectively the hidden state of a neuron in the hidden layer at time $t$. The hidden state represents the value of the activation function at time $t$, which is calculated based on the previous hidden state $s_{t-1}$ and the input at time $t$. $s_t$ can be computed using different methods for different recurrent neural networks. For most commonly used RNNs, we have $s_t = f(Ws_{t-1} + W_{in}x_t)$ where $f(x) = \frac{e^x-e^{-x}}{e^x+e^{-x}}$ or $f(x) = \max(0, x)$. However, in more elaborate types of RNNs, such as long short-term memory algorithm [77] which will be specified in Subsection III-D, each neuron needs to decide what to keep in and what to erase from the hidden state.
- **Output signal**: $y_t$ is the output of a RNN at time $t$, representing the output signal.

Here, we can see that the basic architecture of RNNs is similar to that of FNNs except for the generation of the input, output, and recurrent weight matrices. Moreover, the hidden state in RNNs depends on both current and historical inputs, which enables RNNs to store the historical information. However, when the architecture of an ANN changes from FNNs to RNNs, traditional training method such as blackpropagation used for FNNs may not be available for RNNs. Hence, next, we introduce training methods suitable for RNNs.

2) **Training in Recurrent Neural Networks**: In the architecture of an RNN, the connections between units will form a directed cycle and, hence, the feedforward gradient descent algorithms such as backpropagation cannot be directly used. This is due to the fact that the error backpropagation pertaining to a backpropagation algorithm requires no cycles in the connections
between the neurons of the ANNs. In consequence, the backpropagation through time algorithm (BPTT) \[78\] is more commonly used for training of RNNs. The BPTT approach unfolds the recurrent network in time, by stacking identical copies of it, and redirecting connections within the network to obtain connections between subsequent copies, as shown in Fig. 6. In consequence, the BPTT algorithm actually transforms an RNN to a FNN, which is amenable for training by a backpropagation algorithm. Since BPTT is similar in operation to the backpropagation algorithm, it is significantly faster for training RNNs than general-purpose optimization algorithms such as evolutionary optimization which some works such as in \[79\] have used. However, due to the cycle connections in RNNs, BPTT may get more often trapped in numerous sub-optimal local minimum compared to a backpropagation algorithm for training FNNs. Moreover, like backpropagation, the gradient in BPTT is also computed based on the complete training set, which may become time consuming if the size of training set is very large.

To overcome these drawbacks in BPTT training, real-time recurrent learning (RTRL) \[80\] can be used to compute the exact error gradient at every time step, which is suitable for online learning tasks. In contrast to the BPTT that unfolds RNNs in time, RTRL propagates error forward in time. From (3), we can see that the gradient value with respect to \(w\) at time \(t\) is \(\frac{\partial E(t)}{\partial w}\). In RTRL, the update of weight \(w\) depends not only on the gradient value at time \(t\) but also on the gradient value at the previous time instant, i.e., \(w_{t+1} = w_t - \gamma \sum_{k=0}^{t} \frac{\partial E(k)}{\partial w}\). In RTRL, the gradient of errors propagates forward in time rather than backward in time as BPTT algorithm and, hence, there is no need to unfold RNNs as the BPTT algorithm. However, the time complexity of RTRL is \(O(N_w^4)\) where \(N_w\) is the number of neurons in the considered RNN. In contrast, BPTT has a time complexity of \(O(N_w^2G)\) where \(G\) is length of the input data.

Training the weights in RNNs can also be modeled as a non-linear global optimization problem. A target function can be formed to evaluate the error of a particular weight vector. In particular, the sum-squared-difference between the predictions and the target values specified in the training sequence is used to represent the error of the current weight vector. Arbitrary global optimization techniques may then be used to minimize the sum-squared-difference. The most common global optimization method for training RNNs is via the use of genetic algorithms \[81\], especially in unstructured networks. Other global optimization algorithms may be used to seek a good set of weights, such as simulated annealing \[82\] or particle swarm optimization \[83\]. Such global optimization algorithms can maximize all weight values at the same time and, hence,
can avoid to reach sub-optimal local minimum. However, the complexity of these algorithms for training RNNs can be very high, particularly, when an RNN has a large number of neurons. For instance, when an RNN has 1000 neurons, then a typical global optimization algorithm needs to optimize $1000 \times 1000$ weight values at the same time, which is clearly more complex than in the case of BPTT.

In summary, training RNNs can be done using a variety of techniques. Each technique has its own advantages and disadvantages in terms of the particular learning tasks (i.e., supervised learning tasks), training data size, training time, and data storage space. Depending on the scenario and target performance needed, one can select the most suitable RNN training algorithm.

3) **Example RNN – Echo State Networks**: Next, to shed more light on how RNNs operate, we introduce an RNN that is conceptually simple and easy to implement, called *echo state networks (ESNs)* [84]. Since their inception, ESNs proved to be a highly practical type of RNNs due to their effective approach for training the neural network [85]. In fact, ESNs reinvigorated interest in RNNs, by making them accessible to wider audiences due to their apparent simplicity. In ESN, the input weight matrix and hidden weight matrix are randomly generated without any specific training. Therefore, ESN needs only to train the output weight matrix. Moreover, ESNs belong to a class of algorithms in the framework of *reservoir computing* [66]. Typically, an ANN algorithm is considered part of the framework of reservoir computing if its input signals are mapped to a higher dimensional dynamics of the randomly generated hidden layers, known as a *reservoir*, and the dynamics of the reservoir are mapped to the desired output signals by using a simple training method such as backpropagation. The main benefit of reservoir computing is that the neural network training is performed only at the readout stage while the input and hidden weight matrices are fixed. ESNs can, in theory, approximate arbitrary nonlinear dynamical system with arbitrary precision, they have an inherent temporal processing capability, and are therefore a very powerful enhancement of linear blackbox modeling techniques in nonlinear domain. Due to ESN’s properties such as training simplicity and ability to record historical information, it has been widely applied for several supervised learning tasks such as audio, time series, mobility, water flow, and content request predictions [86]–[89], reinforcement learning tasks such as navigating through a maze [90], classification, and regression. In wireless networks, ESNs will admit many natural applications, such as content prediction, resource management, and mobility pattern estimation, as will be clear from Section IV. Next, the specific architecture
and training methods for ESNs are introduced.

- **Architecture of an Echo State Network:** ESNs use an RNN architecture with only one hidden layer. The generation of an ESN can be given as follows:

  - **Generation of a Reservoir:** Generate a large random reservoir that is represented by the tuple \((W_{\text{in}}, W, \alpha)\) where \(\alpha\) is known as the leaking rate which can be seen as the speed of the reservoir update dynamics, discretized in time. As we mentioned previously, the dynamical system in reservoir computing is known as a reservoir and, hence, in ESN, input and hidden weight matrixes are jointly known as the reservoir. Setting the leaking rate \(\alpha\) must match the speed of the dynamics of hidden state \(s_t\) and output \(y_t\). Here, \(W_{\text{in}}\) and \(W\) is generated randomly. In particular, \(W\) is a sparse matrix while \(W_{\text{in}}\) is a dense matrix. The generation of \(W_{\text{in}}\) and \(W\) are determined by the training data and other ESN parameters. If one ESN uses discrete bi-valued distribution, i.e., \((-0.5, 0.5)\), to generate \(W_{\text{in}}\) and \(W\), then the ESN tends to have a slightly less rich signal space (there is a non-zero probability of identical neurons), but might render the analysis of what is happening in the reservoir easier. To enable ESNs can store historical information, the reservoir should satisfy the so-called *echo state property* which means that the hidden state \(s_t\) should be uniquely defined by the fading history of the input \(x_0, x_1, \ldots, x_t\). This is in contrast to traditional RNNs such as FNNs that need to adjust the weight values of the neurons in hidden layers, ESNs only need to guarantee the echo state property. To guarantee the echo state property of an ESN, the spectral radius of \(W\) should be smaller than 1. However, for some training data, the echo state property will hold even when the spectral radius of \(W\) is larger than 1. The scaling of \(W_{\text{in}}\) is another key method to optimize an ESN. In order to have a small number of freely adjustable parameters, all elements in \(W_{\text{in}}\) are scaled using a single scaling value. If the input signal contributes to the task in very different ways, it is advised to optimize their scalings separately. For very linear tasks, \(W_{\text{in}}\) should be small, letting units operate around 0 and, hence, their activation function is virtually linear. For large \(W_{\text{in}}\), the neurons will get easily close to their \(-1\) and 1 values acting in a more nonlinear, binary switching manner.

- **ESN Implementation:** Run ESN using the training input \(x_t\) and collect the corresponding hidden states \(s_t\). Normalization of input data \(x_t\) can keep the input \(x_t\) bounded and avoid the hidden weight matrix being infinity. In general, the input data from the beginning of
the training will be discarded and not used for training $W_{\text{out}}$ since it may introduce an unnatural starting state which is not normally visited once the network has learnt the task. The amount of input data to discard depends on the memory of the ESN.

- **Training Output weight matrix**: Compute the output weight matrix $W_{\text{out}}$ from the reservoir using a training algorithm such as gradient descent or ridge regression (explained next) to minimize the mean square error (MSE) between the targeted output and action output.

- **Generate Output**: Use the trained network on new input data $x$ computing $y_t$ by employing the trained output weights $W_{\text{out}}$.

Given the components of ESNs, we will next introduce the activation value of each neuron. Even though the input and hidden weight matrices are fixed (randomly), all neurons of ESNs will have their own activation values (hidden state). As opposed to the classical RNNs in which the hidden state depends only on the current input, in ESNs, the hidden state will be given by:

$$\tilde{s}_t = f(W[1; s_{t-1}] + W_{\text{in}}x_t),$$  \hspace{1cm} (7)

$$s_t = (1 - \alpha) + \alpha \tilde{s}_t,$$  \hspace{1cm} (8)

where $f(x) = \frac{e^x - e^{-x}}{e^x + e^{-x}}$ and $[; ;]$ represents a vertical vector (or matrix) concatenation. The model is also sometimes used without the leaky integration, which is a special case for $\alpha = 1$ and, hence, $\tilde{s}_t = s_t$. From (7), we can see that the scaling of $W_{\text{in}}$ and $W$ determines the proportion of how much the current state $s_t$ depends on the current input $x_t$ and how much on the previous state $s_{t-1}$. Here, a feedback connection from $y_{t-1}$ to $s_t$ can be applied to the ESNs, which is defined as a weight matrix $W_{\text{fb}}$. Hence, (7) can be rewritten as $\tilde{s}_t = f(W[1; s_{t-1}] + W_{\text{in}}x_t + W_{\text{fb}}y_{t-1})$.

Based on the hidden state $s_t$, the output signal of ESN can be given by:

$$y_t = W_{\text{out}}[1; s_t; x_t].$$  \hspace{1cm} (9)

Here, an additional nonlinearity can be applied to (9), i.e., $y_t = \tanh(W_{\text{out}}[1; s_t; x_t])$.

- **Training in Echo State Networks**: The objective of the training process in ESNs is to minimize the MSE between the targeted output and actual output. When this MSE is minimized, the actual output will be the target output which can be given by $y^D_t = W_{\text{out}}[1; s_t; x_t]$ where $y^D_t$ is the targeted output. Therefore, the training objective is to find an optimal $W_{\text{out}}$ that enables $W_{\text{out}}[1; s_t; x_t]$ is equal to $y^D_t$. In contrast to conventional recurrent neural networks that require gradient-based learning algorithms, such as BPTT mentioned in Subsection III-A2 to adjust all
input, hidden, and output weight matrices, ESNs only need to train the output weight matrix with simple training methods such as ridge regression. The most universal and stable solution to this problem is via the so-called ridge regression approach also known as regression with Tikhonov regularization [91], which can be given by:

$$W_{\text{out}} = y_t^D \left[1; s_t; x_t\right]^T \left(\left[1; s_t; x_t\right] \left[1; s_t; x_t\right]^T + \theta I\right)^{-1},$$

where $I$ is an identity matrix and $\theta$ is a regularization coefficient which should be selected individually for a concrete reservoir based on validation data. When $\theta = 0$, the ridge regression will become a generalization of a regular linear regression.

The simplest way to train $W_{\text{out}}$ is using the well-known least mean squares (LMS) algorithm in [92], which is a stochastic gradient descent algorithm. At every time step $t$, the LMS algorithm changes $W_{\text{out}}$ in the direction of minimizing the instantaneous squared error $\|y_t^D - y_t\|^2$. LMS is a first-order gradient descent method, locally approximating the error surface with a hyperplane. However, this approximation in LMS is not always accurate. In particular, the curvature of the error surface is very different in different directions. To overcome this disadvantage, a learning algorithm named the recursive least squares (RLS), can be used for training ESNs. RLS, is insensitive to the detrimental effects of eigenvalue spread and exhibits a much faster convergence. Demonstrations of RLS for ESNs are presented in [87] and [93]. The backpropagation-decorrelation in [94] and FORCE learning algorithm in [95] are two other powerful methods for online training of single-layer output with feedback connections. Hence, the output weight matrix of each ESN can be optimized using different training methods. One can select the most suitable ESN training algorithm according to the scenario and target performance needed.

In summary, RNNs (including ESNs) are a type of ANNs in which the connections between neurons will form a directed cycle. This small change enables RNNs to store historical input informations. Therefore, RNNs can be used to process the tasks that cannot be done in one step such as mobility pattern prediction and speech recognition. However, since the RNNs can store historical information, the hidden state of each neuron depends both the current and historical inputs, which makes the training of RNNs much more complex and time consuming than traditional ANNs such as FNNs. To reduce the training complexity, a type of RNNs that only has one hidden layer and needs only to train the output weight matrix is developed, known as ESNs. Due to ESN’s properties such as training simplicity, it has been widely used for several
supervised learning tasks such as predictions, as well as reinforcement learning tasks such as robot control, classification, and regression.

B. Spiking Neural Network

Another important type of ANNs is the so-called spiking neural networks (SNNs). In contrast to other ANNs such as FNNs and RNNs that simply use a single value to denote the activations of neurons, SNNs use a more accurate model of biological neural networks to denote the activations of neurons. Meanwhile, the change of the neurons’ activations will directly affect the training methods for SNNs. Next, we will first specifically introduce the architecture of SNNs. Then, the training methods are formulated. Finally, we give an example for SNNs, called liquid state machine.

1) Architecture of a Spiking Neural Network: The architecture of neurons in SNNs is much similar to the neurons in biological neural networks. Therefore, we will first introduce how the neurons operate in a real-world biological neural networks. Then, we formulate the model of the neurons in SNNs.

In biological neural networks, neurons use spikes to communicate with each other. Incoming signals alter the voltage of a neuron and when the voltage reaches above a threshold-value, the neuron sends out an action potential. Such an action potential is a short (1 ms) and sudden increase in voltage that is created in the cell body or soma. Due to the form and nature of this process, as shown in Fig. 7, we refer to it as a spike or a pulse. This spike in the form of a 1-2 ms pulse travels through the axon, which is linked to the dendrite of another neuron via synapses.
The incoming spike influences the receiving neuron’s membrane potential, which causes the neuron to fire a spike. A spike can have either a positive or a negative impact on the receiving neuron, also called postsynaptic neuron. The positive one is called postsynaptic potential (PSP) and the negative one is inhibitory postsynaptic potential. After sending out a spike, the neuron enters a short moment of rest, the refractory period, in which it cannot send out a spike again. For SNNs, the use of such spikes can significantly improve the dynamics of the networks and, hence, it can model central nervous system and study the operation of biological neural circuits. Since the neurons in SNNs are modeled based on the spike, as opposed to other ANNs such as FNNs and RNNs, SNNs have two major advantages over traditional neural networks: fast real-time decoding of signals and high information carriage capacity via adding temporal dimension. Therefore, an SNN can use fewer neurons to accomplish the same task compared to traditional neural networks and it can also be used for real-time computations on continuous streams of data which means that both the inputs and outputs of an SNN are streams of data in continuous time. However, the training of SNNs will be more challenging (and potentially more time consuming) than traditional ANNs due to their complex spiking neural models.

The basic architecture of an SNN consists of the following components:

- **Input layer** that consists of a number of neurons, representing the signal transmitted in the neurons.
- **Output layer** that consists of a number of neurons, representing the desired output signal.
- **Input weight matrix** that represents the strength of the connections between the neurons in the input layer and the neurons in the hidden layer;
- **Neuron weight matrix** that represents the strength of the connections between the neurons.
- **Output weight matrix** that represents the strength of the connections between the neurons in the output layer and the neurons in the hidden layer.
- **An activation value** that represents a biological neuron’s membrane potential.

In the previously discussed ANNs, such as RNNs and FNNs, the weight matrix values are non-negative which means that if two neurons are connected, then the activation value of each neuron will increase. However, in SNNs, the weight matrix values of SNNs can be negative due to the inhibitory neurons, which improves the dynamics of the neuron system. A spiking neural model is used to represent how a neuron processes the spikes. There are many models to describe
the spiking neurons such as Hodgkin-Huxley model in [96], integrate-and-fire model in [97], theta model in [98], Lzhikevich model in [99], and spike response model in [100]. The most common spiking neuron models are based on the integrate-and-fire and leaky-integrate-and-fire (LIF) paradigm due to its low complexity. Hence, we will introduce the LIF model [101] here in detail as it also serves as a building block for understanding other types.

• **LIF Spiking Neuron Model:** For different spiking neuron models, the spike transmission process will be different. We first introduce the spike transmission process In LIF model. Then, the mathematical LIF model is formulated. In the LIF model, the activation value of a neuron normally starts at a resting value, similar to the resting potential in biological neural networks. Then, whenever the neuron receives a spike, the activation value is raised. Once the activation value crosses the firing threshold, the neuron fires, sending a signal to its all associated neurons. After firing, the neuron observes a relative refractory period during which the activation value is reset to a value lower than its resting value, named the reset value [102]. The LIF model is based on, and most easily explained by, principles of electronics as shown in Fig. 8. From Fig. 8, we can see that a spike travels down the axon and is transformed by a low-pass filter, which converts the short pulse into a current $I_t$. The driving current can be split into two components, $I = I_R + I_C$. The first component is the resistive current $I_R$ that passes through the linear resistor $R$. Here, $R$ models the membrane resistance. $I_R$ can be calculated from Ohm’s law as $I_R = \frac{u_R}{R}$ where $u_R$ is the voltage across the resistor, analogous to the membrane potential. The second component $I_C$ charges the capacitor $C$. Here, $C$ models the membrane capacitance. Based on
the definition of the capacity, \( C = \frac{q}{u} \) with \( q \) being the charge, the capacitive current is given by \( I_C = C \frac{du}{dt} \) and, hence, the total current will be given by:

\[
I_t = C \frac{du}{dt} + \frac{u - u_{\text{rest}}}{R},
\]

(11)

where \( u - u_{\text{rest}} = u_R \) with \( u_{\text{rest}} \) being the resting value. In LIF model, the driving current \( I_t \) represents the input signals of each spiking neuron at time \( t \). We introduce the time constant \( \tau = RC \) of the neuron membrane, modeling the voltage leakage and multiply (11) by \( R \), which can be given by:

\[
\tau \frac{du}{dt} = I_t - u.
\]

(12)

Based on (12), we can calculate the activation potential at time \( t \). In particular, given a constant input current \( I_t = I \) and \( u_0 = u_{\text{rest}} \), (12) can be solved by:

\[
u_t = u_{\text{rest}} + RI \left[ 1 - \exp \left( -\frac{t}{\tau} \right) \right],
\]

(13)

where \( u_t \) is the voltage across the capacitor, analogous to the activation value of a spiking neuron.

In the LIF model, the form of an action potential \( u_t \) is not described explicitly. Instead, when \( u_t \) reaches a certain threshold \( \vartheta \) (spiking threshold), it is instantaneously reset to a lower value \( u_{\text{rest}} \) (reseting value) which can be given by:

\[
u_t = u_{\text{rest}}, \quad t_f < t \leq t_f + \Delta_{\text{abs}},
\]

(14)

where \( t_f \) is the time that a spiking neuron triggers a spike, \( u_{t_f} = \vartheta \), and \( \Delta_{\text{abs}} \) is the absolute refractory period. During the absolute refractory period, \( s_t \) might be clamped to \( u_{\text{rest}} \) and the leaky integration process is re-initiated following a delay of \( \Delta_{\text{abs}} \) after the spike. The leaky integration process described by (11) will start again with the initial value \( u_{\text{rest}} \). Whenever the activation value of a neuron is either higher or lower than its resting potential, its activation value slowly increases or decreases towards the resting value. This is where the leaky part in the LIF name comes from. In fact, equations (11)-(14) describe how the activation of a spiking neuron changes once the neuron receives the signals. From (14), we can see that the activation value \( u_t \) of a spiking neuron is time-varying which is different from the previously discussed ANNs such as FNNs in which the activation value is calculated based on (1). In fact, this is the reason that SNNs can model biological neural networks more accurately than traditional RNNs since in biological neural networks, the activation value is also time varying. However, the varied activation value will result in new challenges for the training of SNNs, as discussed next.
2) **Training in Spiking Neural Networks**: As previously mentioned, the activation value of each spiking neuron is time varying and, hence, in SNNs, the connection strength (weight value) between two neurons is also time varying. The case of weight strengthening (weakening) is called potentiation (depression). If the impact of a change lasts up to a few hours, it is called long term while the impact of a change lasts up to a few minutes scale is called short term. Therefore, the classification of an impact of a weight change can be given by: long term potentiation (LTP) or depression (LTD), short term potentiation (STP) or depression (STD). Neurobiological research has also increasingly demonstrated that connection strength adjustment in networks of spiking neurons is sensitive to the presence and precise timing of spikes [103]. The most commonly used algorithm to adjust the connection strength between neurons is the so-called **spike-timing dependent plasticity (STDP)**. In STDP, it is first assumed that the weight change between presynaptic neuron \(i\) and postsynaptic neuron \(j\) is \(\Delta w_{ij}\), which depends on the relative timing between presynaptic spike arrivals and postsynaptic spikes. Let \(t^z_i\) be the presynaptic spike arrival times at synapse, where \(z = 1, 2, 3, \ldots\), is the index of the presynaptic spike. Similarly, let \(t^n_j\) be the postsynaptic spike arrival time at synapse, where \(n = 1, 2, 3, \ldots\), is the index of the postsynaptic spike. For instance, \(t^2_i\) represents the arrival time of the second spike. The weight change \(\Delta w_{ij}\) can be given by:

\[
\Delta w_{ij} = \sum_{f=1}^{N} \sum_{n=1}^{N} Z \left( t^f_i - t^n_j \right),
\]

where

\[
\Delta w_{ij} = \begin{cases} 
F_+ (w_{ij}) \exp \left( -\frac{|x|}{\tau_+} \right), & \text{if } x > 0, \\
-F_- (w_{ij}) \exp \left( -\frac{|x|}{\tau_-} \right), & \text{if } x \leq 0,
\end{cases}
\]

\(t^f_i - t^n_j\) is the temporal difference between the postsynaptic and the presynaptic spikes. \(F_\pm (w_{ij})\) is the dependence of the update on the current weight of the synapse \(w_{ij}\), and \(N\) is the number of spikes considered for updating the weight. \(\tau_{\pm}\) is the time constant (i.e., \(\tau_+ = 10\) ms and \(-\tau_- = -10\) ms). From (15), we can see that if the presynaptic spikes of a neuron occur immediately before the postsynaptic spikes, then the weight value will be larger and the connection strength will be stronger. Based on the STDP rule, one can control the weight values of LTP and LTD. In summary, STDP is the most commonly used training algorithm in SNNs which uses temporal windows for controlling the weight LTP and LTD. Its apparent simplicity has led to some to propose that it is a universal “first rule” for training SNNs. Different shapes of STDP windows
have been used in recent literature such as in [104]–[106]. However, different forms of STDP are often used in a synapse-specific manner which means that one STDP algorithm may be only available for one type of spiking neuron model. Moreover, STDP only focuses on spike timing and, hence, it ignores other factors that contributes to weight adjustment such as firing rate and dendritic depolarization.

To overcome the disadvantages of STDP algorithm, it is possible to develop training algorithms for adjusting weights of the spiking neurons in SNN. In contrast to the STDP algorithm that only focuses on the temporal domain of SNNs, these training algorithms jointly consider both temporal domain and the complexity of SNNs. The best-known learning rules for training SNNs are the class of error-backpropagation rules for supervised learning. SpikeProp [107], a supervised learning algorithm for SNNs and derived from backpropagation, which uses spike response model neurons to transfer the information in spike-times. Theta neuron learning, another supervised learning method creates a continuous and cyclic model via mapping quadratic-integrate-and-fire neurons to theta neuron, a one-dimensional nonlinear neuron model [107]. There also exist training rules for unsupervised learning of SNNs. A number of approaches for unsupervised learning in spiking neural networks have been developed, based mostly on variants of Hebbian learning [103], such as winner-takes-all learning algorithm in [108]. In [109], a training algorithm that performs unsupervised clustering in a simple two-layer network of spike response model neurons is proposed. Due to the dynamic activation values of spiking neurons, the complexity for training SNNs is much higher than that of training traditional ANNs such as FNNs or RNNs. To reduce the training complexity of SNNs and keep the dynamics of spiking neurons, one promising solution is to develop a spiking neuron network that needs to only train the output weight matrix, like ESNs in RNNs. Next, we will specifically introduce this type of SNNs, named liquid state machine.
3) **Example SNN - Liquid State Machine**: A liquid state machine (LSM) is a reservoir computing algorithm that is conceptually simple and easy to implement as part of an SNN. The architecture of an LSM consists of only two components: liquid and readouts, as shown in Fig. 9. Here, the liquid represents a spiking neural network with LIF model neurons and the readout function is FNNs. The term liquid stems from the analogy between the operation of the LSM and dropping a stone into water. The falling stone will generate ripples in the water. The input in LSM, that represents the falling stone, is converted into a spatio-temporal pattern activation that is representative of ripples. For an LSM, the connections between the neurons in the liquid is randomly generated which allows LSM to possess a recurrent nature that turns the time-varying input into a spatio-temporal pattern. In contrast to the general SNNs that needs to adjust weight values of all neurons, the LSM needs to only train the comparatively simple FNN of the readout function. In particular, simple training method for FNNs such as the feedforward propagation algorithm can be used for training SNNs to minimize the errors between the desired output signal and the actual output signal, which enables LSM to be widely applied for practical applications. Due to LSM’s spiking neurons, it can perform machine learning tasks on continuous data like general SNNs but, it is possible to train it using effective and simple algorithms. Next, we specifically introduce the LSM architecture that consists of a liquid model and FNN readout functions.

- **Liquid Model**

  In LSM, the liquid is made up of a large number of spiking LIF model neurons, located in a virtual three-dimensional column. For instance, Maass in [110] used an architecture composed of 3 by 3 by 15 neurons. The liquid has two important functions in the classification of time-series data. First, its fading memory is responsible for collecting and integrating the input signal over time. Each one of the neurons in the liquid keeps its own state, which gives the liquid a strong fading memory. The activity in the network and the actual firing of the neurons, can also last for a while after the signal has ended, which can be viewed as another form of memory. Second, in the liquid of an LSM, different input signals are separated, allowing for the readout to classify them. This separation is hypothesized to happen by increasing the dimensionality of the signal. For example, if the input signal has 20 input channels, this is transformed into 135 \((3 \times 3 \times 15)\) signals and states of the neurons in the liquid. The generation of a liquid for LSM can be done as follows:
• **Neurons**: The liquid consists of both excitatory and inhibitory neurons. Excitatory neurons will increase the membrane potential of their postsynaptic neurons, making them more likely to fire. Inhibitory neurons on the other hand will decrease the membrane potential making the postsynaptic neurons less likely to fire. In particular, one can set 20% randomly selected inhibitory neurons and 80% excitatory neurons.

• **Connections from input to liquid**: for every pair of input signal and liquid neuron there is a chance of being connected, e.g., 30% in [110].

• **Connections between neurons in the liquid**: The connections between the neurons are allocated in a stochastic manner, with a probability of a connection being placed between neurons $a$ and $b$ as given by [110]:

$$P_{ab} = C \exp \left( -\frac{d_{ab}^2}{\lambda^2} \right),$$

where $C$ is a constant that depends on the type of both neurons. For example, for excitatory-excitatory connections, $C = 0.3$, for inhibitory-excitatoty connections 0.2, for excitatory-inhibitory connections $C = 0.4$ and $C = 0.1$ for inhibitory-inhibitory connections. $d_{ab}$ is the distance between neuron $a$ and $b$. $\lambda$ is a parameter that controls the average number of connections as well as the average distance between connected neurons. The stochastic nature of the creation of the connections between neurons makes a liquid highly likely creates a loop and, hence, a recurrent network is created.

• **Connections from liquid to readout functions**: All neurons in a liquid will connect to the readout functions.

**Readout Model**

The readout of an LSM consists of one or more FNNs that use the activity state of the liquid to approximate a specific function. The purpose of the readout is to build the relationship between the dynamics of the spiking neurons and the desired output signals. The inputs of the readout networks are so-called readout-moments. These are snapshots of the liquid activity taken at a regular interval. The readout can include membrane potentials or spikes or both. Whatever measure is used, the readout represents the state of the liquid at some point in time. In fact, the readout function such as FNNs takes the liquid dynamics such as membrane potentials or spikes as input, and considers the desired output signal as its output. Then, the readout function can be trained using traditional training methods used for FNNs, mainly backpropagation. There
are also some variations on backpropagation such as Levenberg-Marquardt [111], which Matlab recommends for speed and performance. Hence, LSM can be trained using different training methods for FNNs. One can select the most suitable LSM training algorithm according to the specific situation. Once the readout function has been trained, the LSM can be used to perform corresponding tasks.

In summary, SNNs constitute a group of ANNs that model the biological neurons more accurately than other types of ANNs. This change enables SNNs to have two advantages over traditional ANNs: fast real-time decoding of signals and high information carriage capacity. SNNs are particularly effective in dealing with continuous data such as spike trains [110]. However, the complexity of general training methods for SNNs can be much higher than traditional ANNs such as FNNs due to the dynamic spiking neurons. To alleviate this training complexity, a special time of SNNs, known as LSM can be used. LSM consists of randomly connected LIF model neurons and the FNNs readout functions, and is one type of reservoir computing algorithms. LSM is particularly popular due to its simple training, since it needs only to train the readout functions. Due to spiking neurons in LSM, it can deal with continuous-time data. Moreover, the readout function that consists of multiple FNNs enables an LSM can perform multiple tasks at the same time.

C. Deep Neural Networks

Thus far, all of the discussed ANNs, including FNN, RNN, and SNN, have assumed a single hidden layer. Such an architecture is typically referred to as a shallow neural network. In contrast, a deep neural network (DNN) is an ANN with multiple hidden layers between the input and
output layers [112], as shown in Fig. 10. Therefore, a DNN models high-level abstractions in data through multiple non-linear transformations and thus learning multiple levels of representation and abstraction [112]. Several types of DNNs exist such as deep convolutional networks, deep RNNs, deep belief networks, deep feedforward networks, deep SNNs, deep Q-learning, deep ESN, deep residual network, and long-short term memory (LSTM). In fact, the reasons that have made the move from conventional, shallow ANNs, towards DNN possible and desirable are summarized as follows:

- **Improved computing capacity:** The advances in hardware development and data processing capabilities are at the core of the recent renewed interest in developing DNNs. In particular, graphics processing units (GPUs), originally designed to render the computer graphics in video games, have been repurposed to execute the data and algorithm crunching required for machine learning at speeds many times faster than traditional processor chips. This in turn has resulted in a faster and more parallelized computation thus decreasing the required processing time.

- **Improved datasets:** The availability of a large amount of data have made the training of DNNs possible.

- **Improved training algorithms and network architectures:** Deep network architectures and training algorithms have evolved over the years. For example, the use of rectified linear units (ReLU) instead of sigmoid or tanh have made training faster. The ReLU activation function is defined as \( f(x) = x \) for \( x > 0 \) and \( f(x) = 0 \) otherwise. It is shown in [113] that the ReLU activation function outperforms the sigmoidal one in DNNs and is six times faster than tanh [114] for reaching the same training error. In fact, one major benefit of the ReLU activation function is the reduced likelihood of the gradient to vanish. Unlike the derivative of the sigmoid function which is always smaller than one, the gradient of the ReLU function is either 0 for input less than 0 or 1 for input greater than 0. Therefore, one can stack as much layers as needed without having the gradient neither vanish nor explode, thus resulting in a faster learning. Another benefit of ReLUs is sparsity which arises when \( x < 0 \). Sigmoids on the other hand are always likely to generate some non-zero value resulting in dense representations. Sparse representations have been shown to be more beneficial than dense representations.
As opposed to a shallow ANNs that have only one hidden layer, a DNN having multiple layers is more beneficial due to the following aspects:

- **Number of neurons:** Generally, a shallow network would need a lot more neurons than a DNN for the same level of performance. In fact, the number of units in a shallow network grows exponentially with the complexity of the task thus requiring a large number of neurons.

- **Task learning:** While shallow networks can be effective to solve small scale problems, they are ineffective when dealing with more complex problems such as image recognition. In fact, the main issue is that shallow networks are very good at memorization, but not so good at generalization. As such, DNNs are more suitable for many real-world tasks which often involve complex problems that are solved by decomposing the function that needs to be learned into a composition of several simpler functions thus making the learning process effective.

It is worth noting that, although DNNs have large capacity to model a high degree of nonlinearity in the input data, a central challenge is that of overfitting. While overfitting can be a challenge in any type of ANN (as shown in Section III), it can be overcome by simple regularization methods [115]. However, in DNNs, it becomes particularly acute due to the presence of a very large number of parameters. To overcome this issue, several approaches, known as regularization, have been designed. These methods modify the learning algorithm so that the test error is reduced at the expense of increased training error. Some of the most commonly used techniques for regularization in DNNs are summarized as follows [115]:

- **Dataset augmentation:** An overfitting model can perform better if the learning algorithm processes more training data. While an existing dataset might be limited, for some machine learning problems there are relatively easy ways of creating synthetic data. Therefore, dataset augmentation refers to the expansion of the available dataset by applying operations which reflect real world variations as close as possible.

- **Early stopping:** Intuitively, as more data is fed to the model, both training and test error go down. After enough passes over training data the model might start overfitting and learning noise in the given training set. In this case, the training error will decrease while the test error gets worse. Therefore, early stopping overcomes overfitting by interrupting the training
process once the performance of the model on a validation set gets worse. A validation set is a set of examples that are not used for neither the training nor the testing process. They are considered to be representative of future test examples. Early stopping is effectively tuning the hyper-parameter "number of epochs/steps".

- **Dropout layer**: At each training iteration, a dropout layer randomly removes some nodes in the network along with all of their incoming and outgoing connections. Dropout can be applied to either the hidden or the input layer. The intuition behind this approach is that the nodes become more insensitive to the weights of the other nodes, and therefore the model becomes more robust. If a hidden unit has to be working well with different combinations of other hidden units, it’s more likely to do work well individually.

- **Weight penalty L1 and L2**: Weight penalty, also known as "weight decay", is a technique that relies on the implicit assumption that a model with small weights is somehow simpler than a network with large weights. Therefore, the penalties try to keep the weights small or non-existent (zero) unless there are large gradients to counteract it, which makes models also more interpretable. The L2 norm penalizes the square value of the weight and tends to drive all the weights to smaller values. On the other hand, the L1 norm penalizes the absolute value of the weight and tends to drive some weights to exactly zero, while allowing some weights to be large.

1) **Training Deep Neural Networks**: DNNs are often much harder to train than shallow ANNs due to the instability of their gradient that occurs when training them with a gradient-based learning methods such as those described in Section II. In such conventional methods, each of the ANN’s weights receives an update proportional to the gradient of the error function with respect to the current weight in each iteration of training. In particular, the weights and the activation functions (or more precisely, their derivatives) that the gradient passes through affect the magnitude of gradients. Here, note that the gradient by the backpropagation algorithm is computed by the chain rule. Therefore, multiplying $n$ of the gradients at each layer the gradients at the "front" layers in an $n$-layer DNN can result in an exponential decrease or increase with $n$ for small value gradients in the range (-1, 1) or large value gradients, respectively. This is obviously not a major problem in conventional shallow ANNs, as they have only one single layer. For example, the tanh derivative is $< 1$ for all inputs except 0 and the sigmoid is always
≤ 0.25 when used as activation functions $f(\cdot)$ in Equation (1). These two problems are known as the vanishing gradient problem and the exploding gradient problem respectively and result in having different layers in DNNs learn at vastly different speeds. For instance, for a vanishing gradient problem, when later layers in the DNN are learning well, early layers often learn almost nothing during training. To overcome this instability, several techniques for training DNNs have been proposed in the literature [116]–[119]. The adaptive learning rate algorithms, multi-level hierarchy, LSTM and Residual networks (ResNets) techniques are summarized as follows:

- **Adaptive learning rate training algorithms**: Flexible optimization algorithms such as Ada-grad, Adam, AdaDelta, RMSProp have been recently proposed for overcoming the vanishing gradient problem in DNNs [116], [120], [121]. These algorithms provide adaptive learning rates and use different rates for different model parameters. For instance, in the RMSprop gradient descent optimization algorithm [116], the learning rate (i.e., gradient) of a particular weight is divided by a running average of the magnitudes of recent gradients for that weight. Here, note that these rates are considered as hyper-parameters and should be tuned on a subset of training data.

- **Multi-level hierarchy**: Jürgen Schmidhuber’s multi-level hierarchy of DNNs [117] is based on pre-training one level at a time through unsupervised learning and then fine-tuning through backpropagation. Here, each level learns a compressed representation of the observations that is fed to the next level.

- **Long short-term memory**: The LSTM architecture is special type of deep RNN that was introduced in 1997 by Hochreiter and Schmidhuber [118]. In the recurrency of the LSTM, the activation function is the identity function with a derivative of 1. Therefore, in LSTM, the backpropagated gradient neither vanishes nor explodes when passing through, but remains constant. Moreover, unlike traditional RNNs in which each hidden node consists of a single activation function, a hidden node in an LSTM structure is a memory cell with three different gates which either pass or block information, i.e., 1 or 0, and thus allowing it to maintain its own cell state. These gates consist of the forget gate, the input gate, and the output gate and a combination of these gates is trained. These gates control the extent to which new content should be memorized, old content should be erased, and current content should be exposed. Normal RNNs take as an input their previous hidden state and the current input, and output
a new hidden state. An LSTM does the same, except that it also takes in its old cell state and outputs its new cell state. Therefore, no matter how 'deep' your network is, or how 'long' the input sequence is, the network can remember those values, as long as those gates are all 1 along the path. Consequently, iterative gradient descent such as backpropagation through time can be used for training LSTMs. In reinforcement learning applications, LSTM can be trained by policy gradient methods, evolution strategies or genetic algorithms. More details on LSTMs will be given in the next subsection.

- **Residual networks:** Residual neural networks have been introduced by Microsoft in 2015 [119]. They yield lower training error (and test error) than their shallower counterparts simply by reintroducing outputs from shallower layers in the network to compensate for the vanishing data. Therefore, as opposed to traditional activation functions that are defined as $y = f(x)$, these functions are given as $y = f(x) + x$ in ResNets. ResNets have been recently applied for image recognition and classification [122].

Alongside the above mentioned techniques for overcoming the instability of the gradient while training DNNs, some useful guidelines for training DNNs are summarized as follows [123], [124]:

- **Training data:** Use a large training dataset to avoid overfitting; one can possibly rely on data augmentation to create new examples if needed.
- **Activation function:** Depending on the task that needs to be learned, one can choose a sigmoid, tanh, ReLu or SoftSign function.
- **Number of hidden units:** There is an inherent tradeoff that must be factored in when choosing the number of hidden units. On the one hand, using too few neurons in the hidden layers results in underfitting which could make it hard for the DNN to detect the signals in a complicated data set. On the other hand, using too many neurons in the hidden layers can either result in overfitting or an increase in the training time that could prevent the training of the DNN.
- **Number of layers:** Selecting the optimal number of layers can be done using an experimental approach in which one would keep on adding layers until the test error stops to improve.
- **Weight initialization:** Weights should be initialized with small random numbers. For instance, when using sigmoid activation functions, if weights are initialized to very large numbers,
then, the sigmoid will saturate resulting into dead neurons. On the other hand, if the weights are very small, then the gradients will also be small. Therefore, it is preferable to choose weights in an intermediate range having an even distribution around a mean value, typically following a uniform distribution.

- **Hyperparameter tuning:** As the number of hyperparameters keeps on increasing, the computation required for grid search increases exponentially. Therefore, given the large number of hyperparameters in DNNs, one should use random search/sampling instead of grid search, for choosing the optimal hyperparameters. A combination of hyperparameters is generally chosen from a uniform distribution within a particular desired range.

- **Stochastic vs. mini-batch learning:** In a stochastic learning approach, the gradients of the weights are tuned after each training sample. This approach introduces noise into the gradients thus making the model less prone to overfitting. Nevertheless, a stochastic learning approach might be relatively slow and inefficient especially given the availability of machines with high computational power capable of performing parallel computations. Therefore, for greater throughput (and faster learning), it is recommended to use mini-batches instead of stochastic learning, as discussed previously in Section II-C. An appropriate batch size should be capable of retaining some noise and simultaneously using the computation power of machines more effectively. Typically, a batch size between 16 and 128 examples is a good choice (exponential of 2). Here, we note that, in online learning, where the model gets the training data as a stream, one has to resort to stochastic learning.

- **Shuffling training examples:** Randomizing the order of training examples (in different epochs, or mini-batches) results in a faster convergence.

- **Number of epochs/training iterations:** In general, training a DNN for multiple epochs will result in a better model. To know the optimal number of required epochs, one could compare the test error with train error; if the gap is decreasing, then keep on training.

Therefore, the above discussion gives a brief overview on DNNs and some guidelines for training them. Next, we elaborate more on LSTM, a special kind of DNN that is capable of storing information for long periods of time and we also overview convolutional neural networks, a popular DNN in many applications.

2) **Example DNN - Long Short Term Memory:**
Fig. 11. Architecture of an LSTM.

<table>
<thead>
<tr>
<th>Input gate</th>
<th>Forget gate</th>
<th>Behavior</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>1</td>
<td>remember the previous value</td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td>add to the previous value</td>
</tr>
<tr>
<td>0</td>
<td>0</td>
<td>erase the value</td>
</tr>
<tr>
<td>1</td>
<td>0</td>
<td>overwrite the value</td>
</tr>
</tbody>
</table>

D. Deep Neural Networks

As mentioned earlier in Subsection III-C1, LSTMs are a special kind of "deep learning" RNNs that are capable of storing information for either long or short periods of time. In particular, the activations of an LSTM network correspond to short-term memory, while the weights correspond to long-term memory. Therefore, if the activations can preserve information over long duration of time, then this makes them long-term short-term memory. An LSTM network contains LSTM units, where each unit has a cell with a state $c_t$ at time $t$. Access to this memory unit, shown in Fig.11, for reading or modifying information is controlled via three gates:

- **Input gate ($i_t$):** controls whether the input to is passed on to the memory cell or ignored.
- **Output gate ($o_t$):** controls whether the current activation vector of the memory cell is passed on to the output layer or not.
- **Forget gate ($f_t$):** controls whether the activation vector of the memory cell is reset to zero or maintained.

Therefore, an LSTM cell makes decisions about what to store, and when to allow reads, writes,
and erasures, via gates that open and close. At each time step $t$, an LSTM receives inputs from two external sources, the current frame $x_t$ and the previous hidden states of all LSTM units in the same layer $h_{t-1}$, at each of the four terminals (the three gates and the input). These inputs get summed up, along with bias factors $b_f$, $b_i$, $b_o$, and $b_c$. The gates are activated by passing their total input through the logistic function. Table I summarizes the various behaviors an LSTM cell can achieve depending on the values of the input and forget gates. Moreover, the update steps of a layer of LSTM units are summarized in the following equations:

$$g_t = f_g(W_f x_t + U_f s_{t-1} + b_f),$$

$$i_t = f_g(W_i x_t + U_i s_{t-1} + b_i),$$

$$o_t = f_g(W_o x_t + U_o s_{t-1} + b_o),$$

$$c_t = g_t \odot c_{t-1} + i_t \odot f_c(W_c x_t + U_c h_{t-1} + b_c),$$

$$s_t = o_t \odot f_h(c_t),$$

where $g_t$, $i_t$, and $o_t$ are the forget, input and output gate vectors at time $t$, respectively. $x_t$ is the input vector, $h_t$ is the hidden/output vector and $c_t$ is the cell state vector (i.e., internal memory) at time $t$. $W_f$ and $U_f$ are the weight and transition matrices of the forget gate, respectively. $f_g$, $\sigma_c$, and $f_h$ are the activation functions and correspond respectively to the sigmoid, tanh and tanh functions. $\odot$ denotes the Hadamard product. Compared to the standard RNN, the LSTM uses additive memory updates and separates the memory $c$ from the hidden state $s$, which interacts with the environment when making predictions.

LSTM is thus suitable for applications involving sequential learning; it can classify, process and predict time series given time lags of unknown size and duration between important events. In what follows, we summarize a number of other variants of LSTM:

- **Bidirectional LSTM**: Unlike conventional LSTMs, bidirectional LSTMs utilize both the previous and future context, by processing the data from two directions with two separate hidden layers. One layer processes the input sequence in the forward direction, while the other processes the input in the reverse direction. This allows the bidirectional LSTM, at each point in the sequence, to have complete and sequential information about all points before and after it. The output of the current time step is then generated by combining both layers’ hidden vector [125].
• **Sequence-to-sequence LSTM (a.k.a. encoder-decoder LSTM):** An encoder LSTM reads the input sequence and transforms it into a rich fixed-length vector representation, which, in turn, is used as the initial hidden state of a decoder LSTM to generate the output sequence from that vector. Therefore, a straightforward application of the LSTM architecture is that it can solve general sequence to sequence problems. Unlike other DNNs that can only be applied to problems whose inputs and targets can be sensibly encoded with vectors of fixed dimensionality, sequence-to-sequence LSTM can solve problems with sequences whose lengths are not known a-priori [126].

• **Peephole LSTM:** This is similar to traditional LSTM, however, with extra connections between the memory cell and the gates, called peepholes. These peepholes allow the gates to depend, not only on the previous hidden state $h_{t-1}$, but also on the previous internal state $c_{t-1}$.

• **Gated Recurrent Unit (GRU):** GRU was introduced by Cho et al. in 2014 [127]. It combines the forget and input gates into a single ”update gate” and merges the cell state and hidden state. Therefore, each GRU cell consists of a reset gate $r$ and an update gate $z$. In particular, the reset gate determines how to combine the new input with the previous memory and the update gate defines how much of the previous memory to keep around. The basic equations that describe its operation are given as follows:

\[
\begin{align*}
    z_t &= \sigma'_g(W_z x_t + U_z h_{t-1} + b_z), \\
    r_t &= \sigma'_g(W_r x_t + U_r h_{t-1} + b_r), \\
    s_t &= z_t \odot h_{t-1} + (1 - z_t) \odot f'_h(W_h x_t + U_h (r_t \odot s_{t-1}) + b_h),
\end{align*}
\]

where $z_t$ and $r_t$ correspond to the update and reset gate vectors respectively. $f'_g$ and $f'_h$ denote, respectively, the sigmoid and hyperbolic tangent functions. Therefore, the basic idea of using a gating mechanism to learn long-term dependencies is similar to that of LSTM, but with the following key differences:

- A GRU has two gates while an LSTM has three gates.
- GRUs do not possess and internal memory ($c_t$) that is different from the exposed hidden state. Moreover, they do not have the output gate that is present in LSTMs.
- The input and forget gates are coupled by an update gate $z$ and the reset gate $r$ is applied directly to the previous hidden state. Thus, the responsibility of the reset gate
in a LSTM is split up into both \( r \) and \( z \).

- One does not have to apply a second nonlinearity when computing the output.

The authors in [128] compare the performance of different variants of LSTM. Results show that existing variants of LSTM do not show any significant improvement in the LSTM performance. Nevertheless, some modifications such as coupling the input and the forget gates or removing peephole connections simplify the LSTM architecture without significantly degrading its performance. Results also show that the forget gate and the output activation function are the most critical components of an LSTM unit.

Next, we give an overview on other types of DNN, in particular, convolutional neural networks and deep spiking neural networks.

### E. Other Types of ANNs

Beyond our previous, detailed discussion of several types of ANNs, in this subsection, we briefly overview other types of ANNs that can potentially be useful for wireless networking problems, such as self-organizing maps, radial basis function neural networks, modular neural networks, and physical neural networks.

- **Self-Organizing Maps:** A self-organizing map (SOM) [129] is a type of ANN that is trained using unsupervised learning to produce a low-dimensional (typically two-dimensional), discretized representation of the input space of the training samples, called a map, and is therefore a method to do dimensionality reduction. SOMs differ from other ANNs as they apply competitive learning as opposed to error-correction learning such as backpropagation. Moreover, SOMs use a neighborhood function to preserve the topological properties of the input space. These unique features of SOMs makes them useful for visualizing low-dimensional views of high-dimensional data and multidimensional scaling. However, SOMs can only focus on one type of data. When using categorical data or several types of data as input, SOMs behave worse than traditional ANNs. In wireless networks, this can be useful for signal processing such as signal recognition, users context processing and users clustering.

- **Radial Basis Function:** A radial basis function (RBF) network [130] is a type of ANN that uses radial basis functions as activation functions for each neuron. Each neuron in the RBF stores an example from the training set as a “prototype”. The output of RBF is a linear combination of radial basis functions of the inputs and neuron parameters. Linearity involved in the functioning
of RBF enables RBF not suffer from local minima. RBF networks have been applied across many applications including function approximation, time series prediction, classification, and system control. However, RBF networks require good coverage of the input space by radial basis functions. RBF centres are determined by the distribution of the input data, but not by the prediction task. As a result, representational resources may be wasted on the input space that are irrelevant to the task. In wireless networks, this can be useful for antenna array signal processing, channel equalization, coding/decoding, and system identification.

- **Modular Neural Networks**: A modular neural network [131] is a type of ANN that comprises of a series of independent neural networks that are moderated by an intermediary. Each independent neural network serves as a module and operates on separate inputs to accomplish some subtasks of the task the network needs to perform. The independent neural networks do not interact with each other. The intermediary is used to take the outputs of each module and process them and, then, produce the output of the network as a whole. One of the major benefits of a modular neural network is the ability to reduce a large, unwieldy neural network to smaller, more manageable components. By dividing a task into several subtasks, the efficiency of performing this task, the training complexity, and the robustness will all be improved. However, in turn, dividing a task into several subtasks may leads to the learning inaccuracy of the modular neural network. Moreover, how to divide a learning task is not easy to determine which will increase the complexity of the modular neural network. Modular Neural Networks have several applications such as pattern profiling problem, classification, prediction and identification.

- **Physical Neural Networks**: Physical neural networks (PNNs) belong to a group of ANN methods that emphasize the reliance on physical hardware as opposed to software alone when simulating a neural network. An electrically adjustable resistance material is used to emulate the activation function of a neural synapse. While the physical hardware emulates the neurons, the software emulates the neural network. Nugent and Molter have shown that universal computing and general-purpose machine learning are possible from operations available through simple memristive circuits. However, one of the most important challenge for PNN is the scaling problem. A PNN is designed for a specific task. When the tasks change, the physical circuit for the PNN must be designed again, which significantly increase the complexity of implement of PNNs. PNN have several applications such as classification and regression.

In summary, different types of ANNs will have different architectures, activation functions,
connection methods, and data storage capacities. Each specific type of ANNs is suitable for dealing with a particular type of data. For example, RNNs are good at dealing with time-related data while SNNs are good at dealing with continuous data. Moreover, each type of ANNs has its own advantages and disadvantages in terms of learning tasks, specific tasks such as time-related tasks or space-related tasks, training data size, training time, and data storage space. Given all of their advantages, ANNs are ripe to be exploited in a diverse spectrum of applications in wireless networking, as discussed next.

IV. APPLICATIONS OF NEURAL NETWORKS IN WIRELESS COMMUNICATIONS

In this section, we first overview the motivation behind developing ANN solutions for wireless communications and networking problems. Then, we introduce the use of ANNs for various types of wireless applications. In particular, we discuss how to use neural networks for unmanned aerial vehicles (UAVs), virtual reality (VR), mobile edge caching and computing, multiple radio access technologies, and internet of things.

A. Artificially Intelligent Wireless Networks using ANNs: An Overview

One of the main advantages of ANNs is their ability to extract, predict, and characterize regular patterns from massive datasets [132]. Since ANNs can capture many kinds of relationships between a variety of time-varying inputs and outputs such as complex non-linear relationships or highly dynamic relationships, they are well suited to tackle problems such as content predictions, pattern recognition, classification, regression, clustering, and fault detection [132]. Owing to these advantages, ANNs have been widely applied within financial services, government, health care, transportation, marketing and sales [133]. More recently, ANNs have started to attract significant attention in the context of wireless communications and networking [2], [28], since the development of smart devices and mobile applications have significantly increased the level at which human users interact with the wireless communication system. Moreover, the developments of computing and caching technology (typically mobile edge caching and computing) makes it possible for base stations (BSs) to store and analyze the human behaviors for wireless communication. In addition, the development of smart cities motivates the use of ANNs to improve the quality of human services and life. Typically, ANNs can be used for
self-driving car and forecasting the weather so as to reduce the interactions between human and devices.

In particular, within wireless communication domains, ANNs admit two major applications. First, they can be used for prediction, inference, and big data analytics purposes. Within this application domain, the artificial intelligence feature that is enabled by ANNs pertains to the ability of the wireless network to learn from the datasets generated by its users, environment, and network devices. For instance, ANNs can be used to analyze and predict the wireless users’ mobility patterns and content requests and, hence, enables the BSs to determine users’ associations and caching content in advance so as to reduce the data traffic load. Here, behavioral patterns pertaining to the users such as their mobility patterns and content requests will significantly affect which content will be cached, at which node in the network and at which time. If one is able to learn and predict the various context information of its users, such as their locations and content requests, then it can intelligently cache the contents that users will request at the mobile edge, in order to reduce the data traffic from the core network. Such predictions lie at the heart of the notion of proactive wireless networking, in which a wireless network can predict its users’ behavior and adapt its operation to such behavior. Beyond mobile edge caching, predictions and inference will be a primary enabler of the emerging IoT and smart cities paradigms. Within an IoT or smart city ecosystem, sensors will generate massive volumes of data that can be used by the wireless network to optimize its resources usage, understand its network operation, monitor failures, or simply deliver smart services, such as intelligent transportation. In this regard, the use of machine learning, in general, and ANNs, in particular, as tools for optimized predictions is imperative. In fact, ANNs will equip the network with the capability to process massive volumes of data and to parse useful information out of this data, as a pre-cursor to delivering smart city services. For example, road traffic data gathered from IoT sensors can be processed using ANN tools to predict road traffic status at various locations in the city. This, in turn, can be used by the wireless network that connects road traffic signals, apparatus, and autonomous/connected vehicle to inform the vehicles of the traffic state and potentially re-route some traffic to respond to the current state of the system. Moreover, ANNs can be used to predict the computation resource requirement that needs to process each task, which enables BSs to allocate computation resource in advance so as to improve the usage of computation resource and the time consuming of tasks processing. Meanwhile, computation resource and mobile edge caching can be jointly
considered. Here, ANNs can be used to predict the tasks’ computation and, then, the system can store the popular computation tasks. Moreover, ANNs can be beneficial for integrating different sensor data from multiple sensors thus facilitating more interesting and complex applications. In particular, ANNs can identify nonintuitive features largely from cross-sensor correlations which can result in a more accurate estimation of a wireless network’s conditions and an efficient allocation of the available resources. Finally, ANNs can be used for spectrum usage predictions which in turn can minimize the occurrence of a congestion event or the under-utilization of the available resources thus enabling a better usage of the radio spectrum. Clearly, by enabling smart data analytics, ANNs will provide wireless networks with a plethora of smart applications and smart approaches to manage their resources.

Second, a key application of ANNs in wireless networks is for enabling *self-organizing network operation* by instilling artificial intelligence at the edge of the network, as well as across its various components (e.g., base stations and end-user devices). Such edge intelligence is a key enabler of self-organizing solutions for resource management, user association, and data offloading. In this context, ANNs can be developed to serve as RL tools [134] that can learn the environment and adopt different solutions as the environment changes, and, hence, enable decision making for the devices and make the network intelligent. For example, in UAV-based wireless networks, each UAV can use ANN-based RL algorithm to adjust its three dimensional location and flying path according to the mobile users’ locations and data rate requirements. Typically, an UAV wants to find an optimal flying path to service the users located in a larger area. An ANN-based RL algorithm can be used to learn the users information such as their locations and data rate, and determine the flying path based on the learned information. Compared to the traditional learning algorithms such as Q-learning that use tables or matrices to record historical data that do not scale well for a network with dense users, ANNs use a non linear function approximation method to find the relationship from the historical information and, hence, ANNs based learning algorithm can be applied for a network with dense users.

Naturally, ANNs can be simultaneously adopted for both prediction and intelligent/self-organizing operation, as the two functions are largely interdependent. For instance, data can help in decision making, while decision making can generate new data. In particular, when considering virtual reality applications over wireless networks, one can use ANN to predict the behavior of users, such as head movement and content requests. These predictions can help an ANN
based RL algorithm to allocate computation resource and spectrum resource to the users hence improving the QoS. Next, we will introduce the specific applications that use ANNs for wireless communications.

B. Wireless Communications and Networking with Unmanned Aerial Vehicles

1) UAVs for Wireless Communication: Providing connectivity from the sky to ground wireless users is an emerging trend in wireless communications [135], as shown in Fig. 12. High and low altitude platforms, drones, and small aircrafts are being considered as candidates for servicing wireless users and, thus, complementing the terrestrial communication infrastructure. Such communication from the sky is expected to be a major component of beyond 5G cellular networks. Compared to terrestrial communications, a wireless system with low-altitude UAVs is faster to deploy, more flexibly reconfigured, and likely to have better communication channels due to the presence of short-range, line-of-sight (LoS) links. In a UAV-based wireless system, the main applications of UAVs can be classified into three categories: aerial BSs, aerial relays, and cellular connected UAVs for information dissemination/data collection. Aerial BSs can be used for coverage extension, mission critical services, and in an event of damaged/overloaded terrestrial BSs. They can also be used as part of a fully-fledged flying wireless cellular network. For instance, UAVs can be used as flying base stations or relays that can dynamically change their locations and, then, service ground users, especially mobile users. Cellular connected UAVs, on
other hand, open up a large number of IoT applications in agriculture, military, mining operations and industrial inspection services. Such applications include real time video streaming, delivery, surveillance and transmitting telematics. Moreover, UAVs can act as relays between a source and a destination in which a LoS link does not exist or for backhaul connectivity of BSs [

However, the utilization of highly mobile and energy-constrained UAVs for wireless communications also introduces many new challenges such as network modeling, communication standards, optimal deployment for UAV, air-to ground channel modeling, energy efficiency, path planing, security and privacy. Compared to the terrestrial BSs that their deployment is static, mostly long-term, and two dimensional, the deployment of UAVs is flexible, short-term, and three-dimensional. Therefore, there is a need to investigate the optimal deployment of UAVs in wireless networks such as fog radio access networks. Second, UAVs can be employed for data collection, delivery, and transmitting telematics, and, hence, there is a need to develop intelligent self-organizing control algorithms to optimize the flying path of UAVs. In addition, due to the air-to-ground channel characteristics (less blockage and high probability for LoS link) and scarce spectrum resource, the unlicensed bands such as mmWave can be used for UAVs. Therefore, how to jointly allocate cellular and unlicensed bands to the users so as to improve the coverage and reduce the transmission delay must be studied.

2) Neural Networks for UAV-Based Wireless Communication: Due to the flying nature of UAVs, they can track the users’ behavior and collect the data such as users’ and vehicles’ data within any distance, at any time or any place, which provides an ideal setting for implementing neural networks techniques. ANNs have three major use cases for UAV-based wireless communication so as to improve the QoS of a wireless network and QoE of the serviced users. First, using ANN-centric RL algorithms, UAVs can be operated in a self-organizing and intelligent manner that can dynamically adjust their locations, flying directions, resource allocation decisions, and path planning. Second, UAVs can leverage ANN algorithms to exploit the collected users’ data such as mobility data to predict the users’ behavior such as mobility patterns and traffic data demand. Based on the behavioral patterns of the users, battery-limited UAVs can determine the optimal locations and design an optimal flying path to service ground users. In addition, using ANNs enables more advanced UAV applications such as environment identification, 360° VR image generation, and content caching. In particular, UAVs can generate 360° images from high altitude and deliver them to ground users. Meanwhile, UAVs can be equipped with the storage
units to store the most popular contents, and, hence, they can directly send the contents that users request to the ground users so as to reduce the data traffic from the core network. To improve the caching efficiency, UAVs can use neural networks to investigate the users’ content request behavior and predict the requested contents. Clearly, within a wireless environment, all of the human behaviors, UAV movement, and the data collected from wireless devices are time related. For instance, certain users will often go to the same office for work at the same time during weekdays. ANNs can effectively deal with time-dependent data and, hence, it is natural to consider the use of ANNs for UAV applications.

However, using ANNs for UAVs faces many challenges. First, for UAVs, the limited battery as well as the limited computation power can significantly constrain the use of ANNs. This stems from the fact that ANNs require a non-negligible amount of time and computation resources to train. For instance, many UAVs are typically powered by a limited-capacity battery. As such, UAVs must consider a trade-off between power consumption for training ANNs and other applications such as servicing users. Moreover, since UAVs can only collect data during a limited-duration time and, hence, they cannot track the objectives such as vehicles or users for a long time. In consequence, they may not able to collect the entire data to train the ANNs, which will lead to the failure of ANNs learning patterns from the collected data. For example, if UAVs only collect the users’ mobility data in a half day, ANNs cannot find any patterns from these data. However, if the UAVs can collect the users mobility data more than one week, it is possible for ANNs to find the patterns from the collected data. In addition, UAVs must collect data in a flying state, even moving state. The UAVs’ movement such as moving speed, moving height, and moving environment will directly affect the accuracy of the collected data and, hence, increases the difficulty for training ANNs.

- Related works: The existing literature has studied a number of problems related to using ANNs for UAVs such as in [136]–[140]. In [136], the authors studied the use of neural networks for trajectory tracking of UAVs. The authors in [137] used a modified grossberg neural network as a self-organizing algorithm for UAVs to avoid obstacle and collision. The work in [138] proposed an a multilayer perceptron based learning algorithm that uses aerial images and aerial geo-referenced images to estimate the positions of UAVs. In [139], a deep neural network based algorithm was proposed for the classification of images collected by drones. In summary, the recent works in [136]–[140] has used ANNs to solve many UAVs problems such as UAV control,
position estimation, image classification. However, most of these existing works such as in [136]–[140] that only focus on the problems related to UAVs do not consider the use of UAVs for wireless communications. For UAV-based wireless communications, ANNs can be used for path planning, deployment, and channel modeling, so as to improve the users’ QoE. Next, we will introduce a specific ANN application for UAV-based wireless communication.

3) Example: An elegant and interesting use of neural networks for UAV-based communication systems is presented in [140] for the study of the proactive deployment of cache-enabled UAVs. The model in [140] considers the downlink of a wireless cloud radio access network (CRAN) servicing a set of mobile users via terrestrial remote radio heads and flying cache-enabled UAVs. The terrestrial remote radio heads (RRHs) transmit over the cellular band and are connected to the cloud’s pool of the baseband units (BBUs) via capacity-constrained fronthaul links. Since each user has its own QoE requirement, the capacity-constrained fronthaul links will directly limit the data rate of the users that request contents from the BBUs. Therefore, cache-enabled UAVs are introduced to service the mobile users with terrestrial RRHs. Each cache-enabled UAV can store a limited number of popular contents that users request. By caching predicted content, the transmission delay from the content server to the UAVs can be significantly reduced as each UAV can directly transmit its stored content to the users.

For the UAV-to-users communication links, the air-to-ground UAV transmissions model that uses millimeter wave (mmWave) frequency spectrum is considered. This is due to the fact that, the UAVs-to-users links have much less blocking effect than terrestrial RRHs-to-users links and, hence, mmWave is suitable. The transmissions from the cloud to the UAVs occurs over wireless fronthaul links using the licensed cellular band. Consequently, the UAVs’ wireless fronthaul links may interfere with the transmission links from the RRHs to the users. In this system, the users can move continuously.

A realistic model for periodic, daily, and pedestrian mobility patterns is considered according to which each user will regularly visit a certain location of interest. The QoE of each user is formally defined as function of each user’s data rate, delay, and device type. The impact of the device type on the QoE is captured by the screen size of each device. The screen size will also affect the QoE perception of the user, especially for video-oriented applications. The goal of [140] is to find an effective deployment of cache-enabled UAVs to satisfy the QoE requirement of each user while minimizing the transmit power of the UAVs. This problem involves predicting,
for each user, the content request distribution and the periodic locations, finding the optimal contents to cache at the UAVs, determining the users’ associations, as well as adjusting the locations and transmit power of the UAVs. ANNs can be used to solve the prediction tasks due to their advantage of dealing with time varying data. For example, for mobility data, certain users will often go to the same office for work at the same time during weekdays. Moreover, ANNs can extract the relationship between the users’ locations and the users’ context information such as gender, occupation, and age. Therefore, ANNs enable UAVs to collect the data of the users’ behaviors, analyze the collected data, and predict the users’ behaviors in an intelligent manner.

Consequently, a prediction algorithm using the framework of ESN with conceptors is developed to find the users’ content request distributions and their mobility patterns. The predictions of the users’ content request distribution and their mobility patterns will then be used to find the user-UAV association, optimal locations of the UAVs and content caching at the UAVs. Since the data of the users’ behaviors such as mobility and content request are time related, an ESN-based approach previously discussed in Subsection III-A3 can quickly learn the mobility pattern and content request distribution without requiring significant training data. Conceptors defined in [141] enable an ESN to perform a large number of mobility and content request patterns predictions. Moreover, new patterns can be added to the reservoir of the ESN without interfering with previously acquired ones. The architecture of the conceptor ESN-based prediction approach is based on the ESN model specified in Subsection III-A3. For content request distribution prediction, the cloud’s BBUs must implement one conceptor ESN algorithm for each user. The input is defined as a user’s context that includes gender, occupation, age, and device type. The output is the prediction of a user’s content request distribution. The generation of the reservoir is done as explained in Subsection III-A3. The conceptor is defined as a matrix used to control the learning of an ESN. For mobility patterns prediction, the input for mobility patterns prediction is defined as the user’s context and current location. The output is the prediction of a user’s location in the next time slots. Ridge regression is used to train the ESNs. The conceptor is also defined as a matrix used to control the learning of an ESN. During the learning stage, the conceptor will record the learned mobility patterns and content request distribution patterns. When the conceptor ESN-based algorithm meets a new input pattern, the conceptor will first determine whether this pattern has been learned. If this new pattern has been learned previously, the conceptor will instruct the ESN to directly ignore this pattern. This, in turn, allows ESN to
Fig. 13. Mobility patterns predictions of conceptor ESN algorithm from [140]. In this figure, the green curve represents the conceptor ESN prediction, the black curve is the real positions, top rectangle \( j \) is the index of the mobility pattern learned by ESN, the legend on the bottom left shows the total reservoir memory used by ESN and the legend on the bottom right shows the normalized root mean square error of each mobility pattern prediction.

Based on the users’ mobility pattern prediction, the BBUs can determine the users association using \( K \)-mean clustering approach. By implementing a \( K \)-mean clustering approach, the users that are close to each other will be grouped into one cluster. In consequence, each UAV services one cluster and the user-UAV association will be determined. Then, based on the UAV association and each user’s content request distribution, the optimal contents to cache at each UAV and the optimal UAVs’ locations can be found. Typically, when the altitude of a UAV is much higher (lower) than the size of its corresponding coverage, the optimal location of the UAV can be found by [140, Theorems 2 and 3]. For more more generic cases, it can be found by the ESN-based RL algorithm given in [142].

In Fig. 13, based on [140], we show how the memory of the conceptor ESN reservoir changes as the number of mobility patterns that were learned varies. Here, the mobility data is from Beijing University of Posts and Telecommunications. In Fig. 13, one mobility pattern represents the users’ trajectory in one day and the colored region is the memory used by the ESN. Fig. 13 shows that the usage of the memory increases as the number of the learned mobility patterns increases. Fig. 13 also shows that the conceptor ESN uses less memory for learning mobility pattern 2 compared to pattern 6. In fact, mobility pattern 2 is similar to mobility pattern 1, and,
hence, the conceptor ESN requires only a small amount of memory to learn pattern 2. However, the conceptor ESN needs to use more memory to learn pattern 6 because the proposed approach can be used to only learn the difference between the learned mobility patterns and the new one rather than to learn the entirety of the new pattern.

Fig. 14 shows how the total transmit power of the UAVs changes as the number of users varies. From Fig. 14, we can observe that the total UAV transmit power resulting from all algorithms increases with the number of users. This is due to the fact that the number of users associated with the RRHs and the capacity of the wireless fronthaul links are limited. Therefore, the UAVs must increase their their transmit power to satisfy the QoE requirement of each user. From Fig. 14, we can also see that the conceptor based ESN approach can reduce the total transmit power of the UAVs of about 16.7% compared to the ESN algorithm used to predict the content request and mobility for a network with 70 users. This is because the conceptor ESN that separates the users’ behavior into multiple patterns and uses the conceptor to learn these patterns, can predict the users’ behavior more accurately compared to the ESN algorithm.

From this example, we illustrated two key uses of ESN: mobility pattern prediction and content request distribution prediction. In particular, ESNs are particularly effective when dealing with time-dependent users behaviors predictions. Based on the information of the predictions, UAVs can move to optimal location and select the most popular content to store at the UAV cache in advance. Simulation results show that the ESNs can predict the users behaviors accurately and the proactive caching and UAVs movement can effectively service the mobile users.
4) **Future Works:** Clearly, ANNs are an important tool for addressing key challenges in UAV based communication networks. Given their capability of dealing with time-independent data, RNNs can optimize the UAVs deployment by predicting the users’ data rate requirements and their mobility patterns. For example, when the UAVs have a knowledge of the users’ locations and their data rate requirements in advance, they can calculate the optimal deployed locations based on the users’ data rate requirements and locations and fly to the optimal deployed locations before the users arrive. Moreover, SNNs can be used for modeling the air-to-ground channel over mmWave. This is because SNNs are good at dealing with continuous data and wireless channel is time-varying and continuous. For ground-to-ground channel over mmWave, one can directly measure the channel parameters according to existing channel model. However, for air-to-ground channel, there is no existing channel model over mmWave, and, hence, we cannot directly measure the channel parameters to build the air-to-ground model over mmWave. However, one can use SNNs to analyze the data collected from UAVs and generate air-to-ground model to fit the collected data. In summary, for UAV-based wireless communications, RNNs can be used for human behaviors prediction such as mobility patterns prediction, SNNs can be used for signal detection and channel modeling, DNNs can be used to classify the collected data and intelligent self-organizing control.

A summary of key problems that can be solved by using ANNs in the context of UAV-based communications is presented in Table II along with the challenges and future works.

C. **Wireless Virtual Reality**

1) **Virtual Reality over Wireless Communication:** Virtual reality (VR) is seen as one of the most promising new technologies that will enable users to experience and interact with virtual and immersive environments through a first-person view. However, if VR devices such as HTC Vive [158] continue to rely on wired connections to a VR control center, such as a computer, for processing the information, then the users will be significantly restricted in the type of actions that they can take and VR applications that they can experience. Moreover, the reliance on wired connectivity can lead to many breaks in presence for the VR users which can degrade the overall user experience. Therefore, there is a need to enable VR technologies being operated via wireless connections, as shown in Fig. 15, to allow better flexibility and more immersive user experience. In particular, when a VR device is operated over a wireless link, the users must send
<table>
<thead>
<tr>
<th>Applications</th>
<th>Existing Works</th>
<th>Challenges</th>
<th>Future works</th>
<th>Solutions</th>
</tr>
</thead>
<tbody>
<tr>
<td>UAV</td>
<td>• UAV control [136], [137].</td>
<td>• Limited power and computation for training ANNs.</td>
<td>• UAV’s path planning</td>
<td>• RNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Position estimation [138].</td>
<td>• Limited flying time for tracking objectives and data collection.</td>
<td>• Control UAVs to charge batteries</td>
<td>• DNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Image classification [139].</td>
<td></td>
<td>• Channel model for UAVs-to-user</td>
<td>• SNN prediction algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Deployment and caching [140].</td>
<td></td>
<td>• Flying time duration prediction</td>
<td>• RNN prediction algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Handover for cellular connected UAVs</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Design multi-hop aerial network</td>
<td></td>
</tr>
<tr>
<td>VR</td>
<td>• Hand gesture recognition [143].</td>
<td>• Errors in collected data.</td>
<td>• VR users' movement</td>
<td>• RNN prediction algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Interactive shape change [144].</td>
<td>• Limited computation resource.</td>
<td>• Manage computation resource and video formats.</td>
<td>• DNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Video conversion [145].</td>
<td>• Limited time for training ANNs.</td>
<td>• VR video coding and decoding.</td>
<td>• CNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Gaze prediction [146].</td>
<td></td>
<td>• Correction of inaccurate VR images.</td>
<td>• CNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Resource allocation [147].</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Computing and Caching</td>
<td>• Architecture for caching [148].</td>
<td>• Data cleaning.</td>
<td>• Analysis of content correlation.</td>
<td>• CNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Cache replacement [149], [150].</td>
<td>• Content classification.</td>
<td>• Content transmission methods.</td>
<td>• RNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Content prediction [151], [140], [89].</td>
<td>• Limited storage of ANNs for recording all types of contents.</td>
<td>• User and tasks clustering.</td>
<td>• CNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Computing demand and time prediction.</td>
<td>• RNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Computing resource allocation.</td>
<td>• RNN based RL algorithm.</td>
</tr>
<tr>
<td>Multi-RAT</td>
<td>• Architecture for caching [148].</td>
<td>• Channel selection.</td>
<td>• Detection of LoS links.</td>
<td>• CNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Cache replacement [149], [150].</td>
<td>• Mobility predictions.</td>
<td>• Antenna tilting.</td>
<td>• DNNs based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Content prediction [151], [140], and [89].</td>
<td>• Channel load estimation.</td>
<td>• Channel estimation.</td>
<td>• SNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>• Load balancing.</td>
<td>• Handover among multi-RAT BSs.</td>
<td>• RNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Mmwave for backhaul.</td>
<td>• DNNs based algorithm.</td>
</tr>
<tr>
<td>IoT</td>
<td>• Model IoT as ANNs [152].</td>
<td>• Limited computation resource.</td>
<td>• Data compression and recovery.</td>
<td>• CNNs based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Failure detection [153], [154].</td>
<td>• Limited energy resource.</td>
<td>• Resource management.</td>
<td>• RNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td>• User activities classification [155].</td>
<td>• Errors in collected data.</td>
<td>• User identification.</td>
<td>• DNNs based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Image detection [156]</td>
<td>• Real-time training for ANNs.</td>
<td>• IoT devices management.</td>
<td>• SNN based algorithm.</td>
</tr>
<tr>
<td></td>
<td>• Tracking accuracy improvement [157].</td>
<td></td>
<td>• Data relationships extraction.</td>
<td>• RNN based RL algorithm.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>• Modeling autonomous M2M communication.</td>
<td>• FNN and SNN based algorithm.</td>
</tr>
</tbody>
</table>
the tracking information that includes the users’ locations and orientations to the BSs and, then, the BSs will use the tracking information to construct $360^\circ$ images and send these images to the users. Therefore, for wireless VR applications, the uplink and downlink transmission must be jointly considered. Moreover, in contrast to the traditional video that consists of $120^\circ$ images, an VR video consists of high-resolution $360^\circ$ vision with three-dimensional surround stereo. This new type of VR video requires much higher data rate than that of traditional mobile video. In addition, as VR images are constructed according to the users’ movement such as head movement or eye movement, the tracking accuracy of the VR system will directly affect the users experience. In summary, the challenges of operating VR devices over wireless networks [36] include tracking accuracy, low delay, high data rate, and effective image compression.

2) Neural Networks for Wireless Virtual Reality: The use of ANNs is a promising solution for a number of challenges related to wireless VR. This is due to the fact that, most of the wireless VR challenges such as low delay and effective image compression are related to VR images. For example, one can reduce the data size of the VR images to reduce the transmission delay. The VR images are constructed based on the users’ movement. Moreover, compared to UAV applications, VR applications depend more on users’ movement, typically head movement. ANNs are good at identifying and predicting users’ movements and, hence, they can be used to improve VR QoS over wireless networks. ANNs have two major applications for wireless VR. First, ANNs can be used to predict the users’ movement such as head movement and eye gaze movement. For example, $360^\circ$ videos provide users with a panoramic view and allow them to freely control their viewing direction. However, a user displays only the visible portion of a
360° video and, hence, transmitting the entire 360° video frame can waste the capacity-limited bandwidth. In VR, all images are constructed based on the users’ locations, orientations, and eye movements. Therefore, using neural networks, one can predict the users’ movement such as locations, orientations, eye and head movements, and, hence, enable the wireless BSs to generate only the portion of the VR image that a user wants to display. Moreover, the predictions of users’ movement can also improve the tracking accuracy of VR sensors. In particular, the BSs will jointly consider the users’ movement predicted by ANNs and the users’ movements collected by VR sensors to determine the users’ movements.

Second, ANNs can be used to develop self-organizing algorithms to dynamically control and manage the wireless VR network thus addressing problems such as dynamic resource management. In particular, ANNs can be used as a reinforcement learning algorithm for the resource allocation and the adjustment of VR image quality and format according to the dynamic characteristics of wireless networks such as users’ data rates. Moreover, an ANN based RL algorithm can be used to solve the problems that cannot be solved by the traditional algorithms such as optimization theory. Typically, an ANN based RL algorithm can be used to solve coupled resource allocation problems such as the joint consideration of user association and spectrum resource allocation. In fact, ANNs can be used for both applications (prediction and self-organizing operation) simultaneously. For example, ANNs can be used to predict the users’ data rate requirements first and, then, it can be used as a RL algorithm to allocate resource for each VR user.

However, using ANNs for VR faces many challenges. First, in wireless VR networks, the data collected from the users may contain some errors that are unknown to the BSs. In consequence, the BSs may use erroneous data to train the ANNs and, hence, the prediction accuracy of the ANN will be significantly affected. Second, due to the large data size of each 360° VR image, the BSs must spend a large amount of computation resource to process VR images. Meanwhile, the training of ANNs will also spend a large amount of computation resource. In consequence, how to effectively allocate the computation resource allocation for processing VR images and training neural networks must be considered. In addition, VR applications require a ultra-low latency while the training of ANNs can be time consuming. Hence, how to effectively train ANNs in a limited time must be considered. In fact, training the ANNs in an offline manner or using ANNs that converge quickly are two promising solutions for handling training in short
time.

- **Related works**: The existing literature has studied a number of problems related to using ANNs for VR such as in [143]–[147], [159]–[161]. In [143] and [159], the authors investigated the use of different ANNs for recognition of hand gestures in VR applications. The authors in [144] developed a ANN based algorithm to allow the users to perform interactive shape changes and view the resultant deformation changes in a virtual environment. In [145], a deep neural network based algorithm is proposed to automatically convert 2D videos and images to a stereoscopic 3D format. The work in [160] proposed a linear regression learning algorithm to predict the users’ head movement in VR applications. In [146], a decision forests learning algorithm is proposed for gaze prediction. The work in [161] developed a neural network based transfer learning algorithm for data correlation aware resource allocation. In summary, the existing literature such as [143]–[147], [159]–[161] has used ANNs to solve the VR problems such as hand gestures recognition, interactive shape changes, video conversion, head movement prediction, and resource allocation. However, with the exception of our work in [147], all of the other works that use ANNs for VR applications are focused on wired VR. Therefore, they do not consider the challenges over wireless VR such as scarce spectrum resources, dynamic wireless channels and data rates. In fact, ANNs can be used for wireless VR to solve the problems such as users movement prediction, computation resource allocation, spectrum management for VR services, and VR image generation. Next, a specific ANNs’ application for VR over wireless network is introduced.

3) **Example**: One key application of using ANNs for wireless VR systems is presented in [147] for the study of resource allocation in cellular networks that support VR users. In this model, BSs act as the VR control centers that collect the tracking information from VR users over the cellular uplink and then send the generated images (based on the tracking information) and accompanying surround stereo audio to the VR users over the downlink. Therefore, this resource allocation problem in wireless VR must jointly consider both the uplink and downlink transmissions. To capture the VR users’ QoS in cellular network, the model in [147] jointly accounts for VR tracking accuracy, processing delay, and transmission delay. The tracking accuracy is defined as the difference between the tracking vector transmitted wirelessly from the VR headset to the SBS and the accurate tracking vector obtained from the users’ force feedback. The tracking vector represents the users’ positions and orientations. The transmission delay consists of the uplink
transmission delay and the downlink transmission delay. The uplink transmission delay represents the time that an SBS uses to receive the tracking information while the downlink transmission delay is the time that an SBS uses to transmit the VR contents. The processing delay is defined as the time that an SBS spends to correct the VR image from the image constructed based on the inaccurate tracking vector to the image constructed according to the accurate tracking vector. In the model of [147], the relationship between the delay and tracking is not necessarily linear nor independent and, thus, multi-attribute utility theory [162] is used to construct the utility function that can assign a unique value to each tracking and delay components of the VR QoS.

The goal of [147] is to develop an effective resource blocks allocation scheme to maximize the users’ utility function that captures the users’ VR QoS. This maximization jointly considers the coupled problems of user association, uplink resource allocation, and downlink resource allocation. Moreover, the VR QoS of each SBS depends not only on its resource allocation scheme but also on the resource allocation decisions of other SBSs. Consequently, the use of centralized optimization for such a complex problem is not possible as it is largely intractable and yields significant overhead. In addition, for VR resource allocation problem, we must jointly consider both uplink and downlink resource allocation, and, thus, the number of actions will be much larger than conventional resource allocation scenarios that consider only the uplink or downlink resource allocation. Therefore, as the number of actions significantly increases, each BS may not be able to collect all information used to calculate the average utility function. To overcome these challenges, an ANN based RL algorithm can be used for self-organizing VR resource allocation. In particular, an ANN based RL algorithm can find the relationship between the users association, resource allocation, and users’ data rates, and, hence, it can directly select the optimal resource allocation scheme after training process. Compared to the traditional RL algorithms such as Q learning that use tables or matrices to recored the one agent’s utility values and actions, an ANN based RL algorithm uses its hidden neurons, typically dynamic reservoir mentioned in Section III, to approximate the relationships between the utility values and actions, which can significantly speed up the training process.

In [147], a ESN based RL algorithm is introduced for VR resource allocation. The ESN-based learning algorithm enables each SBS to predict the value of VR QoS resulting from each resource allocation scheme without having to traverse all resource allocation schemes. The architecture of the ESN-based self-organizing approach is based on the ESN model specified
in Subsection III-A3. To use ESNs as an RL algorithm, each row of the ESN’s output weight matrix is defined as one action. Here, one action represents one type of resource allocation. The input of each ESN is the current strategies of the action selection of all BSs. The generation of reservoir is presented in Subsection III-A3. The output is the estimated utility value. In the learning process, at each time slot, each BS will implement one action according to the current action selection strategy. After the BSs perform their selected actions, they can get the actual utility values. Based on the actual utility values and the utility values estimated by ESNs, each BS can adjust the values of the output weight matrix of an ESN according to (10). As time elapses, the ESN can accurately estimate the utility values for each BS and, hence, it can find the relationship between the resource allocation schemes and the utility values. Based on this relationship, each BS can find the optimal action selection strategy that maximizes the average VR QoS for its users.

In simulation results of [147], Fig. 16, shows how the average delay of each user varies as the number of BSs changes. From Fig. 16, we can see that as the number of BSs increases, the transmission delay for each serviced user increases. This is due to the fact that as the number of SBSs increases, the number of users located in each SBS’s coverage decreases and, hence, the average delay increases. However, as the number of SBSs increases, the increase of the delay becomes slower due to the increase in interference. This stems from the fact that as the number of BSs continues to increase, the spectrum allocated to each user increases and, hence, the delay
of each user will continue to decrease. However, the increase of the interference from the BSs to the users will hinder the decrease of the delay due to the spectrum allocation. Fig. 16 also shows that the ESN-based algorithm achieves up to 19.6% gain in terms of average delay compared to the Q-learning algorithm for the case with 6 SBSs. Fig. 16 also shows that the ESN-based approach allows the wireless VR transmission to meet the VR delay requirement that includes both the transmission and processing delay (typically 20 ms [163]). These gains stem from the adaptive nature of ESNs. Fig. 17 shows the number of iterations needed till convergence for both the ESN-based learning approach and Q-learning. In this figure, we can see that, as time elapses, the total VR QoS utilities for both the ESN-based algorithm and Q-learning increase until convergence to their final values.

Fig. 17 also shows that the ESN-based algorithm needs 19 iterations to reach convergence while Q-learning needs 25 iterations to reach convergence. Hence, the ESN-based learning algorithm achieves 24.2% gain in terms of the number of the iterations needed to reach convergence compared to Q-learning. This is due to the fact that the ESN can store the BSs’ actions and their corresponding total utility values.

From this example, we illustrated the use of ESN as an RL algorithm for self-organizing resource allocation in wireless VR. An ESN-based RL algorithm enables each BS to allocate downlink and uplink spectrum resource in a self-organizing manner that adjusts the resource allocation according to the dynamical environment. Moreover, an ESN-based RL algorithm
can use the approximation method to find the relationship between each BS’s actions and its corresponding utility values, and, hence, an ESN-based RL algorithm can speed up the training process. Simulation results show that an ESN-based RL algorithm enables each BS can achieve the delay requirement of VR transmission.

4) Future Works: In summary, ANNs are a promising tool to address challenges in wireless VR applications. In fact, the above application of ANNs for spectrum resource allocation can be easily extended to other resource allocation such as computation resource, caching resource, video formats, and video quality. In particular, DNNs are a good choice for implementing RL algorithms. This is due to the fact that DNNs have multiple hidden layers that can store more actions and utility values compared to other ANNs. Hence, DNNs can use these stored information to find the relationship between the actions and utility values. Moreover, RNNs can be used to predict and detect the VR users’ movement such as eye movement and head movement and, hence, we can construct the VR images based on the prediction in advance, which can reduce the time spends to construct the VR images. The data of the users’ movements are all time-dependent and, hence, RNNs are a good choice for performing such tasks. Note that, the prediction of users’ movement will directly affect the VR images that send to the users at each time slot and, hence, the learning algorithm must complete the training process in a short time. In consequence, we should use RNNs that are easy to train for the prediction of the users’ movement. Finally, CNNs can be used for VR video compression, recovery, coding, and decoding so as to reduce the data size of each transmitted VR video and improve the QoS for each VR user. This is because CNNs are good at storing large amount of data and exact the patterns from the data. A summary of key problems that can be solved by using ANNs in wireless VR system is presented in Table II along with the challenges and future works.

D. Mobile Edge Caching and Computing

1) Mobile Edge Caching and Computing: Due to rapid increase of devices connected via wireless networks, mobile traffic is expected to reach an increase of 60 percent of total network traffic. The majority of this mobile traffic stems from video and streaming applications [164]. This explosive growth of data traffic will congest current transmission links, which motivates the use of mobile edge caching and computing technologies, as shown in Fig. 18. Caching at the edge of wireless networks enables the network devices (BSs and end-user devices) to store
the most popular contents to reduce the data traffic (content transmission), delay, and bandwidth usage, and improve the energy efficiency and the utilization of users’ context information and social information. Recently, one can jointly consider cache placement and content delivery, which is known as coded caching. Coded cache enables network devices can create multicasting opportunities for a certain content via coded multicast transmissions and, hence, coded caching can significantly improve the bandwidth efficiency [165]. However, designing effective caching strategies for wireless networks faces many challenges that include optimized cache placement, cache update, and content popularity analytics. In addition to caching, the wireless network’s edge devices can be used for effective, low-latency computation, using the emerging paradigm of mobile edge computing [166]. The basic premise of mobile edge computing is to exploit local resources for computational purposes (e.g., for VR image generation or sensor data processing), in order to avoid high-latency transmission to remote cloud servers. Mobile edge computing, which includes related ideas such as fog computing, can reduce the overall computational latency by reducing the reliance on the remote cloud while effectively offloading computation resources across multiple local and remote devices. The key challenge in mobile edge computing is to optimally allocate computational tasks across both edge devices (e.g., fog nodes) and the remote data servers, in a way to optimize latency. Finally, it is worth noting that, recently, there has been also works [167] that jointly combine caching and computing. The cache is used to store
the most popular and basic computation tasks. Based on the caching computation results, the
network should determine the optimal computation resource allocation to globally minimize the
network latency. However, optimizing mobile edge computing faces many challenges such as
computing placement, computing resource allocation, computing tasks assignment, end-to-end
latency minimization, and minimization of the energy consumption for the devices.

2) Neural Networks for Mobile Edge Caching and Computing: The use of ANNs is a promis-
ing solution for the challenges of mobile edge caching and computing. This is due to the fact that
challenges such as cache placement and cache update are all dependent on the users’ behaviors
such as users’ content request. For example, the cache placement depends on the users’ locations
while the cache update depends on the frequency of a user requesting a content. Human behavior
is highly predictable by ANNs and, hence, ANNs are a promising solution for enabling effective
mobile edge caching and computing. In essence, ANNs have three major applications for mobile
edge caching and computing. First, ANNs can be used for prediction and inference purposes.
For example, ANNs can be used to predict the users’ content request distributions, mobility
patterns, and content request frequency. The content request distribution and content request
frequency can be used to determine which contents to store at the users or SBSs. The mobility
patterns can be used to determine the users’ association and, hence, affect the cache replacement
and the stored contents. Furthermore, ANNs can also be used to extract the social informations
from the collected data. In particular, ANNs can learn the users’ interests, activities, and their
interactions. By exploiting the correlation between users’ data, their social interests, and their
common interests, the accuracy of predicting future events such as users’ geographic locations,
next visited cells, and requested contents can be dramatically improved [168]. For example,
ANNs can be used to predict the users’ interests. The users that have the same interests may have
high probability to request the same content. Therefore, the system operator can cluster the users
that have the same interests and store the popular contents they may request. Similarly, ANNs
can be used to predict the computation requirement for each task. Based on this prediction, the
network devices can schedule the computing resource in advance to minimize the global latency.

Second, ANNs can also be used as an effective clustering algorithm to classify the requested
contents and the users based on their activities such as content request and, thus, it can help the
system operator to determine which contents to store at a storage unit and improve the usage of
the caching contents. Typically, one users content request will change as time varies. In general,
the caching content will be updated for a long time (i.e., one day) and, hence, the system operator must determine which contents to cache by reviewing all the collected content requests. ANNs such as CNNs can be used to store the content request information and classify the large amount of content requests for caching content update. In fact, both ANNs’ applications (prediction and clustering application) can be worked in a cooperative manner. For example, ANNs can first be used to predict the users’ content request distributions, and then, it can be used to classify the users that have the similar content request distributions. In addition, ANNs can also be used to classify the computing tasks to assign the same type of computing tasks to the same computers, which will speed up the time consuming of computation. Finally, ANNs can also be used for intelligently scheduling the computing tasks to different computing centers. Typically, ANNs can be used as an RL algorithm to learn each computing center’s state such as remaining computation resource, and then, allocate computing tasks based on the learnt information to reduce the time consuming of computation. However, using ANNs for mobile edge caching and computing faces many challenges. Data cleaning is an essential part of the data analysis process for mobile edge processing. For example, to predict the users’ content requests, the data processing system should be capable of reading and extracting the useful data from huge and disparate data sources. For example, one user’s content request depends on its age, job, and locations. The data processing system should be able to extract the useful data to train the ANNs. In fact, the data cleaning process usually takes more time than the learning process. For example, the number of contents that users may request can be in the order of millions and, hence, the data processing system should select appropriate contents to analyze and predict the users’ content request behaviors. For caching, the most important use of ANNs is to predict the users’ content requests which directly determines the caching update. However, each user can request a large amount of contents such as sport video, carton show, and news. For each content, it may have different formats and different resolutions. Hence, for each user, the total number of the requested contents will be significantly large. However, the memory of an ANN is limited and, hence, each ANN can record only a limited number of requested contents. In consequence, an ANN must be able to select the most important contents for content request prediction so as to help the network operator to determine which contents to store at mobile edge cache.

- **Related works:** The existing literature has studied a number of problems related to using neural networks for caching such as in [89], [140], [148]–[151], [169]. The authors in [148]
proposed a big data-enabled architecture to investigate proactive content caching in 5G wireless networks. In [149] and [150], neural networks are used to determine the cache replacement. The authors in [151] used neural networks for the predictions of the web pages that are most likely to be re-accessed. The authors in [169] developed a data extraction method using the Hadoop platform to predict content popularity. The authors in [140] and [89] developed an ESN-based learning algorithm to predict the users’ mobility patterns and content request distributions. In summary, the existing works such as in [89], [140], [148]–[151], [169] have used ANNs to solve the caching problems such as cache replacement, content popularity prediction, and content request distribution prediction. However, ANNs can also be used to analyze the content correlation in time or in space, the content transmission methods, users’ clustering method based on the content requests. For mobile edge computing, in general, no existing works use ANNs to solve the wireless communication problems related to mobile edge caching. In fact, ANNs can be used to solve the problems related to mobile edge caching. Typically, ANNs can solve the problems such as the prediction of the computing resource demand, computing resource allocation, and the estimation of the computing time for a task. Next, we introduce a specific ANNs’ application for mobile edge caching.

3) Example: One illustrative application for the use of ANNs for mobile edge caching is presented in [89] which studies the problem of proactive caching in CRANs. In this model, users are served by the RRHs which are connected to the cloud pool of the BBUs via capacity-constrained wired fronthaul links. The RRHs and users are all equipped with storage units that can be used to store the most popular contents that users request. RRHs which have the same content request distributions are grouped into a virtual cluster and service the users using zero-forcing dirty paper coding. Content request distribution for a particular user represents the probabilities of the contents that the user may request. Virtual clusters are connected to the content servers via capacity-constrained wired backhaul links. The total transmission rate of backhaul (fronthaul) links is equally allocated to the contents needed to transmit over backhaul (fronthaul) links. Each user has a periodic mobility pattern and will regularly visit a certain location. Since the cache-enabled RRHs and BBUs can store the requested contents, the content transmission links can be given by: a) content server-BBUs-RRH-user, b) cloud cache-BBUs-RRH-user, c) RRH cache-RRH-user, and d) remote RRH cache-remote RRH-BBUs-RRH-user. We used the notion of effective capacity [170] to capture the maximum content transmission rate of a channel as
the transmission QoS is given. The effective capacity is a link-layer channel model that can be used to measure a content transmission over multiple hops. In particular, the effective capacity can be used to measure a content transmission from the BBU s to the RRHs, then from RRHs to the users. The effective capacity of each content transmission depends on the link that uses to transmit the content and the actual link capacity between the user and associated RRHs.

The goal of [89] is to develop an effective framework for content caching and RRH clustering in an effort to reduce the network’s interference and offload the traffic of the backhaul and fronthaul based on the predictions of the users’ content request distributions and periodic mobility patterns. To achieve this goal, a QoS and delay optimization problem is formulated, whose objective is to maximize the long-term sum effective capacity. This optimization problem involves predicting the content request distribution and periodic location for each user, and finding optimal contents to cache at the BBU s and RRHs. To predict the content request distribution and mobility patterns for each user, an ESN based learning algorithm is used. The model of the ESN-based algorithm has been specified in Subsection III-A3. For each user, the BBU s must implement one ESN algorithm for content request distribution prediction and another ESN algorithm for mobility pattern prediction.

For content request distribution prediction, the input of the developed ESN is a user’s context which includes content request time, week, gender, occupation, age, and device type. The output is the predicted content request distribution. The ESN model consists of the input weight matrix, output weight matrix, and the recurrent weight matrix that have been specified in Section III-A3. A linear gradient descent approach is used to train the output weight matrix. For mobility pattern prediction, the input of the developed ESN is current location of each user and the output is the vector of locations that a user is predicted to visit for the next steps. In contrast to the recurrent matrix that is a sparse matrix and generated randomly, as specified in Subsection III-A3, the recurrent matrix of the ESN used for mobility prediction contains only $W$ non-zero elements. This simplified recurrent matrix can speed up the training process of ESNs. An offline manner using ridge regression is used to train the output weight matrix.

Based on the users’ content request distribution and locations, the cloud can estimate the users’ RRH association and, hence, determine each RRH’s content request distribution then cluster the RRHs into several groups. To determine the contents that must be cached at cloud, the cloud needs to update the content request distribution of each user to compute the distribution of the
Fig. 19. Error as the number of iteration varies.

Fig. 20. Sum effective capacity as function of the number of RRHs.

requested content that must be transmitted via fronthaul links based on the associated RRH cache. However, within a period, the BBUs cannot record the updated content request distributions for all of the users as this will lead to a massive amount of data. Therefore, a sublinear approach is proposed to determine the contents stored at cloud cache. Theorem 2 in [89] proves that the ESN-based algorithm will reach an optimal solution to the content caching.

Using the simulation results of [89], in Fig. 19, we show how the error of the ESN-based estimation changes as the number of iterations varies. In Fig. 19, $\lambda$ is the learning rate of the ESN based algorithm. From Fig. 19, we can see that, as the number of iterations increases, the
error of the ESN-based estimation decreases. Fig. 19 also shows that the ESN approach requires less than 50 iterations to estimate the content request distribution for each user. This is due to the fact that ESNs need to only train the output weight matrix, thus, they exhibit a fast training phase, as discussed in Subsection III-A3. Fig. 19 also shows that for all learning rates, an ESN based algorithm yields a very accurate prediction, with an error that does not exceed 0.43% for $\lambda = 0.03$. Fig. 20 shows how the sum of the effective capacities of all users in a period changes with the number of RRHs. From Fig. 20, we can see that, as the number of the RRHs increases, the effective capacities of all algorithms increase as the users become closer to their RRHs. From Fig. 20, we can also see that the ESN approach can yield up to 21.6% and 24.4% of improvements in the effective capacity compared to random caching with clustering and random caching without clustering, respectively, for a network with 512 RRHs. This is due to the fact that the ESN-based algorithm can effectively use the predictions of the ESNs to determine which content to cache.

This example illustrated the use of ESN for content request distribution prediction and mobility patterns prediction. An ESN based learning algorithm can quickly learn the users mobility patterns and content requests by adjusting only the output weight matrix. Moreover, it can enables the BBU{s} to quickly obtain the users locations and content request distribution due to its simple training process. Based on the predictions, the BBU{s} can determine the user association and caching contents in advance. These results show that the ESN-based algorithm can accurately predict the users’ mobility patterns and content request distribution, and, hence, it can provide a useful solution for proactive caching content replacement.

4) Future Works: Clearly, ANNs are an important tool for solving challenges in mobile edge caching and computing applications, especially for content request prediction and computing tasks prediction. In fact, CNNs that are good at storing huge amount of data can be used to investigate the content correlation in time domain and space domain. Based on the content correlation, each BS can store the contents that are the most related to other contents and, hence, the caching efficiency and hit ratio can be improved. Moreover, RNNs can be used as a self-organizing RL algorithm to allocate computing resource. In particular, for computing resource allocation, one computing task can be allocated to several computing center as well as one computing center can handle several computing tasks. Hence, in contrast to the spectrum allocation, one user can only receive the spectrum from one BS, one computing task can be
allocated to several computing centers, which makes the computing resource problem more complex than the resource allocation problem. RNNs that can record series of the computing resource allocation that BSs have been performed, and, hence, RNN based RL algorithm can quickly find the optimal computing resource allocation scheme to minimize the global latency. Meanwhile, RNNs based RL algorithm can also be used to jointly optimize the cache replacement and content delivery. To achieve this purpose, the actions of the RL algorithm must be jointly combined the content delivery methods with the cache update methods. In addition, RNNs can also be used to predict the demand of the computation resource for each user due to their advantages on dealing with time-dependent data. A summary of key problems that can be solved by using ANNs in wireless VR system is presented in Table II along with the challenges and future works.

E. Spectrum Management and Co-existence of Multiple Radio Access Technologies (Multi-RAT)

1) Co-existence of Multiple Radio Access Technologies: To cope with the unprecedented increase in mobile data traffic and realize the envisioned 5G services, significant enhancement of per-user throughput and overall system capacity is required. Such an enhancement can be achieved through advanced PHY/MAC/network technologies and efficient methods of spectrum management. In fact, one of the main advancements in the network design for 5G networks relies on the integration of multiple different radio access technologies (RATs) under one system, as shown in Fig. 21. With multi-RAT integration, a mobile device can potentially transmit data on multiple radio interfaces such as LTE and WiFi, at the same time, thus improving its performance [171]. Moreover, a multi-RAT network allows fast handover between different
RATs and, thus, providing seamless mobility experience for users. Therefore, the integration of different RATs results in an improvement in the overall utilization of the available radio resources and, thus, an increase in the system’s capacity. It also guarantees a consistent service experience for different users irrespective of the served RAT and facilitates the management of the resulting network.

Spectrum management is also regarded as another key component of 5G networks. Unlike early generations of cellular networks that operate on the sub-6 GHz (microwave) licensed band only, 5G networks are expected to transmit over the conventional sub-6 GHz band, unlicensed spectrum and the 60 GHz mmWave frequency band [172]. Here, note that, the classical LTE microwave licensed band is reliable, however, limited and hence is a scarce resource. On the other hand, mmWave has a huge bandwidth if it’s not blocked and can provide multi-gigabit communication services, however, the uncertainty and dynamic channel variations of the mmWave band makes it unreliable. In fact, mmWave will constitute a main part for spectrum management in 5G networks. In particular, mmWave will enable ultra-high speed content download with low latency, will be used as a micro radio access network for cost-effective backhauling of small cell BSs in dense urban scenarios, and will be used for vehicular-to-vehicular and vehicular-to-everything communications thus allowing automated driving by exchanging high definition dynamic map information between cars and roadside units. On the other hand, the unlicensed bands are best effort. Therefore, a multi-mode BS operating over the licensed, unlicensed and mmWave frequency bands can exploit the different characteristics and availability of the frequency bands thus providing robust and reliable communication links for the end users. However, to reap the benefits of multi-mode BSs, spectrum sharing will be crucial. A particular example of spectrum sharing is LTE in unlicensed spectrum (LTE-U) or licensed-assisted access using LTE (LTE-LAA). The goal of LTE-U is to allow the opportunistic transmission of LTE over the unlicensed bands. It will leverage unlicensed spectrum as a complement to licensed spectrum to offload best-effort traffic through the carrier aggregation framework, while critical control signalling, mobility, voice and control data will always be transmitted on the licensed band. Nevertheless, the problems of traffic balancing across the licensed and unlicensed bands [173], coexistence of LTE with WiFi [142], and coexistence of different LTE-U operators need to be addressed.

2) Neural Networks for Spectrum Management and Multi-RAT: ANNs are regarded as an attractive solution approach for tackling various challenges that arise in multi-RAT scenarios. To
leverage the advantage of such multi-RAT networks, ANNs can allow the smart use of different RATs wherein a SBS can learn when to transmit on each type of frequency band based on the network conditions. For instance, ANNs allow multi-mode BSs to steer their traffic flows between the mmWave, the microwave, and the unlicensed band based on the availability of a LoS link, congestion on the licensed band and the availability of the unlicensed band. Moreover, in LTE-WLAN link aggregation (LWA) scenarios, ANNs allow cellular devices to learn when to operate on each band or utilize both links simultaneously. For instance, delay-tolerant data traffic can be offloaded to WLAN in case of congestion on the licensed band.

Moreover, ANNs can provide multi-mode BSs with the ability to learn the appropriate resource management over different RATs or spectrum bands in an offline manner and thus offering an autonomous and self-organizing operation with no explicit communication among different BSs, once deployed. For instance, ANNs can be trained over large datasets which take into account the variations of the traffic load over several days for scenarios where the traffic load of WiFi access points (WAPs) can be characterized through a particular model [174]. Here, note that cellular data traffic networks exhibit statistically fluctuating and periodic demand patterns, especially applications such as file transfer, video streaming, and browsing [175]. ANNs can also accommodate the users’ mobility pattern to predict the availability of a LoS link thus allowing the transmission over the mmWave band. In particular, they can be trained to learn the antenna tilting angle based on the environment changes in order to guarantee a LoS communication link with the users and, thus, enabling an efficient communication over the mmWave spectrum. Moreover, ANNs enable multiple BSs to learn how to form the multi-hop, mmWave links over backhaul infrastructure, while properly allocating resources across those links in an autonomous manner. To accommodate with the changes in the traffic model and/or users’ mobility pattern, ANNs can also be combined with online machine learning [176] by properly re-training the weights of the developed learning mechanisms. Multi-mode BSs can thus learn the traffic pattern over time and thus predict the future channel availability status. With proper network design, ANNs allow operators to improve the network performance by reducing the probability of occurrence of a congestion event while ensuring a degree of fairness to other technologies in the network, such as the wireless local area network (WLAN).

A proactive resource management of the radio spectrum for multi-mode BSs can also be achieved using ANNs. In a proactive approach, rather than reactively responding to incoming
demands and serving them when requested, multi-mode BSs can predict traffic patterns and determine future off-peak times on different spectrum bands so that incoming traffic demand can be properly allocated over a given time window. In an LTE-U system, for instance, a proactive coexistence mechanism enables future delay-intolerant data demands to be served within a given prediction window ahead of their actual arrival time thus avoiding the underutilization of the unlicensed spectrum during off-peak hours. This in turn allows an increase in the LTE-U transmission opportunity as well as a decrease in the collision probability with WAPs and other BSs in the network.

- **Related works:** In this context, several exiting work have adopted different learning techniques in order to tackle various challenges that arise in multi-RAT networks [177]–[183]. The authors in [177] propose a fuzzy-neural system for resource management among different access networks. In [178], the authors propose a cross-system learning framework, based on Boltzmann-Gibbs learning, in order to optimize the long-term performance of multi-mode BSs, by steering delay-tolerant traffic towards WiFi. A reinforcement learning based approach was proposed in [179] to deal with the problem of RAT selection. The authors in [180] propose a supervised neural network approach, based on a multi layer perceptron (MLP), for the classification of the users’ transmission technology in a multi-RAT system. In [181], the authors propose a hopfield neural network scheme for multi-radio packet scheduling. The problem of resource allocation with uplink-downlink decoupling in an LTE-U system has been investigated in [182] in which the authors propose a decentralized scheme based on ESN. The authors in [183] propose a distributed approach based on Q-learning for the problem of channel selection in an LTE-U system. Nevertheless, this prior work [177]–[183] consider a reactive approach in which data requests are first initiated and, then, resources are allocated based on their corresponding delay tolerance value. In particular, existing works do not consider the predictable behavior of the traffic and thus do not account for future off-peak times during which data traffic can be steered among different RATs.

Here, note that, ANNs are suitable for learning the data traffic variations over time and thus predicting the future traffic load. In particular, LSTM cells are capable of storing information for long periods of time and hence learning the long-term dependency within a given sequence. Predictions at a given time step are influenced by the network activations at previous time steps thus making LSTMs an attractive solution for proactively allocating the available resources in
multi-RAT systems. In what follows, we summarize the work in [184], in which we developed a deep reinforcement learning scheme, based on LSTM memory cells, for allocating the resources in an LTE-U network over a fixed time window $T$.

3) Example: An interesting application of DNN, in the context of LTE-U and WiFi coexistence, is presented in [184]. The work in [184] considers a network composed of several LTE-U BSs belonging to different LTE operators, several WAPs and a set of unlicensed channels on which LTE-U BSs and WAPs can operate on. The LTE carrier aggregation feature, using which the BSs can aggregate up to five component carriers belonging to the same or different operating frequency bands, is adopted. We consider a time domain divided into multiple time windows of duration $T$, each of which consisting of multiple time epochs $t$. Our objective is to proactively determine the spectrum allocation vector for each BS at $t = 0$ over $T$ while guaranteeing long-term equal weighted airtime share with WLAN. In particular, each BS learns its channel selection, carrier aggregation, and fractional spectrum access over $T$ while guaranteeing long-term airtime fairness with WLAN and other LTE-U operators. A contention-based protocol is used for channel access over the unlicensed band. The exponential backoff scheme is adopted for WiFi while the BSs adjust their CW size (and thus the channel access probability) on each of the selected channels based on the network traffic conditions while also guaranteeing a long-term equal weighted fairness with WLAN and other BSs.

The proactive resource allocation scheme in [184] is formulated as a noncooperative game in which the players are the BSs. Each BS must choose which channels to transmit on along with the corresponding channel access probabilities at $t = 0$ for each $t$ of the next time window $T$. This, in turn, allows BSs to determine future off-peak hours of WLAN on each of the unlicensed channels thus transmitting on the less congested channels. Each BS can therefore maximize its total throughput over the set of selected channels over $T$ while guaranteeing long-term equal weighted fairness with WLAN and other BSs. To solve the proposed game, a DNN framework based on LSTM cells was proposed. To allow a sequence to sequence mapping, we consider an encoder-decoder model as described in Section III-D. In this model, the encoder network maps an input sequence to a vector of a fixed dimensionality and then the decoder network decodes the target sequence from the vector. In our scheme, the input of the encoder is essentially a time series representation of the historical traffic load of the BSs and WAPs on all the unlicensed channels. The learned vector representation is then fed into a multi-layer perceptron (MLP) that
Fig. 22. The average throughput gain for LTE-U upon applying a proactive approach (with varying $T$) as compared to a reactive approach [184].

summarizes the input vectors into one vector, thus accounting for the dependency among all the input time series vectors. The output of the MLP is then fed into different separate decoders which allows each BS to reconstruct its predicted action sequence over $T$.

To train the proposed network, the REINFORCE algorithm in [185] is used to compute the gradient of the expected reward with respect to the policy parameters, and the standard gradient descent optimization algorithm [186] described in Subsection II-C is adopted to allow the model to generate optimal action sequences for input history traffic values. In particular, we consider the RMSprop gradient descent optimization algorithm [116], an adaptive learning rate approach, wherein the learning rate of a particular weight is divided by a running average of the magnitudes of recent gradients for that weight.

The proposed proactive resource allocation scheme in [184] is compared with a reactive approach for three different network scenarios. Fig. 22 shows that for very small values of $T$, the proposed scheme does not yield any significant gains. However, as $T$ increases, BSs have additional opportunities for shifting part of the traffic into the future and thus the gains start to become more pronounced. For example, from Fig. 22, we can see that, for 4 BSs and 4 channels, the proposed proactive scheme achieves an increase of 17% and 20% in the average airtime allocation for LTE-U as compared to the reactive approach, respectively. Here, note that the gain of the proposed scheme, with respect to the reactive approach, keeps on increasing until...
it reaches the maximum achievable value, after which it remains almost constant.

4) Future Works: In summary, we have shown that ANNs can improve the total network performance for a multi-RAT system. They allow different technologies to predict the behavior of other nodes in the network and thus allocating the resources based on the future network state. Moreover, ANNs can allow multi-mode BSs to learn on which spectrum band to transmit on based on different network conditions. Here, note that, the above application of ANNs to LTE-U systems can be easily extended to a multi-mode network in which the BSs transmit on the licensed, unlicensed, and mmWave spectrum. In fact, given their capability of dealing with time series data, RNNs can enhance mobility and handover in highly mobile wireless environments by learning the mobility patterns of users thus decreasing the ping-pong effect among different RATs. For instance, a predictive mobility management framework can address critical handover issues, including frequent handover, handover failure, and excessive energy consumption for seamless handover in emerging dense multi-RAT wireless cellular networks. ANNs can also predict the QoS requirements, in terms of delay and rate, for future offered traffic. Moreover, they can predict the transmission links’ conditions and thus schedule users based on the links’ conditions and the QoS requirements. Therefore, given the mobility patterns, transmission links’ conditions and QoS requirements for each user, BSs can learn how to allocate different users on different bands such that the total network performance, in terms of delay and throughput, is optimized.

An interesting application of DNNs for mmWave is antenna tilting. In particular, DNNs are capable of learning several features of the network environment and thus predicting the optimal tilt angle based on the availability of a LoS link and data rate requirements. This in turn improves the users’ throughput thus achieving high data rate. Moreover, LSTMs are capable of learning long time series and thus can allow BSs to predict the link formation for the mmWave backhaul network. In fact, the formation of this backhaul network is highly dependent on the network topology and traffic conditions. Therefore, given the dynamics of the network, LSTMs enable BSs to dynamically update the formation of the links among each others based on the changes in the network.
F. Internet of Things

1) The Internet of Things: In the foreseeable future, it is envisioned, that trillions of machine-type devices such as wearables, sensors, connected vehicles, or mundane objects will be connected to the Internet, forming a massive IoT ecosystem [187], as shown in Fig. 23. Typically, IoT system allows the devices operating in a self-organizing manner and transmit data to other devices with minimal human intervention. Moreover, IoT facilitates the devices connected with each other over wireless links and, hence, they can collect and exchange real-time information to provide intelligent services. In addition, the IoT will allow delivering innovative services and solutions in the realms of smart cities, smart grids, smart homes, and connected vehicles that could provide a significant qualitative improvement in people’s lives. For example, IoT technology can be used to intelligently manage all the city’s systems such as local departments’ information, schools, libraries, transportation, hospitals, water supply, and electricity systems so as to improve the efficiency of services. Meanwhile, retailers, restaurant chains and makers of consumer goods can use data from smartphones, wearable technologies and in-home devices to do targeted marketing and promotions. However, the practical deployment of the IoT system still faces many challenges such as data analytics, computation, transmission capabilities, connectivity end-to-end latency, compatibility, longevity, security, and privacy. In particular, how to connect so many devices with a certain latency requirement will be one of the biggest challenges of the future IoT, and it will completely defy the structure of current centralized communication models and the underlying technologies. One promising approach is to move some of the tasks to the edge, such as using fog computing models where IoT devices take charge of critical operations and cloud servers take on data gathering and analytical responsibilities. Moreover, for each IoT
device, the computation resource and power are limited. Hence, how to allocate computation resource and power for all IoT devices to achieve the data rate and latency requirements is another challenge.

2) Neural Networks for Internet of Things: The use of ANNs is a promising solution for the challenges of IoT. ANNs have four major applications for IoT. First, IoT devices in dynamic environments will change rapidly over time. First, ANNs enable the IoT system can extract important patterns and relationship from the amounts of data sent by the IoT devices. For example, ANNs can be used to discover important correlations among data to improve data recovery. Second, using ANNs based RL algorithms, IoT devices can be operated in a self-organizing and intelligent manner that can make intelligent decision and adapt autonomous control in such environments. For instance, an IoT device using ANN based RL algorithm can dynamically select the most suitable frequency band for communication according to the environment. Third, IoT devices such as sensors may be used for collecting new knowledge about unreachable, dangerous locations (e.g., waste water monitoring) in exploratory applications. ANNs can be used to learn the newly knowledge from the observation data. Finally, one of the main goals of the IoT is to improve the life quality of humans and reduce the interaction between human and IoT devices, and hence, ANNs can be used to predict the users behavior to provide advanced information for IoT devices. For example, ANNs can be used to predict the time that human will come home, and, hence, adjust the control strategy for IoT devices at home. However, using ANNs for IoT faces many challenges. First, in IoT, both the energy and computation resource are limited. Therefore, one should consider the tradeoff between the consumption of training ANNs and the accuracy requirement of the ANNs based learning algorithm. In particular, the higher the required accuracy, the higher the computational requirements, and the higher energy consumptions. Second, within an IoT ecosystem, the collected data may have different structure and even contain several errors. Therefore, when we use data to train ANNs, we should consider how to classify the data and deal with the flaws in the data. Typically, ANNs in IoT must be able to tolerate the flaws in data.

- Related works: The existing literature such as in [152]–[157] has studied a number of problems related to using ANNs for IoT. In [152], the authors use a framework to treat a IoT network holistically as an ANNs to reduce delivery latency. The authors in [153] and [154] used the back-propagation neural network for sensor failure detection in IoT network.
In [155], eight machine learning algorithms such as DNNs and ANNs are tested for human activities classification, robot navigation, body postures and movements. In [157], the authors used Laguerre neural network-based approximate dynamic programming scheme to improve the tracking efficiency in IoT network. The authors in [156] developed a streaming hardware accelerator for convolutional neural network to improve the accuracy of image detection in IoT network. In summary, the existing works such as in [152]–[157] have used ANNs to solve the IoT problems such as IoT network modeling, failure detection, human activities classification, and tracking accuracy improvement. However, ANNs can also be used to analyze the data correlation in time or in space, human activity prediction, and intelligent IoT devices management such as resource allocation and communication methods. Next, we introduce a specific ANNs’ application for IoT.

3) Example: One illustrative application for the use of ANNs for IoT is presented in [152] which studies how to improve the communication quality by mapping IoT networks to ANNs. In [152], the IoT network represents the wireless sensor network. Two objective functions are considered for IoT network. One of the objective functions is to minimize the overall cost of communication between the devices mapped to the neurons in the input layer and the devices mapped to the neurons in the output layers. The overall cost represents the total transmit power of all devices used to transmit the signals. Another one of the objective functions is to minimize the expected transmit time to delivery the signals.

To minimize the total transmit power and expected transmit time for IoT network, the basic idea of [152] is to train an ANN using the objective functions such as the function of the expected transmit time and, then, map the IoT network to the ANN. FNNs mentioned in Section III are used for mapping IoT networks, since FNNs transmit the information in only one direction, forward, from the input nodes, through the hidden nodes, and to the output nodes. First, one must identify devices that want to send signals and the devices that are being prepared to receive signals. The IoT devices that want to send signals are mapped to the neurons in the input layers. The IoT devices that are being prepared to receive the signals are mapped to the neurons in the output layers. The other IoT devices used to forward the signals are mapped to the neurons in the hidden layers. Second, the FNN is trained in an offline manner using the objective functions such as the function of the total transmit power of all devices. In [152], the IoT network devices are mapped into neurons and wireless links into connections between neurons, and, hence, we need
to find a method to map the trained FNN to the IoT network. Since the computation resource of each IoT device is limited, the IoT devices with different computational resources will map to different number of neurons. For example, an IoT device that has more computational resources can map to a larger number of neurons. Moreover, to ensure the integrity of the mapping model, each neuron can only map to one of the IoT devices. Given that there are several ways to map the IoT network to the trained FNN, the optimal mapping is formulated as an integer linear program which is then solved using CPLEX [152]. Finding the optimal mapping represents that the IoT network is finding the optimal connections for IoT devices. Hence, if the IoT network can find the optimal connections for all devices based on the objective functions, the transmit power and expected transmit time can be reduced. Simulation results show that the mapping algorithm can achieve significant gains in terms of total transmit power and expected transmit time compared to the centralized algorithm. This is because the IoT network uses FNNs to approximate the objective functions and find the optimal devices’ connections.

From this example, we illustrated the use of FNN to minimize the total transmit power and expected transmit time for IoT network. Since FNNs transmit information in only one direction from the input layer to hidden layer, then to the output layer, the IoT devices that want to send signals can be mapped to the neurons in input layer. Meanwhile, the devices that are being prepared to receive signals can be mapped to the neurons in output layer. Then, the FNN is trained and an optimal mapping from the FNN to the IoT network is found. Simulation results show that using FNNs to map the IoT networks so as to find the optimal devices’ connections can reduce the total transmit power and expected delivery time for all IoT devices.

4) Future Works: In summary, ANNs are an important tool for solving a variety of challenges in the IoT, particularly in terms of big data analytics and smart operation. In fact, except of using FNNs to map IoT devices, FNNs can also be used for other mapping IoT applications. For example, the input layer can be used to map the IoT devices and the output layer can be used to computing centers. Then, one can find an optimal computing tasks allocation via FNN mapping. Moreover, ANNs can be used for data compression and recovery so as to reduce both the size of the data needs to transmit and end-to-end devices latency. To compress the data, an ANN needs to exact the most important features from the data and these features can be used to present the compressed data. In particular, CNNs can be used for data compression and recovery in space domain while RNNs can be used for data compression and recovery in time domain.
This is because CNNs are good at extracting the patterns and features from large amount data such as one image and RNNs are good at finding the relationships from the time-dependent series data. In addition, DNNs can be used for user recognition, which enables the IoT devices can provide specific services for a particular user. Typically, DNNs that have multiple hidden layers can store more information related to a user compared to other ANNs and, hence, DNNs can use one user’s information such as hairstyle, clothes, and behaviors to identify a user so as to provide particular services for a specific user. A summary of key problems that can be solved by using ANNs in IoT system is presented in Table II along with the challenges and future works.

G. Other Applications

Beyond our previous, detailed discussion of several applications of ANNs, in this subsection, we briefly introduce other applications of ANNs such as physical layer design, vehicular networks, smart cities, and security.

- **Physical Layer Design:** The physical layer is the first layer of the open system interconnection model, which deals with bit-level transmission between different devices and supports electrical or mechanical interfaces connecting to the physical medium for communication. The physical layer is aimed at consolidating the hardware requirements of a wireless network to enable the successful transmission of data. The physical layer provides services such as modulation, bit-by-bit delivery, coding, forward error correction, and transmission mode control. To improve the services of the physical layer, ANNs can be used in many applications [29]. For example, FNNs can be used to model the end-to-end transmission, especially when the channel model of the transmission is unknown. This is due to the fact that the FNN can map the transmitter to the input layers, receiver to the output layers, and the values of channel gain to the hidden layers, and hence, there is no need to know the specific mathematical channel model. Moreover, SNNs that are good at dealing with continuous data can be used for signal detection and correction so as to improve the quality of the signal transmission. SNNs can be used to learn the signal patterns and use the learnt patterns to check the new received signals. Then, the signal errors can be detected and corrected. In addition, DNNs and CNNs can be used for modulation classification due to their capability of storing and extracting information from the data. The existing works such as [29]–[33] have studied a number of problems related to the use of ANNs for physical layer
design such as signal detection in [29] and [33], learning transmitted symbols in [30], channel decoding in [31] and [32]. In summary, ANNs can be used for physical layer design in many applications such as channel decoding and coding, signal detection and correction, modulation control and classification, and carrier sensing and collision detection.

- **Vehicular Networks**: Vehicular networks are migrating from theory to practice, due to the manufacturers’ interest in providing new on-road services such as video and music, to their clients [188]. In vehicular networks, some vehicles can be selected to act as temporary content providers for delivering popular contents to potential nearby vehicles using vehicle-to-vehicle (V2V) communication. Recently, with the development of AI applications, autonomous vehicles is considered as an inevitable trend for future vehicles development. Autonomous vehicles use a variety of techniques to detect their surroundings, such as radar, laser light, GPS, and computer vision, and intelligently identify appropriate navigation paths, as well as obstacles and relevant signage from the collected information. In fact, ANNs have the potential to be highly beneficial when integrated into an autonomous vehicle’s software architecture. For example, ANNs can use the data collected by the autonomous vehicles to learn the behaviors of other vehicles on the road and recognize the navigation paths, obstacles, and signages. Moreover, ANNs can also be used for speech recognition, gesture recognition, eye tracking and driver monitoring, virtual assistance, and natural language interfaces in autonomous vehicles. In addition, ANNs can also be used for vehicle communication condition evaluation so as to select the most suitable vehicles for content transmission. In particular, ANNs must be used to predict the vehicles’ speed, the distance, the power, the contents that has been stored, and the route. Based on these predicted informations, one vehicles can select the optimal vehicles for content transmission. The existing literature [189]–[193] has been studied a number of problems related with the use of ANNs for autonomous vehicles, such as driver behavior modeling [189], mobility speed prediction [190], objective classification [191], vehicles classification [192], and steering autonomous vehicles [193]. However, most of these existing works such as in [189]–[193] that are focus on only the autonomous vehicles do not consider the issues related to wireless communications. In fact, these existing studies such as mobility speed prediction and driver behavior modeling can also be used for V2V communications. For example, one can use the prediction of the mobility speed to infer the time duration of a wireless link that can be used to transmit data.

- **Smart Cities**: A smart city is a complex ecosystem characterized by the intensive use of
information and communications technologies (ICT) such as IoT technology, aiming to make cities more attractive, more sustainable, and unique places for innovation and entrepreneurship [194]. ICT allows city officials to interact directly with the community and the city infrastructure, and to monitor what is happening in the city, how the city is evolving, and how to enable a better quality of life. The various systems in smart cities such as transportation, hospitals, water supply systems will generate tremendous amount of data that can be leveraged for safe, efficient applications and services for city residents. The management of this voluminous data is fundamental to the realization of smart cities. Therefore, ANNs based algorithms may be a promising method for the management and investigation of the voluminous data in smart cities. In particular, DNNs can be used as a self-organizing RL algorithms to control the street lights so as to reduce the energy consumption. Typically, a DNN-based RL algorithm can learn the time of day and the location where traffic activities are low, and, hence, turn off the lights to save energy. Moreover, ANNs such as RNNs can be used to predict the traffic activities such as vehicular traffic and pedestrian traffic. Based on these predictions, a DNN-based RL algorithm can dynamically adjust the signals so as to relieve congestion and clear isolated backups caused by collisions. In addition, we can consider the communication devices as an IoT networks and, hence, the solutions that use ANNs to solve IoT problems also applies to the communication problems in smart cities. The existing works such as [194]–[197] studied several problems related to the use of ANNs for solving smart cities problems such as data management [194], urban traffic flow prediction [195], water demand prediction [196], and parking availability prediction [197].

V. CONCLUSION

In this paper, we have provided one of the first comprehensive tutorials on the use of machine learning, in general, and neural networks, in particular, for enabling a variety of applications in tomorrow’s wireless networks. In particular, we have presented a comprehensive overview on a number of key types of neural networks, that include feed-forward, recurrent, spiking, and deep neural networks. For each type, we have overviewed the basic architecture and training procedure, as well as the associated challenges and opportunities. Then, we have provided a panoramic overview on the variety of wireless communication problems that can be addressed using ANNs. In particular, we have investigated many emerging applications that include unmanned aerial
vehicles, wireless virtual reality, mobile edge caching and computing, Internet of Things, and spectrum management. For each application, we have provided the main motivation for using ANNs along with their associated challenges while also providing a detailed example for a use case scenario. Last, but not least, for each individual application, we have provided a broad overview on future works that can be addressed using ANNs.

REFERENCES


