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FURTHER UPDATES TO THE PRELIMINARY DRAFT NEW REPORT ITU-R M. [IMT.MULTIMEDIA]

Capabilities of the terrestrial component of IMT-2020 for multimedia communications

1 Introduction

During the 43rd meeting of Working Party 5D, working document addressing the new multimedia related capabilities of IMT-2020 was further developed and upgraded to PDNR, with a view to its approval in the 44th meeting of WP5D.

2 Proposal

In this document, some further changes are proposed to the PDNR, particularly for the addition of the DECT standard of IMT-2020 in section 6. In addition some other minor changes are also made in the title of section 7.7 and a new section 7.8 has been added.

The table of contents has also been updated.

The proposed changes to the working document are in the attachment. These are highlighted in yellow.

Attachment: 1

¹ ITU-APT FOUNDATION OF INDIA (<u>https://itu-apt.org</u>)

Annex 3.5 to Working Party 5D Chairman's Report

PRELIMINARY DRAFT NEW REPORT ITU-R M.[IMT.MULTIMEDIA]

Capabilities of the terrestrial component of IMT-2020 for multimedia communications

(Question ITU-R 262/5)

(20YY)

TABLE OF CONTENTS

Page
I ugu

14

1	Scope
2	Introduction
3	Relevant ITU-R Recommendations and Reports4
4	Acronyms4
5	Trends and demands of applications for multimedia content supported by IMT-2020 technologies
6	Overview of the technical characteristics of IMT-2020 technologies for multimedia communications
6.1	3GPP 5G NR
6.2	DECT NR+10
7	Use cases11
7.1	Ultra-high-definition multimedia content11
7.2	Virtual reality (VR) panoramic video11
7.3	Augmented reality (AR)11
7.4	Entertainment live streaming
7.5	Live commerce
7.6	Smart venue
7.7	Live video production, streaming and distribution
8	Capabilities of multimedia communications supported by IMT-2020 technologies
8.1	Capabilities of multimedia communications14
8.2	Multimedia communications supported by IMT-2020 technologies23
9	Case studies
9.1	News live report

9.2	Smart Venues	. 29
9.3	Sports event live broadcast	.30
9.4	Interactive cloud gaming	.31
10	Summary	.31

1 Scope

This Report addresses the capabilities of IMT-2020 to distribute multimedia content such as video, audio, text and graphics, including support for real-time multimedia interactive applications. This report also addresses the capabilities of IMT-2020 user devices and base stations to support such multimedia communications with low latency and wider transmission bandwidth.

This new report complements Report <u>ITU-R M.2373</u> on "Audio-visual capabilities and applications supported by terrestrial IMT systems" which addresses the capabilities of IMT systems for delivering audio-visual services to the consumers and also covers some aspects of production of audio-visual content.

2 Introduction

Multimedia applications include network video, digital magazine, digital newspaper, digital radio, social media, mobile TV, digital TV, touch media, etc., that are enabled by IMT-2020 technologies. Beyond the traditional media service, the new media application not only supports accurate delivery of content, but also supports real-time interaction and real-time uploading of user-generated content. The users can be both consumers and producers of new media content.

These applications for multimedia content include but are not limited to:

- audio-visual applications,
- network video applications,
- digital online magazine applications,
- digital online newspaper applications,
- internet radio applications,
- social media applications,
- mobile internet TV applications,
- touch media applications,
- online information distribution applications,
- on-demand video applications,
- imaging and audio distribution applications,
- content dissemination applications,
- file delivery application,
- real time uploading of multimedia content,
- electronic classroom presentation technology,
- full motion video conferencing.

This report covers the application of IMT technology to the specific applications mentioned above. For details of applications of the Broadcasting service for multimedia, please refer to the list of ITU-R Recommendations and ITU-R Reports in section 3 below.

3 **Relevant ITU-R Recommendations and Reports**

Recommendation ITU-R BT.1833 – Broadcasting of multimedia and data applications for mobile reception by handheld receivers

Recommendation ITU-R BT.2016 – Error-correction, data framing, modulation and emission methods for terrestrial multimedia broadcasting for mobile reception using handheld receivers in VHF/UHF bands

Recommendation ITU-R M.2083 – Framework and overall objectives of the future development of IMT for 2020 and beyond

Recommendation ITU-R M.2150 – Detailed specifications of the terrestrial radio interfaces of International Mobile Telecommunications-2020 (IMT-2020)

Report ITU-R BT.2049 – Broadcasting of multimedia and data applications for mobile reception

Report ITU-R BT.2295 – Digital terrestrial broadcasting systems

Report ITU-R M.2373 – Audio-visual capabilities and applications supported by terrestrial IMT systems

4 Acronyms

	•
2K Resolution	Content having 2 000 pixels in horizontal resolution
4K Resolution	3 840 wide x 2 160 tall pixels = 8.30 megapixels
8K Resolution	7 680 wide x 4 320 tall pixels = 33.20 megapixels
AMF	Access and Mobility Management Function
AR	Augmented Reality
CA	Carrier Aggregation
CLI	Cross-Link Interference
CLI-RSSI	CLI Received Signal Strength Indicator
CN	Core Network
CSI	Channel Status Information
DC	Dual Connectivity
DSS	Dynamic Spectrum Sharing
E1	Interface between gNB-CU-CP and gNB-CU-UP
eMBB	Enhanced Mobile Broadband
eNB	Evolved Node B – Radio Base Station used in 4G LTE
F1	Interface Interface is between gNB-CU and gNB-DU
FDD TDD CA	Carrier Aggregation between TDD and FDD bands
FR2	Frequency Band above 24.250 GHz
GBR	Guaranteed Bit Rate
gNB	Next Generation Node B – Radio Base Station used in 5G NR
GPU	Graphic Processing Unit

- 5 -5D/1668 (Annex 3.5)-E

HARQ	Hybrid Automatic Repeat Request
HD	1 020 tall \times 1 920 wide pixels = 2 million pixels
HPHT	High Power High Tower
HSDPA	High Speed Downlink Packet Access
HSUPA	High Speed Uplink Packet Access
IoT	Internet of Things
IPTV	Internet Protocol Television
L1/L2	Control Signaling
LAN-VN	Local Area Network-Virtual Network
LTE	Long Term Evolution
MBMS	Multimedia Broadcast Multicast Service
MCG	Master Call Group
MIMO	Multiple Input and Multiple Output
mMTC	Massive Machine Type Communications
MPMT	Medium Power Medium Tower
MR	Mixed Reality $-$ VR $+$ AR
MR-DC	Multi-Radio Dual Connectivity
ACK/NACK	Used in Data Transmission
NG-RAN	New Generation Radio Access Network
NPN	Non-Public Network (Private Network)
NR	New Radio O-FDMA Orthogonal Frequency Division Multiple Access
PAPR	Peak to Average Power Ratio
PCF	Policy Control Function
PDSCH	Physical Downlink Shared Channel
PDU	Protocol Data Unit
PLMN	Public Land Mobile Network
PRACH	Preamble Random Access Channel
PTM	Point to Multipoint
PTP	Point to Point
PUCCH	Physical Uplink Control Channel
QFI	QoS Flow ID
QR	Code A type of Bar Code
RAN	Radio Access Network
RIM	Remote Interference Management in 5G
RIT	Radio Interface Technology

- 6 -
5D/1668 (Annex 3.5)-E

RRC	Radio Resource Control
SC-FDMA	Single Carrier- Frequency Division Multiple Access
SFN	Single Frequency Network
SMF	Session Management Function
SRIT	Set of Radio Interface Technology
SRS	Sounding Reference Signal
SRS-RSRP	SRS-Reference Signal Received Power
TRP	Transmission and Reception Points
UE	User Equipment
UHD 5K	Resolution 5 000 pixels wide
UMTS	Universal Mobile Telecommunications Service
UPF	Use Plane Function
URLLC	Ultra-Reliable Low Latency Communications
UTRAN	UMTS Terrestrial Radio Access Network
V2X	Vehicle to Everything
VR	Virtual Reality
XRM	External Resource Management

5 Trends and demands of applications for multimedia content supported by IMT-2020 technologies

IMT-2020 is expected to revolutionize the mobile experience with much faster, always-on, always connected, and responsive mobile Internet.

By end of 2025 video traffic is estimated to account for 69 percent of all mobile data traffic, a share that is forecast to increase to 79 percent by 2027² Multimedia is an effective method of communicating information because it enriches presentations, retains the audience's attention, and allows multiple and flexible interaction. And it² is also estimated that the global media market over cellular networks will go up to \$420 bn in 2028 (\$124 bn in the US), a CAGR of 9.8% over 10 years. Consumer² spend for video, music, and games on mobile will nearly double by 2028 to reach almost \$150 bn globally (\$29 bn in the US). Many Multimedia applications are already in use and many more such applications are being developed quite rapidly. The trend to develop latest

content/uploads/sites/11/2018/10/ovum%E2%80%93intel%E2%80%935g%E2%80%93ebook.pdf

² Source: Ericsson Mobility Report, June 2022.

https://www.ericsson.com/49d3a0/assets/local/reports-papers/mobility-report/documents/2022/ericsson-mobility-report-june-2022.pdf.

For additional information refer to the interactive graph at <u>Ericsson Mobility Visualizer - Mobility</u> <u>Report - Ericsson</u>.

Source: OVUM "How 5G will transform the business of media and entertainment", October, 2018 https://newsroom.intel.com/wp-

multimedia application towards XR (eXtended Reality), high-definition video, real-time multimedia interaction and real-time uploading of self-produced content.

Virtual reality (VR) being capable of developing immersive and interactive video content, now becoming a key technology to upgrade or even subvert traditional media industries. Its application fields include video games, event live streaming, video entertainment, healthcare, real estate, retail, education, engineering and public safety.

The augmented reality (AR) (enhanced version of VR) applications are also being developed very rapidly and to create a new way for people to connect with media through virtual items, virtual characters, and augmented contextual information. The hardware and the ecosystem are gradually maturing.

Mixed reality also referred as hybrid reality is the merging of AR and VR world to produce new environments and visualizations where physical and digital objects can co-exist and interact in real time.

XR is a newly added term and defined as an umbrella for all the three realities namely VR, AR and MR. XR refers to all real-and-virtual combined environments and human-machine interactions generated by computer technology and wearables.

With the rapid development of IMT-2020 and AR, new businesses including B2B and B2C based on mobile applications have become the important exploration areas. The main serving enterpriselevel users are automobiles, machinery manufacturing, real estate, TV stations, publishing houses and exhibition halls. From the perspective of application scenarios, commodity advertising and early childhood education may be the first popular scenarios at present.

Video distribution supported by IMT-2020 is evolving to higher resolution e.g. 4K resolution, whether it is on-demand, live streaming, or video surveillance, etc., thereby improving user experience on information transmission and image recognition.

The interactive user experience - selection of loading videos, switching channels and other viewing operations through the user device while receiving multimedia content - is another developing trend. Users expect operations to complete virtually instantaneously. In order to meet user expectations, initial video loading should be completed within 1s and the channel switching within 500 ms³.

Considering the complex visual object perception and processing of VR, it is a trend to move the processing functions to the cloud, then the video streams and action commands are delivered via IMT-2020 wireless technologies to VR terminal. Cloud VR utilizes the powerful processing power of the cloud to reduce the weight and size of the VR terminal and improve user experience, which helps to reduce the complexity and cost of the terminal and greatly help to promote scale development. The cloud VR application scenarios require a large user experienced data rate of around 100 Mbit/s to ensure a high-definition video experience (> 2K), and a network delay of $5 \sim 10 \text{ ms}$ (MTP delay of 20 ms) to eliminate user motion sickness effects.

With the continuous development of mobile augmented reality, users have increasingly higher requirements for augmented reality application experience: smooth presentation, real-time interaction, and persistent operation, which pose challenges to the computing power and media processing capabilities of mobile terminal. Efficient invoking the hardware capabilities of mobile

³ <u>https://www-file.huawei.com/-</u>

[/]media/CORPORATE/PDF/white%20paper/White_Paper_on_the_Experience_driven_4K_Bearer_ Network.pdf, Tables 2-3 and 2-4.

terminals, quick identification and capture of augmented reality targets in different business execution environments, and superimposing the smooth display of augmented reality content of various media types in real time, all greatly affect the user experience. Cloud AR can balance the computing power demand and cost. In the future, with the help of the Cloud AR platform, the great computing power required by rendering videos or pictures can be provided with lower cost. It is predicted that the Cloud AR is future direction of AR development, which provides increased computing power while reducing terminal costs. At the same time, the flexible deployment method of the Cloud AR platform will help to achieve low network delay which is very important for AR applications.

Further, co-working with edge cloud services, IMT-2020 can connect the user to a high-definition virtual world on their mobile device. Edge Cloud is an architecture that is used to decentralise (processing) power to the edges (clients/devices) of the networks. Traditionally the computing power of servers are used to perform tasks such as data minimisation or to create advanced distributed systems. Within the cloud model, such intelligent tasks can be performed by servers, so that a normal mobile with less or almost no computing power can be used.

Live events with High Definition (HD) and Ultra-High Definition (UHD) can be streamed via an IMT-2020 radio network with higher throughput. HD and UHD TV content can be accessed on mobile devices without any interruptions through IMT-2020 higher user experienced data rate. The entertainment industry will hugely benefit from IMT-2020 wireless networks, which are expected to provide high resolution, and high dynamic range video streaming without interruption. Cloud AR and Cloud VR with HD or UHD video can be supported with higher user experienced data rate and low latency. HD virtual reality games are becoming popular while IMT-2020 network can offer a better real-time interactive gaming experience. It is expected that with the support of IMT-2020 technologies, an amazing virtual experience will bring to the users and the above mentioned multimedia services will become the basic services in the future mobile Internet.

6 Overview of the technical characteristics of IMT-2020 technologies for multimedia communications

6.1 3GPP 5G NR

Three primary 5G NR use cases defined by 3GPP are:

- Enhanced Mobile Broadband (eMBB): data-driven use cases requiring high data rates across a wide coverage area.
- Ultra-reliable Low Latency Communications (URLLC): strict requirements on latency and reliability for mission critical communications, such as remote surgery, autonomous vehicles or the Tactile Internet.
- Massive Machine Type Communication (mMTC): need to support a very large number of devices in a small area, which may only send data sporadically, such as Internet of Things (IoT) use cases.

In 2017-18, 3GPP, via Release-15, for the first time introduced 5G specification (5G Phase-I) in three phases. The first phase focused primarily on mobile broadband for non-standalone (NSA) 5G architecture. The second phase introduced for standalone (SA) 5G architecture. The third phase introduced an architecture for migration from 4G to 5G. Its primary focus was enabling enhanced Broadband (eMBB), as offer very high uplink throughput, lower latency and/or higher capacity to support real-time multi-media information/interaction and real-time uploading of self-media content.

In 2019-20, 3GPP, via Release-16 (5G Phase-II), further improved the features of 5G (Release 15) and incorporated Dynamic Spectrum Sharing (DSS), Network Slicing and other features designed for private 5G Network. These developments marked a new focus on enterprise and business centric capabilities. Other capabilities included New Radio (NR) based access to unlicensed spectrum (NR-U) and satellite access. The completion of Release 16 occurred in two stages, Stage-1 in late 2019 for the physical layer aspects and Stage - 2 in late 2020 for the higher layer aspects.

In 2019-20, 3GPP Release 17 further improved the features of the 5G network including DSS and private 5G network capabilities and introduced significant enhancements in the Next-Generation Radio Access Network (NG-RAN) and 5G Core Network (5GCN).

Interference coordination is one of key characteristics of IMT-2020 technologies. If non-synchronized frame structures are deployed for public and non-public networks, interference between gNB and gNB or UE and UE need to be avoided as it would cause big impacts on the overall system performance. 3GPP Release-16 extends interference mitigation techniques to certain operational environments, such as those in non-synchronized TDD networks.

Carrier Aggregation (CA) and Dual Connectivity (DC) are the straightforward and effective ways to boost uplink throughput by directly increasing the uplink frequency resources and possibly cell capacity. With CA technology, a UE can receive or transmit on one or multiple contiguous/non-contiguous component carriers. The CA was firstly introduced in LTE Release 10. CA in IMT-2020 was specified from 3GPP Release 15 and support maximum 16 carriers with 400 MHz each, thus it can support up to 6.4 GHz bandwidth. It is continuously developed in 3GPP Release 16, Release 17 and so on to support more flexible schedule and interference mitigation. The Multi-Radio Dual connectivity (MR-DC) was introduced in 3GPP Release 15. It is comprised of LTE+NR DC, NR+NR DC, NR+LTE DC etc. The RATs utilized in the first node and second node can be different. It is also continuously enhanced in 3GPP Release 15, Release 17 and so on to improve radio resource efficiency and reduce latency.

UL MIMO is another technology to improve up link data rate. It was specified for IMT-2020 from 3GPP Release 15. It introduces CSI feedback and reference signalling design which are more flexible than LTE. With high resolution codebook and beam forming characteristics, the data throughput and capacity of IMT-2020 system are expected to be met. In 3GPP Release 16, Release 17, it continuously developed CSI feedback, Coordinated Multiple Points (CoMP), beam management, power control, PAPR reference etc. to offer higher radio resource efficiency and communication performance for new media e.g. XR applications.

The Multicast Broadcast Services (MBS), including "LTE-based 5GBroadcast"⁴ and NR MBS, capabilities of 3GPP 5G-SRIT and 5G RIT MBS enables high-quality multimedia applications within a wide range of services and platforms providing a significant improvement in performance, quality of service and user experience. IMT-2020 specification supporting Multicast and Broadcast services has been developed step-by-step.

IMT-2020 technologies support group transmission for Multicast and Broadcast Services (MBS) delivery⁵. Resource efficient delivery of multicast/broadcast services is introduced for services of

⁴ "LTE-based 5G Broadcast" is the term utilized by 3GPP during the specification development phase. It has been included one of the three radio interfaces of IMT-2020, 3GPP 5G-SRIT, and it is adopted in Recommendation ITU-R M.2150-1 (02/2022) – *Detailed specifications of the terrestrial radio interfaces of International Mobile Telecommunications-2020 (IMT-2020)*.

⁵ Sourced from 3GPP TR21.917.

Point-To-Multipoint nature, such as for example public safety and mission critical services, V2X applications, IPTV, live video, software delivery over wireless and IoT applications.

For broadcast communication service, the same service and the same specific content data are provided simultaneously to all UEs in a geographical area. A broadcast communication service is delivered to the UEs using a broadcast session. A UE can receive a broadcast communication service in RRC_IDLE, RRC_INACTIVE and RRC_CONNECTED state.

For multicast communication service, the same service and the same specific content data are provided simultaneously to a dedicated set of UEs. A multicast communication service is delivered to the UEs using a multicast session. A UE can receive a multicast communication service in RRC_CONNECTED state with mechanisms such as Point-To-Point (PTP) and/or Point-To-Multipoint (PTM) delivery. HARQ feedback/retransmission can be applied to both PTP and PTM transmission.

The "LTE-based 5G Broadcast" finished in 3GPP Release 16 is based on legacy LTE eMBMS that can be traced back to 3GPP Release 8&9 which lay down the foundation for cellular broadcast. "LTE-based 5G Broadcast" is able to apply 100% of radio resources of one or more radio carriers for the delivery of broadcast/multicast content-with inter-site distances of up to 200 km.

The 3GPP-developed LTE based 5G Broadcast technology is also used as the radio part of the socalled "LTE-based 5G terrestrial broadcast system", described in ETSI TS 103 720 V1.1.1. and Recommendation ITU-R BT.2016-3 as "System L".

The NR MBS in 3GPP Release 17⁶ enables such a service over a specific geographic area which further enables a more efficient and effective delivery system for real-time and streaming multicast/broadcast content. NR MBS is able to carry IP multicast as well as Ethernet multicast packets with better QoS support using dynamic delivery mode switching.

6.2 DECT NR+

Another IMT-2020 technology DECT NR+ (Digital Enhanced Cordless Telecommunication New Radio+), recently approved by ITU-Ris fully specified by ETSI and Release 1 of the technical specification has already been published (ETSI TS 103 636, Part 1 to Part 5). In addition, ETSI is also working on a harmonized standard EN 301 406-2 "Digital Enhanced Cordless Telecommunications (DECT); Harmonized Standard for access to radio spectrum; Part 2: DECT-2020 NR", which is part 2 of a multi-part deliverable covering the access to radio spectrum of the different DECT radio interfaces. The EN is currently in the ETSI approval process and is expected to be cited in the Official Journal in Q1 2024.

DECT NR+ is not targeted to replace DECT but will be a new technology standard Following parameter of DECT NR+ for the URLLC usage scenario are important for meeting multimedia requirements (ETSI TR 103 810).

Parameter	Direction	Value	Unit
User plane latency	Uplink and Downlink	<mark>0.11 - 0.96</mark>	<mark>ms</mark>
Control plane latency	Not applicable	<mark>2.1 16.83</mark>	<mark>ms</mark>
Reliability	<mark>Uplink or Downlink</mark>	<mark>> 99.999</mark>	<mark>%</mark>

Table 1: Technical performance of DECT NR+

⁶ Refer the 3GPP TR21.917 clause 6.3.4.

Maximum bandwidth	Not applicable	<mark>221.184</mark>	<mark>MHz</mark>
<mark>Scalability</mark>	Not applicable	<mark>1.728 - 221.184</mark>	<mark>MHz</mark>

7 Use cases

This section provides examples of recently emerged multimedia use cases and reflects the current trend of moving towards a more interactive and dynamic environments.

7.1 Ultra-high-definition multimedia content

The visual and audio applications are being developed towards Ultra-high-definition (UHD) multimedia content. For instance, in the concert, the 4K-based multi-screen and multi-camera can capture extremely vivid details such as subtle "micro-expression" changes, bringing a strong visual impact to the audience. Meanwhile, the HDR and wide colour gamut technology can perfectly present the effect of live lighting and stage design to the audience. In band performances, multicamera allows the audience to follow their favourite players. For large concerts (such as orchestras), multi-channel audio transmission with Hi-Fi music quality with inaudible noise and distortion, and a flat frequency response within the human hearing range, can allow listeners to experience a truly immersive experience.

Huge user experienced data rate and the low transmission latency are typical transmission objectives from above scenarios. With the communication capability of IMT-2020, the above scenarios can be supported.

7.2 Virtual reality (VR) panoramic video

Virtual Reality panoramic video is developed and extended on the technology of 720-degree or 360-degree panorama. It converts static panoramic pictures into dynamic video images, and can be viewed at any angle of 360 degrees from left to right, up and down, so that we have a truly immersive feeling. VR Panoramic video is not a single static panoramic picture, but has depth of field, dynamic image and sound, etc., and also has sound and picture alignment, sound and picture synchronization. It can be applied to multiple multimedia applications, e.g. network video applications, digital online magazine applications, digital online newspaper applications, social media applications, on-demand video applications, mobile internet TV applications etc. Compared with the traditional 720-degree panorama, VR panoramic video has a huge leap in quality, quantity, form and content. But it requires very higher user experienced data rate, lower latency and higher availability and multiple information sources synchronization capabilities. With the support of IMT-2020 capable of high data rate and ultra-reliability and low latency, the VR panoramic video can be delivered to the user.

7.3 Augmented reality (AR)

Augmented reality (AR) allows the virtual world on the screen to be combined and interacted with the real-world scene through the calculation of the position and angle of the camera image and the addition of image analysis technology. AR can be utilized in multiple scenarios. For example, AR can help visualize construction projects. Computer-generated structural images can be superimposed onto a real partial view of the property prior to constructing a physical building on top of it. AR systems have been used as collaborative tools for design and planning in a build environment. For example, AR can be used to create augmented reality maps, buildings and data sources projected onto a desktop for collaborative viewing by built environment professionals. In educational settings, AR has been used to supplement standard curriculum. Text, graphics, video

and audio can be overlaid into the student's real-time environment. Textbooks, flashcards, and other educational reading materials may contain embedded "markers" or triggers that, when scanned by AR devices, provide supplemental information to students presented in multimedia formats.

The combination display of virtual and real objects, motion tracking, network communication, fusion rendering and human-computer interaction are the key elements of AR which require very high use experienced data rate, higher accuracy positioning and tracing, and real-time interaction between human and computer. With the capabilities of IMT-2020 especially for uplink higher throughput capability, the AR is expected to be developed step by step.

7.4 Entertainment live streaming

Live streaming which is an important information distribution application has gradually become a very popular form of entertainment. The popularity of live streaming is no less than the media that traditional entertainment platforms rely on (such as TV or Radio). For instance, top tier live streamers can attract millions of viewers when they are streaming. However, unlike those produced programs, there is no fixed schedule for live streaming, which means the live streamers could start their streaming anytime, anywhere. This characteristic will trigger an explosive surge in online viewing within a geographic area or within a period of time, which brings great technical challenges to the network transmission of the streaming. Moreover, the way of live streaming is evolving as well. For instance, the streaming video is evolving to ultra-high definition and immersive, which requires the higher video bit rate and stricter latency tolerance.

IMT-2020 network empowered streaming can be benefit to solve above issues. By integrating communication and information distribution storage and processing supported by IMT-2020, the mass access and the explosive traffic peak challenges caused by super-influencer across the country or region can be effectively addressed.

7.5 Live commerce

The term "Live commerce" refers to the use of live webcast technology to carry out new sales methods such as online display of merchandise, customer Q&A, and shopping guide sales through the Internet platform. Manufacturers usually use professional platforms to build their live streaming booth to sell a variety of products by streamers. Unlike traditional home shopping channel, live commerce could interact real-time with customers. Considering the characteristics of live commerce, the following interactions are needed:

- The live streamers can receive the feedback from the viewer and response further with supplemental product introduction, which help enhance the viewer's confidence in the quality of the product.
- The live streamers can use the virtual reality (VR) / augmented reality (AR) to provide product trials and related functions, allowing customers to "experience" the product.
- The viewer can order and pay for the products online in real-time through the payment methods of live streamers (Payment link/QR code, etc.).
- The live streamers can conduct online prize draws or mini-games, to improve viewer stickiness.

Therefore, for live commerce application, distributing low-latency and high-quality live video to a large number of viewers, and real-time interaction between the live streamer and the viewers are the key objectives for IMT-2020.

7.6 Smart venue

With the construction of IMT-2020 networks, the concept of "Smart Venue" that empower large venues with IMT-2020 networks and provide a new viewing experience to live events or activities

has emerged. In top events, new viewing methods such as multi-view point and free-view point have been introduced, bringing a different experience to the website viewer and spectator. Taking "multi-view point" as an example, by deploying multiple cameras on stadium, content service providers can allow the viewer watch the game with different viewing angles from the regular streaming.

In addition to providing a diverse experience for viewers, another core concept of "Smart Venue" is to provide spectator with live services. For instance, during a game or event, the spectator can choose a different perspective from the current seat to watch the game via the client. During the intermission, the spectator could watch the replay of highlights or behind-the-scenes.

Generally, a large event often has tens or even hundreds of thousands of spectators, and millions of viewers. Such a dense population leads to huge challenges for communication network coverage, communication latency and connectivity capacity.

IMT-2020 empowered streaming could provide multimedia distribution service to users in or out of the certain area e.g. the venue, which will effectively overcome the wired network transmission bandwidth limitation and greatly reduce the cost of venue network construction. Meanwhile, the venue owners or event organizers will be able to leverage 5G empowered streaming to bring viewers (or spectators) an unprecedented experience.

7.7 Live video production, streaming and distribution

The production of traditional large sports events is mainly implemented by sportscast production systems and equipment such as sportscast vans with wired communication. However, ultra-high-definition sports video has gradually become the mainstream, and the content of live events is becoming more and more complex with a large number of cameras, microphones and sensors working at the same time, a large amount of data needs to be transmitted back to the sportscast production control centre. In addition, as events become more globalized, sportscast resources need to be quickly collected across the world in a very short period of time. The event live streaming production is gradually developing towards centralized, remote and lightweight.

IMT-2020 has the characteristics of low latency, high bandwidth and wide coverage, which could improve the sportscast production capability, optimize resource allocation, and efficiently distribute real-time content from multi-viewing point, leaderboard and other content generated on the event to viewers. In particular, the high bandwidth and low latency transmission of IMT-2020 enables the data of some independent camera (such as Video Adjudication System) to be collected and delivered on time. For cameras with special mobility (such as Spidercam), IMT-2020 will expand its moving range, help it working more flexible and efficient.

Specifically, the following aspects could be enhanced:

- Short-distance transmission from multi-camera to live streaming system: multiple cameras (such as Steadicam, panoramic VR, AR hardware, etc.) are simultaneously transmitted to the sportscast van through IMT-2020 for live streaming production. The Single-channel high-definition video (mainly $1920 \times 1080i$) is encoded with a bit rate of 15 Mbit/s, and the uplink bit rate of the transmission channel will not be lower than this value. The relative delay between multiple cameras (same signal types) is less than 40 ms to meet synchronization requirements.
- Lightweight deployment of live streaming equipment: IMT-2020 has extremely low delay for both DL&UL transmission, multiple signals (such as video/audio/data) can be transmitted to the remote production center through wireless transmission. Therefore, the transmission of control, tally, and voice signals can ensure extremely low delay in both uplink and downlink. These devices only need to be connected to IMT-2020

network, whereas in the past they required thousands of cables to connect to each other, which avoids the complicated deployment process.

Multi-content streaming services: Through high speed streaming, various content (such as multi-camera, VR, Box score, etc.) can be distributed to a large number of spectators in real time, allowing them to fully grasp the details of the event and provide a better viewing experience.

7.8 Live Audio Production

In Programme Making and Special Events (PMSE) wireless applications are during the live professional audio/video productions such as concerts, musicals, or other staging of entertainment, meetings, conferences, cultural and education activities, trade fairs, sport, religious, and other public or private events.

Audio PMSE applications require mission critical low latency audio transmission in an indoor or outdoor service area. During an event, performers rely on receiving a personalized audio mix of the event streamed back to their in-ear monitoring device. An in-ear monitoring device receives an audio stream ranging between 200 kbit/s. The maximum end-to-end latency tolerated by most of professional performers between their wireless microphone (audio input) and their in-ear monitoring device (audio output) is 4 ms.

Currently the audio PMSE industry utilize analogue and digital technologies employing narrowband modulation techniques with each link typically occupying a bandwidth of 200 kHz⁷. IMT 2020 technology capabilities for reliability, latency, synchronicity and spectral efficiency must align with the requirements of professional audio transmission to cater for increased demand in audio PMSE applications.

8 Capabilities of multimedia communications supported by IMT-2020 technologies

8.1 Capabilities of multimedia communications

Capabilities of multimedia communications are introduced based on the 3GPP TS 22.261⁸, 3GPP TS 22.263⁹ and 3GPP TR38.913¹⁰. They includes not only broadcast, multicast communication methods, but also unicast to flexible and efficient support UHD, AR/VR, live streaming production and distribution, live commerce, and smart venue etc.

⁷ <u>III</u> ETSI EN 300 422-1, V2.2.1: "Wireless Microphones; Audio PMSE up to 3 GHz, Part 1: Audio PMSE Equipment up to 3 GHz; Harmonised Standard for access to radio spectrum"

⁸ 3GPP TS22.261 V18.6.1 (2022-06) - Service requirements for the 5G system.

⁹ 3GPP TS22.263 V17.4.0 (2021-06) - Service requirements for video, imaging and audio for professional applications (VIAPA).

¹⁰ 3GPP TR38.913 V17.0.0 (2022-04) - Study on scenarios and requirements for next generation access technologies.

8.1.1 IMT-2020 capabilities for real-time multimedia interaction and media content uploading

Interactive multimedia allows the user to control, combine and manipulate a variety of media types, such as text, computer graphics audio and video materials, animation and virtual reality.

To enable interactive task completion during voice conversation, IMT-2020 is capable of supporting low-delay speech coding for interactive conversational services (refer 3GPP TS22.261, 100 ms, one-way mouth-to-ear).

Table 1¹¹ gives a capability example of IMT-2020 to support high data rate and traffic density scenario for an interactive audio and video application in indoor hotspot.

TABLE 1

High data rate and traffic density scenario of IMT-2020 for an interactive audio and video application

Scenario	Experienced data rate (DL)	Experienced data rate (UL)	Area traffic capacity (DL)	Area traffic capacity (UL)	Overall user density	Activity factor	UE speed	Coverage
Indoor hotspot	1 Gbit/s	500 Mbit/s	15 Tbit/s/km ²	2 Tbit/s/km ²	250 000/km ²	Note 2		Office and residential (Note 1)

NOTE 1: A certain traffic mix is assumed; only some users use services that require the highest data rates.

Table 2¹² provides capabilities of IMT-2020 to support AR/VR high data rate and low latency usages.

¹¹ 3GPP TS 22.261 Table 7.1-1.

¹² 3GPP TS22.261 Table 7.6.1-1.

- 16 -5D/1668 (Annex 3.5)-E

TABLE 2

Capabilities of IMT-2020 to support AR/VR high data rate and low latency service

Use Cases	Char	acteristic parameter (KPI)	Influence quantity			
	Max allowed end-to-end latency	Service bit rate: user- experienced data rate	Reliability	# of UEs	UE Speed	Service Area (Note 2)
Cloud/Edge/Split Rendering (Note 1)	5 ms (i.e. UL+DL between UE and the interface to data network) (Note 4)	0,1 to [1] Gbit/s supporting visual content (e.g. VR based or high definition video) with 4K, 8K resolution and up to 120 frames per second content.	99,99 % in uplink and 99,9 % in downlink (Note 4)	-	Stationary or Pedestrian	Countrywide
Gaming or interactive data exchanging (Note 3)	10 ms (Note 4)	0,1 to [1] Gbit/s supporting visual content (e.g. VR based or high definition video) with 4K, 8K resolution and up to 120 frames per second content.	99.99% (Note 4)	≤[10]	Stationary or Pedestrian	20 m x 10 m; in one vehicle (up to 120 km/h) and in one train (up to 500 km/h)
Consumption of VR content via tethered VR headset (Note 6)	[5 to 10] ms (Note 5)	0.1 to [10] Gbit/s (Note 5)	[99.99%]	-	Stationary or Pedestrian	_

NOTE 1: Unless otherwise specified, all communication via wireless link is between UEs and network node (UE to network node and/or network node to UE) rather than direct wireless links (UE to UE).

NOTE 2: Length x width (x height).

NOTE 3: Communication includes direct wireless links (UE to UE).

NOTE 4: Latency and reliability KPIs can vary based on specific use case/architecture, e.g., for cloud/edge/split rendering, and can be represented by a range of values.

NOTE 5: The decoding capability in the VR headset and the encoding/decoding complexity/time of the stream will set the required bit rate and latency over the direct wireless link between the tethered VR headset and its connected UE, bit rate from 100 Mbit/s to [10] Gbit/s and latency from 5 ms to 10 ms.

NOTE 6: The performance requirement is valid for the direct wireless link between the tethered VR headset and its connected UE.

NOTE7: The [] means that it is better to be supported but not has to.

The capabilities of IMT-2020 to support professional low-latency periodic deterministic audio transport service which are used for Video and audio production applications are listed in Table 3¹³.

¹³ 3GPP TS22.263 Table 6.2.1-1.

- 17 -5D/1668 (Annex 3.5)-E

TABLE 3

Capabilities of IMT-2020 to support professional low-latency periodic deterministic audio transport service

Profile	# of active UEs	UE Speed	Service Area	E2E latency (Note 1)	Transfer interval (Note 1)	Packet error rate (Note 2, Note 3)	Data rate UL	Data rate DL
Music Festival	200	10 km/h	500 m × 500 m	750 µs	250 µs	10-6	500 kbit/s	-
	100	10 km/h	500 m × 500 m	750 µs	250 µs	10-6	-	1 Mbit/s
Musical	30	50 km/h	$50 \text{ m} \times 50 \text{ m}$	750 µs	250 µs	10-6	500 kbit/s	-
	20	50 km/h	$50\ m\times 50\ m$	750 µs	250 µs	10-6	-	1 Mbit/s
	10	-	$50\ m\times 50\ m$	750 µs	250 µs	10-6	-	500 kbit/s
Semi-	10	5 km/h	$5 \text{ m} \times 5 \text{ m}$	750 µs	250 µs	10-6	100 kbit/s	-
professional	10	5 km/h	$5 \text{ m} \times 5 \text{ m}$	750 µs	250 µs	10-6	-	200 kbit/s
	2	-	$5 \text{ m} \times 5 \text{ m}$	750 µs	250 µs	10-6	-	100 kbit/s
AV production	20	5 km/h	$\begin{array}{c} 30 \text{ m} \times 30 \\ \text{m} \end{array}$	750 µs	250 µs	10-6	1.5 Mbit/s	-
	10	5 km/h	$\begin{array}{c} 30 \text{ m} \times 30 \\ \text{m} \end{array}$	750 µs	250 µs	10-6	-	3 Mbit/s
Audio Studio	30	-	$\begin{array}{c} 10 \text{ m} \times 10 \\ \text{m} \end{array}$	750 µs	250 µs	10-6	5 Mbit/s	-
	10	5 km/h	$\begin{array}{c} 10 \text{ m} \times 10 \\ \text{m} \end{array}$	750 µs	250 µs	10-6	-	1 Mbit/s

NOTE 1: Transfer interval refers to periodicity of the packet transfers. It has to be constant during the whole operation. The value given in the table is a typical one, however other transfer intervals are possible as long as the end-to-end latency is $\leq (1 \text{ ms} - \text{Transfer interval})$.

NOTE 2: Packet error rate is related to a packet size of (transfer interval × data rate). Packets that do not conform with the end-to-end latency are also accounted as error.

NOTE 3: The given requirement for a packet error rate assumes a uniform error distribution. The requirement for packet error rate is stricter if packet errors occur in bursts.

The capabilities of IMT-2020 to support low latency Video applications are listed in Table 4¹⁴.

¹⁴ 3GPP TS22.263 Table 6.2.1-3.

- 18 -5D/1668 (Annex 3.5)-E

TABLE 4

Capabilities of IMT-2020 to support low latency video

	UEs	UE Speed	Service Area	E2E latency	Packet error rate (Note 1)	Data rate UL	Data rate DL
Uncompressed UHD video	1	0 km/h	1 km ²	400 ms	10 ⁻¹⁰ UL 10 ⁻⁷ DL	12 Gbit/s	20 Mbit/s
Uncompressed HD video	1	0 km/h	1 km ²	400 ms	10 ⁻⁹ UL 10 ⁻⁷ DL	3 .2 Gbit/s	20 Mbit/s
Mezzanine compression UHD video	5	0 km/h	1 000 m ²	1 s	10 ⁻⁹ UL 10 ⁻⁷ DL	3 Gbit/s	20 Mbit/s
Mezzanine compression HD video	5	0 km/h	1 000 m ²	1 s	10 ⁻⁹ UL 10 ⁻⁷ DL	1 Gbit/s	20 Mbit/s
Tier one events UHD	5	0 km/h	1 000 m ²	1 s	10 ⁻⁹ UL 10 ⁻⁷ DL	500 Mbit/s	20 Mbit/s
Tier one events HD	5	0 km/h	1 000 m ²	1 s	10 ⁻⁸ UL 10 ⁻⁷ DL	200 Mbit/s	20 Mbit/s
Tier two events UHD	5	7 km/h	1 000 m ²	1 s	10 ⁻⁸ UL 10 ⁻⁷ DL	100 Mbit/s	20 Mbit/s
Tier two events HD	5	7 km/h	1 000 m ²	1 s	10 ⁻⁸ UL 10 ⁻⁷ DL	80 Mbit/s	20 Mbit/s
Tier three events UHD (Note 2)	5	200 km/h	1 000 m ²	1 s	10 ⁻⁷ UL 10 ⁻⁷ DL	20 Mbit/s	10 Mbit/s
Tier three events HD (Note 2)	5	200 km/h	1 000 m ²	1 s	10 ⁻⁷ UL 10 ⁻⁷ DL	10 Mbit/s	10 Mbit/s
Remote OB	5	7 km/h	1 000 m ²	6 ms	10 ⁻⁸ UL 10 ⁻⁷ DL	200 Mbit/s	20 Mbit/s

The capabilities of IMT-2020 to support UHD streaming via airborne base stations are listed in Table 5^{15} .

¹⁵ TS22.261 Table 6.2.1-4.

- 19 -5D/1668 (Annex 3.5)-E

TABLE 5

Capabilities of IMT-2020 to support UHD via airborne base stations

Profile	# of active UEs	UE Speed	Service Area	E2E latency	Packet error rate (Note 1)	Data rate UL	Data rate DL
NPN ground to air UHD up Link	10	500 km/h	700 km ² x 6 000 m (Note 2)	40 ms	10 ⁻⁸ UL 10 ⁻⁷ DL	100 Mbit/s	20 Mbit/s
NPN ground to air HD up link	10	500 km/h	700 km ² x 6 000 m (Note 2)	40 ms	10 ⁻⁸ UL 10 ⁻⁷ DL	80 Mbit/s	20 Mbit/s
NPN air to ground UHD down Link	2	500 km/h	700 km ² x 6 000 m (Note 2)	40 ms	10 ⁻⁷ UL 10 ⁻⁸ DL	20 Mbit/s	100 Mbit/s
NPN air to ground HD down link	2	500 km/h	700 km ² x 6 000 m (Note 2)	40 ms	10 ⁻⁷ UL 10 ⁻⁸ DL	20 Mbit/s	80 Mbit/s
NPN radio Camera UHD	10	200 km/h	1 km ²	3 ms	10 ⁻⁸ UL 10 ⁻⁷ DL	100 Mbit/s	20 Mbit/s
NPN radio camera HD	10	200 km/h	1 km ²	3 ms	10 ⁻⁸ UL 10 ⁻⁷ DL	80 Mbit/s	20 Mbit/s

NOTE 1: Packets that do not conform with the end-to-end latency are also accounted as error. The packet error rate requirement is calculated considering 1 500 B packets, and 1 packet error per hour is $10^{-5}/(3*x)$, where x is the data rate in Mbit/s. NOTE 2: 6 000 m = height but in a cone formation (i.e. ground coverage with a circle of diameter 30 km).

Further, with development of immersive multiple modal integrated with AR/VR, IMT-2020 are capable of supporting the communication KPIs listed in Table 7.1-6¹⁶, which will greatly enrich and enhance user experience of the immersive multimedia service.

¹⁶ 3GPP TS22.261 Table 7.11-1.

- 20 -5D/1668 (Annex 3.5)-E

TABLE 7.1-6

Capabilities of IMT-2020 to support immersive multimedia communication service

Use Cases	Charact	eristic parame	ter (KPI)	Influence quantity			Remarks
	Max allowed end-to-end latency	Service bit rate: user- experienced data rate	Reliability	Message size (byte)	UE Speed	Service Area	
Immersive multi- modal VR (UL: device \rightarrow application sever)	5 ms (Note 2)	16 kbit/s -2 Mbit/s (without haptic compression encoding); 0.8-200 kbit/s (with haptic compression encoding)	99.9% (without haptic compression encoding) 99.999% (with haptic compression encoding) [40]	1 DoF: 2-8 3 DoFs: 6-24 6 DoFs: 12-48 More DoFs can be supported by the haptic device	Stationary or Pedestrian	typically < 100 km ² (Note 5)	Haptic feedback
	5 ms	< 1 Mbit/s	99.99% [40]	1500	Stationary or Pedestrian	typically < 100 km ² (Note 5)	Sensing information e.g. position and view information generated by the VR glasses
Immersive multi- modal VR	10 ms (Note 1)	1-100 Mbit/s	99.9% [40]	1500	Stationary or Pedestrian	typically < 100 km ² (Note 5)	Video
(DL: application sever \rightarrow device)	10 ms	5-512 kbit/s	99.9% [40]	50	Stationary or Pedestrian	typically < 100 km ² (Note 5)	Audio
	5 ms (Note 2)	16 kbit/s - 2 Mbit/s (without haptic compression encoding); 0.8-200 kbit/s (with haptic compression encoding)	99.9% (without haptic compression encoding) 99.999% (with haptic compression encoding) [40]	1 DoF: 2-8 3 DoFs: 6-24 6 DoFs: 12-48	Stationary or Pedestrian	typically < 100 km ² (Note 5)	Haptic feedback

- 21 -
5D/1668 (Annex 3.5)-E

Use Cases	Characteristic parameter (KPI)			Influence quantity			Remarks
	Max allowed end-to-end latency	Service bit rate: user- experience d data rate	Reliability	Message size (byte)	UE Speed	Service Area	
Immersive multi- modal navigation applications Remote Site → Local Site (DL)	50 ms [39]	16 kbit/s -2 Mbit/s (without haptic compressio n encoding); 0.8-200 kbit/s (with haptic compressio n encoding)	99.999% [40]	1 DoF: 2 to 8 10 DoF: 20 to 80 100 DoF: 200 to 800	Stationary or Pedestrian	≤ 100 km ² (Note 5)	Haptic feedback
	<400 ms [39]	1-100 Mbit/s	99.999% [40]	1500	Stationary/ or Pedestrian,	$\leq 100 \text{ km}^2$ (Note 5)	Video
	<150 ms [39]	5-512 kbit/s	99.9% [40]	50	Stationary or Pedestrian	$\leq 100 \text{ km}^2$ (Note 5)	Audio
	<300 ms	600 Mbit/s	99.9% [40]	1500	Stationary or Pedestrian	$\leq 100 \text{ km}^2$ (Note 5)	VR
Immersive multi- modal navigation applications Local Site → Remote Site (UL)	<300 ms	12 kbit/s [26]	99.999% [40]	1500	Stationary or Pedestrian	$\leq 100 \text{ km}^2$ (Note 5)	Biometric / Affective
	<400 ms [39]	1-100 Mbit/s	99.999% [40]	1500	Workers: Stationary/ or Pedestrian, UAV: [30- 300 mph]	$\leq 100 \text{ km}^2$ (Note 5)	Video
	<150 ms [39]	5-512 kbit/s	99.9 % [40]	50	Stationary or Pedestrian	$\leq 100 \text{ km}^2$ (Note 5)	Audio
	<300 ms	600 Mbit/s	99.9 % [40]	1500	Stationary or Pedestrian	$\leq 100 \text{ km}^2$ (Note 5)	VR

NOTE 1: Motion-to-photon delay (the time difference between the user's motion and corresponding change of the video image on display) is less than 20 ms, and the communication latency for transferring the packets of one audio-visual media is less than 10 ms, e.g. the packets corresponding to one video/audio frame are transferred to the devices within 10 ms.

NOTE 2: According to IEEE 1918.1 [40] as for haptic feedback, the latency is less than 25 ms for accurately completing haptic operations. As rendering and hardware introduce some delay, the communication delay for haptic modality can be reasonably less than 5 ms, i.e. the packets related to one haptic feedback are transferred to the devices within 10 ms.

NOTE 3: Haptic feedback is typically haptic signal, such as force level, torque level, vibration and texture.

NOTE 4: The latency requirements are expected to be satisfied even when multimodal communication for skillset sharing is via indirect network connection (i.e., relayed by one UE to network relay).

NOTE 5: In practice, the service area depends on the actual deployment. In some cases a local approach (e.g. the application servers are hosted at the network edge) is preferred in order to satisfy the requirements of low latency and high reliability.

Beyond these above capabilities of IMT-2020, refer 3GPP TS22.263, following capabilities are developed:

- 1) IMT-2020 is capable of enabling an NPN for video, imaging and audio for professional applications.
- 2) IMT-2020 network is capable of providing a time reference information to a 3rd party application acting as a master clock with an accuracy of 1 microsecond.
- 3) IMT-2020 is capable of securely reconnecting within a short period of time (<1s) from UE starting first network reconnection attempt after the UE has detected a UE network connection loss.
- 4) IMT-2020 is capable of supporting uplink and downlink service continuity maintaining acceptable performance requirements while switching between co-located PLMN and NPN (e.g., due to mobility).
- 5) IMT-2020 is capable of supporting service continuity maintaining acceptable performance requirements: for an uplink stream while performing traffic steering, switching, and splitting among co-located PLMN(s) and NPN(s); for downlink while switching between co-located PLMN and NPN.

8.1.2 IMT-2020 capabilities for broadcast/multicast in multimedia communication

The capability examples of multimedia broadcast/multicast supported by IMT-2020 are:

- 1 to support operation of downlink only broadcast/multicast over a specific geographic area (e.g., a cell sector, a cell or a group of cells);
- 2 to support operation of a downlink only broadcast/multicast system over a wide geographic area in a spectrally efficient manner for stationary and mobile UEs;
- 3 to enable the operator to reserve 0% to 100% of radio resources of one or more radio carriers for the delivery of broadcast/multicast content;
- 4 to allow the UE to receive content via a broadcast/multicast radio carrier while a concurrent data session is ongoing over another radio carrier.
- 5 to support broadcast/multicast of UHD streaming video (e.g., 4K/8K UHD);
- 6 to allow the operator to configure and broadcast multiple quality levels (i.e., video resolutions) of broadcast/multicast content for the same user service in a stand-alone IMT-2020 based broadcast/multicast system;
- 7 to support parallel transfer of multiple quality levels (i.e., video resolutions) of broadcast/multicast content for the same user service to the same UE taking into account e.g., UE capability, radio characteristics, application information;
- 8 to support parallel transfer of multiple multicast/broadcast user services to a UE.
- 9 to be able to setup or modify a broadcast/multicast service area within [1s]¹⁷;
- 10 be able to apply QoS, priority and pre-emption to a broadcast/multicast service area;
- 11 to support downlink parallel transfer of the same content, via broadcast/multicast and/or unicast, such that all receiver group members in a given area receive the media at the same time according to user perception;
- 12 to support a mechanism to inform a media source of relevant changes in conditions in the system (e.g., capacity, failures);
- 13 to provide means for a media source to provide QoS requirement requests to the broadcast/multicast service;

¹⁷ The [] means it is better to support this KPI, but not has to.

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- 14 to provide means for the broadcast/multicast service to inform the media source of the available QoS, including modification of available QoS characteristics and availability of the broadcast/multicast service;
- 15 to support broadcast/multicast of voice, data and video group communication, allowing at least 800 concurrently operating groups per geographic area;
- 16 to support delivery of the same UE-originated data in a resource-efficient manner in terms of service bit rate to UEs distributed over a large geographical area;
- 17 to allow a UE to request a communication service to simultaneously send data to different groups of UEs at the same time;
- 18 to allow different QoS policy for each group the UE communicates with.

And refer 3GPP TR38.913, following capabilities for broadcast/multicast have been described:

- 1 to support existing Multicast/Broadcast services (e.g., download, streaming, group communication, TV, etc.);
- 2 to support dynamic adjustment of the Multicast/Broadcast area based on e.g. the user distribution or service requirements;
- 3 to support concurrent delivery of both unicast and Multicast/Broadcast services to the users;
- 4 to support efficient multiplexing with unicast transmissions in at least frequency domain and time domain;
- 5 to support static and dynamic resource allocation between Multicast/Broadcast and unicast; the new RAT is needed to in particular allow support of up to 100% of DL resources for Multicast/Broadcast (100% meaning a dedicated MBMS carrier);
- 6 to cover large geographical areas up to the size of an entire country in SFN mode with network synchronization. It is needed to also support local, regional and national broadcast areas;
- 7 to support Multicast/Broadcast services for fixed, portable and mobile UEs;
- 8 to leverage usage of RAN equipment (hard- and software) including e.g., multi-antenna capabilities (e.g., MIMO) to improve Multicast/Broadcast capacity and reliability;
- 9 to support on-demand establishment of UE to UE, multicast and broadcast private communication between member UEs of the same network;
- 10 to allow member UEs of local area virtual network to join an authorized multicast session;
- 11 to support for real time and non-real time multimedia services and applications with advanced Quality of Experience (QoE).

8.2 Multimedia communications supported by IMT-2020 technologies

The capabilities can supported by different IMT-2020 technical components e.g. radio interface, architecture enhancements. This section describes examples of these technical components.

8.2.1 Radio Access Network

The radio technical components of IMT-2020 to support multimedia communication can be found in Recommendation ITU-R M.2150. Follows are the example of them.

8.2.1.1 Flexible and dynamic resources allocation

IMT-2020 can utilize flexible and dynamic resources allocation to enable both Multicast and Broadcast in use cases of public safety and mission critical, V2X applications, transparent IPv4/IPv6 multicast delivery, IPTV, software delivery over wireless network, group communications and IoT applications, instead of broader area broadcast like services.

With this technology, RAN basic functions of broadcast/multicast for UEs in RRC_CONNECTED state are to be supported with better reliability and better service continuity. For multicast communication service, the same service and the same specific content data are provided simultaneously to a dedicated set of UEs (i.e., not all UEs in the Multicast service area are authorized to receive the data). A multicast communication service is delivered to the UEs using a multicast session. A UE can receive a multicast communication service in RRC_CONNECTED state with mechanisms such as point to point (PTP) and/or point to multi-point (PTM) delivery, with a balance between network efficiency.

HARQ feedback/retransmission can be applied to both PTP and PTM transmissions. For PTM transmission there can be two HARQ feedback schemes: UE specific feedback and NACK only feedback, depending on network decision based on level of reliability requirement or network resources. Network is able to dynamically change Multicast service delivery between PTM and PTP with service continuity for a given UE.

Basic mobility with service continuity is supported. Unicast like mobility mechanism is designed to offer the basic mobility, e.g., the Multicast session resources are to be established along with UE's mobility in the target RAN node which supports the MBS service.

For broadcast communication service, the same service and the same specific content data are provided simultaneously to all UEs in a geographical area (i.e., all UEs in the Broadcast service area are authorized to receive the data). A broadcast communication service is delivered to the UEs using a broadcast session. A UE can receive a broadcast communication service in RRC_IDLE, RRC_INACTIVE and RRC_CONNECTED state.

There are other enhancement including, UE are able to receive MBS with simultaneous operation in unicast reception; enhancement on RAN network interface like Xn, F1, and E1 interfaces; support on dynamic broadcast area etc.

8.2.1.2 UL enhancement

A) UL MIMO

While UL MIMO offers the capability for reduction in overhead and/or latency, high-speed vehicular scenarios (e.g. a UE traveling at high speed on highways) at FR2 require more aggressive reduction in latency and overhead – not only for intra-cell, but also for L1/L2 centric inter-cell mobility. This also includes reducing the occurrence of beam failure events. Besides of it, enhancements for enabling panel-specific UL beam selection were investigated and specified continuously. This offers some potential for increasing UL coverage including. Then, channels other than PDSCH can benefit from multi-TRP transmission (as well as multi-panel reception) which also includes multi-TRP for inter-cell operations. This includes some new use cases such as UL dense deployment within a macro-cell and/or heterogeneous-network-type deployment scenarios. And the SRS can be further enhanced for capacity and coverage.

B) UL Carrier Aggregation

Carrier aggregation was developed the unaligned frame boundary capability to provide more flexible beginning TX frame structure configurations among different carriers to offer higher uplink throughput and lower latency. Tx switching is to specify the dynamic switching mechanisms among two uplink carriers. Unaligned frame boundary and Tx switching can be implemented together to achieve larger uplink throughput for TDD CA operation. In one cell group, it supports PUCCH carrier switching semi-statically or dynamically. This could reduce the latency of PUCCH transmissions significantly for CA operation.

C) UL Dual connectivity (DC)

Dual connectivity is capable of Uplink power controlling i.e. limiting UE's transmission power to assure edge user's communication QoS, migration interference among the users, and reducing energy consumption of the UE. Uplink DC is also capable of earlier measurement and fast recovery from MCG failure to reduce latency. Beyond these, it is capable of indicating UE entering the third state to maintain context of the UE which will reduce configuration overhead.

D) Interference coordination

Interference coordination includes two aspects, i.e., cross-link interference (CLI) and remote interference management (RIM). For CLI, it is left to network implementation for gNB-gNB interference. The UE-UE interference coordination is capable of such as SRS-RSRP/CLI-RSSI based layer-3 CLI measurement and reporting, and network coordination mechanism for CLI with inter-gNB exchange of intended UL/DL configuration. RIM targets to migrate the interference occurring in specific weather conditions with the distance between aggressor gNB and victim gNB hundreds of kilometres. The RIM reference signal (RS) based on PRACH preamble-like RS is introduced for better interference measurement while the detailed remote interference mitigation mechanisms are left to implementation.

8.2.1.3 LTE-based 5G Broadcast

IMT-2020 can deliver audio-visual services (including free-to-air services) in single frequency network (SFN) on stand-alone infrastructure, usually in High-Power High-Tower Single Frequency Networks to support larger inter-site distance (e.g., to allow cell radii of up to 100 km).

To allow cell radii of up to 100 km, numerology enhancement with a new 0.37 kHz subcarrier spacing and CP duration ~300 μ s was introduced to support broadcast in medium power medium tower (MPMT) & HPHT. To allow reception with UE mobility up to 250 km/h, numerology enhancement with a new 2.5 kHz wider sub-carrier spacing, CP duration ~100 μ s with better Doppler resiliency. There is other related enhancement to support broader coverage for the control information like dedicated reference signals (RS) accompany each numerology, enhanced subframe structure, control channel, and less dense RS pattern with reducing overheads.

8.2.2 Architecture Enhancements

The architecture enhancement of IMT-2020 which also related with the radio part are described in the following.

8.2.2.1 Unicast enhancement to support real-time multimedia interaction and uploading

The unicast architecture follows IMT-2020 architectural principles as defined in 3GPP TS 23.501 and 3GPP TS 26.501¹⁸ to support real-time multimedia interaction and uploading. Following architecture enhancements are supported in 5GS.

¹⁸ 3GPP TS26.501 V17.3.0 (2022-09) - 5G Media Streaming (5GMS); General description and architecture

- 26 -5D/1668 (Annex 3.5)-E

5G-XR Application Provider provides a 5G-XR Aware Application on the UE to make use of a 3GPP 5G-XR client and network functions via 5G media streaming model which is introduced in 3GPP TS26.501.

The 5G QoS model supports both:

- OoS Flows that require guaranteed flow bit rate (GBR OoS Flows).
- and OoS Flows that do not require guaranteed flow bit rate (Non-GBR OoS Flows).

The OoS model also supports Reflective OoS. A OoS Flow ID (OFI) is used to identify a OoS Flow in IMT-2020. User Plane traffic assigned to the same QoS Flow within a Protocol Data Unit (PDU) Session receives the same traffic forwarding treatment (e.g. scheduling, admission threshold). The QFI may be dynamically assigned or may be equal to the 3GPP 5G QoS Identifier (5QI). A QoS Flow may either be 'GBR', 'Non-GBR' or "Delay Tolerant GBR" depending on its OoS profile and it contains OoS parameters as follows:

For each QoS Flow, the QoS profile includes the QoS parameters:

- 3GPP 5G OoS Identifier (5OI); and
- Allocation and Retention Priority (ARP).
- For each Non-GBR QoS Flow only, the QoS profile can also include the QoS parameter:
 - Reflective QoS Attribute (RQA).
 - For each GBR QoS Flow only, the QoS profile also include the QoS parameters:
 - Guaranteed Flow Bit Rate (GFBR) uplink (UL) and downlink (DL); and
 - Maximum Flow Bit Rate (MFBR) UL and DL; and
- In the case of a GBR OoS Flow only, the OoS profile can also include one or more of the QoS parameters:
 - Notification control:
 - Maximum Packet Loss Rate UL and DL.

PDU Set is introduced to optimize the delivery of XRM service in 5GS. A PDU Set is composed of one or more PDUs carrying the payload of one unit of information generated at the application level (e.g., a frame or video slice for XRM Services), which are of same importance at application layer. All PDUs in a PDU Set are needed by the application layer to use the corresponding unit of information. In some cases, the application layer can still recover parts of the information unit, when some PDUs are missing.

The following key issues are currently under investigating:

WT#1: Enhancements for supporting multi-modality service: Study whether and how to enable delivery of related tactile and multi-modal data (e.g., audio, video and haptic data related to a specific time) with an application to the user at a similar time, focusing on the need for policy control enhancements (e.g. QoS policy coordination).

WT#2: Enhancements of network exposure to support interaction between 5GS and application:

- WT#2.1: Study whether and how interaction between AF and 5GS is needed for application synchronization and QoS policy coordination among multiple UEs or between multiple QoS flows per UE.
- WT#2.2: Study exposure of 5GS QoS information (e.g., QoS capabilities) and network conditions to the Application to enable quick codec/rate adaptation help to provide desired QoE (e.g. such as assist in alleviating 5GS congestion).

WT#3: Study whether and how the following QoS and policy enhancements for XR service and media service transmission are performed:

- WT#3.1: Study the traffic characteristics of media service enabling improved network resources usage and QoE.
- WT#3.2: Enhance QoS framework to support media units granularity (e.g., video/audio frame/tile, Application Data Unit, control information), where media units consist of PDUs that have the same QoS requirements.
- WT#3.3: Support differentiated QoS handling considering different importance of media units. e.g., eligible drop packets belong to less important media units to reduce the resource wasting.
- WT#3.4: Whether and how to support uplink-downlink transmission coordination to meet RTT (Round-Trip Time) latency requirements between UE and N6 termination point at the UPF.
- WT#3.5: Potential policy enhancements to minimize the jitter, focusing on i.e. requirement provisioning from AF, extension of PCC rule.

WT#4: Study potential enhancements of Mobility and power management considering traffic pattern of media services:

- WT 4.1: Void.
- WT 4.2: Power saving enhancement e.g. support trade-off of throughput/latency/reliability considering device battery life, whether and how to enhance CDRX, considering XR/media traffic pattern.

8.2.2.2 MBS architecture

The MBS architecture follows IMT-2020 architectural principles as defined in 3GPP TS 23.501¹⁹, enabling distribution of the MBS data from the 5GS ingress to NG-RAN node(s) and then to the UE. The main architecture capabilities are summarized as follow:

The MBS architecture provides:

- Efficient usage of RAN and CN resources, with an emphasis on radio interface efficiency;
- Efficient transport for a variety of multicast and broadcast services.

MBS traffic is delivered from a single data source (e.g. Application Service Provider) to multiple UEs. Depending on many factors, there are several delivery methods which may be used to deliver the MBS traffic in the 5GS. The 5G MBS also provides functionalities such as local MBS service, authorization of multicast MBS and QoS differentiation. Between 5GC and NG-RAN, there are two possible delivery methods to transmit the MBS data:

- 5GC Individual MBS traffic delivery method: This method is only applied for multicast MBS session. 5GC receives a single copy of MBS data packets and delivers separate copies of those MBS data packets to individual UEs via per-UE PDU sessions, hence for each such UE one PDU session is required to be associated with a multicast session. The 5GC Individual MBS traffic delivery method supports mobility when there is an NG-RAN deployment with non-homogeneous support of 5G MBS.
- 5GC Shared MBS traffic delivery method: This method is applied for both broadcast and multicast MBS session. 5GC receives a single copy of MBS data packets and

¹⁹ 3GPP TS23.501 V17.6.0 (2022-09) - System architecture for the 5G System (5GS).

delivers a single copy of those MBS packets to an NG-RAN node, which then delivers the packets to one or multiple UEs. The 5GC Shared MBS traffic delivery method is applied in all 5G MBS deployments.

For the multicast session, a single copy of MBS data packets received by the CN may be delivered via 5GC Individual MBS traffic delivery method for some UE(s) and via 5GC Shared MBS traffic delivery method for other UEs.

There are some new network functionalities to support MBS in 5GS:

- MB-SMF, Supporting MBS session management (including QoS control).
- MB-UPF, user plane function for MBS with Packet filtering of incoming downlink packets for multicast and broadcast flows, and QoS enforcement.

Other network functions e.g., SMF to support UE join/leave operation of Multicast session, AMF to support group paging, and signaling routing, PCF to provide policy information and QoS handling for MBS session are also supported in 5GS.

9 Case studies

Followings are intended to provide examples of IMT-2020 capabilities to support multimedia content on demand transmission, real-time multimedia interactive in different scenarios.

9.1 News live report

FIGURE 1

5G based interview with holographic remote location on the same screen²⁰



From 2019, 5G based multimedia technology was adopted to support the live reports of the two sessions. At 2022, Xinhua News Agency launched the first immersive screen fusion interview the With 5G-based multimedia interaction and broadcast technology.

It provided 5G network transmission and cloud-based production of ultra-high-definition videos to media users. Through 5G multimedia services, visitors can not only use the 5G network to return ultra-high-definition video interviews, but also produce program content on the cloud platform. It makes full use of the characteristics of 5G large bandwidth and ultra-low delay, thus the news reports can use 4K, VR and other video services to improve the picture's quality, to smooth the interaction, to ensure high-quality user experience and to diversify needs in all aspects. Based on 5G transmission, the all-round director can realize the integration of the core functions of the studio, such as the input of multiple SDI camera signals, HDMI signals, and IP signals, real-time director

²⁰ <u>http://www.zgjx.cn/2022-03/09/c_1310506639.htm</u>

- 29 -5D/1668 (Annex 3.5)-E

switching, program production, and graphic packaging. It can experience and display highly integrated and efficient broadcast program. And through 5G transmission, it can realize multi-camera indoor and outdoor multi-signal access live broadcast, providing real-time experience of virtual studio.

9.2 Smart Venues

FIGURE 2

Smart Venues for 2022 Beijing Winter Olympics²¹



Located in Beijing and Zhangjiakou City, Hebei Province, the 87 venues and the roads connecting the venues for 2022 Beijing Winter Olympics were full coverage of 5G networks. This is also the largest commercial use of 5G networks in the history of the Olympic Games. With the help of 5G network, many Winter Olympics games e.g. alpine skiing which operated in high mountains with high-speed competitions and difficult-to-capture images are turn into visual feasts in a clear and timely manner. Many of these wonderful videos and pictures are presented through 5G live broadcast technology.

- For the audience, the picture supported by 5G is more high-definition and richer. For example, at the opening ceremony of the Winter Olympics, some Bird's Nest venue shots and athletes entering the arena were captured by the Olympic Broadcasting Service (OBS) through the 5G network that fully covers the Bird's Nest venues. And for the first time it is broadcasted in 4K ultra-high-definition, and some important events are broadcasted in 8K which has higher requirements for transmission bandwidth, e.g. at least 130Mbit/s or more. Only gigabit-level 5G networks can guarantee image quality. In addition, the combination of the VR camera and the 5G network allows the audience to enjoy the game from a 180-degree free angle of view, e.g.the audience in the stands, the environment of the stadium... a lot of spatial information that would have been omitted in the past can be displayed this time, allowing the audience to see the game from different angles.
- The Winter Olympics has built a dedicated 5G media network to serve media live broadcasts and other special activities. 4K/8K picture quality, free viewing angles, and "bullet time" and other images are all broadcast through low-latency, high-bandwidth 5G media dedicated networks. Even in venues with a high density of people, the

²¹ The photo of National Bobsleigh Center is sourced from <u>https://news.cctv.com/2021/02/02/ARTIo9biaVx8daT9R7Gtp9cw210202.shtml;</u>

Photo of National Speed Skating Stadium is sourced from Xinhua News Agency reporter Ju Huanzong.

network capacity can still meet media needs to upload the multimedia content quickly. Additional, on the Beijing-Zhangjiakou high-speed railway connecting Beijing and Zhangjiakou, the two major Winter Olympic competition areas, 356 5G base stations were deployed to construct the 3.5G 200M high-speed rail network, thus even at a speed of up to 350 km/h and across mountains and mountains, global media workers can watch the game in real time and use 4K live broadcast throughout the whole process without any lag. During the competition, many media reporters need to work in the mountains, while the 5G backpack works like a mobile signal station can follow the footsteps of the camera reporters to ensure the mobile network signal in sub-zero temperatures. Thus it allows high-definition video pictures to be uploaded to the cloud in real time, which enable the background staff directly edits, produces, and broadcasts on time to provide the wonderful scenes.

9.3 Sports event live broadcast

FIGURE 3

2019 International Gymnastics Federation Chengdu Parkour World Cup²²



In the program "2019 International Gymnastics Federation Chengdu Parkour World Cup" from April 6th to 7th, 2019, China Media Group (CMG) realized the live broadcast of VR sports events based on 5G network transmission. This is the first time that 5G technology and ultra-highdefinition VR production technology have been combined and applied to the broadcast of sports events. Considering the characteristics of parkour, the technical team took advantage of 5G's advantages of low latency, higher data rate, and large capacity to design the VR video live broadcast solution, technically shortening the time delay of real-time VR video signals from collection and production to distribution and improving its performance. In addition, the refined production of event-related VR video content covers the scenery of the host city, highlights around the stadium, VR video aerial photography, and the unique first-person perspective of contestants. The game-related content is displayed in all directions and from multiple angles, enabling viewers to experience the game atmosphere at close range.

In the solution, the multi-angle videos were shot in the competition area, collected and stitched in real time via the 5G network. Then, the high-definition VR video signal is transmitted to the 5G media application laboratory in Beijing's Guanghua Road office through the 5G base station and network. After the VR video were produced including switching, subtitle packaging, and special

http://www.nrta.gov.cn/module/download/downfile.jsp?classid=0&filename=cf359da741ac4bdba2 5f4b77b9cfc160.pdf

²² National Radio and Television Administration (PRC) 5G Advanced Video – White Paper on VR Video Technology (2020)

- 31 -5D/1668 (Annex 3.5)-E

effects in the media production platform, the media content integrated publishing platform distributes to CCTV5 APP client via 5G network for live broadcast.

9.4 Interactive cloud gaming

ShouGang No. 1 Furnace Paradise²³

FIGURE 4

Integrated with 5G, cloud XR, multiple layer's computing network, the traditional sports competitions are merged with modern virtual reality technology, build the century-old industrial site of ShouGang No. 1 blast furnace into the largest XR technology experience park in North China, turn the abandoned industrial site into the entrance of the universe, build the capital's high-tech cultural industry, and form 5G entertainment new benchmark. Through 5G+cloud XR technology, gamers only need to wear a lightweight head-mounted display device to enter the metaverse venue and experience the subversive and innovative immersive experience.

The "wide-area and large-space VR interaction" scenario supported by IMT-2020, further provides differentiated computing power for cloud XR gaming, saving GPU resources by up to 50%, processing delay can be reduced to 35 ms, and end-to-end communication delay is less than 20 ms, ensuring the high performance of cloud XR gaming services experience. Through the intelligent scheduling of IMT-2020 resources, it greatly reduces the initial CAPX investment and OPAX of network operation in every venues. At the same time, the number of customers is increased at least 30% with new interactive games (e.g. Player versus Player, Player versus Environment, skydiving game, etc.). And with IMT-2020 supporting the social and interactive capabilities in different locations, the different venues that was originally independent of each other can be upgraded to a "metaverse" entrance, realize the online and offline linkage of experiencers, support multiple gamers to interact in the metaverse space.

10 Summary

This report summarizes various capabilities of terrestrial component of IMT-2020 for Multimedia communications.

Multimedia is an immersive technological way of presenting information that combines audio, video, images, and animations with textual data. Multimedia applications include network video, digital magazine, digital newspaper, digital radio, social media, touch media, etc., and can be easily enabled using IMT-2020. New emerging technologies such as Virtual Reality (VR) and Augmented Reality (AR) are becoming key technologies to upgrade the traditional multimedia industries.

The IMT-2020 capabilities can support the evolving interactive multimedia communication with the capabilities not only broader bandwidth, higher data rate, but also lower latency and higher reliability.

²³ <u>http://www.5gaia.org.cn/FourthBloomingCup/News/Detail/360;</u> https://www.laoyaoba.com/n/805626

The typical technologies are flexible and dynamic resources allocation, uplink enhancement e.g. UL MIMO, UL Carrier Aggregation and Dual connectivity, and related architecture improvement, which can connect the user to a high-definition video, real-time multimedia interaction virtual world on their mobile device.

Live events with high definition and ultra-high definition content can be streamed via IMT-2020 radio network with higher throughput. HD and UHD content (e.g. news, sport event) can be real-time produced and on demand distributed to mobile devices without any interruptions through IMT-2020 higher user experienced data rate and low latency. The entertainment industry will hugely benefit from IMT-2020 wireless networks, which are expected to enable HD virtual reality games with a better real-time interactive gaming experience, and high dynamic range video streaming without interruption. Cloud AR and Cloud VR with HD or UHD video can be supported with higher user experienced data rate and low latency supported by IMT-2020.

It is foreseeable that with the support of IMT-2020 technology, it will gradually bring consumers more amazing virtual experience.
