



TCP and MP-TCP in 5G mmWave Networks

Future 5G networks will likely include mmWave radio access communication links, because of their potential multi-gigabit-per-second capacity. However, these frequencies are characterized by highly dynamic channel conditions, which lead to wide fluctuations in the received signal quality. This article explains how the end-to-end user experience in mobile mmWave networks could be affected by a suboptimal interaction between the most widely used transport protocol, TCP, and mmWave links. It also provides insights on the throughput-latency tradeoff when Multipath TCP (MP-TCP) is used judiciously across various links, such as Long-Term Evolution (LTE) and mmWave.

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The next generation of mobile networks (5G) will be standardized before 2020, to address mobile Internet traffic's growth. According to the Next-Generation Mobile Network (NGMN) alliance, 5G networks shall provide the following¹: a user bitrate of at least 50 megabits per second (Mbit/s) at cell edges, with gigabits per second (Gbit/s) peaks in the most favorable conditions; ultra-low end-to-end latency (possibly below 10 ms); high reliability and availability of communications; and support for low-power massive machine-type communications (MTC).

MmWave communications will play a major role in meeting the throughput target of 5G networks. At frequencies

above 10 gigahertz (GHz), indeed, there is a high availability of contiguous spectrum that can be allocated to cellular networks. However, mmWaves present numerous challenges and issues that researchers must address to make this technology market-ready. In fact, these frequencies suffer from high isotropic path loss and blockage by most solid materials – for example, buildings, cars, and even the human body² – which could result in service unavailability (that is, an outage). The high antenna gain provided by using multiple antenna techniques, such as beamforming and massive multiple input, multiple output (MIMO), can make up for the path loss; but how to provide a reliable service in the

presence of frequent blockages is still an open question. In particular, a sudden transition of a mmWave link from a line-of-sight (LOS) to a non-line-of-sight (NLOS) channel state generates wide fluctuations in the signal-to-interference-plus-noise ratio (SINR; in the order of 30 dB, according to Sundeep Rangan and colleagues²), and therefore the capacity offered changes drastically, thereby compromising the end-to-end user experience.

These extreme propagation conditions demand a new design of the physical and medium access control (MAC) layers, but also have an impact on the interplay with the higher layers of the protocol stack. The most widely used reliable data transport protocol is TCP, which however was designed in a different historical context, and according to different needs and constraints. TCP considers packet losses as an implicit notification of congestion on the link, and therefore reduces the sending rate trying to relieve it. However, the lossy nature of mmWave links might trigger the TCP congestion control mechanisms even if there is no actual congestion. This might yield a suboptimal end-to-end performance and waste the great potential of mmWave links.

With this in mind, here we provide an overview of the impact of mmWave communications on the performance of TCP, outlining which are the main novelties that the usage of a mmWave radio access network (RAN) introduces in end-to-end connections at the transport layer. We also consider how adopting the recent Multipath TCP (MP-TCP) option can improve the performance of TCP on these kinds of links. Finally, we suggest possible future research directions. A more technical discussion with additional results is provided elsewhere.³

Recent Advances in TCP

TCP was designed in the 1980s as a connection-oriented and reliable protocol that provides end-to-end connectivity over multiple hops and congestion control (CC). Reliability is enabled by a retransmission mechanism, based on the acknowledgments received by the TCP transmitter. Congestion control is implemented by different algorithms that increase and/or decrease the maximum amount of unacknowledged data that the sender is allowed to transmit (the congestion window), reacting to network events such as packet losses. The original congestion

control algorithms has had several evolutions: the Request for Comments (RFC) 5681⁴ describes the most recent one, and there are 13 TCP variants currently implemented in the Linux kernel.⁵ In particular, the latest versions of the major operating systems use TCP CUBIC as the default.⁶ This TCP version was designed in order to react more promptly to packet losses, and to restore the connection to the highest available rate in a shorter time than legacy designs based on TCP New Reno. In the latter, after a packet loss, the sender decreases its congestion window by half, and then increases it by one packet in each round-trip time (RTT), thus requiring a very long time to fill a high capacity link. With TCP CUBIC, instead, the growth is independent of the RTT, which makes it possible to reach a higher throughput more quickly than in New Reno, while behaving fairly toward flows using other versions of TCP.⁶

Another trend in modern cellular networking is to exploit the presence of multiple network interfaces in mobile devices. A typical example is the usage of MP-TCP for vertical and seamless handovers between third- and fourth-generation (3G/4G) cellular networks and Wi-Fi hotspots.⁷ MP-TCP is an extension of TCP that enables multipath transport – that is, the application communicates with a traditional TCP socket, which transparently handles multiple subflows on different interfaces, such as Wi-Fi, the cellular network, and Ethernet. MP-TCP is currently under discussion in the IETF,^{8,9} and is designed around three main goals.¹⁰ First, it should improve the throughput, in the sense that it should perform at least as well as a single path TCP (SP-TCP) flow on the best path available. Second, it should not use more resources than standard TCP flows. Finally, it should steer more packets towards less congested paths.

The core of MP-TCP is its congestion control algorithm, that manages the congestion window of each different subflow in a coupled or uncoupled manner. With an uncoupled congestion control, each subflow is independent, thus the congestion window is updated separately in each path. With a coupled approach, instead, the congestion window of the different subflows is increased and decreased in a coordinated manner, considering the congestion of all the available subflows. By coupling the different subflows, Costin Raiciu and colleagues¹⁰ claim that it is possible to reach the aforementioned

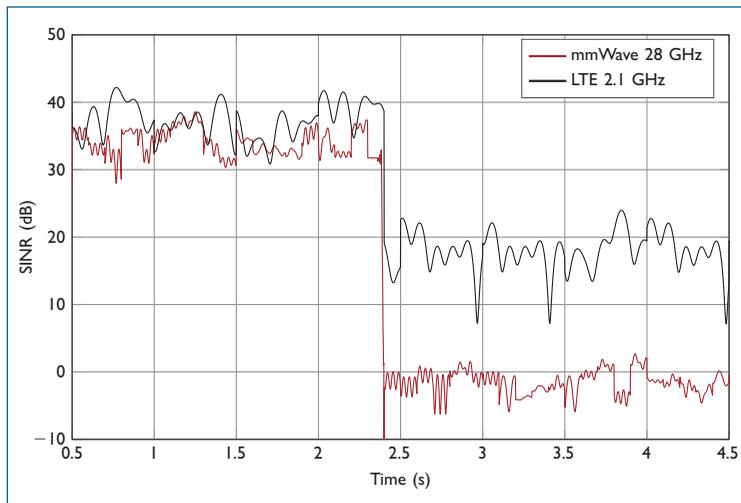


Figure 1. The ns-3 simulated signal-to-interference-plus-noise ratio (SINR) for a mmWave and Long-Term Evolution (LTE) link, with the user equipment moving at 2 meters per second (m/s) from (45, 0) to (55, 0), with the evolved node base at coordinates (75, 50). The traces are generated using the channel model for the mmWave frequencies (described elsewhere^{2,15}) and the ns-3 LTE channel model.

MP-TCP design goals. The first coupled congestion control (CC) that they proposed,¹⁰ however, received criticism,^{11,12} because it transmits too much traffic on congested paths and is unfriendly with respect to SP-TCP. Ramin Khalili and colleagues¹¹ presented the Opportunistic Linked Increases Algorithm (OLIA), which is designed to overcome these two issues. However, according to Qiuyu Peng and colleagues,¹² OLIA presents unresponsiveness problems with respect to congestion changes in the subflows. The Balanced Linked Adaptation algorithm (BALIA)¹² addresses both the problems of the original CC and those of OLIA. In particular, the parameters of the protocol are derived through a theoretical analysis of the performance of multipath congestion control algorithms. However, these schemes are based on the legacy design of Reno and New Reno congestion control algorithms, and thus share with them the drawbacks we described in the previous paragraph.

TCP Performance on mmWave Networks

In this article, we discuss the main issues that arise when TCP and MP-TCP are deployed in

networks with mmWave links, showing how state-of-the-art transport protocols interact with this technology.

The mmWave Channel

Because TCP is the most widely used transport protocol, it is important to understand the interactions that exist between mobile networks (for example, wireless channels) and the TCP performance. In wireless networks, the loss of a packet is not caused necessarily by congestion, but instead might be due to a sudden (and possibly only temporary) drop in signal quality. In related work,^{13,14} the authors study the behavior of TCP in relation to a complex mobile network such as LTE, showing the following: first, as the distance between the user equipment (UE) and the evolved node base (eNB) increases, the TCP throughput degrades; and second, how TCP is affected by network events such as handovers.

MmWave networks are expected to reach an order of magnitude higher throughput than current systems, thanks to the larger bandwidth available, but present more troublesome propagation conditions. As an example, Figure 1 shows a comparison between the time evolution of the SINRs of a mmWave link at 28 GHz and an LTE link at 2.1 GHz for a UE that moves at 2 meters per second (m/s) at an average distance of 75 meters from the eNB and switches from a LOS to a NLOS condition. The main differences between the mmWave and the LTE channels are as follows:

- The LOS to NLOS path loss transitions are deeper for mmWave. Below 6-GHz frequencies, these are of the order of 10–15 dB, while for the mmWave channel the fluctuation can exceed 30 dB. Therefore, the available capacity changes dramatically. Moreover, mmWave networks will be small cell networks, and mmWave links are sensitive to blockage from foliage, the human body, moving obstacles and so on, thus the transitions from LOS to NLOS for mmWave connections will be much more frequent than in LTE.² Furthermore, given the shorter range of mmWave communication, it is more probable for mmWave links than for LTE ones to experience an outage (where no signal is received) because of shadowing.
- The mmWave channel has a shorter coherence time, which results in variations of the

channel on the order of hundreds of microseconds,² that are faster than those in current mobile networks. As Figure 1 shows, the transitions of the LTE channel due to fading are much smoother than those of the mmWave link.

Currently, these issues are mainly addressed in the RAN, but they also impact the performance of transport layer protocols. To date, there are only few published results on the performance of TCP over 5G mmWave links. There are several reasons for this. The first is the lack of mobile testbeds that scale beyond a single mmWave link. There are indeed commercial prototypes that can be used to set up a mmWave link between a MIMO transmitter and receiver, but the focus of these products is on the single connection. The deployment of large testbeds is limited by the lack of a definitive standard and by the cost and complexity of prototyping the electronics for these links. Moreover, an end-to-end system composed of a mmWave RAN, a core network (CN), and the Internet with remote servers has an extremely high complexity, and this prevents an insightful theoretical analysis of the system. Finally, an analytical channel model that represents the correlation of fading over time in mmWave links has not been developed yet.

As mmWave networks and 5G are not currently standardized, a good compromise between the realism of a testbed and the convenience of a theoretical analysis is system-level simulation, which can provide deep insights on a complex system while avoiding the need for expensive dedicated hardware. In this article, we use an extension¹⁶ of the NYU mmWave¹⁷ module of ns-3, a widely used open source network simulator, together with the TCP/IP stack of the Linux kernel, to provide a realistic insight on the interplay between TCP and mmWave links.

Link-Level Retransmissions and TCP

One of the main issues of mmWave links is the highly dynamic channel that, because of blockages, experiences wide fluctuations of the perceived SINR and therefore is characterized by unreliable transmissions. Current and future mobile networks deploy different retransmission mechanisms to mitigate the effects of packet loss and increase the mobile devices'

throughput. When using mmWave links, these retransmission protocols become a key element in hiding the highly dynamic and consequently unstable behavior of the channel to higher-layer transport protocols such as TCP.

The MAC layer uses a Hybrid Automatic Repeat reQuest (HARQ) mechanism. When the physical layer at the receiver receives a packet, but detects the presence of some errors that prevent reliable decoding, it asks for a retransmission. The sender then transmits additional redundancy that helps retrieve the packet without any error.¹⁸ Moreover, in 3GPP-like networks (such as LTE), there is a layer on top of the MAC layer that might perform additional link-level retransmissions – the radio link control (RLC) layer – that will also likely be a part of the 5G protocol stack.¹⁹ Because the number of retransmissions at the MAC layer is usually limited (typically only three attempts are performed), the RLC layer acknowledged mode (AM) offers another way of recovering lost packets. Thanks to periodic acknowledgments from the receiver, the RLC AM sender knows which packets were lost and can retransmit them. The number of attempts that RLC AM can perform is also limited. The RLC unacknowledged mode (UM) instead does not perform any retransmission in addition to those of the HARQ at the MAC layer. Both HARQ and RLC AM operate based on information related to the link, and with a greater timeliness with respect to TCP, which uses packet losses to detect congestion and operates on the larger timescale of retransmission time-outs (RTOs), of the order of a second.

Menglei Zhang and colleagues presented a first study of TCP in mmWave cellular networks.²⁰ The authors studied different scenarios, with either a statistical channel model, based on real measurements in New York City, or a raytracing model, with traces from a route in Bristol. They observed that, in the presence of long outage events, neither HARQ nor RLC AM retransmissions are able to mask the channel losses to the TCP sender, which triggers an RTO event, after which it might take a long time to recover the full capacity. Moreover, because of the presence of these retransmission mechanisms, TCP does not promptly track the mmWave link state when the channel between the UE and the eNB switches from an LOS to NLOS condition. This issue is more marked in mmWave than in current cellular networks,

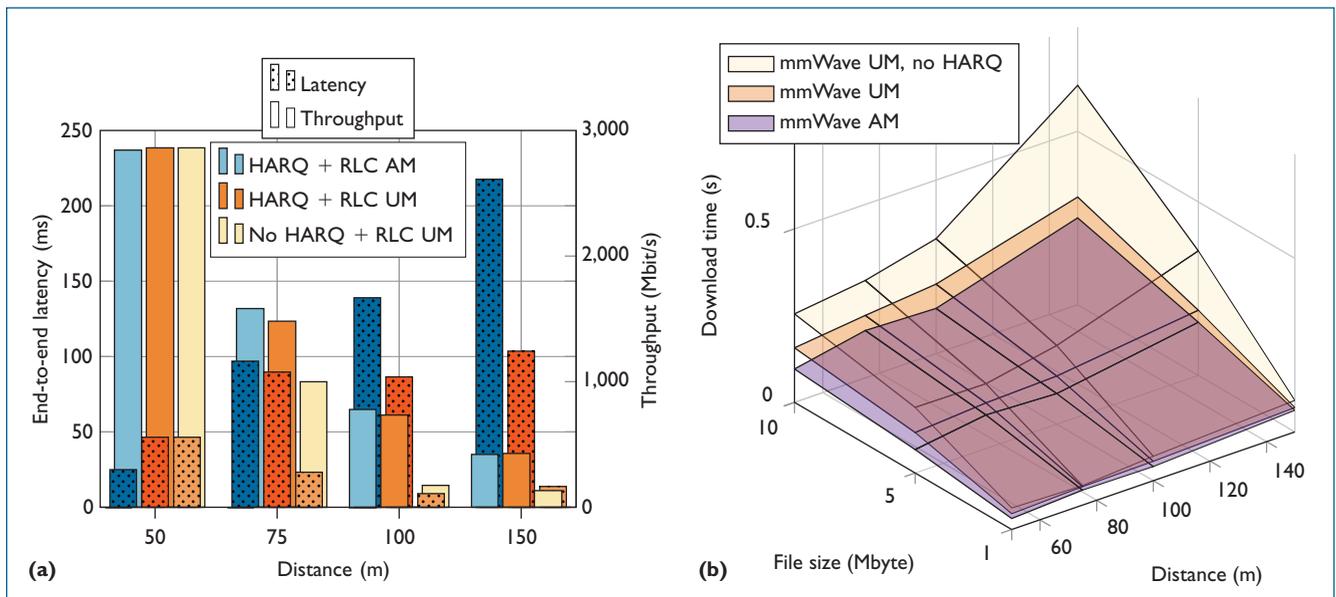


Figure 2. TCP performance on mmWave links, with and without link-level retransmissions. (a) TCP latency (dotted bars) and throughput, with and without the different retransmission mechanisms of the mmWave protocol stack. (b) Download time as a function of the file size and of the distance, for TCP with and without lower-layer retransmissions. (AM stands for acknowledged mode; HARQ stands for Hybrid Automatic Repeat reQuest; RLC stands for radio link control; and UM stands for unacknowledged mode.)

because LOS to NLOS transitions are much more frequent and provide wider variations in the link capacity.² The consequence is that the packets queue in the RAN buffers, and latency increases. This is an instance of the well-known bufferbloat problem²¹: because of the large size of buffers in network devices, packets are not dropped but queued and eventually transmitted. While this improves the throughput of the mmWave link, it also increases the overall latency.

This tradeoff can be better understood by considering the results in Figure 2a, where we use the iPerf network bandwidth measurement tool on top of TCP CUBIC with and without link-level retransmissions, at a different distance d between the UE and the eNB. At a small distance (that is, $d = 50$ m), the achievable TCP throughput is high even without retransmissions, because the UE is in LOS with very high probability. As the distance increases, two different behaviors can be observed, if retransmissions are used: the achievable throughput is from 1.5 to 3 times higher than without retransmissions; however, the latency increases, especially when MAC and RLC layer

retransmissions are combined at very large distances.

If we consider the download of small files (from 1 to 10 Mbytes) using `wget`, which can be representative of web browsing, then, as Figure 2b shows, the difference between the download times with RLC AM (both RLC retransmissions and HARQ) and with RLC UM (HARQ only) is more noticeable than that between the values of Figure 2a, showing that, for short-lived TCP sessions, it is important to perform retransmissions as quickly as possible – that is, at a layer as close to the radio link as possible.

The design goals of 5G networks, nonetheless, demand both high throughput and low latency. However, as shown in Figure 2a and other work,²⁰ when the distance between eNB and UE is high or there are blockages and an NLOS channel, a high TCP throughput can be reached only with retransmissions, thus increasing latency. The latency-throughput tradeoff is an issue that must be solved before mmWave networks can be considered market-ready. A first step, proposed in related work,^{22,23} is a cross-layer approach that makes TCP aware

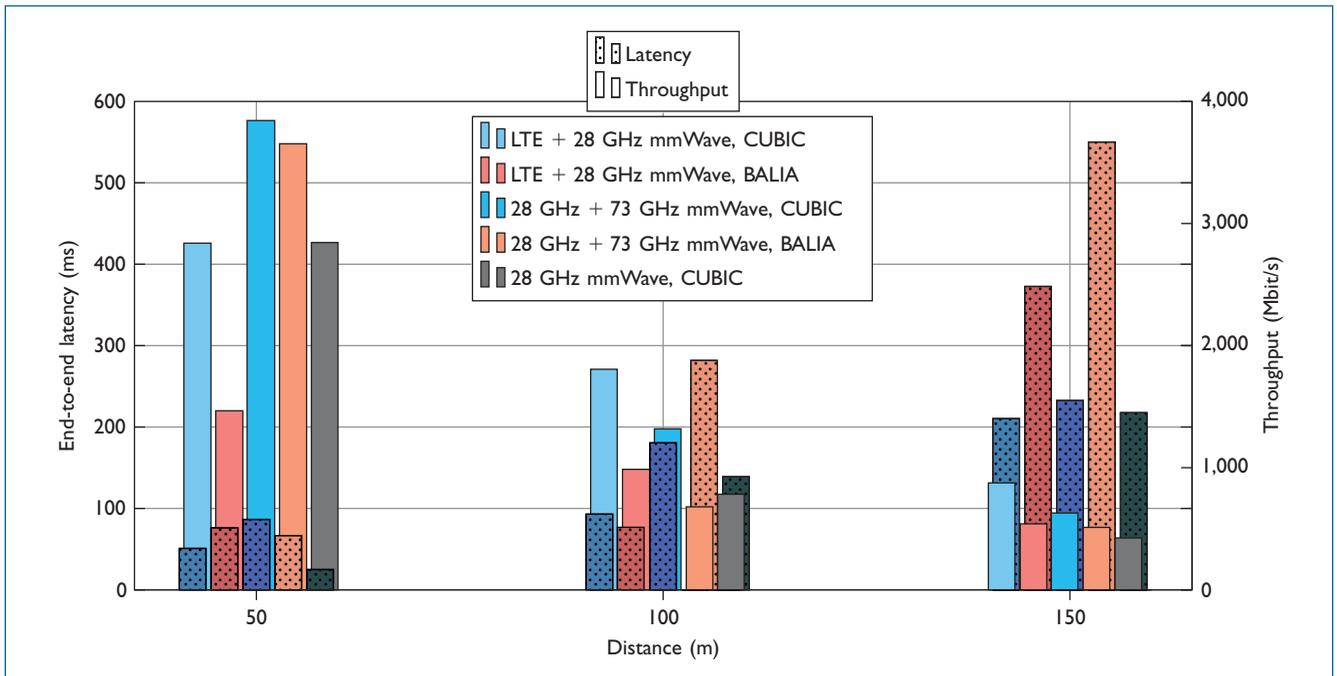


Figure 3. Multipath TCP (MP-TCP) throughput and latency for different distances d and different MP-TCP options. The rightmost bar in each group shows the performance of a single path TCP (SP-TCP) connection with TCP CUBIC over a 28-GHz mmWave link as a reference. (BALIA stands for Balanced Linked Adaptation algorithm.)

of the actual channel conditions. A second solution might involve the replacement of TCP with other transport protocols, or the redesign of HARQ and RLC-layer retransmissions in the 3GPP stack. Finally, an ultra-dense deployment could decrease the probability of having NLOS links toward mmWave eNBs; however, in this case the issue becomes a robust management of mobility, handovers, and cell association.¹⁵

Multipath TCP in mmWave Networks

5G devices will likely be connected to multiple Radio Access Technologies (RATs) or to multiple mmWave eNBs.¹⁵ This is why it is interesting to analyze the performance of MP-TCP over mmWave and LTE links, since it could be used as an end-to-end solution for multi-connectivity. However, the complex interactions that exist between the reliability of the different paths and the different congestion control algorithms require a deep understanding of MP-TCP performance before deploying it on mmWave links.

First, we consider whether using LTE or mmWave as a secondary subflow yields a higher throughput. In Figure 3 we show the

throughput performance of different MP-TCP congestion control algorithms over different connections, with respect to the baseline of a SP-TCP connection with TCP CUBIC on a mmWave link. When the UE has a high probability of being in LOS (that is, for $d \leq 50$ m), the solution with MP-TCP on mmWave-only links outperforms SP-TCP, with a gain between 800 Mbit/s and 1 Gbit/s (about 30–40 percent). The LTE link, instead, has a much smaller rate than a mmWave link when $d \leq 50$ m, and therefore the throughput of MP-TCP on LTE and mmWave subflows is close to or worse than that of the reference SP-TCP. However, the performance of an LTE link is less dependent on the distance d than that of a mmWave link; thus, for higher distances, MP-TCP over LTE and mmWave links performs better than MP-TCP with only mmWave connections. The 73-GHz mmWave subflow, indeed, provides a larger theoretical capacity than the LTE one, but has a lossy behavior that penalizes the overall throughput, except for small distances. As Figure 3 shows, for $d = 150$ m, MP-TCP with LTE and 28 GHz mmWave offers a gain of more than 450 Mbit/s

(about 100 percent) with respect to SP-TCP (that is, more than the LTE link throughput), without increasing the latency. This shows that link diversity –that is, the presence of the secondary and reliable LTE path – improves the overall connection's throughput.

Second, we compare the performance of two different congestion control algorithms for MP-TCP: BALIA (that is, a coupled CC algorithm), and CUBIC (an uncoupled algorithm). As Figure 3 shows, MP-TCP with the coupled BALIA CC algorithm fails to meet the first target of MP-TCP design, because in many cases its throughput is lower than that of SP-TCP. When coupling a mmWave and an LTE link, indeed, the congestion control algorithm sees the losses on the 28-GHz mmWave connection as congestion, and steers the whole traffic to the LTE subflow, degrading the end-to-end connection's performance. Instead, the uncoupled congestion control algorithm is unaffected by this issue, because each path behaves independently. However, in this case the MP-TCP flow might be unfriendly to SP-TCP flows on shared bottlenecks.

Therefore, the definition of a CC algorithm that meets the three design goals of MP-TCP for mmWave links is still an open area of research. The current CC algorithms either waste the potential gain given by multi-connectivity, because of suboptimal balancing between flows, or harm the fairness among network users. Therefore, before deploying these algorithms in future mmWave networks, a proper congestion control algorithm should be designed and thoroughly tested.

We provided an overview of the issues that might arise from the complex interactions between TCP – either single or multipath – and mmWave links. The interplay between communications at such high frequencies and legacy transport protocols could be a limitation that prevents the full exploitation of the potential of mmWave communications. For mmWave, it is fundamental to mask the channel losses to the higher TCP layer with link-level retransmissions to reach high throughput, but these mechanisms introduce additional latency. An important contribution could be given by multi-connectivity through MP-TCP at the transport layer, where path diversity offers promising results, but a suitable CC algorithm has not been identified yet.

An optimized interaction between transport protocols and mmWave networks will be a fundamental enabler of 5G networks. In particular, the main issue is related to how to reduce latency while maintaining a high and stable throughput while using TCP and MP-TCP over mmWave networks, so that the end users can effectively exploit the huge amount of resources available in fifth-generation cellular networks.

To reach this goal, in the next few years the following open research questions must be addressed. How do we design reliable handover mechanisms that do not impact TCP performance in high mobility scenarios? How can we optimally exploit multi-connectivity, either with MP-TCP, or with different solutions at a lower layer in the RAN, also studying the tradeoff between energy consumption and very high throughput? And how can we combine the stability and reactivity of TCP CUBIC with a MP-TCP congestion control algorithm coupled over 5G mmWave and legacy links? □

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