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Contents

Foreword

This Technical Specification (TS) has been produced by the 3rd Generation Partnership Project (3GPP).

The contents of the present document are subject to continuing work within the TSG and may change following formal TSG approval. Should the TSG modify the contents of the present document, it will be re-released by the TSG with an identifying change of release date and an increase in version number as follows:

Version x.y.z

where:

x   the first digit:

     1   presented to TSG for information;

     2   presented to TSG for approval;

     3   or greater indicates TSG approved document under change control.

y   the second digit is incremented for all changes of substance, i.e. technical enhancements, corrections, updates, etc.

z   the third digit is incremented when editorial only changes have been incorporated in the document.

1 Scope

This TS defines the service requirements from users' and operators' perspective for the support of IP multimedia applications.

IP multimedia applications are supported by IP multimedia sessions in the IM CN Subsystem. IP multimedia sessions use IP connectivity bearers (e.g. GPRS as a bearer). Examples of IP multimedia applications include speech communication, real time multimedia applications, shared online whiteboards etc.

This TS, in general, does not standardise usage of IP multimedia applications, but instead identifies the requirements to enable their support.

In order to align IP multimedia applications wherever possible with non-3GPP IP applications, the general approach is to adopt non-3GPP IP based solutions.

The existing legacy tele- and supplementary services shall not be re-standardised as IP multimedia applications, and multimedia equivalent applications may be created with toolkits.

2 References

The following documents contain provisions which, through reference in this text, constitute provisions of the present document.

- ? References are either specific (identified by date of publication, edition number, version number, etc.) or non-specific.
- ? For a specific reference, subsequent revisions do not apply.
- ? For a non-specific reference, the latest version applies.
- ? A non-specific reference to an ETS shall also be taken to refer to later versions published as an EN with the same number.

2.1 Normative references

[1] TS 22.003: " CS Teleservices supported by a PLMN".

[2] TS 22.011: "Service Accessibility".

[3] TS 22.060: "General Packet Radio Service (GPRS) stage 1".

[4] TS 22.066: "Support of Mobile Number Portability (MNP)".

[5] TS 22.101: "Service principles".

[6] TS 22.105: "Services and Service Capabilities".

[7] TS 22.121: "3rd Generation Partnership Project; Technical Specification Group Services and System Aspects; The Virtual Home Environment"

[8] TS 22.129: "Handover requirements between UMTS and GSM and other Radio Systems".

[9] RFC2543: "SIP: Session Initiation Protocol"

[10] TS 22.078: "; Customised Applications for Mobile network Enhanced Logic (CAMEL); Service definition - Stage 1"

[11] TS 22.057: "; Mobile Execution Environment (MExE); Service description, Stage 1"

[12]   TS 22.038: "3rd Generation Partnership Project; Technical Specification Group Services and System Aspects; USIM/SIM Application Toolkit (USAT/SAT); Service description; Stage 1"

[13]   TS 22.127: "3rd Generation Partnership Project; Technical Specification Group Services and System Aspects; Stage 1 Service Requirement for the Open Service Access (OSA)

3 Definitions and abbreviations

3.1 Definitions

For the purposes of this TS the following definitions apply:

IM CN subsystem: (IP Multimedia CN subsystem) comprises of all CN elements for the provision of IP multimedia applications over IP multimedia sessions

IP multimedia application: an application that handles one or more media simultaneously such as speech, audio, video and data (e.g. chat text, shared whiteboard) in a synchronised way from the user's point of view. A multimedia application may involve multiple parties, multiple connections, and the addition or deletion of resources within a single IP multimedia session. A user may invoke concurrent IP multimedia applications in an IP multimedia session.

IP multimedia service: an IP multimedia service is the user experience provided by one or more IP multimedia applications.

IP multimedia session: an IP multimedia session is a set of multimedia senders and receivers and the data streams flowing from senders to receivers. IP multimedia sessions are supported by the IP multimedia CN Subsystem and are enabled by IP connectivity bearers (e.g. GPRS as a bearer). A user may invoke concurrent IP multimedia sessions.

Local service: services which are provided by a current serving (home or visited) network.

3.2 Abbreviations

For the purposes of this TS the following abbreviations apply:

API	Application Programming Interface
CAMEL	Customised Application for Mobile Enhanced Logic
CN	Core Network
CS	Circuit Switched
GPRS	General Packet Radio Service
IM	IP Multimedia
IP	Internet Protocol
MExE	Mobile Execution Environment
OSA	Open Service Architecture
OA&M	Operations, Administration and Maintenance
QoS	Quality of Service
SAT	SIM Application Toolkit
SIP	Session Initiation Protocol
UE	User Equipment
VHE	Virtual Home Environment
WWW	World Wide Web

4 Introduction

IP has opened up a whole range of communication applications, which may allow operators to develop totally new value added applications as well as to enhance their existing solutions. The open architecture and platforms supported by IP

and operating systems may lead to applications and new opportunities that are more difficult to replicate using a standard switched centralised solution.

A complete solution for the support of IP multimedia applications (including voice communications) shall be available. The solution consists of terminals, GERAN or UTRAN radio access networks and GPRS evolved core network. One of the main objectives for 3GPP specifications is to ensure that the availability and behaviour of these IP applications when used via the 3GPP mobile access is at least as good as when used via other mobile access types.

5 High level requirements

Support for IP multimedia sessions shall be provided in a flexible manner to allow operators to differentiate their services in the market place as well customise them to meet specific user needs. This shall be provided by the use of service capabilities in both networks and terminals, for the creation and support of IP multimedia applications.

The following high level requirements shall be supported for IP multimedia applications:-

? negotiable QoS for IP multimedia sessions both at the time of a session establishment as well as during the session by the operator and the user

? negotiable QoS for individual media components in an IP multimedia session both at the time of establishing a media component as well as when the media component is active by the operator and the user

? end to end QoS for voice at least as good as that achieved by the circuit-switched (e.g. AMR codec based) wireless systems shall be enabled

? support of roaming, negotiation between operators for QoS and for Service Capabilities is required. Such negotiation should be automated rather than manual, e.g., when another operator adds new service capabilities.

? possibility for a network operator to implement IP Policy Control for IP multimedia applications.

? IP multimedia sessions shall be able to support a variety of different media types. A set of media types shall be identified to ensure interoperability (e.g. default codec selection and header compression).

? within each IP multimedia session, one or more IP multimedia applications shall be supported

? the possibility for IP multimedia applications to be provided without a reduction in privacy, security, or authentication compared to corresponding GPRS and circuit switched services

? support for interworking between the packet and circuit switched services, and with PSTN, ISDN and Internet

? roaming shall be supported enabling users to access IP multimedia services provisioned by the:-

    ? Home Environment

    ? Serving Network

? access independence shall be supported. It is desirable that an operator should be able to offer services to their subscribers regardless of how they obtain an IP connection (e.g. GPRS, fixed lines, LAN).

? It shall be possible to support session-related internet applications that have been developed outside the 3GPP community.

? It shall be possible to limit the view of an operator's network topology to authorised entities.

6 Standardised service capability approach

IP multimedia applications shall, as a principle, not be standardised, allowing operator specific variations. It shall be possible to enable rapid service creation and deployment using service capabilities.

It is important that commercially available IP multimedia applications are supported. In general compatibility shall be with these IP multimedia applications instead of building 3GPP-specific solutions.

The following options shall be available in the 3GPP standards to enable service delivery:

?   an architectural framework shall be created that enables maximum flexibility in the end user device and network servers, similar in concept to that used in the Internet.

This framework shall enable an operator to efficiently deploy IP multimedia applications in a network-agnostic manner without having to wait for these applications or additional enabling technology, to be standardised in 3GPP.

?   service capabilities (enhanced to control IP multimedia applications), which will allow IP multimedia applications to be deployed in a vendor independent manner

CAMEL [10], MExE [11], SAT [12] and OSA [13], which are the identified service capabilities of VHE in 22.121 [7], should be improved to support IP multimedia applications, e.g. additions to APIs, service capability features, service capability servers, user profile etc.

?   registration mechanisms which allow the network to understand the limitations of the mobile and thereby take appropriate actions.

Note: There is a concern that with a large variety of toolkits to create applications, service interworking between terminals and networks may be compromised and needs to be addressed.

7   User service requirements

IP multimedia sessions provide the ability for users to invoke IP multimedia applications to send and receive (where applicable) voice and data communications, even when roaming. This includes interworking with existing voice and data networks for both fixed (e.g. PSTN, ISDN, internet etc.) and mobile users.

The IM CN subsystem shall support interworking with existing fixed and mobile voice and IP data networks, including PSTN, ISDN, Mobile and Internet.

When an IM session originates or terminates in a CS telephony call, the experience of the CS telephony network user should not substantially differ from that of a call between two CS telephony network users in terms of aspects such as the delay to set-up communications and the total permissible delay in transporting speech between the end users. The IM CN subsystem does not necessarily have to support all services offered by the CS telephony network.

Visited network provided services give the opportunity for the visited network to offer services of a local nature to the visiting user and gain additional revenue from their usage by inbound roamers.

7.1   Identifying IP multimedia application subscriptions

There is no requirement to support standardised subscription mechanisms for IP multimedia applications.

IP multimedia applications may require to be provisioned and configured by users and operators. Since the source and variety of IP multimedia applications are not standardised, the specific feature codes to provision, enable and configure IP multimedia applications cannot be standardised either. Thus there are no requirements on the network capabilities to support provisioning and configuration for specific IP multimedia applications.

Note: The standardised service capabilities, personalised Internet web pages and evolving IP mechanisms may be used to allow user (self) provisioning, configuration and enabling of IP multimedia applications.

7.2 Access to the IM CN subsystem

7.2.1 Access control

The IM CN subsystem shall be able to verify at any time that the user is entitled to use the resources of the IM CN subsystem.

7.3 Capability negotiation

The IP multimedia applications shall be able to negotiate their capabilities to identify and select the available media components, QoS etc. of IP multimedia sessions. It shall be possible for the capability negotiation to take place on invocation, acceptance and during an IP multimedia session (e.g. following a change in UE capabilities, change in media types etc.). Capability negotiation may be initiated by the user, operator or an application on behalf of them.

In order to support the user's preferences for IP multimedia applications, the capability negotiation shall take into account the information in the user profile whenever applicable.

7.4 Redirect

It shall be possible for the user to identify an alternative destination for IP multimedia sessions or individual media of an IP multimedia session. This may be initiated by the sending or receiving party during an IP multimedia session. Redirect may also occur prior to an IP multimedia session being established (e.g. the user's preference is that the IP multimedia session is unconditionally redirected).

7.5 Invoking an IP multimedia session

The user shall be able to invoke one or more IP multimedia sessions. The user shall also be able to activate concurrent IP multimedia applications within each IP multimedia session.

7.5.1 Identification of entities

Both telecom and internet numbering and addressing schemes shall be supported. IP multimedia communication establishment (both mobile originating and terminating) depending on originator shall be able to be based on E.164 (e.g. +1 23 456 789) or SIP URL (sip:my.name@company.org) [9].

It shall be possible for the network operator to use

- the same E.164 number for IP multimedia sessions and CS speech telephony (TS11) [1]

- a different E.164 number if desired for IP multimedia sessions

This allows customers who originally had only an E164 MSISDN to retain the same number for receiving communications in the IM domain and also in the CS domain when outside IM coverage.

7.5.2 Negotiation at IM session invocation

It shall be possible for the capability negotiation to take place at the time of the IP multimedia session invocation. Refer to subclause 7.3 for further details on capability negotiation on IP multimedia session invocation.

7.5.3 Emergency communications

See [5] for further details.

7.6 Handling of an incoming session (by the terminating entity)

7.6.1 Presentation of session originator identity

It shall be possible to present the identity of the session originator (see 7.5.1) subject to it not being suppressed by the session originator.

7.6.2 Negotiation of an incoming session

Interaction with the user profile shall be supported, and additionally direct interaction with the user may be required. Refer to subclause 7.3 for further details on capability negotiation on an incoming IP multimedia session.

7.6.3 Accepting or rejecting an incoming session

It shall be possible for the user to either accept or reject an incoming IP multimedia session. Further, it shall also be possible for the user to accept only a subset of the offered media, not have any of the media offered to him at all etc.

7.7 Handling of an ongoing session

7.7.1 User modification of media in an ongoing session

The user shall be able to negotiate the addition or deletion of media components of IP multimedia applications during an IP multimedia session. Refer to subclause 7.3 for further details on capability negotiation during an IP multimedia session.

7.7.2 Suspending and resuming of an ongoing session

It shall be possible for the user to suspend an IP multimedia session, and resume that IP multimedia session at a later time.

7.8 Ending a session

The user shall be able to end an IP multimedia session at any time during the session.

7.9 Local services

When users roam outside the Home Environment, as well as being able to access VHE features with a similar look and feel to the home network, they should also be able to access services of a local nature offered to them by the visited network.

Such services offered by the visited network could include, for example:

? Access to local numbering plans;

? Address translation;

? Services dependent on the geographical location of the user, e.g. traffic information.

Visited network offered services would probably be non-subscription services, although they may be chargeable, with charges collected via the Home Environment according to the usual arrangements for roaming.

Annex A   Example IP multimedia application scenarios (informative)

The following example scenarios describe the personalised handling of individual media in multimedia applications (note that this list is neither complete nor exhaustive):-

1. The user is in a voice communication, and receives an incoming IP video communication. The user decides not to accept the communication, but diverts the incoming video to a messaging system. Further, the user is given an indication that there is a video message in his mail box

2. The user is in a voice communication, and receives an incoming video communication. The user decides to accept the communication but wishes to switch between the two communications.

3. The user is idle in a network and not involved in a communication. The user modifies his user profile to divert all voice communications other than those from high priority, pre-identified callers (e.g. his boss). In this scenario all emails and text messages continue to be received regardless of the sender.

4. On receiving a communication, the calling party's identity is displayed (if not restricted) and user shall be able to decide whether to accept the communication, or divert to a messaging system. The user shall be able to request media handling of the communication (e.g. media splitting to different destinations, media conversion).

5. The user is busy in a communication when receiving an incoming communication, but responds to the originating party that he will respond later. The user may request that the originating party's details (if not restricted) are stored with a reminder in user's profile.

6. The user is in a voice communication, and receives an incoming IP video communication. The user decides not to accept the communication, but diverts the incoming video to a messaging system. Further, the user is given an indication that there is a video message in his mail box

7. The user is in a voice communication, and receives an incoming video communication. The user decides to accept the communication but wishes to switch between the two communications.

8. The user is idle in a network and not involved in a communication. The user modifies his user profile to divert all voice communications other than those from high priority, pre-identified callers (e.g. his boss). In this scenario all emails and text messages continue to be received regardless of the sender.

9. On receiving a communication, the calling party's identity is displayed (if not restricted) and user shall be able to decide whether to accept the communication, or divert to a messaging system. The user shall be able to request media handling of the communication (e.g. media splitting to different destinations, media conversion).

10. The user is busy in a communication when receiving an incoming communication, but responds to the originating party that he will respond later. The user may request that the originating party's details (if not restricted) are stored with a reminder in user's profile.

11. Hi-fi sound (nuances, character of voice)
    Person(s): Marketing Manager, Rita
    Situation: She is at a launch party for some customers in London. In the break she listens to her messages and one from another customer in Tokyo gets her attention. He just wants her to call, but doesn't say if it is urgent or not.
    Solution: Due to the excellent sound quality of the terminals involved and the messaging system, she picks up the faint irritation in his voice and decides to call him immediately. It was urgent and she could remedy the situation easily by emailing the information from her built in PDA storage. The customer was relieved as he was just going in to a very important meeting.
    Benefit(s): Good sound quality gives more information to base judgements on, i.e. emulates real life meetings better.

12. Stereo sound (nuances, character of voice plus positions, sound-scapes)
    Person(s): Purchase Officer, Gustavo
    Situation: Participates in a conference to discuss purchase of a new kind of steel for the factory in Rio. As he is on the road he calls from his hotel room in Sydney. The conference is in the head office in Rio. The local department has invited the two final contenders to have them argue their cases. The two companies are positioned at the

different ends of the table. One of the groups is presenting and mentions something about deliveries. A side remark is barely audible, "we can't deliver that quality and that quantity this year !" Who gave this remark?

Solution: The excellent sound quality together with the stereoscopic sound gives Gustavo the information he needed. It was the other group that gave the remark. The decision was made for him at that point. He gave the order to the presenting group right after they finished a very good presentation that told him everything he wanted to hear. The setup at the head office was done with two synchronized 3G phones at each end of the table.

Benefit(s): Stereoscopic sound gives even more information than just hi-fi sound to base judgements on, i.e. emulates real life meetings better.

13. Conference/chat with "private rooms"

Person(s): A project team at an IT company: Rick, Diana, Ted, Sven and Liu
They are based in different cities.

Situation: The project team has one of their weekly reporting meetings using their mobile communicators. In the middle of the meeting, Rick and Diana get lost in a lengthy arguing on some detailed design matters that bores the rest of the team. Ted, the moderator, finds that it is nevertheless necessary to give Rick and Diana some minutes to finish their discussion, so he decides to not interrupt them. At the same time Sven remembers that he need to remind Liu to send a report to him on the latest findings from her research work.

Solution: The team use a conference/chat service with the new facility "private rooms". This allows Sven to direct a few words in privacy to Liu. Sven activates easily this feature by the GUI of his communicator. Liu is immediately notified by the GUI of her communicator that Sven is now talking privately with her (this is necessary to avoid embarrassing misunderstandings that could occur if Liu would answer Sven in the "common room" instead of in the new "private room" that Sven has created).

Since the voices of all conference members are synthetically mapped in a stereophonic projection, Liu is able to hear what Sven is saying, even though he speaks simultaneously with the other team members (the communicator will not automatically adjust the sound volume of the "common room", since it cannot know if Liu is more interested in Sven's comments or in continuing to listen to the other team members).

Benefit(s): This service emulates virtual presence in a conference room in the best possible way without adding more visionary technologies like holographic projections, etc. The synthetic stereophonic sound projection provides good possibilities for a conference member to discriminate unwanted voices even if the meeting situation is informal and spontaneous and everyone are talking at the same time. The flexible possibilities to create one or more "private rooms" make it easy to make private comments to selected colleagues. The easy-to-use and fast responding GUI makes the needed end-user effort to create a new "room" so low, that it feels natural to use the function even for exchanging just a few quick words.

Alternative use: Exchange the IT project team with a gang of teenagers that are planning what to do in the weekend. The service works perfectly well also in that scenario and provides the same benefits.

Additional features: Easy GUI controlled addition of new participants (can be initiated by any of the participants), including addressing, notification/invitation, etc. (cf. "outgoing call" in PSTN). GUI notification of new incoming session invitations (cf. "incoming call" in PSTN) and possibility to choose action as desired (incorporate the "calling party" in the existing conference session, creating a new separate session, rejecting the invitation, diverting it to a messaging system, etc.) Whiteboarding and/or application sharing.

14. Multiplayer mobile gaming with voice channel

Person(s): Joe (age 15), Blenda (age 14), Fredric (age 15) and all their "cyber friends" in the Shoot-n-Shout v.14.0 community

Situation: In the legendary multiplayer game Shoot-n-Shout v.14.0 the most popular game mode is a team competition. The idea is simply to shoot down the members of the concurring teams. There are always a lot of active game sessions in CyberSpace. At a web/WAP service operated by the game application provider, interested potential players can choose a game session and also find other gamers to form a team with. There is a text chat service where potential team-mates can learn to know each other.

Joe, Blenda and Fredric meet on the web/WAP chat and decide to form a team to take up the fight in one of the Shoot-n-Shout sessions. They are preparing a game strategy in advance through the text chat service, but when they have started the battle it takes too long time to type text, so they the will need another way to communicate with each other.

Solution: The game application provider makes use of a conference/chat service with "private rooms" in order to provide a multi-player voice service to the players of Shoot-n-Shout. When a game starts there is one "common room" where all players can talk (or rather shout) to each other and one "private room" for each team. Players in a team can also dynamically create more "private rooms" if they only like to talk to one (or a few) of their friends. (See the conference/chat scenario for details.)

The volume (and stereophonic position) of the players voices when they are using the "common room" is controlled so that it matches the virtual surroundings in the game environment. As an example, players that are behind a wall

will only be heard as a vague whisper in the distance.

Benefit(s): A voice channel will enhance the gaming experience for several popular network games.

15. Application sharing with voice commentary

Person(s): Marketing Manager, Rita and Media expert, Jones

Situation: The launch of a new campaign for some customers in London. Last minute feedback is that one of the customers is expecting the latest gadget to be included, even if its only a prototype. Rita knows it's not included in the presentation and she has no information with her.

Solution: Rita calls Jones, the media guru they employed for design of their important presentations. He has the information and some pictorials. He sends them over into Rita's PowerPoint application and they edit the new slide together as they discuss the textual information to be included.

Benefit(s): The process is extremely interactive and the session takes only 5 minutes thanks to the broadband connection and the fact that they don't need to Ping-Pong the pictures and the text back and forth. (Emphasize mobile or fixed access as required). The customer is happy and a Letter of Intent is signed.

Comments: By adding voice and pictures in an interactive session we achieve both effectiveness and interaction, two desired components.

16. Emergency location with voice conversation, navigation and picture transfer

Person(s): Ma Beth, her children and the pet dog Bobby

Situation: The family is out driving in the country side and they take a turn on the slippery country road a bit too fast. They slide down into the ditch. Bobby the dog in the back of the van gets a heavy box of books on top of his left paw. It may be broken, and you can tell it certainly hurts from the loud yelps that come out in a rushed stream. The rest of the family is ok. They were all buckled up.

Solution: Ma Beth reaches for her 3G communicator as soon as she has recovered from the initial shock. She calls 112 (911 or similar). The answer comes after 23 seconds and the operator immediately confirms the identity and the location of the van. Ma Beth is a bit taken aback by this quick information and has to think for awhile, then confirms the location as possibly correct. She then states the problem and she gets connected to a vet that asks a few pertinent questions. She can show a close up picture of the dog's left paw and the vet confirms a possible (95%) broken leg just above the paw. He gives a few quick instructions and sends her a map of the closest emergency animal hospital. The map shows her current position and soon displays the quickest way to get to the hospital. Well there, Bobby is taken care of and things are looking up. Even the kids are smiling now that the dog is calm and free from pains, and he looks so funny with his little cast.

Benefit(s): The initial call transfers emergency information to the operator automatically. This ensures minimum delay to correct action. The Communicator transfers the picture that gives enough information to make a very accurate and fast assessment of the situation. Then the map transfer and display on the terminal together with the current position gives clear information and directions for Ma Beth to drive and make the right turns at every corner. In her still half-shocked state she can drive to the hospital without hesitation about where to go. Very reassuring for all parties including the dog that gets fastest possible help.

Comments: The call is initially just a voice call but evolves with the best of positioning in emergency situations and navigational aid together with picture and graphics transfer.

17. The Real Virtual Theatre and Foyer Chat room – Fixed Network example

Person(s): Theatre going "cultural" group with one member (Bob) in a hospital bed.

Situation: The group is watching the play and are utterly fascinated by the first act. When they come out into the foyer in the break they remember Bob. They really want to share this first act with him since they know Shakespeare's Midsummer Night's Dream is his favorite.

Solution: Bob uses the theatre's online streaming service via the hospital network. (At only half the price of a theatre ticket!). The play displays in color and stereo surround sound on his bedside TV set. In the break his friends call him up from the theatre chat room. The chat room is equipped with 3D sound pick up and local display screens with streaming facilities. They set up the streaming from one of the screens to be synchronized with Bob's bedside equipment. Their voices are also mixed into the sound streams as they talk. Bob now gets both the playbacks from the first act and his friends' voices in 3D surround sound. Bob's voice is projected close to the screen as if he was standing leaning on the bench right there. His voice is very clear and full of emotions as he speaks to the various playbacks. Both parties can control the playbacks and watch their own selection in a second window on the screen.

Benefit(s): Bob can pick up every nuance in the lively discussion, including the whispered comments from Greta in the back. The group is almost feeling Bob's presence because of the emotional clarity and distinct position of his voice. As both parties have control and visibility of the streaming sessions, it is very effective and very interactive.

Comments: Experiential services are sought after. This one can be a bit exclusive because of the equipment requirements, but the uses are many.

18. Mobile synchronized MM container

    <u>Person(s)</u>: The married couple Bill and Christine and their daughter Linda

    <u>Situation</u>: Bill is on a business travel to Spain. He calls his wife Christine every night using his MMM terminal. Often Christine is answering at home using her Screenphone, but this particular evening Christine has arranged a baby-sitter for their children so she could go to a restaurant with some friend. When Bill is calling, she is sitting on the commuter train on her way home. Bill often show some pictures during his calls (both live pictures showing the environment where he is at the moment and pictures that he has been taking during the day with his separate digital camera).

    Today, their talk starts off as a common voice conversation. After a while Bill likes to show Christine the lovely sunset view that he can see from his hotel room, so he make some snapshots with the built-in camera of his terminal and sends them in real-time mode to Christine. Christine likes to show one of them to their little daughter Linda when she comes home.

    <u>Solution:</u> With a quick gesture on the touchscreen of Christine's MMM terminal, she instantly moves the selected picture from the real-time session window to the "multimedia container" icon. All the contents of the "container" is automatically mirrored between the MMM terminal and her home server. In this way, Christine can easily pick up the picture from her Screenphone at home. If Linda is at sleep when Christine comes home, she can wait until tomorrow.

    <u>Benefit(s)</u>: The "multimedia container" can be used for every type of MM content that one likes to have available both at home and at another location. This "container paradigm" is very intuitive and stimulates the use of images, video clips etc. for a multitude of purposes. The "container" can be used both for transferring content from the MMM terminal to the home server (as in this scenario) and in the opposite direction.

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