

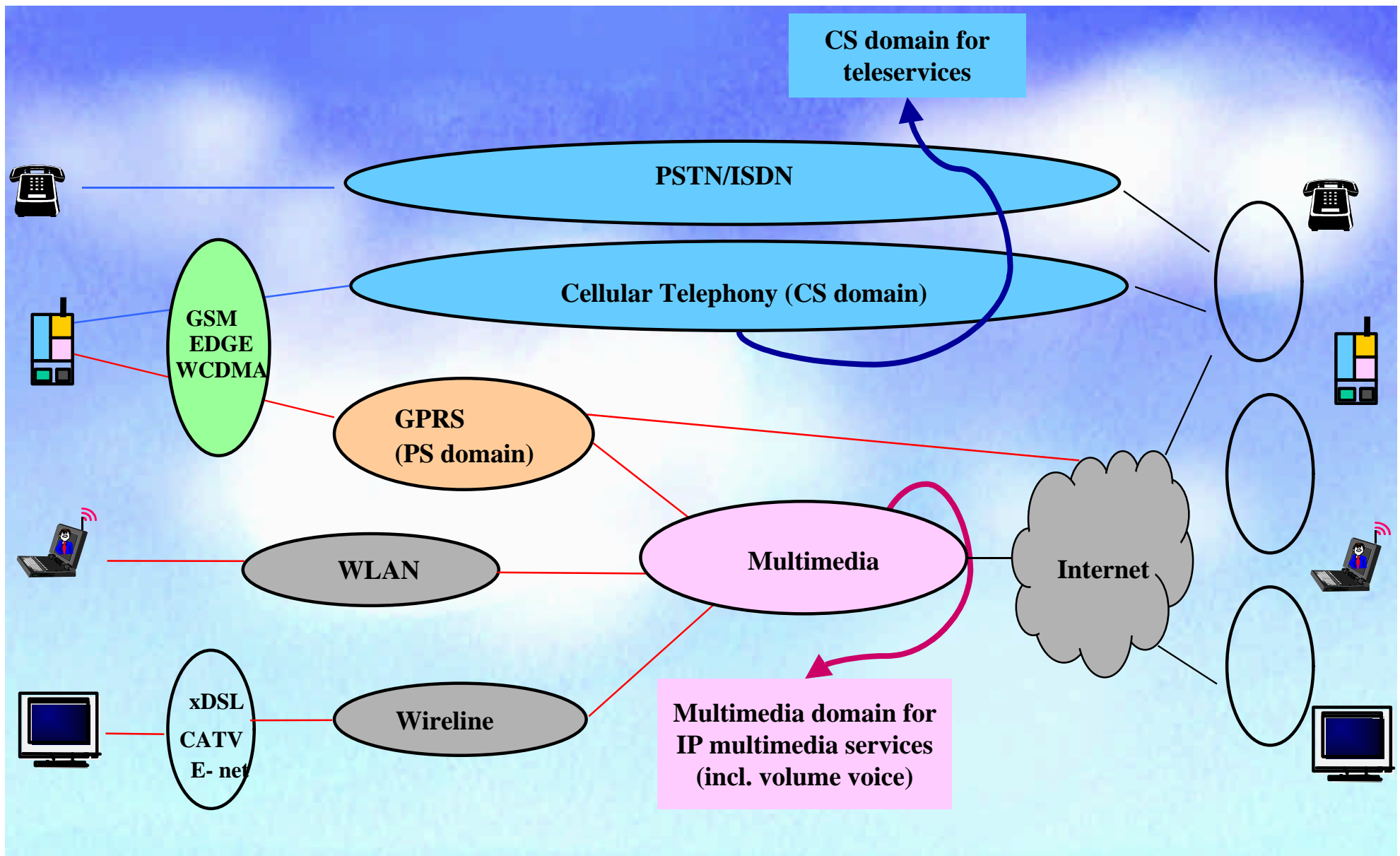
Architectural principles

3GPP All IP Workshop

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- to enable cost effective high quality IP-based multimedia services (including wireless voice over IP as a volume service)
- rationalized teleservices with service continuity of GSM/UMTS teleservice speech (and GSM/UMTS CS data)

2 different service domains, 2 different purposes



- Cost effective high quality IP-based multimedia services, including VoIP, is the fundamental requirement for “All IP” networks.
- The market driver for “All IP” is the development of mobile IP based services supporting **new** end user needs.
- Few market drivers for copying all “old” CS end-user services to new IP services.
 - An end-user does not care which CC protocol is used. ISUP/ISDN based CC, such as 24.008, is mature, widespread and appropriate for teleservice speech. Redoing ISUP/ISDN within IP multimedia call control will hinder the IP multimedia development.
- The battle of VoiceOverIP vs. GSM/UMTS-TeleserviceSpeech
 - Current PSTN/PLMN voice business are rationalised and costs are cut.
 - Development of methods to carry the PSTN/PLMN traffic over an IP-transport.
 - Dual mode mobile devices, combining telephony and GPRS-based always-on packet data, provides mobile end-users with multimedia feeling.
 - IP-telephony is still very much in its infancy, especially for mass deployment in public carrier class networks.
 - The massive footprint of the PLMN/PSTN/ISDN and specifically the strength of today’s Cellular Telephony.
- GSM/UMTS-TeleserviceSpeech will be a tough competitor to voice on IP

UMTS network architecture

Service network

Services/application layer

Applications and Services

Standardized API

Core network
CC/SM/MM

Control layer

Teleservice

Internet service

Internet services uses:

- UMTS PS domain (backwards compatible)
- IP Multimedia domain (new service domain)

Switching/routing

Connectivity layer

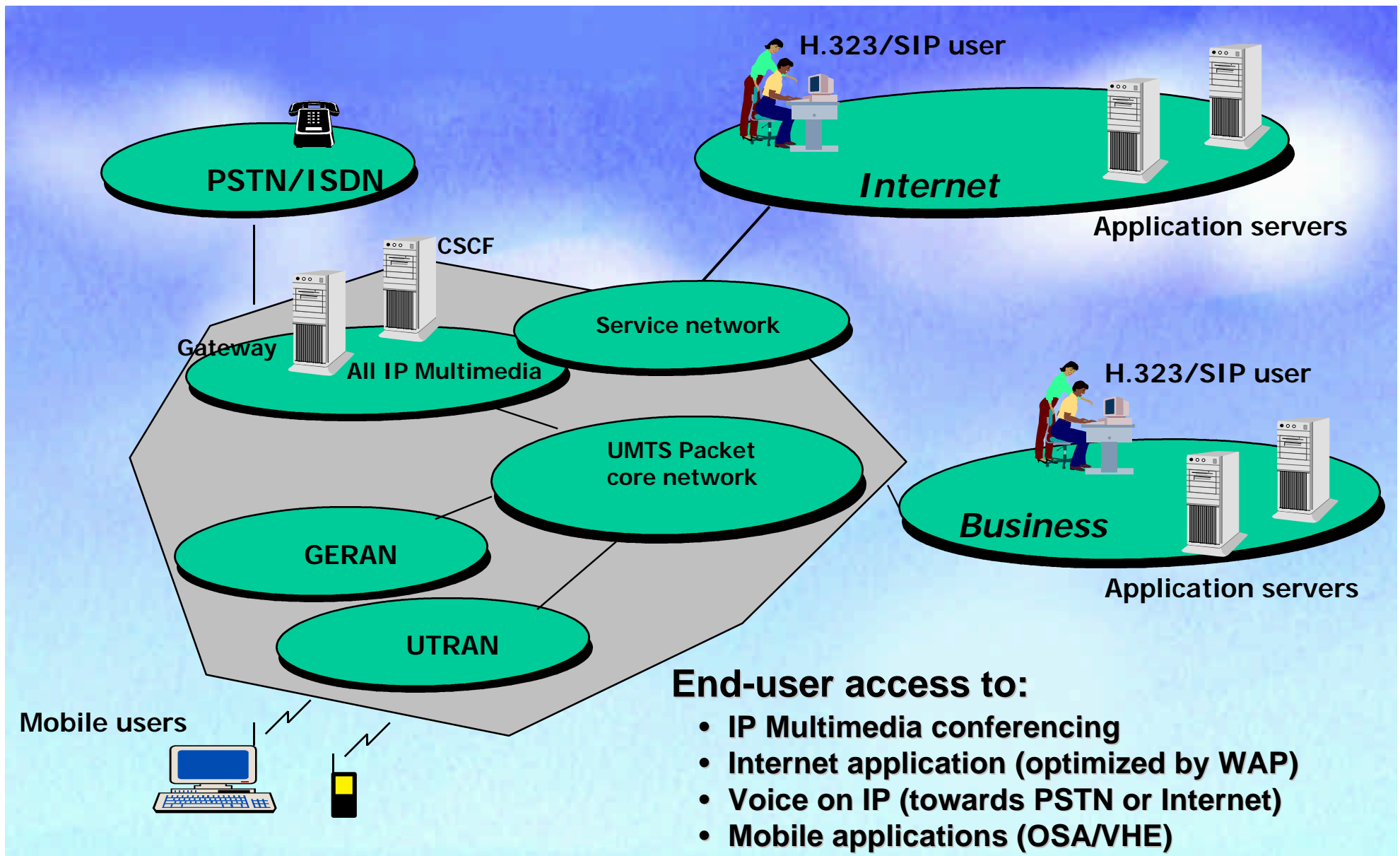
Teleservices uses:

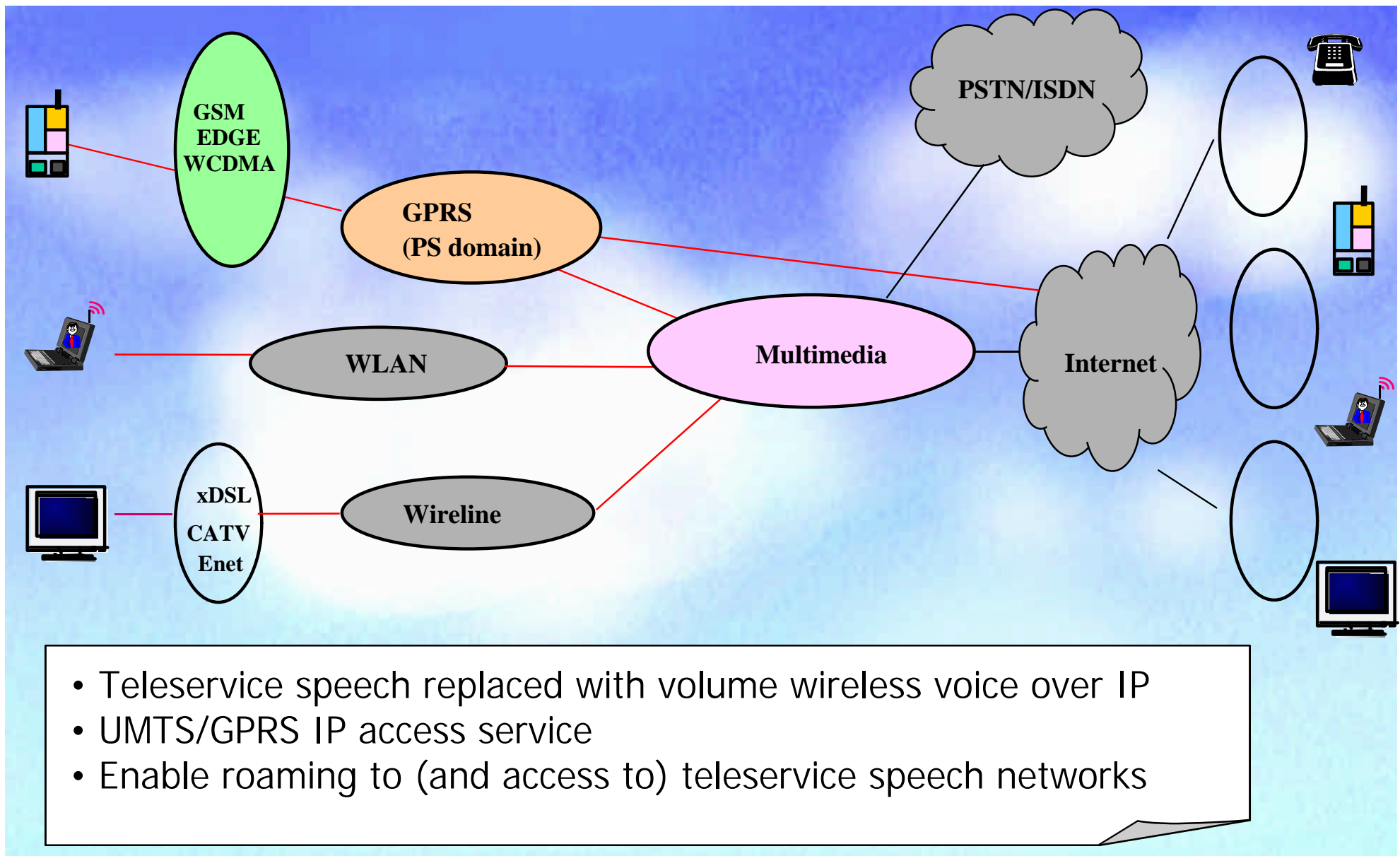
- UMTS CS domain
- Backwards compatibility a strong requirement

Access network

UTRAN

GERAN





- Teleservice speech replaced with volume wireless voice over IP
- UMTS/GPRS IP access service
- Enable roaming to (and access to) teleservice speech networks

Basic requirements:

- IP end-to-end
- IP Multimedia includes call control of one (or several) types (H.323, SIP is the main candidates)
- Shall be possible to access from various access networks
 - Must thus follow mainstream IP and Internet standards
- Support for IP QoS
- Support for global roaming
- Radio optimizations shall allow IP based volume voice services
 - Header compression, Header stripping
 - General means for unequal bit protection (remotely located Codec)
 - GSM speech is the benchmark

End - user perspective:

- GSM teleservices (and CS data) continues
- Full backwards compatibility

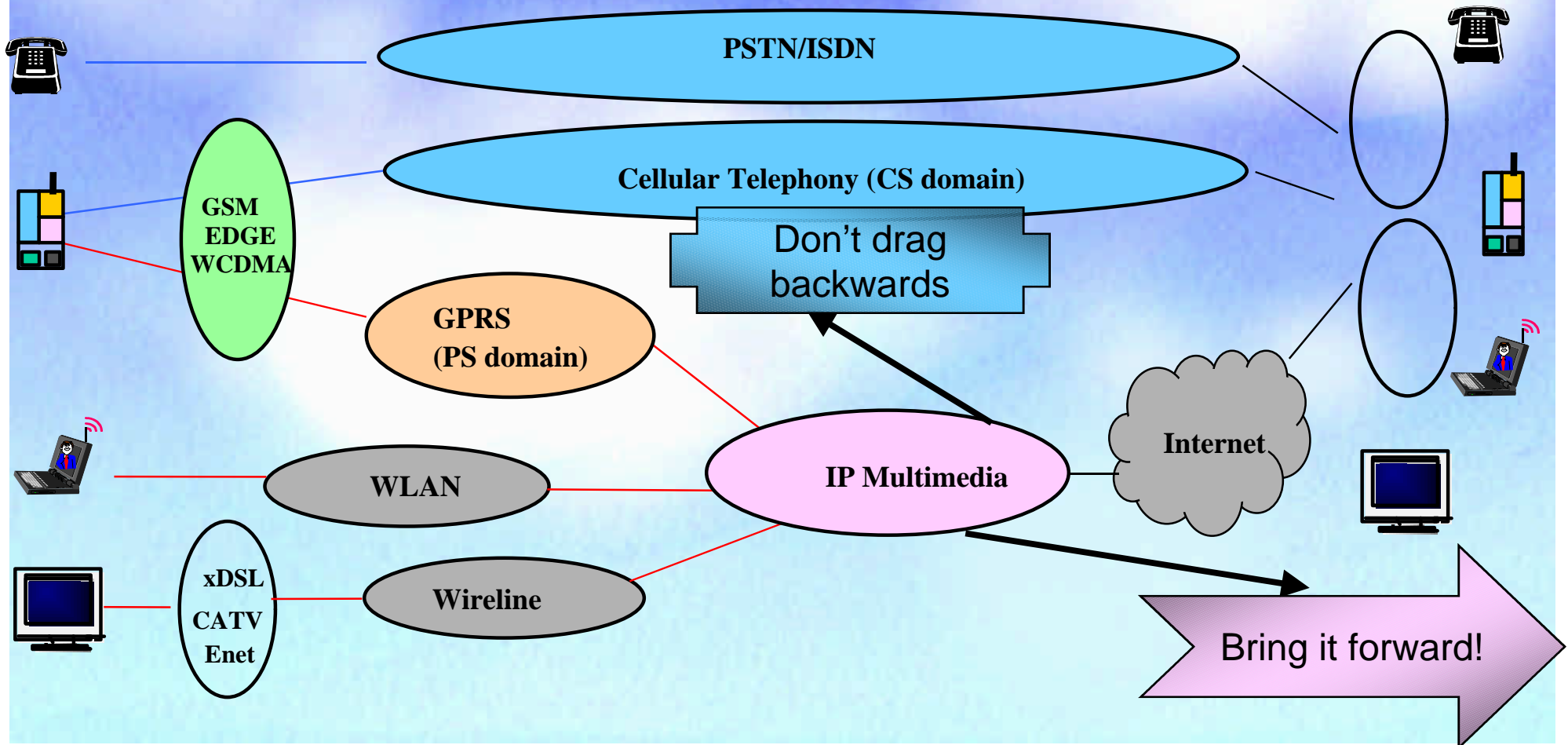
Operator perspective:

- Increased network flexibility for transport of user traffic and signaling
- Develop IP transport of user traffic and signaling
- Develop signaling network based on IP transport and IP addressing
- Allow migration from today's networks

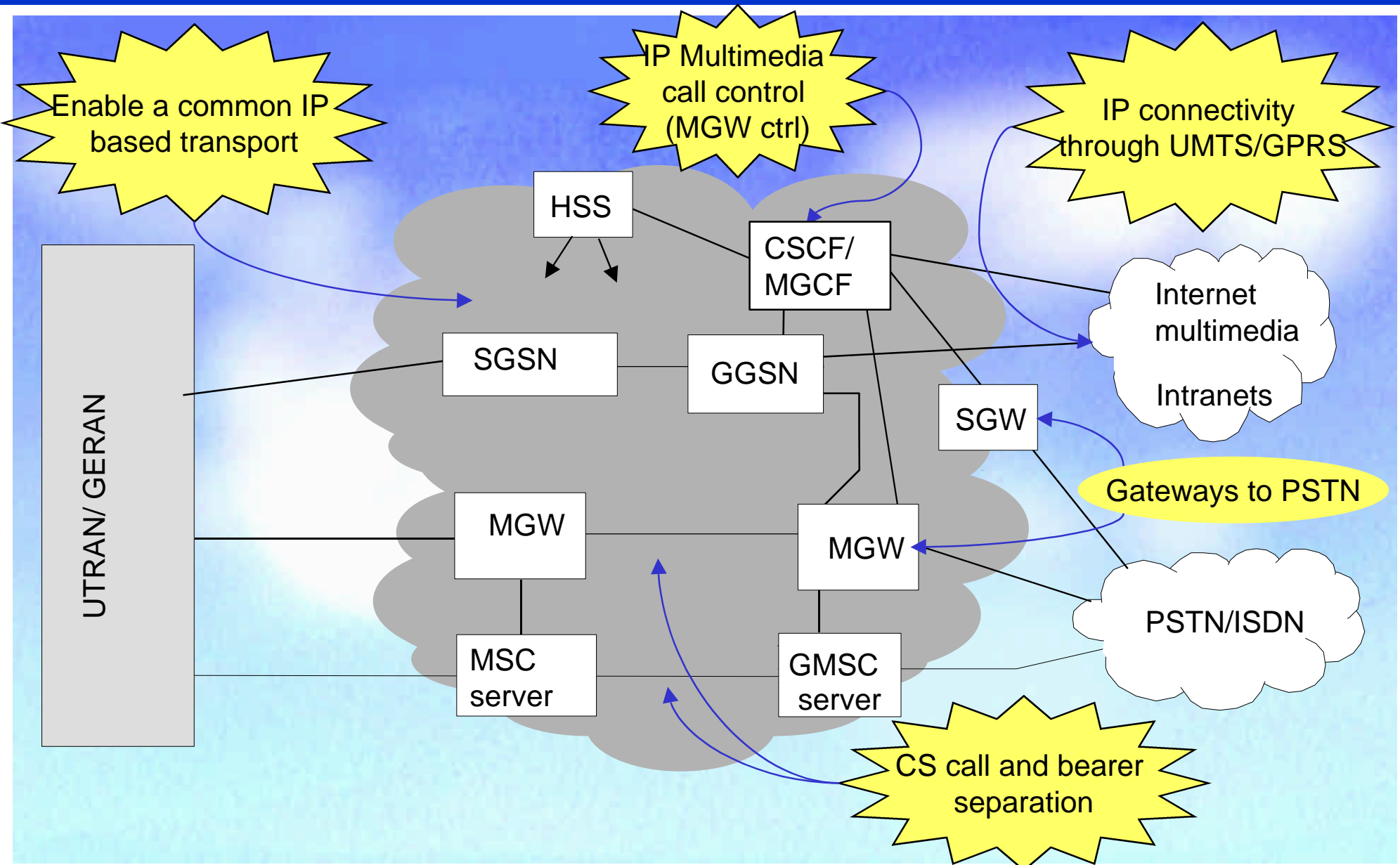
Technology enablers:

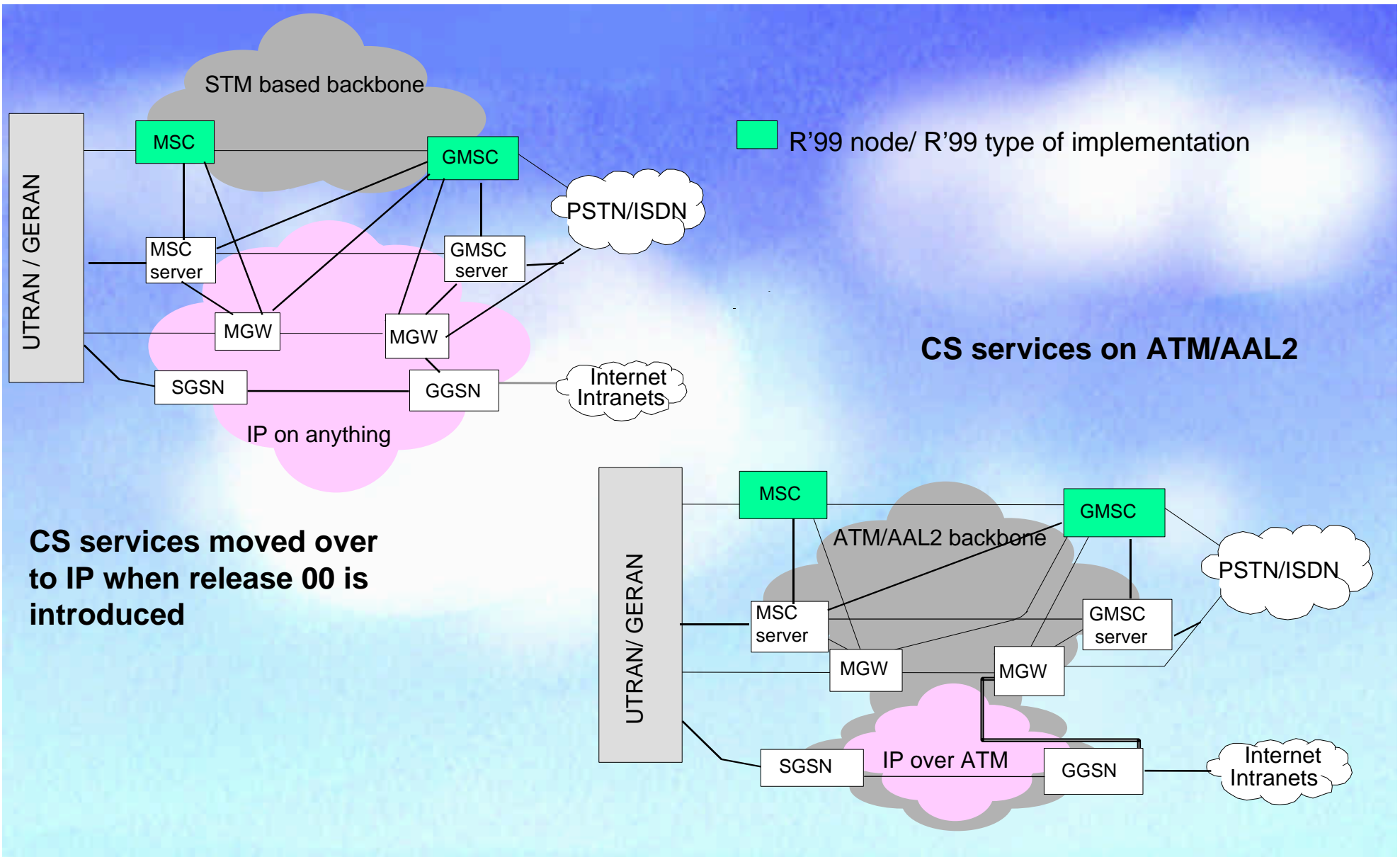
- Transport independence
- Call and bearer separation
- IP QoS

- “All IP” standardization objective
- New IP based end-user services
 - IP transport for existing services



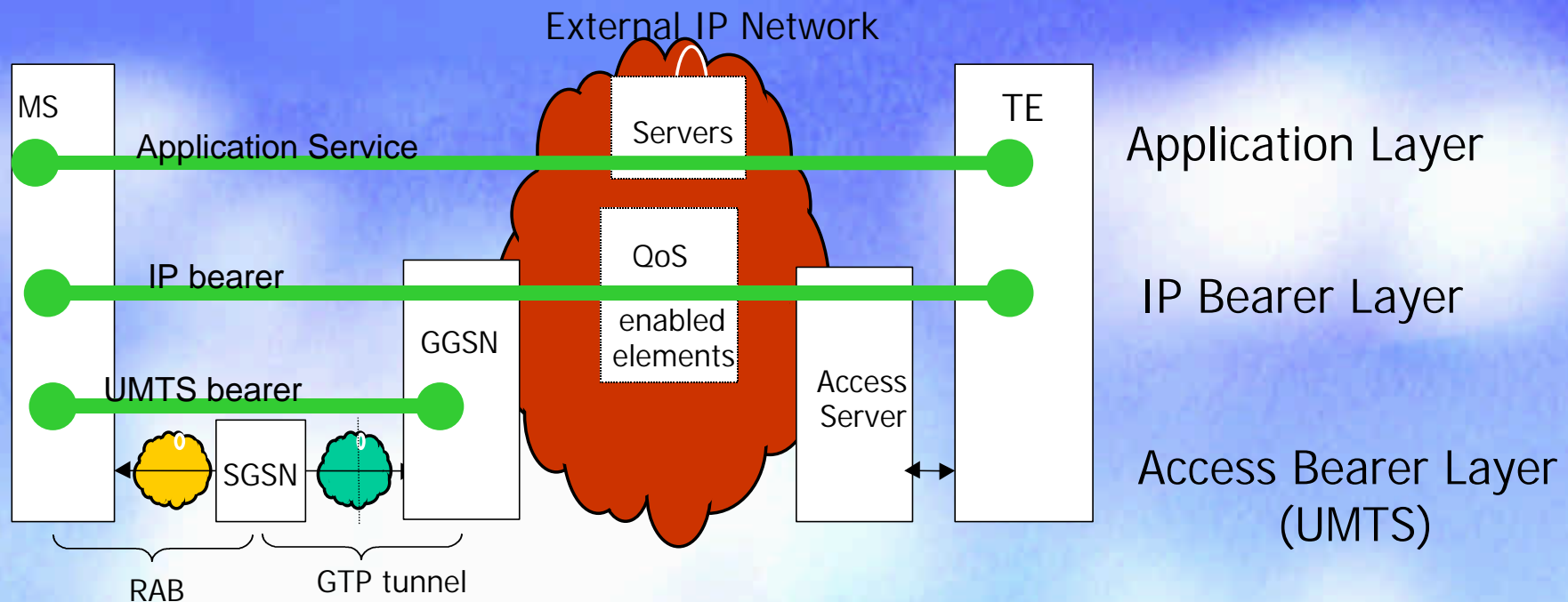
Technology enablers





QoS requirements

- **QoS guarantees from the MS up to/including the UMTS edge (Gi)**
- **End-to-end QoS based on open IP QoS methods (bandwidth broker, QoS policy framework etc.)**
- **IP transmission resources**
 - efficiently utilized for voice, conversational multimedia and data traffic
 - QoS control on a transport independent manner
 - scalable traffic management mechanisms
- **User/application control of radio- and IP QoS through API**
- **Mapping from application to radio bearer is one key for success**



- QoS requests originates at different layers
- The layers need to be coordinated to provide end-to-end QoS
- Proper mapping all the way from application (W3C, H.323) to radio bearer (dB) is key to success

- **Ensure end users experience to be consistent (unaware of change in domains-within or outside of home operator's environment)**
- **Minimize dependencies between Home and Visiting networks regarding call/session control and services triggering**
- **Provide flexibility for bearer path optimization**
- **Support for multiple addressing schemes**
- **Allow better support for Operator differentiation**

Used for originating calls/session and for reachability (H.323/SIP)

Service/application specific mobility

Reachability functions such as mobile IP

Network specific mobility

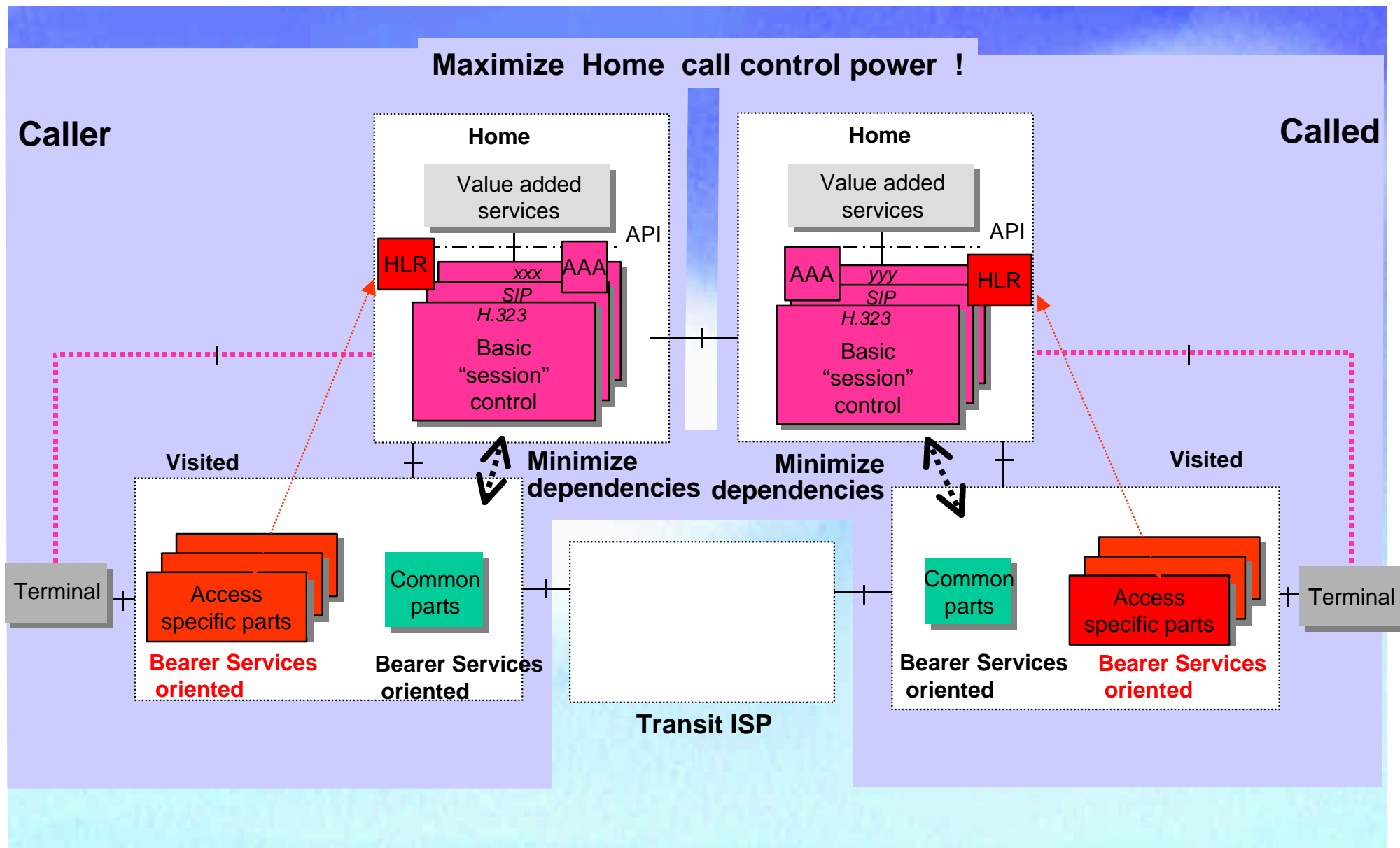
Access granting functionality (AAA):
• Login for internet access
• Login to application

Generic IP roaming framework

PLMN roaming
GSM/UMTS handover

Access specific mobility

Minimise coupling between Home and Visited domain.



The goal for “All IP” standardization effort is to

- ❖ enable cost effective high quality IP-based multimedia services (including wireless voice over IP as a volume service)
- ❖ rationalized teleservices with service continuity of GSM/UMTS teleservice speech (and GSM/UMTS CS data)

● Key elements

- New IP based end-user services
- Enable IP transport for existing services
- Access independence
- Global roaming

● Technology enablers

- End-to-end QoS architecture
- Radio resource optimization
- Transport independence
- Call and bearer separation