

ETSI TS 126 445 V15.3.2 (2020-12)



**Universal Mobile Telecommunications System (UMTS);
LTE;
Codec for Enhanced Voice Services (EVS);
Detailed algorithmic description
(3GPP TS 26.445 version 15.3.2 Release 15)**



ReferenceRTS/TSGS-0426445vf32

Keywords

LTE,UMTS

ETSI

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Siret N° 348 623 562 00017 - NAF 742 C
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