Resource Allocation Modeling for Fine-Granular Network Slicing in Beyond 5G Systems

Zhaogang SHU††, Tarik TALEB††, and JaeSeung SONG†††

SUMMARY Through the concept of network slicing, a single physical network infrastructure can be split into multiple logically-independent Network Slices (NS), each of which is customized for the needs of its respective individual user or industrial vertical. In the beyond 5G (B5G) system, this customization can be done for many targeted services, including, but not limited to, 5G use cases and beyond 5G. The network slices should be optimized and customized to stitch a suitable environment for targeted industrial services and verticals. This paper proposes a novel Quality of Service (QoS) framework that optimizes and customizes the network slices to ensure the service level agreement (SLA) in terms of end-to-end reliability, delay, and bandwidth communication. The proposed framework makes use of network softwarization technologies, including software-defined networking (SDN) and network function virtualization (NFV), to preserve the SLA and ensure elasticity in managing the NS. This paper also mathematically models the end-to-end network by considering three parts: radio access network (RAN), transport network (TN), and core network (CN). The network is modeled in an abstract manner based on these three parts. Finally, we develop a prototype system to implement these algorithms using the open network operating system (ONOS) as a SDN controller. Simulations are conducted using the Mininet simulator. The results show that our QoS framework and the proposed resource allocation algorithms can effectively schedule network resources for various NS types and provide reliable E2E QoS services to end-users.

key words: QoS, Network Softwarization, Network Slicing, Reliable Communications, SDN, NFV, 5G, and Beyond 5G.

1. Introduction

The ongoing advances in the 5G system have led to thriving mobile communications by targeting broad services and applications. In contrast to previous generations, which were consumer-based technologies, 5G and beyond systems, besides regular customers, target vertical industries, such as industry 4.0, virtual reality (VR), and the Internet of Things (IoT). The 5G system mainly targets three use cases [1]: i) Enhanced mobile broadband (eMBB) that targets regular customers by enabling high-data-rate access to mobile services and multimedia contents; ii) Ultra-reliable low-latency communications (URLLC) that target mission-critical connectivity verticals including, but not limited to, autonomous driving and industry 4.0, which have rigorous requirements on reliability, latency, and availability; and iii) Massive machine-type communications (mMTC) carrying highly dense and immensely connected devices deployed in IoT scenarios. Currently, the 5G system is being deployed in many countries, and it is time for both academia and industry to shift their attention to its successor, namely the sixth generation (6G) system. The latter should satisfy industrial verticals targeted by information and communications technology (ICT) in 2030.

Tremendous efforts have been carried out towards the softwarization of mobile networks in 5G. This has led to the popular concept of network slicing. Some level of success has been achieved in adopting IT and software engineering principles in the design of 5G network slices. The outcome is the 5G Service Based Architecture, consisting of network functions that are virtualized and instantiable on cloud/virtualization platforms. Despite the achieved progress, the current design of the 5G network architecture is yet far from being truly cloud-native. The network slices considered in the context of 5G are still static, following three types of blueprints, each defined for a specific 5G use case (i.e., eMBB, URLLC, or mMTC). They do not reflect, nor do they meet the frequently changing requirements of provisioned services and the dynamicity of the respective end users. Furthermore, network slices, in the context of 5G, are limited to the core network; i.e., the radio access network is simply shared among multiple network slices. An important expectation at beyond 5G (B5G) is to achieve the vision of dynamic and fine-granular network slicing. To attain true customization of network slices with fine granularity, it becomes vital to perform resource allocation according to the dynamics of the underlying infrastructure, the dynamics of the provisioned services, and the spatio-temporal behavior.
of the end-users. This underpins the focus of this paper.

While we are in the early stage to determine all the targeted 5G use cases, some initiatives have started defining the next use cases. According to [2], the ICT 2030 is expected to support use cases related to virtual and augmented reality communications, holographic-type communications, digital twin, ubiquitous intelligence, tactile Internet, and multi-sense experiences. These use cases can be categorized into four main classes: eMBB Plus (eMBBBPlus), Big Communications (BigCom), Holographic and Tactile Communications (HTC), and Secure URLLC (SURLLC) [3]. The eMBBBPlus aims to replace eMBB by providing a high quality of experience (QoE) and considering other services, such as accurate indoor positioning. BigCom aims to provide excellent coverage by allowing customers to communicate everywhere with high bandwidth. Meanwhile, HTC mainly targets 3D holographic communications and the Tactile Internet to allow real-time transfer of senses. Last but not least, SURLLC, besides the reliability and latency, targets security. On the other hand, six different use cases have been considered in [4]. Besides the three use cases targeted by the 5G system, three new use-cases have been considered: i) ubiquitous MBB (uMBB), similar to BigCom, targets the ubiquitous global connectivity of eMBB; ii) Ultra-reliable low-latency broadband communication (ULBC) targets verticals that require high throughput in addition to reliability and reduced latency, such as collaborative virtual reality and holographic communications; and iii) massive ultra-reliable low-latency communication (mULC) that simultaneously considers the features of mMTC and URLLC for the benefits of industrial verticals with dense sensors and actuators.

The aforementioned use cases have challenging key performance indicators (KPIs) in terms of network bandwidth, end-to-end (E2E) latency, and network reliability and availability. These challenges call for a beyond 5G architecture with high customization and elasticity to adapt to various scenarios. Network softwarization technologies, including network function virtualization (NFV), software-defined networking (SDN), and network slicing (NS) [5], will play a crucial role in the planned architecture due to their advantages for overcoming the everyday challenges. Regarding the latter, fine granular network slicing is the cornerstone. It has been generally agreed that applications’ diverse requirements can no longer be satisfied by the “one-size-fits-all” model of conventional mobile networks. NS’s fundamental idea is to partition the common physical network infrastructure into multiple logically independent or virtual networks for different verticals, such as industry 4.0, autonomous driving, and smart city applications. From this perspective, NS is similar in spirit to the concept of Infrastructure as a Service (IaaS) in cloud computing. Effectively, every virtualized network function (VNF) is created, modified, and destroyed as per the NS owner or users’ changing needs, and independently without disrupting other ongoing network slices [6].

While network slices are created on cloud systems that do not necessarily guarantee high reliability, it is vital for the success and wide adoption of services running on network slices to be reliable. In this vein, it is mandatory to ensure NS reliability by guaranteeing the desired QoS. Network softwarization enablers, including SDN, NFV, and NS, will play an essential role in overcoming the challenges associated with reliability [7]. By separating the network control and data planes, SDN provides a softwarization solution to flexibly change networking rules according to different user requirements and circumstances. Operators program SDN controllers to accordingly customize the functionalities of packet forwarding elements (e.g., switches and open virtual switches). Complementarily to SDN, NFV provides an abstract mechanism to virtualize the network functions, whereby traditional network devices or function blocks are realized by software and deployed on demand in places of interest (i.e., in the local server or remote cloud). Accordingly, SDN and NFV help in creating virtual networks along with VNFs on top of common physical infrastructure and managing them in a flexible and economical way [8]. In this regard, it is essential to efficiently integrate SDN and NFV to develop a reliable QoS-aware NS management framework. This defines the core objective of this paper.

Recently, quite a few research studies have addressed this topic. Some of them presented conceptual frameworks and discussed their applicability to 5G and beyond services [6, 7, 9, 10]. However, they did not discuss the details of the implementation. Some studies focused on the design of network slicing mechanisms for only a part of the 5G network, such as Radio Access Network or Core Network [11, 12]. These works developed their respective NS frameworks for parts of the 5G network and did not consider the whole network topology of 5G to meet the end-to-end (E2E) user requirements. Many other works concentrated on domain-specific applications, such as medical applications [13], multimedia use cases [14], and the Internet of Things [15]. However, these studies created virtual networks for only specific application scenarios without considering their particular QoS parameters, such as bandwidth, latency (jitter), packet loss rate, and reliability [16]. Admittedly, establishing a QoS-sensitive connection is one of the most critical and challenging tasks for NS in beyond 5G [17]. Although many studies have been conducted on QoS provisioning in traditional networks, it is not easy to apply these solutions directly in virtual network slices due to the heterogeneous nature of the underlying physical infrastructure and the implementation complexity associated with network softwarization.
To design a comprehensive solution for reliable network slicing in beyond 5G, we propose a QoS framework for fine granular NS that makes use of both SDN and NFV. An abridged version of this paper has been published in [18]. In this work, we described the fundamental concept of the envisioned framework and conducted basic experiments to validate the framework. In comparison to the work presented in [18], this study makes the following extensions. First, it derives network requirements of users and accordingly devises a beyond 5G network resource model. Second, three kinds of resource allocation algorithms for NS have been designed and developed. Third, more experiments are conducted to validate the proposed algorithms. The experiment results show the proposed algorithms’ efficiency for achieving their desired objectives in terms of E2E reliability, QoS, and delay.

The remainder of this paper is organized as follows. In Section II, we discuss previous research works related to network resource modeling and allocation algorithms for NS. The envisioned QoS framework is then presented in Section III. The formal model of network resources in beyond 5G and the envisioned resource allocation algorithms are described in details in the same section. Section IV presents the prototype implementation based on the ONOS controller. It also introduces and discusses the performance evaluation conducted in the Mininet environment. Finally, some concluding remarks and future research directions are presented in Section V.

2. Related Work

This section will discuss different research work on networking slicing in 5G and beyond, mainly focusing on relevant network resource allocation algorithms. We organize the solutions into groups, highlighting the advantages and pitfalls of each. We also list the different features suggested by these solutions and compare them to the solution proposed herein.

2.1 QoS frameworks based on SDN

Research on SDN-based QoS provisioning has received considerable attention, as SDN is considered as a promising network softwarization technology. Generally, QoS provisioning is one of the crucial functionalities of the SDN technology. Thus, related works have mainly focused on developing a SDN controller module, considering traffic engineering, multi-path routing algorithms, and load balancing issues [19–23]. For instance, Tomovic et al. suggested a SDN controller framework with QoS provisioning for multimedia applications [19]. In this framework, four key function blocks – resource monitoring, route calculation, call admission control and resource reservation – were integrated into controllers for enabling the QoS functionality. Dutra et al. proposed a solution that enables E2E QoS based on multi-path in SDN [21]. An extension of this work was published in [20]. These solutions allow the operators to allocate network resources through the queue features in OpenFlow so that over-provisioning of bandwidth resources is reduced or eliminated. The primary purpose of these works is to obtain fine-grained and dynamic control of network resources using SDN to prevent the limitations of legacy networks. In fact, the architecture of traditional networks is not suitable to adapt to the changing requirements of users. However, most of these solutions are designed in a centralized controller in a single mode and for small scale networks. It is difficult to apply them to large-scale Beyond 5G (B5G) networks, which are known for their heterogeneity and complexity. Nevertheless, these works provide valuable guidelines for designing the QoS provisioning framework in B5G, such as network resource allocation algorithms and mechanisms for interactions with other network function modules within the SDN controllers.

2.2 QoS frameworks to support specific applications

Each network application has its own QoS requirements. Therefore, many researchers have concentrated on QoS-sensitive network solutions for specific applications, such as cloud data center networks [24], smart grid networks [25], energy networks [26], and Internet of Things [27]. Besides, Chekired et al. designed a framework to enhance the QoS of an autonomous driving application [11]. This framework contained a network and an NS system model to improve transmission efficiency and satisfy the low latency constraint. Bektas et al. provided an E2E solution by allocating RAN network resources, creating NS for mission-critical applications via a novel scheduler [12]. A common feature of all these works is their focus on a single application domain, hence rendering them not applicable to other applications with different requirements.

2.3 QoS support for reliable NS

There have been studies on QoS provisioning for network slices [28–30]. These studies make use of the programmability of both control and data planes. Baldini et al. [31] described how the control plane of SliceNet is involved in meeting the E2E needs of different 5G verticals. Ricart-Sanchz et al. [32] proposed a QoS-aware NS framework based on hardware acceleration to support data plane programmability [32]. Another important issue is resource deployment, and management policy for E2E NS [33,34]. Solutions proposed in [10,35] leveraged edge/fog computing for optimizing the E2E NS. For instance, Lu Ma et al. [36] proposed a framework to manage 5G services by leveraging NFV, SDN and edge computing. The framework was designed to provide distributed and on-demand
Table 1: Comparison between existing solutions and our proposed framework.

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deployment of network functions and create service-guaranteed network slices based on optimal workload allocation. As a promising technology, artificial intelligence (AI) was also explored to manage network resources [37,38].

2.4 Solutions comparison

To recognize the advantages and disadvantages of existing solutions and compare them against our envisioned framework, we use some typical features. These features include programmability, i.e., thanks to the flexibility offered by SDN, dynamic multi-path routing, cooperation of multiple SDN controllers, NFV in cloud, edge computing, NS and AI-based algorithm. Table I depicts a comparison between the proposed solutions and the related works with respect to the aforementioned features. In the table, 'Y' means that a solution supports or offers that feature, while 'N' represents the opposite. It shall be noted that not all solutions in the same group support a specific feature. For example, in the solution group of "QoS Service or framework based on SDN", only the solution presented in [21] has the feature of dynamic multi-path routing. From this comparison, we state that our proposed framework covers most of the typical features except 'AI-based algorithm'. This demonstrates the added value of this work compared to the existing ones.

3. Proposed QoS framework for reliable NS in B5G system

In this section, we first introduce the proposed framework’s design goals, and then describe the overall framework in detail. Then, we present the network resource model and resource allocation algorithms of NS adopted by our framework.

3.1 Framework design goals

After a deep investigation, we have determined that an ideal QoS framework for NS should ensure the features mentioned below. We have designed and developed our framework to achieve these features.

1. Flexibility and reliability:
   Flexibility and reliability are significant for the QoS framework of NS in a B5G system. Softwarization is one of the most important B5G enablers, in which VNFs can be highly customized according to the traffic needs. The NS should be customized to ensure the service level agreement (SLA) in terms of bandwidth, latency, jitter, and packet loss. Moreover, a smart life cycle management (LCM) of NS should be designed to meet the desired SLA. Similarly, the reliability of QoS refers to the stability needed to meet QoS requirements. For some latency-sensitive applications, such as autonomous driving and industrial manufacturing, network stability is crucial for ensuring the functionality of the system and human safety.

2. QoS as a service for multi-tenants:
   The second goal of the proposed framework is to provide QoS as a service (QaaS). Similar to the software as a service (SaaS) concept in cloud computing, QaaS is enabled by SDN and NFV technologies to change the B5G network industry’s business model. In this case, multi-tenants can transparently use the same service provider’s infrastructure while ensuring NS’s isolation and SLA. Also, this concept provides the flexibility to apply the pay-as-you-go (PAYG) model of cloud computing, whereby customers pay only for what
they have effectively used. The tenants can rent QoS from the main provider and the middle-providers, which work as an agent between the main provider and end users.

3. **Integrating both SDN and NFV technologies:**

SDN provides a distributed architecture to control and manage all physical network resources, which is well-aligned with the QoS framework’s scalability and flexibility. However, SDN can not effectively handle the life cycle of NS. Therefore, SDN and NFV must be integrated to manage virtualized resources and the life cycle of NS.

3.2 Framework overview

Fig. 1 shows the overall framework intended to achieve the design goals described above. The B5G network infrastructure is divided into three parts: radio access network (RAN), transport network (TN), and core network (CN). End users (e.g., mobile phones, webcams, smart industrial devices, vehicles, etc.) connect to B5G through the wireless base stations in RAN. In contrast to the previous generations, the B5G system is expected to host services close to the users by leveraging the multi-access edge computing (MEC) capability. In this case, many network functions are implemented as VNFs in MEC, enabling the softwarization of RAN.

Most industrial SDN controllers (e.g. ONOS and OpenDayLight) provide RESTful north-bound application programming interfaces. Different NSs with specific QoS requirements can be created based on these interfaces. According to the scope of network resources needed by NS, we define three types of NS as shown in Fig. 1. A Type 1 NS is created in a single RAN, implicating that the packet forwarding path from the source to the destination must be constrained in that RAN. A Type 2 NS is designed with a packet forwarding path such as RAN-TN-RAN, while a Type 3 NS is created with a packet forwarding path such as RAN-TN-CN-TN-RAN. Therefore, a Type 3 NS covers more network nodes and needs more network resources than a Type 1 NS and a Type 2 NS. When tenants apply for a network slice, they should consider both requirements, besides the cost, to decide which type of NS is the best choice.

3.3 Network resource model and allocation algorithm for reliable NS

In this subsection, we describe the network resource model and mathematically formulate the proposed solutions.

1. **Formulation of network resources in 5G:**

Let \( G(V, E) \) represent a network topology managed by a SDN controller, in which \( V \) is the set of nodes and \( E \) is the set of links in this network. Let \( b(i,j) \) and \( l(i,j) \) denote the available bandwidth and transmission latency between nodes \( i \) and \( j \), respectively; \( i,j \in V \) and \( (i,j) \in E \) in network topology \( G(V, E) \). For each \( G(V, E) \) that contains \( n \) nodes, the following \( n \times n \) matrix represents the available bandwidth between any two nodes.

\[
B_G = \begin{bmatrix}
  b_{(1,1)} & b_{(1,2)} & \cdots & b_{(1,n)} \\
  b_{(2,1)} & b_{(2,2)} & \cdots & b_{(2,n)} \\
  \vdots & \vdots & \ddots & \vdots \\
  b_{(n,1)} & b_{(n,2)} & \cdots & b_{(n,n)}
\end{bmatrix}
\]  

(1)

Since \( b_{(i,j)} \) always equals \( b_{(j,i)} \), \( B_G \) is a symmetric matrix. Furthermore, if two nodes that are not di-
rectly connected, we cannot obtain the bandwidth value exactly, because it is possible that there are multiple paths with different bandwidth values between them. Therefore, we assume that each element $b(i,j)$ of this matrix satisfies the following condition.

$$b(i,j) = \begin{cases} \text{bandwidth} & \text{if } (i \neq j) \text{ & connects directly } \\ 0 & \text{otherwise} \end{cases}$$ \tag{2}$$

Similarly, the following $n \times n$ symmetric matrix represents the transmission latency between any two nodes.

$$L_G = \begin{bmatrix} l_{(1,1)} & l_{(1,2)} & \ldots & l_{(1,n)} \\ l_{(2,1)} & l_{(2,2)} & \ldots & l_{(2,n)} \\ \vdots & \vdots & \ddots & \vdots \\ l_{(n,1)} & l_{(n,2)} & \ldots & l_{(n,n)} \end{bmatrix}$$ \tag{3}$$

where

$$l_{(i,j)} = \begin{cases} \text{latency} & \text{if } (i \neq j) \text{ & connects directly } \\ 0 & \text{otherwise} \end{cases}$$ \tag{4}$$

We assume that the SDN controller can measure and collect all the data of $B_G$ and $L_G$ in its own management domain. However, according to our framework, multiple SDN controllers possibly coordinate to measure the QoS between different network management domains for Type 2 and Type 3 NS, respectively. For example, the forwarding path of a Type 2 NS looks like RAN-TN-RAN, so we should acquire the network status data between RAN and TN. In this vein, the following two $m \times n$ matrices represent the bandwidth and transmission latency between the network $G_1 = (V_1, E_1)$ and $G_2 = (V_2, E_2)$, respectively.

$$B_{G_1 \times G_2} = \begin{bmatrix} b_{(1,1)} & b_{(1,2)} & \ldots & b_{(1,n)} \\ b_{(2,1)} & b_{(2,2)} & \ldots & b_{(2,n)} \\ \vdots & \vdots & \ddots & \vdots \\ b_{(m,1)} & b_{(m,2)} & \ldots & b_{(m,n)} \end{bmatrix}$$ \tag{5}$$

$$L_{G_1 \times G_2} = \begin{bmatrix} l_{(1,1)} & l_{(1,2)} & \ldots & l_{(1,n)} \\ l_{(2,1)} & l_{(2,2)} & \ldots & l_{(2,n)} \\ \vdots & \vdots & \ddots & \vdots \\ l_{(m,1)} & l_{(m,2)} & \ldots & l_{(m,n)} \end{bmatrix}$$ \tag{6}$$
Similarly, when \( i \in V_1 \) does not connect directly to \( j \in V_2 \), \( b_{(i,j)} \) and \( l_{(i,j)} \) should equal to zero. Generally, \( G_1 \) and \( G_2 \) are far away from each other geographically so that there are very few direct links between them. Thus, it is clear that both \( B_{G_1 \times G_2} \) and \( L_{G_1 \times G_2} \) are large sparse matrices, since most of their elements are zero.

2. **E2E QoS requirements for reliable NS:**

We define \( R_{NS} = (EU_s, EU_d, B_{min}, L_{max}) \) as the requirements of a QoS-sensitive NS. In this case, \( EU_s \) represents the source end users, while \( EU_d \) represents the destination end users, \( B_{min} \) and \( L_{max} \) denote the minimum bandwidth and the maximum latency of this NS, respectively. This means that the bandwidth between \( EU_s \) and \( EU_d \) should not be less than \( B_{min} \), and maximum transmission latency between them should not be longer than \( L_{max} \). For a Type 1 NS, both \( EU_s \) and \( EU_d \) connect with the nodes of the same RAN domain managed by a SDN controller, which is represented as \( G_{RAN} = (V_{RAN}, E_{RAN}) \). We assume that the SDN controller can detect two nodes \( s \in V_{RAN} \) and \( d \in V_{RAN} \) that connect \( EU_s \) and \( EU_d \) directly, respectively. If a specific forward path from \( s \) to \( d \), described as \( P_{(s \rightarrow d)} = s, v_1, v_2, \ldots, v_n, d \), can meet the QoS requirements of a Type 1 NS, then we have the following conditions:

\[
B_{min} \leq \min\{b_{(s,v_1)}, b_{(v_1,v_2)}, \ldots, b_{(v_{n-1},v_n)}, b_{(v_n,d)}\}
\]

\[
L_{max} \geq l_{(s,v_1)} + \sum_{i=1}^{n-1} l_{(v_i,v_{i+1})} + l_{(v_n,d)}
\]

For a Type 2 NS, the source and destination end users are located in different RANs that connect via the same TN. In this situation, the forward path looks like the following:

\[
P_{(s \rightarrow d)} = s, v_1^1, v_2^1, \ldots, v_m^1, v_1^2, v_2^2, \ldots, v_n^2, v_1^3, v_2^3, \ldots, v_k^3, d
\]

where

\[
v_i^1 (1 \leq i \leq m) \in V_{RAN}^1
\]

\[
v_i^2 (1 \leq i \leq n) \in V_{TN}^2
\]

\[
v_i^3 (1 \leq i \leq k) \in V_{RAN}^3
\]

The forwarding path for a Type 2 NS has the form RAN-TN-RAN and should meet the following QoS requirements.

\[
B_{min} \leq \min\{b_{(s,v_1^1)}, b_{(v_1^1,v_2^1)}, \ldots, b_{(v_{m-1}^1,v_m^1)}, b_{(v_m^1,v_1^2)}, b_{(v_1^2,v_2^2)}, \ldots, b_{(v_{n-1}^2,v_n^2)}, b_{(v_n^2,v_1^3)}, b_{(v_1^3,v_2^3)}, \ldots, b_{(v_{k-1}^3,v_k^3)}, b_{(v_k^3,d)}\}
\]

\[
L_{max} \geq l_{(s,v_1^1)} + \sum_{i=1}^{m-1} l_{(v_i^1,v_{i+1}^1)} + l_{(v_m^1,v_1^2)} + \sum_{i=1}^{n-1} l_{(v_i^2,v_{i+1}^2)}
\]

\[
+ l_{(v_n^2,v_1^3)} + \sum_{i=1}^{k-1} l_{(v_i^3,v_{i+1}^3)} + l_{(v_k^3,d)}
\]

Given the QoS requirements of a Type 2 NS, we can easily deduce the QoS requirements of a Type 3 NS. Obviously, it should look more complicated since a Type 3 NS has a longer forwarding path than a Type 2 NS, as described before.

3. Allocation algorithms for reliable NS

Allocating E2E network resources for different types of NS is a challenging task, which should leverage available network resources in the physical network and dynamic user requirements. To solve this problem, we propose corresponding network resource allocation algorithms for NS, which can effectively schedule network resources to satisfy the QoS requirements of different NS types. The core idea of the algorithms is to iterate Dijkstra algorithm to find multiple forwarding paths for a specific NS so that the flows of the same NS are forwarded along multiple paths to get to the destination. With this algorithm, SDN controller splits one big QoS requirement into several smaller ones. We first focus on the situation of a Type 1 NS, where all the end users connect to the same RAN. The input to this algorithm is the QoS requirements of a Type 1 NS, \( R_{NS} = (EU_s, EU_d, B_{min}, L_{max}) \) and the available network resource information of the RAN, i.e., \( B_{G_{RAN}(V_{RAN}, E_{RAN})} \) and \( L_{G_{RAN}(V_{RAN}, E_{RAN})} \). We assume that the SDN controller has the ability to find the \( EU_s \) to connect the RAN through the node \( s \in V_{RAN} \), and \( EU_d \) to connect the RAN thanks to the node \( d \in V_{RAN} \). Therefore, the next key step is how to find one or more forwarding paths form \( s \) to \( d \) in the network topology of RAN to satisfy the QoS requirements \( B_{min} \) and \( L_{max} \) as shown in Expression (7). Algorithm 1 describes the process based on the mathematical formulation mentioned above.

In this algorithm, we supply four input parameters \( R_{NS}, B_{G_{RAN}}, L_{G_{RAN}} \) and \( MaxPathNumber \). The first three parameters are self-explanatory, so we explain only the fourth one, \( MaxPathNumber \). As described above, when a single forward path can not meet the requirements of a NS for some reason, such as a lack of sufficient bandwidth or too much transmission latency, we should try to
find multiple paths that can cooperate to serve this NS. Theoretically, it is possible to find many paths unless the Dijkstra algorithm fails. However, we do not need too many forward paths for a single NS, because the more paths we have for a single NS, the more configuration we need to handle in the network, which probably increases the management cost and impacts the reliability of the whole network. In general, the optimal number of paths depends on the current network status. Therefore, we set a parameter \( \text{Max\_path\_number} \) that enables operators to flexibly decide the maximum number of paths in terms of the real network situation.

Following the initialization, Algorithm 1 gets into the while iteration where there is an important judgment statement, which is \( B_{\min}^t > b_{(v_r,v_{r+1})} \) \& \& \( L_{\max} \geq L_{\text{total}} \). It deals with the situation that the path, generated by the Dijkstra algorithm, cannot meet the bandwidth requirement but can meet the latency requirement of the NS. In this case, we will recompute the path using the Dijkstra algorithm to get a new path to satisfy the new bandwidth \( B_{\min}^t \leftarrow (B_{\min}^t - b_{(v_r,v_{r+1})}) \), in which \( b_{(v_r,v_{r+1})} \) is the bandwidth of the previous path. Before recomputing the path using the Dijkstra algorithm, the bandwidth matrix \( B_{\text{G}\_\text{RAN}} \), denoted by matrix (1), should be updated by subtracting the value of \( b_{(v_r,v_{r+1})} \) from all the related edge nodes of the previous path so that it may generate a different path compared to the previous one. Another important statement is \( L_{\max} < L_{\text{total}} \), which handles the situation, that the total latency of the path can not satisfy the requirement of the NS. The solution is to determine the edge nodes in the path which have maximum latency, and remove them from the network topology \( G_{\text{RAN}}(V_{\text{RAN}},E_{\text{RAN}}) \), as well as to update \( B_{\text{G}\_\text{RAN}} \) and \( L_{\text{G}\_\text{RAN}} \), denoted by matrices (1) and (3), respectively. Then, the Dijkstra algorithm will probably generate different paths in the next iteration.

After the while iteration breaks, we should check two conditions \( n = 0 \) and \( B_{\min} \geq \sum_{i=1}^{n} b_{p_i} \). The first condition means that the Dijkstra algorithm failed to find any available path, and the second condition represents that it found one or more paths, but the bandwidth sum of these paths did not meet the requirements of the NS. Finally, if the whole algorithm successfully runs, it will return one or multiple paths and the corresponding bandwidth they can provide, which is represented by \( \bigcup_{1 \leq i \leq n} [P^i_{(s \rightarrow d)}, b_{p_i}] \). Otherwise, it will return empty set, which means it failed to find any available path.

For a Type 2 NS, we adopt a hop-by-hop policy to process the problem of network resource allocation based on Algorithm 1. Algorithm 2 describes this policy in detail. Algorithm 2 requires many more parameters than Algorithm 1 because the forwarding path takes the more complex form RAN-TN-RAN. In addition to the

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**Algorithm 1:** Network resource allocation for Type 1 NS.

**Input:**
- \( R_{NS} = (EU_s, EU_d, B_{\min}, L_{\max}) \)
- \( B_{G\_\text{RAN}}(V_{\text{RAN}},E_{\text{RAN}}) \)
- \( L_{G\_\text{RAN}}(V_{\text{RAN}},E_{\text{RAN}}) \)
- \( \text{Max\_path\_number} \)

**Output:**
- \( \bigcup_{1 \leq i \leq n} [P^i_{(s \rightarrow d)}, b_{p_i}] \) or \( \emptyset \)

1. Set \( i = 1 \) and \( n = 0; \)
2. Set \( B_{\min}^t = B_{\min}; \)
3. Get \( s \in V_{\text{RAN}} \) and \( d \in V_{\text{RAN}} \) from \( EU_s, EU_d; \)
4. while true do
5. Run Dijkstra algorithm with \( s \) and \( B_{G\_\text{RAN}} \)
6. Get a shortest path \( P^i_{(s \rightarrow d)} \) from \( s \) to \( d; \)
7. if \( P^i_{(s \rightarrow d)} \neq \emptyset \) \& \& \( i \leq \text{Max\_path\_number} \) then
8. \( b_{(v_r,v_{r+1})} \leftarrow \min(b_{(v_1,v_2)}, b_{(v_1,v_2)}, ...); \)
9. \( b_{(v_n-1,v_n)} \leftarrow b_{(v_n,v_d)} \) in \( P^i_{(s \rightarrow d)}; \)
10. \( L_{\text{total}} \leftarrow t_{(v_1,v_{r+1})} + \sum_{j=1}^{l} t_{(v_j,v_{j+1})} + t_{(v_n,d)}; \)
11. else
12. \( n \leftarrow (i - 1); \)
13. goto 20;
14. end
15. if \( B_{\min}^t \leq b_{(v_r,v_{r+1})} \) \& \& \( L_{\max} \geq L_{\text{total}} \) then
16. \( b_{p_i} \leftarrow b_{(v_r,v_{r+1})}; \)
17. \( n \leftarrow i; \)
18. goto 20;
19. end
20. if \( B_{\min}^t > b_{(v_r,v_{r+1})} \) \& \& \( L_{\max} \geq L_{\text{total}} \) then
21. \( b_{(v_r,v_{r+1})} \leftarrow \min(b_{(v_1,v_2)}, b_{(v_1,v_2)}, ...); \)
22. \( b_{(v_n-1,v_n)} \leftarrow b_{(v_n,v_d)} \) in \( P^i_{(s \rightarrow d)}; \)
23. \( B_{\min}^t \leftarrow (B_{\min}^t - b_{(v_r,v_{r+1})}); \)
24. Update \( B_{G\_\text{RAN}} \) by subtracting the \( b_{(v_r,v_{r+1})} \) along the path \( P^i_{(s \rightarrow d)}; \)
25. \( b_{p_i} \leftarrow b_{(v_r,v_{r+1})}; \)
26. \( i \leftarrow (i + 1); \)
27. goto 5;
28. end
29. if \( L_{\max} < L_{\text{total}} \) then
30. \( t_{(v_r,v_{r+1})} \leftarrow \max(t_{(v_1,v_2)}, t_{(v_1,v_2)}, ...); \)
31. \( t_{(v_n-1,v_n)} \leftarrow t_{(v_n,v_d)} \) in \( P^i_{(s \rightarrow d)}; \)
32. Remove edge \( (v_x,v_{x+1}) \in E_{\text{RAN}} \)
33. from \( G_{\text{RAN}}(V_{\text{RAN}},E_{\text{RAN}}); \)
34. Update \( B_{G\_\text{RAN}} \) and \( L_{G\_\text{RAN}}; \)
35. goto 5;
36. end
37. end
38. if \( n = 0 \) \& \& \( B_{\min} \geq \sum_{i=1}^{n} b_{p_i} \) then
39. Return \( \emptyset; \)
40. else
41. Return \( \bigcup_{1 \leq i \leq n} [P^i_{(s \rightarrow d)}, b_{p_i}] \);
QoS requirements \( R_{NS} \) of the NS, we also feed the Algorithm 2 with all the bandwidth and latency data of two RANs and one TN, such as \( B_{G_{RAN}^{1}}, B_{G_{TN}}, B_{G_{RAN}^{2}}, L_{G_{RAN}^{1}} \) and so on. The output of Algorithm 2 is the path sequence from \( EU_{s} \) to \( EU_{d} \) represented by \( P_{E_{RAN}^{1}} \cup P_{E_{TN}} \cup P_{E_{RAN}^{2}} \). The core idea of Algorithm 2 is to consider each RAN or TN as a hop, where we use Algorithm 1 to compute and generate an internal path for each independently. If all internal paths can be generated successfully, they will be assembled to form a complete path sequence as the target output. Otherwise, it will return an empty set. Specifically, we first define two edge node sets that connect directly from \( RAN_{1} \) to \( TN \) and from \( TN \) to \( RAN_{2} \), denoted by \( E_{(RAN_{1} \rightarrow TN)} \) and \( E_{(TN \rightarrow RAN_{2})} \). The two edge sets can be built from \( B_{G_{RAN}^{1} \times G_{TN}} \) and \( B_{G_{TN} \times G_{RAN}^{2}} \) using the filtering conditions \( b(x,y) \geq B_{min} \) and \( l(x,y) \leq L_{max} \). If both edge sets are empty, then we try to select \( (x_{1}, y_{1}) \in E_{(RAN_{1} \rightarrow TN)} \) and \( (x_{2}, y_{2}) \in E_{(TN \rightarrow RAN_{2})} \) to construct a full path \( s \rightarrow x_{1} \rightarrow y_{1} \rightarrow x_{2} \rightarrow y_{2} \rightarrow d \). Through the \textit{For each} iteration, it will search all possible paths and find the target path to meet the QoS requirements.

The network resource allocation algorithm for a Type 3 NS is similar to that of a Type 2 NS, except that the former needs to handle more hops than the latter. It is easy to work out the algorithm for a Type 3 NS; the latter. It is easy to work out the algorithm for a Type 2 NS.

4. Experimental evaluation

In this section, we first introduce the experimental environment and then evaluate the performance of the algorithms described in Section III.

4.1 Experiment Environment Setup

We conduct the experiments on two computers, each with an Intel multi-core i5-4300 CPU and 8 Gb RAM. The two computers are connected directly through network interfaces with a speed of 1Gbps. We select ONOS as our experimental SDN controller, Mininet as the network topology emulation tool and OVS as the virtual switch in the network nodes. ONOS runs on one computer and Mininet/OVS runs on another computer. ONOS is a highly-modular distributed controller, which can be deployed in a large-scale network to form a cluster of controllers. Each controller can manage its own part of the network nodes, communicating with other controllers to keep the consistency of the real-time network status. In addition, ONOS provides many flexible north-bound RESTful interfaces to allow users to develop a new application module, which can be easily integrated into ONOS. In order to evaluate the feasibility of the proposed QoS framework, we develop a prototype system as an ONOS application module which implements the following functions:

- Collect the bandwidth and latency data between any two nodes that connect directly.
- Compute the shortest path according to QoS requirements based on Algorithm 1 and Algorithm 2 for various packet flows. Generally, the packet flow is recognized by the pair of source IP address and destination IP address, so we take it as a NS in this context.
- Allocate bandwidth for a specific flow along the shortest path, which can be handled by ONOS remotely based on the queue feature of the OVS port.
- Steer a specific flow into a specific queue of the OVS port through modifying the flow table in the OVS. Each queue can be configured with a minimum and maximum bandwidth. When a flow

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**Algorithm 2:** Network resource allocation for a Type 2 NS.

<table>
<thead>
<tr>
<th>Input:</th>
</tr>
</thead>
<tbody>
<tr>
<td>( R_{NS} = (EU_{s}, EU_{d}, B_{min}, L_{max}) )</td>
</tr>
<tr>
<td>( B_{G_{RAN}}, B_{G_{TN}} )</td>
</tr>
<tr>
<td>( B_{G_{RAN} \times G_{TN}} )</td>
</tr>
<tr>
<td>( B_{G_{TN} \times G_{RAN}} )</td>
</tr>
<tr>
<td>( L_{G_{RAN}}, L_{G_{TN}}, L_{G_{RAN}^{2}} )</td>
</tr>
<tr>
<td>( L_{G_{RAN} \times G_{TN}}, L_{G_{TN} \times G_{RAN}} )</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Output:</th>
</tr>
</thead>
<tbody>
<tr>
<td>( P_{E_{RAN}^{1}} \cup P_{E_{TN}} \cup P_{E_{RAN}^{2}} ) or ( \Phi )</td>
</tr>
</tbody>
</table>

1. Set \( E_{(RAN_{1} \rightarrow TN)} = \Phi \);
2. Set \( E_{(TN \rightarrow RAN_{2})} = \Phi \);
3. Get \( s \in V_{RAN}^{1} \) and \( d \in V_{RAN}^{2} \) from \( EU_{s}, EU_{d} \);
4. if \( b(x,y) \geq B_{min} \) and \( l(x,y) \leq L_{max} \) do
5. \( E_{(RAN_{1} \rightarrow TN)} \leftarrow E_{(RAN_{1} \rightarrow TN)} \cup \{y(x,y)\} \);
6. end
7. if \( b(x,y) \neq 0 \) and \( l(x,y) \leq L_{max} \) do
8. \( E_{(TN \rightarrow RAN_{2})} \leftarrow E_{(TN \rightarrow RAN_{2})} \cup \{x(x,y)\} \);
9. end
10. if \( b(x,y) \geq B_{max} \) and \( l(x,y) \leq L_{max} \) then
11. \( E_{(RAN_{1} \rightarrow TN)} \leftarrow E_{(RAN_{1} \rightarrow TN)} \cup \{x(x,y)\} \);
12. return \( \Phi \);
13. if \( b(x,y) \neq 0 \) and \( l(x,y) \leq L_{max} \) then
14. \( E_{(TN \rightarrow RAN_{2})} \leftarrow E_{(TN \rightarrow RAN_{2})} \cup \{y(x,y)\} \);
15. end
16. if \( b(x,y) \neq 0 \) and \( l(x,y) \leq L_{max} \) then
17. \( E_{(RAN_{1} \rightarrow TN)} \leftarrow E_{(RAN_{1} \rightarrow TN)} \cup \{x(x,y)\} \);
18. end
19. if \( b(x,y) \neq 0 \) and \( l(x,y) \leq L_{max} \) then
20. \( E_{(TN \rightarrow RAN_{2})} \leftarrow E_{(TN \rightarrow RAN_{2})} \cup \{y(x,y)\} \);
21. if \( L_{total} \leq L_{max} \) then
22. \( E_{(RAN_{1} \rightarrow TN)} \leftarrow E_{(RAN_{1} \rightarrow TN)} \cup \{x(x,y)\} \);
23. end
24. return \( \Phi \);
25. end
is steered into a specific queue, the transmission speed of this flow will be constrained between the minimum and maximum bandwidth configured for this queue, which is guaranteed by the packet schedule algorithm in OVS, such as HTB (Hierarchy Token Bucket) or HFSC (Hierarchical Fair Service Curve). For example, let’s assume there are two ports p1 and p2 in an OVS node and two flows f1 and f2 belonging to two different NS, respectively. If both f1 and f2 pass through this OVS from p1 to p2, we can create two queues q1 and q2 for port p2. Then, f1 and f2 can be steered into q1 and q2 when they enter this OVS from p1 and leave from p2. Consequently, the transmission speed of f1 or f2 should not be less than the minimum bandwidth and should not exceed the maximum bandwidth of the corresponding queue. In this way, ONOS and Mininet/OVS can be used to simulate the network performance when different types of NS are created in various network environments.

4.2 Performance Evaluation

This evaluation assesses the processing time of the algorithms and the bandwidth utilization of the networks after every NS request. The processing time is particularly important for users when they request a new NS with specific requirements. However, the processing time may be related with many factors, such as the number of network nodes, current network state, the position of end users, the computing efficiency of the underlying algorithm, etc. In the beginning, there is enough available bandwidth in the network, so the algorithms can find a forwarding path for a new NS very quickly. As the available bandwidth and the number of links in the network decrease, the algorithms will intuitively take a longer time to find suitable paths, and may even fail to find one. In addition, the size and type of network topology also affect the execution time of the algorithms to some extent. Therefore, by investigating the processing time for creating a new NS and the bandwidth utilization of networks, we can observe the performance of the proposed algorithms for different types of NS and different network statuses, which may provide some useful guidance in practice. In order to observe the relationship between the processing time and the related factors, we measure the processing time and the bandwidth utilization of networks by sending 100 creation requests for new NS continuously in the same network topology, which is designed as a full-mesh network. Any two nodes in a full-mesh network connect directly and the reason why we choose full-mesh network is to better observe the behavior of the algorithms for different NS types.

We first test the situation of requesting new Type 1 NS in a full-mesh network that contains 20 and 100 OVS nodes. The bandwidth and latency matrix of the network is generated randomly with the constraints \( b(i,j) \in [50 \text{ Mbps}, 100 \text{ Mbps}] \) and \( l(i,j) \in [1 \text{ ms}, 10 \text{ ms}] \). Fig. 2 shows the network topology and experiment environment for Algorithm 1.

For each Type 1 NS request, the QoS requirements \( R_{NS} = (EU_s, EU_d, B_{min}, L_{max}) \) are also generated randomly, which means that the positions of the end users are changing randomly. As for the minimum bandwidth and maximum latency, we have the constraints of \( B_{min} \in [1 \text{ Mbps}, 5 \text{ Mbps}] \) and \( L_{max} \in [50 \text{ ms}, 100 \text{ ms}] \). Fig. 3 presents the processing time for the Type 1 NS request based on the above situation. In the figure, the Y-axis denotes the processing time in milliseconds and the X-axis shows the 100 requests for the creation of a new Type 1 NS, numbered from 1 to 100. The straight line represents the processing time in the full-mesh network with 20 OVS nodes, while the triangular dotted line shows the processing time in case of the network with 100 OVS nodes. For the sake of convenience, we call the full-mesh network with 20 OVS nodes as Network A and the full-mesh network with 100 OVS nodes as Network B. We can see that as the number of requests for new NS increases, the processing time of Network A increases from 2.5 ms to 15 ms at an approximately linear rate. This shows that the available network resources in the same network are decreasing, and the SDN controller takes longer time to create a new NS, because the forwarding path of the new NS includes more OVS nodes, even sometimes needs multiple sub-paths to satisfy the requirements. We also observe that the processing time for Network B increases faster than that for Network A. As expected, the reason is that the target path in Network B becomes more complicated than that of Network A with the decrease of network resources in it. Moreover, there

Fig. 2: Network topology and experiment environment for evaluating the performance of Algorithm 1.
is a slight fluctuation in both lines, which means that sometimes a new request requires less time than the previous one. For example, the processing time of the 74th request in Network B is about 35 ms while that of the 77th request is about 28 ms. We believe this may be caused by the random positions of the end users as input parameters of Algorithm 1.

Next, we measure the bandwidth utilization of the underlying network. The bandwidth utilization of a network is defined as the mean value of the bandwidth utilization of all the links in the network. Fig. 4 shows the bandwidth utilization after creating each of the 100 envisioned Type 1 NS in case of a RAN (i.e., as we focus on Type 1 NS) consisting of 20 nodes and 100 nodes, respectively. As expected, the bandwidth utilization of RAN increases as the number of requests for new NS grows. From the figure, it becomes also clear that the bandwidth utilization of 20 nodes increases more quickly than that of 100 nodes. At the end, after the 100th request of a new NS, the bandwidth utilization of 20 nodes RAN gets to about 60 percent, while the bandwidth utilization of 100 nodes RAN gets to only about 38 percent. The reason is that the more nodes a network has, the more paths a new NS can select from this network. Therefore, the usage of links becomes more balanced and more bandwidth remains available.

Similarly to the evaluation method for Type 1 NS, we conduct experiments of requesting new Type 2 NS in a more complicated network topology that contains three full-mesh networks, which represent RAN1, TN, and RAN2, respectively. RAN1 and TN are connected only by two links, which mimics the long-distance fiber connection between two network domains in a real network environment. Similarly, TN and RAN2 are also connected by two links directly. We set the bandwidth and latency of the links between two full-mesh networks as 200 Mbps and 1 ms. The bandwidth and latency matrices of the three full-mesh networks are generated randomly in the same way used in the Type 1 NS. Fig. 5 shows the network topology for evaluating the performance of Algorithm 2.

For each Type 2 NS request, the QoS requirements $R_{NS} = (EUs, Eu_d, B_{min}, L_{max})$ are also generated randomly with the constraints $B_{min} \in [1\, Mbps, 5\, Mbps]$ and $L_{max} \in [50\, ms, 100\, ms]$. Fig. 6 presents the relationship between the processing time and the number of requests for Type 2 NS based on this situation.

In Fig. 6, as described before, we call the network with 20 OVS nodes as Network A, which means each network domain (RAN1, TN and RAN2) contains 20 OVS nodes in this scenario. Similarly, each network domain of Network B contains 100 OVS nodes. From the figure, we observe that the processing time for Type
2 NS experiences a sudden decrease when the number of requests for new NS is between 40 and 50. For Network A, the processing time decreases from about 60 ms to 18 ms at the 50th request. For Network B, the processing time decreases from about 140 ms to 25 ms at the 40th request. The main reason for this relates to the policy of the search path in Algorithm 2. When the bandwidth of one link between two network domains is consumed, Algorithm 2 will choose another link to find an E2E path. In this case, the edge nodes for the two network domains are changed, which probably produces a shorter path inside each network domain and finally reduces the processing time. We can also make a quantitative analysis. The bandwidth of one link between two network domains is 200 Mbps, and the constraint of NS requirements is \( B_{\text{min}} \in [1 \text{ Mbps}, 5 \text{ Mbps}] \), therefore it is reasonable to consume all the bandwidth of one link when the number of requests is about 40 to 50.

Next, we measure the bandwidth utilization of Network B when applying Algorithm 2. Fig. 7 shows the bandwidth utilization of RAN1, RAN2 and TN, respectively after creating each of the 100 Type 2 NS. In Fig. 7, we observe that the bandwidth utilization of TN increases more quickly than that of RAN1 and RAN2 is similar as the number of Type 2 NS requests increases, but the bandwidth utilization of TN increases more quickly than that of RAN1 and RAN2. At the end, after the 100th request for a new NS, the bandwidth utilization of TN gets to about 65 percent, while the bandwidth utilization of RAN1 and RAN2 reaches only about 50 percent and 55 percent, respectively. We believe that the reason for this is that all the edge nodes of TN are fixed when the new NS is created, so the NS paths in the TN are located intensively, which will probably lead to higher bandwidth utilization.

To validate Algorithm 3 for Type 3 NS, we construct a more complicated network topology. Fig. 8 shows the network topology for evaluating the performance of Algorithm 3. As shown in Fig. 8, the network topology contains five full-mesh networks, including RAN1, RAN2, TN1, TN2, and CN. RAN1 and TN1, RAN2, and TN2 are each connected only by two direct links, and we set the bandwidth and latency of these links to 200 Mbps and 1 ms. TN1 and CN, and TN2 and CN are also connected by two links, and we set the bandwidth and latency of these links to 500 Mbps and 1 ms. Similarly, the bandwidth and latency matrices of the five full-mesh networks are generated randomly in the same way used in the Type 2 NS, and every full-mesh network contains 20 or 100 OVS nodes.

For each Type 3 NS request, the QoS requirements \( R_{NS} = (EU_s, EU_d, B_{\text{min}}, L_{\text{max}}) \) are also generated...
Type 3 NS experiences a sudden rise when the number of requests for Type 3 NS increases from about 80 ms to 180 ms at the 28th request. We believe the main reason for this relates to the network topology for Algorithm 3. When the number of network domains is greater than three and the number of new NS requests reaches a certain threshold, the number of iterations of Algorithm 3 will sharply increase, which consequently consumes more processing time. Another experimental result is that the processing times of new Type 3 NS in case of Network A and B are very similar, which indicates that the scale of nodes in the network is not an important factor for the processing time due to the high complexity of Algorithm 3.

In the same way, we measure the bandwidth utilization in case of Network B (i.e., consisting of 100 nodes in each domain) for Algorithm 3. Fig. 10 shows the bandwidth utilization of RAN1, RAN2, TN1, TN2, and CN, respectively, after creating each of the 100 Type 3 NS. In Fig. 10, we notice that the bandwidth utilizations of RAN1 and RAN2 are similar as the number of Type 3 NS requests increases, but the bandwidth utilizations of TN1 and TN2 are slightly higher than those of RAN1 and RAN2. The bandwidth utilization of CN is the highest among all the network domains. At the end, after the 100th request of Type 3 NS, the bandwidth utilization of CN gets to about 78 percent, while the bandwidth utilization of the other network domains reaches between 50 percent and 60 percent. The reason for this is that all the edge nodes of CN are fixed. When a new NS is created, the NS paths in the TN or CN are located intensively, which will probably lead to higher bandwidth utilization. Therefore, in a real network, we should improve the structure of the network topology to eliminate these problems. For example, we can simplify the structure of TN and CN, or increase the bandwidth of their inner links.

5. Conclusion

In this paper, we proposed a QoS framework for fine-granular network slicing in Beyond 5G systems that is based on SDN and NFV. By dividing the network into three parts, namely RAN, TN, and CN, we described the function modules of the QoS framework in each domain. We then defined a requirement model of the B5G network resources and presented three different types of NS. Following the formulation of the network resources in NS, algorithms for network resource allocation for different NSs were proposed. Finally, we conducted simulations to evaluate the performance of these algorithms. The results showed that the proposed algorithms could satisfy the QoS requirements of different end users and may provide useful guidance to create and deploy NS in B5G. Future work will involve applying the algorithms in a real network and improving the algorithms according to the obtained feedback.

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References


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